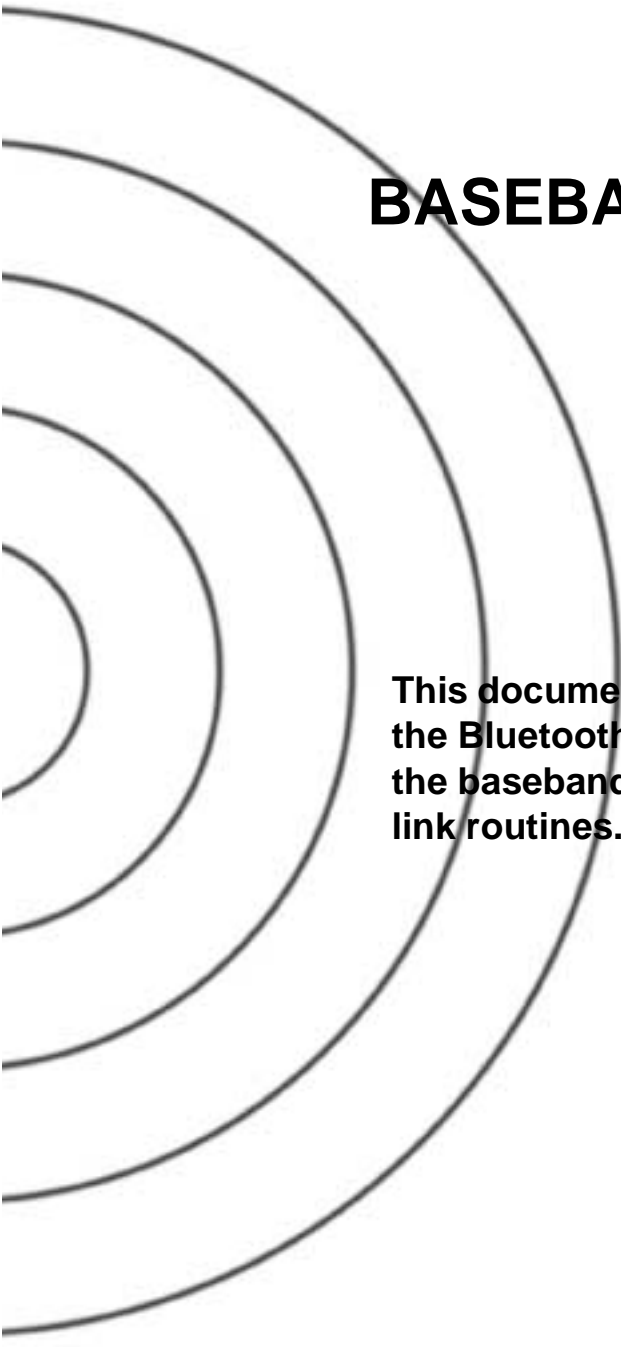


Part B

BASEBAND SPECIFICATION



This document describes the specifications of the Bluetooth link controller which carries out the baseband protocols and other low-level link routines.





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1 GENERAL DESCRIPTION

Bluetooth is a short-range radio link intended to replace the cable(s) connecting portable and/or fixed electronic devices. Key features are robustness, low complexity, low power, and low cost.

Bluetooth operates in the unlicensed ISM band at 2.4 GHz. A frequency hop transceiver is applied to combat interference and fading. A shaped, binary FM modulation is applied to minimize transceiver complexity. The symbol rate is 1 Ms/s. A slotted channel is applied with a nominal slot length of 625 μ s. For full duplex transmission, a Time-Division Duplex (TDD) scheme is used. On the channel, information is exchanged through packets. Each packet is transmitted on a different hop frequency. A packet nominally covers a single slot, but can be extended to cover up to five slots.

The Bluetooth protocol uses a combination of circuit and packet switching. Slots can be reserved for synchronous packets. Bluetooth can support an asynchronous data channel, up to three simultaneous synchronous voice channels, or a channel which simultaneously supports asynchronous data and synchronous voice. Each voice channel supports a 64 kb/s synchronous (voice) channel in each direction. The asynchronous channel can support maximal 723.2 kb/s asymmetric (and still up to 57.6 kb/s in the return direction), or 433.9 kb/s symmetric.

The Bluetooth system consists of a radio unit (see [Radio Specification](#)), a link control unit, and a support unit for link management and host terminal interface functions, see [Figure 1.1 on page 41](#). The current document describes the specifications of the Bluetooth link controller, which carries out the baseband protocols and other low-level link routines. Link layer messages for link set-up and control are defined in the [Link Manager Protocol on page 183](#).

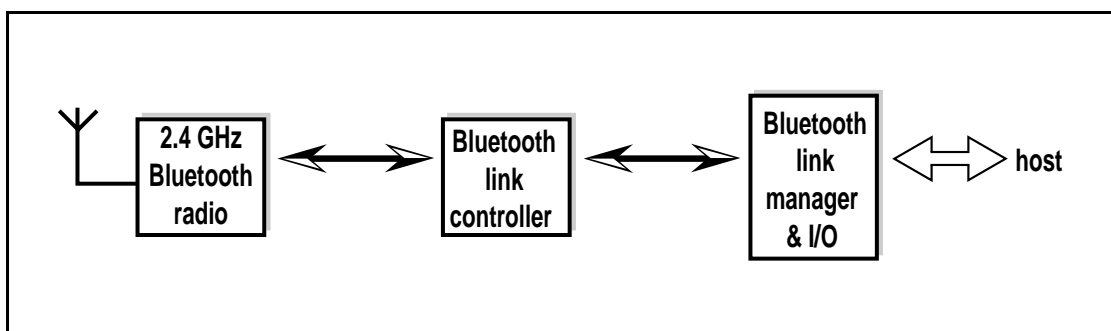


Figure 1.1: Different functional blocks in the Bluetooth system

The Bluetooth system provides a point-to-point connection (only two Bluetooth units involved), or a point-to-multipoint connection, see [Figure 1.2 on page 42](#). In the point-to-multipoint connection, the channel is shared among several Bluetooth units. Two or more units sharing the same channel form a **piconet**. One Bluetooth unit acts as the master of the piconet, whereas the other unit(s)



acts as slave(s). Up to seven slaves can be active in the piconet. In addition, many more slaves can remain locked to the master in a so-called parked state. These parked slaves cannot be active on the channel, but remain synchronized to the master. Both for active and parked slaves, the channel access is controlled by the master.

Multiple piconets with overlapping coverage areas form a **scatternet**. Each piconet can only have a single master. However, slaves can participate in different piconets on a time-division multiplex basis. In addition, a master in one piconet can be a slave in another piconet. The piconets shall not be frequency-synchronized. Each piconet has its own hopping channel.

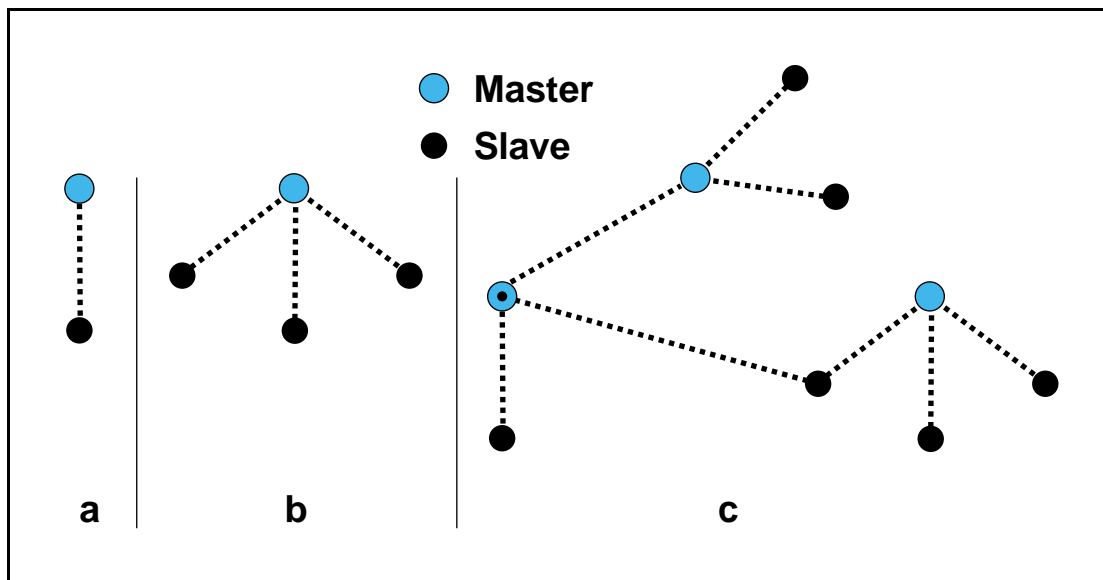


Figure 1.2: Piconets with a single slave operation (a), a multi-slave operation (b) and a scatternet operation (c).

2 PHYSICAL CHANNEL

2.1 CHANNEL DEFINITION

The channel is represented by a pseudo-random hopping sequence hopping through the 79 or 23 RF channels. The hopping sequence is unique for the piconet and is determined by the Bluetooth device address of the master; the phase in the hopping sequence is determined by the Bluetooth clock of the master. The channel is divided into time slots where each slot corresponds to an RF hop frequency. Consecutive hops correspond to different RF hop frequencies. The nominal hop rate is 1600 hops/s. All Bluetooth units participating in the piconet are time- and hop-synchronized to the channel.

2.2 TIME SLOTS

The channel is divided into time slots, each 625 μ s in length. The time slots are numbered according to the Bluetooth clock of the piconet master. The slot numbering ranges from 0 to $2^{27}-1$ and is cyclic with a cycle length of 2^{27} .

In the time slots, master and slave can transmit packets.

A TDD scheme is used where master and slave alternatively transmit, see [Figure 2.1 on page 44](#). The master shall start its transmission in even-numbered time slots only, and the slave shall start its transmission in odd-numbered time slots only. The packet start shall be aligned with the slot start. Packets transmitted by the master or the slave may extend over up to five time slots.

The RF hop frequency shall remain fixed for the duration of the packet. For a single packet, the RF hop frequency to be used is derived from the current Bluetooth clock value. For a multi-slot packet, the RF hop frequency to be used for the entire packet is derived from the Bluetooth clock value in the first slot of the packet. The RF hop frequency in the first slot after a multi-slot packet shall use the frequency as determined by the current Bluetooth clock value. [Figure 2.2 on page 44](#) illustrates the hop definition on single- and multi-slot packets. If a packet occupies more than one time slot, the hop frequency applied shall be the hop frequency as applied in the time slot where the packet transmission was started.

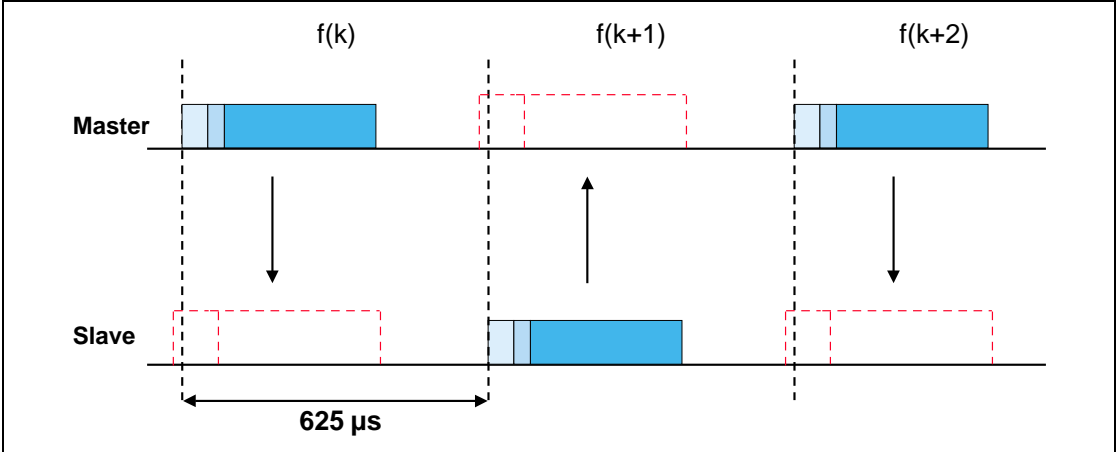


Figure 2.1: TDD and timing

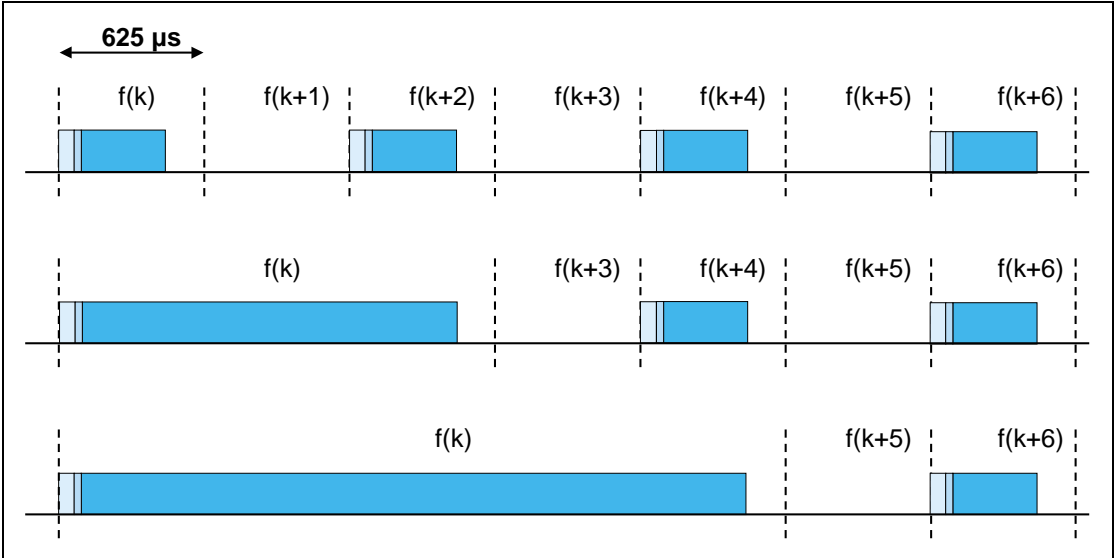


Figure 2.2: Multi-slot packets

3 PHYSICAL LINKS

3.1 GENERAL

Between master and slave(s), different types of links can be established. Two link types have been defined:

- Synchronous Connection-Oriented (SCO) link
- Asynchronous Connection-Less (ACL) link

The SCO link is a point-to-point link between a master and a single slave in the piconet. The master maintains the SCO link by using reserved slots at regular intervals. The ACL link is a point-to-multipoint link between the master and all the slaves participating on the piconet. In the slots not reserved for the SCO link(s), the master can establish an ACL link on a per-slot basis to any slave, including the slave(s) already engaged in an SCO link.

3.2 SCO LINK

The SCO link is a symmetric, point-to-point link between the master and a specific slave. The SCO link reserves slots and can therefore be considered as a circuit-switched connection between the master and the slave. The SCO link typically supports time-bounded information like voice. The master can support up to three SCO links to the same slave or to different slaves. A slave can support up to three SCO links from the same master, or two SCO links if the links originate from different masters. SCO packets are never retransmitted.

The master will send SCO packets at regular intervals, the so-called SCO interval T_{SCO} (counted in slots) to the slave in the reserved master-to-slave slots. The SCO slave is always allowed to respond with an SCO packet in the following slave-to-master slot unless a different slave was addressed in the previous master-to-slave slot. If the SCO slave fails to decode the slave address in the packet header, it is still allowed to return an SCO packet in the reserved SCO slot.

The SCO link is established by the master sending an SCO setup message via the LMP protocol. This message will contain timing parameters such as the SCO interval T_{SCO} and the offset D_{SCO} to specify the reserved slots.

In order to prevent clock wrap-around problems, an initialization flag in the LMP setup message indicates whether initialization procedure 1 or 2 is being used. The slave shall apply the initialization method as indicated by the initialization flag. The master uses initialization 1 when the MSB of the current master clock (CLK_{27}) is 0; it uses initialization 2 when the MSB of the current master clock (CLK_{27}) is 1. The master-to-slave SCO slots reserved by the master and the slave shall be initialized on the slots for which the clock satisfies the following equation:



$$\text{CLK}_{27-1} \bmod T_{\text{SCO}} = D_{\text{SCO}} \quad \text{for initialization 1}$$

$$(\overline{\text{CLK}}_{27}, \text{CLK}_{26-1}) \bmod T_{\text{SCO}} = D_{\text{SCO}} \quad \text{for initialization 2}$$

The slave-to-master SCO slots shall directly follow the reserved master-to-slave SCO slots. After initialization, the clock value $\text{CLK}(k+1)$ for the next master-to-slave SCO slot is found by adding the fixed interval T_{SCO} to the clock value of the current master-to-slave SCO slot:

$$\text{CLK}(k+1) = \text{CLK}(k) + T_{\text{SCO}}$$

3.3 ACL LINK

In the slots not reserved for SCO links, the master can exchange packets with any slave on a per-slot basis. The ACL link provides a packet-switched connection between the master and all active slaves participating in the piconet. Both asynchronous and isochronous services are supported. Between a master and a slave only a single ACL link can exist. For most ACL packets, packet retransmission is applied to assure data integrity.

A slave is permitted to return an ACL packet in the slave-to-master slot if and only if it has been addressed in the preceding master-to-slave slot. If the slave fails to decode the slave address in the packet header, it is not allowed to transmit.

ACL packets not addressed to a specific slave are considered as broadcast packets and are read by every slave. If there is no data to be sent on the ACL link and no polling is required, no transmission shall take place.

4 PACKETS

4.1 GENERAL FORMAT

The bit ordering when defining packets and messages in the *Baseband Specification*, follows the *Little Endian format*, i.e., the following rules apply:

- The *least significant bit* (LSB) corresponds to b_0 ;
- The LSB is the first bit sent over the air;
- In illustrations, the LSB is shown on the left side;

The linkcontroller interprets the first bit arriving from a higher software layer as b_0 ; i.e. this is the first bit to be sent over the air. Furthermore, data fields generated internally at baseband level, such as the packet header fields and payload header length, are transmitted with the LSB first. For instance, a 3-bit parameter $X=3$ is sent as $b_0b_1b_2$ over the air where 1 is sent first and 0 is sent last.

The data on the piconet channel is conveyed in packets. The general packet format is shown in [Figure 4.1 on page 47](#). Each packet consists of 3 entities: the access code, the header, and the payload. In the figure, the number of bits per entity is indicated.

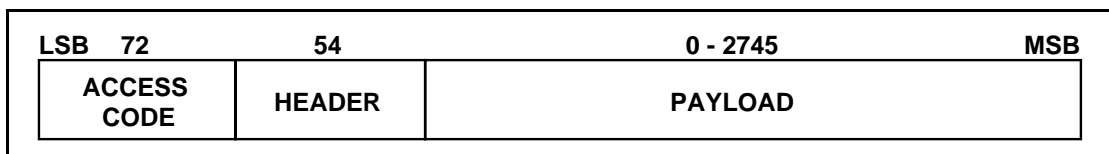


Figure 4.1: Standard packet format.

The access code and header are of fixed size : 72 bits and 54 bits respectively. The payload can range from zero to a maximum of 2745 bits. Different packet types have been defined. Packets may consist of the (shortened) access code only (see [ID packet on page 55](#)), of the access code – header, or of the access code – header – payload.

4.2 ACCESS CODE

Each packet starts with an access code. If a packet header follows, the access code is 72 bits long, otherwise the access code is 68 bits long. This access code is used for synchronization, DC offset compensation and identification. The access code identifies all packets exchanged on the channel of the piconet: all packets sent in the same piconet are preceded by the same channel access code. In the receiver of the Bluetooth unit, a sliding correlator correlates against the access code and triggers when a threshold is exceeded. This trigger signal is used to determine the receive timing.

The access code is also used in paging and inquiry procedures. In this case, the access code itself is used as a signalling message and neither a header nor a payload is present.

The access code consists of a preamble, a sync word, and possibly a trailer, see [Figure 4.2 on page 48](#). For details see [Section 4.2.1 on page 48](#).

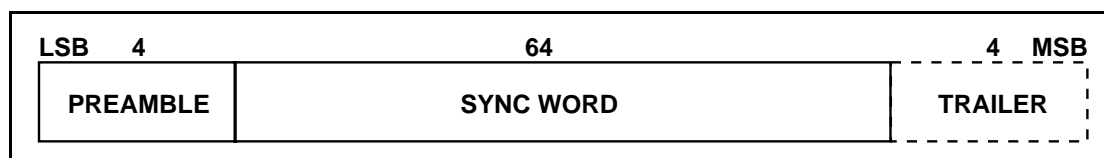


Figure 4.2: Access code format

4.2.1 Access code types

There are three different types of access codes defined:

- Channel Access Code (CAC)
- Device Access Code (DAC)
- Inquiry Access Code (IAC)

The respective access code types are used for a Bluetooth unit in different operating modes. The channel access code identifies a piconet. This code is included in all packets exchanged on the piconet channel. The device access code is used for special signalling procedures, e.g., paging and response to paging. For the inquiry access code there are two variations. A general inquiry access code (GIAC) is common to all devices. The GIAC can be used to discover which other Bluetooth units are in range. The dedicated inquiry access code (DIAC) is common for a dedicated group of Bluetooth units that share a common characteristic. The DIAC can be used to discover only these dedicated Bluetooth units in range.

The CAC consists of a **preamble**, **sync word**, and **trailer** and its total length is 72 bits. When used as self-contained messages without a header, the DAC and IAC do not include the trailer bits and are of length 68 bits.



The different access code types use different Lower Address Parts (LAPs) to construct the sync word. The LAP field of the BD address is explained in [Section 13.1 on page 142](#). A summary of the different access code types can be found in [Table 4.1 on page 49](#).

Code type	LAP	Code length	Comments
CAC	Master	72	See also Section 13.2 on page 142
DAC	Paged unit	68/72*	
GIAC	Reserved	68/72*	
DIAC	Dedicated	68/72*	

Table 4.1: Summary of access code types.

*. length 72 is only used in combination with FHS packets

4.2.2 Preamble

The preamble is a fixed zero-one pattern of 4 symbols used to facilitate DC compensation. The sequence is either 1010 or 0101, depending whether the LSB of the following sync word is 1 or 0, respectively. The preamble is shown in [Figure 4.3 on page 49](#).



Figure 4.3: Preamble

4.2.3 Sync Word

The sync word is a 64-bit code word derived from a 24 bit address (LAP); for the CAC the master’s LAP is used; for the GIAC and the DIAC, reserved, dedicated LAPs are used; for the DAC, the slave unit LAP is used. The construction guarantees large Hamming distance between sync words based on different LAPs. In addition, the good auto correlation properties of the sync word improve on the timing synchronization process. The derivation of the sync word is described in [Section 13.2 on page 142](#)



4.2.4 Trailer

The trailer is appended to the sync word as soon as the packet header follows the access code. This is typically the case with the CAC, but the trailer is also used in the DAC and IAC when these codes are used in FHS packets exchanged during page response and inquiry response procedures.

The trailer is a fixed zero-one pattern of four symbols. The trailer together with the three MSBs of the syncword form a 7-bit pattern of alternating ones and zeroes which may be used for extended DC compensation. The trailer sequence is either 1010 or 0101 depending on whether the MSB of the sync word is 0 or 1, respectively. The choice of trailer is illustrated in [Figure 4.4 on page 50](#).



Figure 4.4: Trailer in CAC when MSB of sync word is 0 (a), and when MSB of sync word is 1 (b).

4.3 PACKET HEADER

The header contains link control (LC) information and consists of 6 fields:

- AM_ADDR: 3-bit active member address
- TYPE: 4-bit type code
- FLOW: 1-bit flow control
- ARQN: 1-bit acknowledge indication
- SEQN: 1-bit sequence number
- HEC: 8-bit header error check

The total header, including the HEC, consists of 18 bits, see [Figure 4.5 on page 51](#), and is encoded with a rate 1/3 FEC (not shown but described in [Section 5.1 on page 66](#)) resulting in a 54-bit header. Note that the AM_ADDR and TYPE fields are sent with their LSB first. The function of the different fields will be explained next.

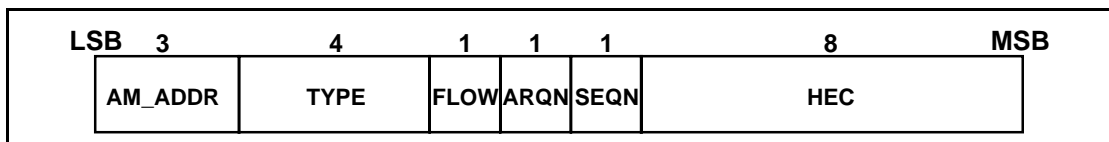


Figure 4.5: Header format.

4.3.1 AM_ADDR

The AM_ADDR represents a member address and is used to distinguish between the active members participating on the piconet. In a piconet, one or more slaves are connected to a single master. To identify each slave separately, each slave is assigned a temporary 3-bit address to be used when it is active. Packets exchanged between the master and the slave all carry the AM_ADDR of this slave; that is, the AM_ADDR of the slave is used in both master-to-slave packets and in the slave-to-master packets. The all-zero address is reserved for broadcasting packets from the master to the slaves. An exception is the FHS packet which may use the all-zero member address but is *not* a broadcast message ([Section 4.4.1.4 on page 56](#)). Slaves that are disconnected or parked give up their AM_ADDR. A new AM_ADDR has to be assigned when they re-enter the piconet.

4.3.2 TYPE

Sixteen different types of packets can be distinguished. The 4-bit TYPE code specifies which packet type is used. Important to note is that the interpretation of the TYPE code depends on the physical link type associated with the packet. First, it shall be determined whether the packet is sent on an SCO link or an ACL link. Then it can be determined which type of SCO packet or ACL packet has been received. The TYPE code also reveals how many slots the current packet will occupy. This allows the non-addressed receivers to refrain from listening to the channel for the duration of the remaining slots. In [Section 4.4 on page 54](#), each packet type will be described in more detail.



4.3.3 FLOW

This bit is used for flow control of packets over the ACL link. When the RX buffer for the ACL link in the recipient is full and is not emptied, a STOP indication (FLOW=0) is returned to stop the transmission of data temporarily. Note, that the STOP signal only concerns ACL packets. Packets including only link control information (ID, POLL and NULL packets) or SCO packets can still be received. When the RX buffer is empty, a GO indication (FLOW=1) is returned. When no packet is received, or the received header is in error, a GO is assumed implicitly. In this case, the slave can receive a new packet with CRC although its RX buffer is still not emptied. The slave shall then return a NAK in response to this packet even if the packet passed the CRC check.

4.3.4 ARQN

The 1-bit acknowledgment indication ARQN is used to inform the source of a successful transfer of payload data with CRC, and can be positive acknowledge ACK or negative acknowledge NAK. If the reception was successful, an ACK (ARQN=1) is returned, otherwise a NAK (ARQN=0) is returned. When no return message regarding acknowledge is received, a NAK is assumed implicitly. NAK is also the default return information.

The ARQN is piggy-backed in the header of the return packet. The success of the reception is checked by means of a cyclic redundancy check (CRC) code. An unnumbered ARQ scheme which means that the ARQN relates to the latest received packet from the same source, is used. See [Section 5.3 on page 67](#) for initialization and usage of this bit.

4.3.5 SEQN

The SEQN bit provides a sequential numbering scheme to order the data packet stream. For each new transmitted packet that contains data with CRC, the SEQN bit is inverted. This is required to filter out retransmissions at the destination; if a retransmission occurs due to a failing ACK, the destination receives the same packet twice. By comparing the SEQN of consecutive packets, correctly received retransmissions can be discarded. See [section 5.3.2 on page 69](#) for initialization and usage of the SEQN bit. For broadcast packets, a modified sequencing method is used, see [Section 5.3.5 on page 71](#).

4.3.6 HEC

Each header has a header-error-check to check the header integrity. The HEC consists of an 8-bit word generated by the polynomial 647 (octal representation). Before generating the HEC, the HEC generator is initialized with an 8-bit value. For FHS packets sent in **master page response** state, the slave upper address part (UAP) is used. For FHS packets sent in **inquiry response**, the default check initialization (DCI, see [Section 5.4 on page 72](#)) is used. In all other cases, the UAP of the master device is used. For the definition of Bluetooth device addresses, see [Section 13.1 on page 142](#).

After the initialization, a HEC is calculated for the 10 header bits. Before checking the HEC, the receiver must initialize the HEC check circuitry with the proper 8-bit UAP (or DCI). If the HEC does not check, the entire packet is disregarded. More information can be found in [Section 5.4 on page 72](#).



4.4 PACKET TYPES

The packets used on the piconet are related to the physical links they are used in. Up to now, two physical links are defined: the SCO link and the ACL link. For each of these links, 12 different packet types can be defined. Four control packets will be common to all link types: their TYPE code is unique irrespective of the link type.

To indicate the different packets on a link, the 4-bit TYPE code is used. The packet types have been divided into four segments. The first segment is reserved for the four control packets common to all physical link types; all four packet types have been defined. The second segment is reserved for packets occupying a single time slot; six packet types have been defined. The third segment is reserved for packets occupying three time slots; two packet types have been defined. The fourth segment is reserved for packets occupying five time slots; two packet types have been defined. The slot occupancy is reflected in the segmentation and can directly be derived from the type code. [Table 4.2 on page 54](#) summarizes the packets defined so far for the SCO and ACL link types.

Segment	TYPE code b ₃ b ₂ b ₁ b ₀	Slot occupancy	SCO link	ACL link
1	0000	1	NULL	NULL
	0001	1	POLL	POLL
	0010	1	FHS	FHS
	0011	1	DM1	DM1
2	0100	1	undefined	DH1
	0101	1	HV1	undefined
	0110	1	HV2	undefined
	0111	1	HV3	undefined
	1000	1	DV	undefined
	1001	1	undefined	AUX1
3	1010	3	undefined	DM3
	1011	3	undefined	DH3
	1100	3	undefined	undefined
	1101	3	undefined	undefined
4	1110	5	undefined	DM5
	1111	5	undefined	DH5

Table 4.2: Packets defined for SCO and ACL link types

4.4.1 Common packet types

There are five common packets. In addition to the types listed in segment 1 of the previous table, there is the ID packet not listed. Each packet will now be examined in more detail.

4.4.1.1 ID packet

The identity or ID packet consists of the device access code (DAC) or inquiry access code (IAC). It has a fixed length of 68 bits. It is a very robust packet since the receiver uses a bit correlator to match the received packet to the known bit sequence of the ID packet. The packet is used, for example, in paging, inquiry, and response routines.

4.4.1.2 NULL packet

The NULL packet has no payload and therefore consists of the channel access code and packet header only. Its total (fixed) length is 126 bits. The NULL packet is used to return link information to the source regarding the success of the previous transmission (ARQN), or the status of the RX buffer (FLOW). The NULL packet itself does not have to be acknowledged.

4.4.1.3 POLL packet

The POLL packet is very similar to the NULL packet. It does not have a payload either. In contrast to the NULL packet, it requires a confirmation from the recipient. It is not a part of the ARQ scheme. The POLL packet does not affect the ARQN and SEQN fields. Upon reception of a POLL packet the slave must respond with a packet. This return packet is an implicit acknowledgement of the POLL packet. This packet can be used by the master in a piconet to poll the slaves, which must then respond even if they do not have information to send.

4.4.1.4 FHS packet

The FHS packet is a special control packet revealing, among other things, the Bluetooth device address and the clock of the sender. The payload contains 144 information bits plus a 16-bit CRC code. The payload is coded with a rate 2/3 FEC which brings the gross payload length to 240 bits. The FHS packet covers a single time slot.

Figure 4.6 on page 56 illustrates the format and contents of the FHS payload. The payload consists of eleven fields. The FHS packet is used in page master response, inquiry response and in master slave switch. In page master response or master slave switch, it is retransmitted until its reception is acknowledged or a timeout has exceeded. In inquiry response, the FHS packet is not acknowledged. The FHS packet contains real-time clock information. This clock information is updated before each retransmission. The retransmission of the FHS payload is thus somewhat different from the retransmission of ordinary data payloads where the same payload is used for each retransmission. The FHS packet is used for frequency hop synchronization before the piconet channel has been established, or when an existing piconet changes to a new piconet. In the former case, the recipient has not been assigned an active member address yet, in which case the AM_ADDR field in the FHS packet header is set to all-zeroes; however, the FHS packet should not be considered as a broadcast packet. In the latter case the slave already has an AM_ADDR in the existing piconet, which is then used in the FHS packet header.

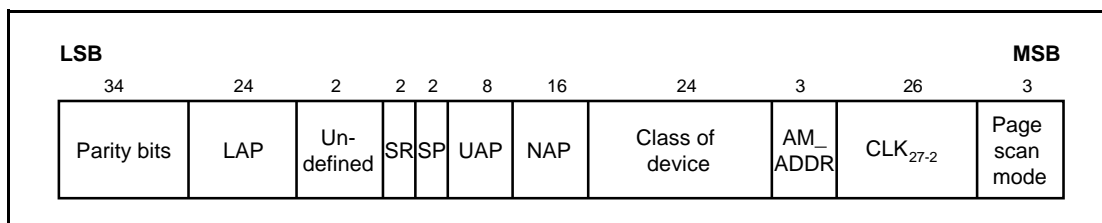


Figure 4.6: Format of the FHS payload

Each field is described in more detail below:

Parity bits	This 34-bit field contains the parity bits that form the first part of the sync word of the access code of the unit that sends the FHS packet. These bits are derived from the LAP as described in section 13.2 on page 142 .
LAP	This 24-bit field contains the lower address part of the unit that sends the FHS packet.
Undefined	This 2-bit field is reserved for future use and shall be set to zero.
SR	This 2-bit field is the scan repetition field and indicates the interval between two consecutive page scan windows, see also Table 4.4 and Table 10.1 on page 97

Table 4.3: Description of the FHS payload



SP	This 2-bit field is the scan period field and indicates the period in which the mandatory page scan mode is applied after transmission of an inquiry response message, see also Table 4.5 and Table 10.6 on page 109 .
UAP	This 8-bit field contains the upper address part of the unit that sends the FHS packet.
NAP	This 16-bit field contains the non-significant address part of the unit that sends the FHS packet (see also section 13.1 on page 142 for LAP, UAP, and NAP).
Class of device	This 24-bit field contains the class of device of the unit that sends the FHS packet. The field is defined in Bluetooth Assigned Numbers (http://www.bluetooth.org/assigned-numbers.htm).
AM_ADDR	This 3-bit field contains the member address the recipient shall use if the FHS packet is used at call setup or master-slave switch. A slave responding to a master or a unit responding to an inquiry request message shall include an all-zero AM_ADDR field if it sends the FHS packet.
CLK₂₇₋₂	This 26-bit field contains the value of the native system clock of the unit that sends the FHS packet, sampled at the beginning of the transmission of the access code of this FHS packet. This clock value has a resolution of 1.25ms (two-slot interval). For each new transmission, this field is updated so that it accurately reflects the real-time clock value.
Page scan mode	This 3-bit field indicates which scan mode is used by default by the sender of the FHS packet. The interpretation of the page scan mode is illustrated in Table 4.6. Currently, the standard supports one mandatory scan mode and up to three optional scan modes (see also “Appendix VII” on page 1025).

Table 4.3: Description of the FHS payload

SR bit format b_1b_0	SR mode
00	R0
01	R1
10	R2
11	reserved

Table 4.4: Contents of SR field

SP bit format b_1b_0	SP mode
00	P0
01	P1
10	P2
11	reserved

Table 4.5: Contents of SP field

Bit format $b_2b_1b_0$	Page scan mode
000	Mandatory scan mode
001	Optional scan mode I
010	Optional scan mode II
011	Optional scan mode III
100	Reserved for future use
101	Reserved for future use
110	Reserved for future use
111	Reserved for future use

Table 4.6: Contents of page scan mode field

The LAP, UAP, and NAP together form the 48-bit IEEE address of the unit that sends the FHS packet. Using the parity bits and the LAP, the recipient can directly construct the channel access code of the sender of the FHS packet.

4.4.1.5 DM1 packet

DM1 serves as part of segment 1 in order to support control messages in any link type. However, it can also carry regular user data. Since the DM1 packet is recognized on the SCO link, it can interrupt the synchronous information to send control information. Since the DM1 packet can be regarded as an ACL packet, it will be discussed in [Section 4.4.3 on page 60](#).

4.4.2 SCO packets

SCO packets are used on the synchronous SCO link. The packets do not include a CRC and are never retransmitted. SCO packets are routed to the synchronous I/O (voice) port. Up to now, three pure SCO packets have been defined. In addition, an SCO packet is defined which carries an asynchronous data field in addition to a synchronous (voice) field. The SCO packets defined so far are typically used for 64 kb/s speech transmission.

4.4.2.1 HV1 packet

The **HV1** packet carries 10 information bytes. The bytes are protected with a rate 1/3 FEC. No CRC is present. The payload length is fixed at 240 bits. There is no payload header present.

HV packets are used for transmission of voice and transparent synchronous data (see also [Link Manager Protocol, section 3.21.1 on page 227](#). HV stands for High-quality Voice. The voice packets are never retransmitted and need no CRC.



An HV1 packet can carries 1.25ms of speech at a 64 kb/s rate. An HV1 packet has therefore to be sent every two time slots ($T_{SCO}=2$).

4.4.2.2 HV2 packet

The **HV2** packet carries 20 information bytes. The bytes are protected with a rate 2/3 FEC. No CRC is present. The payload length is fixed at 240 bits. There is no payload header present.

An HV2 packet carries 2.5ms of speech at a 64 kb/s rate. An HV2 packet has therefore to be sent every four time slots ($T_{SCO}=4$).

4.4.2.3 HV3 packet

The **HV3** packet carries 30 information bytes. The bytes are not protected by FEC. No CRC is present. The payload length is fixed at 240 bits. There is no payload header present.

An HV3 packet carries 3.75ms of speech at a 64 kb/s rate. An HV3 packet has therefore to be sent every six time slots ($T_{SCO}=6$).

4.4.2.4 DV packet

The **DV** packet is a combined data - voice packet. The payload is divided into a voice field of 80 bits and a data field containing up to 150 bits, see [Figure 4.7](#). The voice field is not protected by FEC. The data field contains up to 10 information bytes (including the 1-byte payload header) and includes a 16-bit CRC. The data field is encoded with a rate 2/3 FEC. If necessary, extra zeroes are appended to assure that the total number of payload bits is a multiple of 10 prior to FEC encoding. Since the **DV** packet has to be sent at regular intervals due to its synchronous (voice) contents, it is listed under the SCO packet types. The voice and data fields are treated completely separate. The voice field is handled like normal SCO data and is never retransmitted; that is, the voice field is always new. The data field is checked for errors and is retransmitted if necessary.

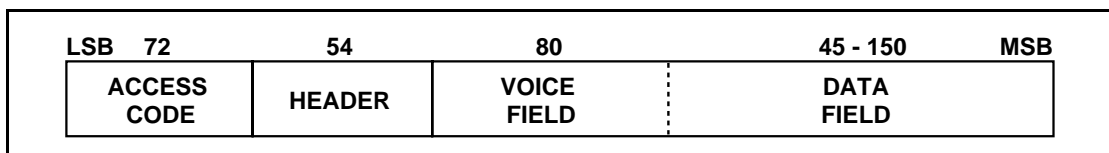


Figure 4.7: DV packet format

4.4.3 ACL packets

ACL packets are used on the asynchronous links. The information carried can be user data or control data. Including the DM1 packet, seven ACL packets have been defined. Six of the ACL packets contain a CRC code and retransmission is applied if no acknowledgement of proper reception is received (except in case a flush operation is carried out, see [Section 5.3.3 on page 70](#)). The 7th ACL packet, the AUX1 packet, has no CRC and is not retransmitted.

4.4.3.1 DM1 packet

The DM1 packet is a packet that carries data information only. DM stands for Data – Medium rate. The payload contains up to 18 information bytes (including the 1-byte payload header) plus a 16-bit CRC code. The DM1 packet may cover up to a single time slot. The information plus CRC bits are coded with a rate 2/3 FEC which adds 5 parity bits to every 10-bit segment. If necessary, extra zeros are appended after the CRC bits to get the total number of bits (information bits, CRC bits, and tail bits) equal a multiple of 10. The payload header in the DM1 packet is only 1 byte long, see [Figure 4.8 on page 62](#). The length indicator in the payload header specifies the number of user bytes (excluding payload header and the CRC code).

4.4.3.2 DH1 packet

This packet is similar to the DM1 packet, except that the information in the payload is not FEC encoded. As a result, the DH1 frame can carry up to 28 information bytes (including the 1 byte payload header) plus a 16-bit CRC code. DH stands for Data – High rate. The DH1 packet may cover up to a single time slot.

4.4.3.3 DM3 packet

The DM3 packet is a DM1 packet with an extended payload. The DM3 packet may cover up to three time slots. The payload contains up to 123 information bytes (including the 2-bytes payload header) plus a 16-bit CRC code. The payload header in the DM3 packet is 2 bytes long, see [Figure 4.9 on page 62](#). The length indicator in the payload header specifies the number of user bytes (excluding payload header and the CRC code). When a DM3 packet is sent or received, the RF hop frequency shall not change for a duration of three time slots (the first time slot being the slot where the channel access code was transmitted).

4.4.3.4 DH3 packet

This packet is similar to the DM3 packet, except that the information in the payload is not FEC encoded. As a result, the DH3 packet can carry up to 185 information bytes (including the two bytes payload header) plus a 16-bit CRC code. The DH3 packet may cover three time slots. When a DH3 packet is sent or received, the hop frequency shall not change for a duration of three time slots (the first time slot being the slot where the channel access code was transmitted).

4.4.3.5 DM5 packet

The DM5 packet is a DM1 packet with an extended payload. The DM5 packet may cover up to five time slots. The payload contains up to 226 information bytes (including the 2-bytes payload header) plus a 16-bit CRC code. The payload header in the DM5 packet is 2 bytes long. The length indicator in the payload header specifies the number of user bytes (excluding payload header and the CRC code). When a DM5 packet is sent or received, the hop frequency shall not change for a duration of five time slots (the first time slot being the slot where the channel access code was transmitted).

4.4.3.6 DH5 packet

This packet is similar to the DM5 packet, except that the information in the payload is not FEC encoded. As a result, the DH5 packet can carry up to 341 information bytes (including the two bytes payload header) plus a 16-bit CRC code. The DH5 packet may cover five time slots. When a DH5 packet is sent or received, the hop frequency shall not change for a duration of five time slots (the first time slot being the slot where the channel access code was transmitted).

4.4.3.7 AUX1 packet

This packet resembles a DH1 packet but has no CRC code. The AUX1 packet can carry up to 30 information bytes (including the 1-byte payload header). The AUX1 packet may cover up to a single time slot.

4.5 PAYLOAD FORMAT

In the previous packet overview, several payload formats were considered. In the payload, two fields are distinguished: the (synchronous) voice field and the (asynchronous) data field. The ACL packets only have the data field and the SCO packets only have the voice field – with the exception of the DV packets which have both.

4.5.1 Voice field

The voice field has a fixed length. For the HV packets, the voice field length is 240 bits; for the DV packet the voice field length is 80 bits. No payload header is present.

4.5.2 Data field

The data field consists of three segments: a payload header, a payload body, and possibly a CRC code (only the AUX1 packet does not carry a CRC code).

1. Payload header

Only data fields have a payload header. The payload header is one or two bytes long. Packets in segments one and two have a 1-byte payload header; packets in segments three and four have a 2-bytes payload header. The payload header specifies the logical channel (2-bit L_CH indication), controls the flow on the logical channels (1-bit FLOW indication), and has a payload length indicator (5 bits and 9 bits for 1-byte and 2-bytes payload header, respectively). In the case of a 2-byte payload header, the length indicator is extended by four bits into the next byte. The remaining four bits of the second byte are reserved for future use and shall be set to zero. The formats of the 1-byte and 2-bytes payload headers are shown in [Figure 4.8 on page 62](#) and [Figure 4.9 on page 62](#).

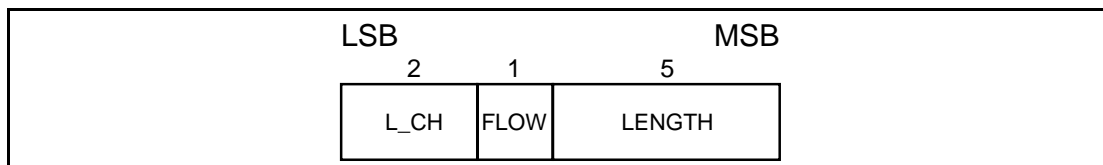


Figure 4.8: Payload header format for single-slot packets.

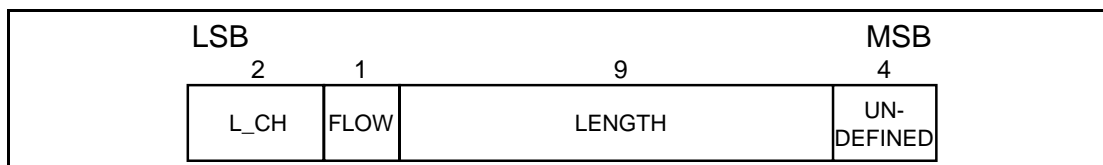


Figure 4.9: Payload header format for multi-slot packets.

The L_CH field is transmitted first, the length field last. In [Table 4.7 on page 63](#), more details about the contents of the L_CH field are listed.



L_CH code b ₁ b ₀	Logical Channel	Information
00	NA	undefined
01	UA/UI	Continuation fragment of an L2CAP message
10	UA/UI	Start of an L2CAP message or no fragmentation
11	LM	LMP message

Table 4.7: Logical channel L_CH field contents

An L2CAP message can be fragmented into several packets. Code 10 is used for an L2CAP packet carrying the first fragment of such a message; code 01 is used for continuing fragments. If there is no fragmentation, code 10 is used for every packet. Code 11 is used for LMP messages. Code 00 is reserved for future use.

The flow indicator in the payload is used to control the flow at the L2CAP level. It is used to control the flow per logical channel (when applicable). FLOW=1 means flow-on ("OK to send") and FLOW=0 means flow-off ("stop"). After a new connection has been established the flow indicator should be set to FLOW=1. When a Bluetooth unit receives a payload header with the flow bit set to "stop" (FLOW=0), it shall stop the transmission of ACL packets before an additional amount of payload data is sent. This amount can be defined as the flow control lag, expressed with a number of bytes. The shorter the flow control lag, the less buffering the other Bluetooth device must dedicate to this function. The flow control lag shall not exceed 1792 bytes (7*256 bytes), but in order to allow units to optimize the selection of packet length and buffer space, the flow control lag of a given implementation is provided in the LMP_features_res message.

If the packets containing the payload flow bit of "stop" is received with a valid packet header but bad payload, the payload flow control bit will not be recognized. The packet level ACK contained in the packet header will be received and a further ACL packet can be transmitted. Each occurrence of this situation allows a further ACL packet to be sent in spite of the flow control request being sent via the payload header flow control bit. It is recommended that Bluetooth units that use the payload header flow bit should ensure that no further ACL packets are sent until the payload flow bit has been correctly received. This can be accomplished by simultaneously turning on the flow bit in the packet header and keeping it on until an ACK is received back (ARQN=1). This will typically be only one round trip time. Since they lack a payload CRC, AUX1 packets should not be used with a payload flow bit of "stop".

The link manager is responsible for setting and processing the flow bit in the payload header. Real-time flow control is carried out at the packet level by the link controller via the flow bit in the packet header (see [Section 4.3.3 on page 52](#)). With the payload flow bit, traffic from the remote end can be controlled. It is allowed to generate and send an ACL packet with payloadlength zero irrespective of flow status. L2CAP start- and continue-fragment



indications (L_CH=10 and L_CH=01) also retain their meaning when the payload length is equal to zero (i.e. an empty start-fragment should not be sent in the middle of an on-going L2CAP packet transmission). It is always safe to send an ACL packet with payload length=0 and L_CH=01. The payload flow bit has its own meaning for each logical channel (UA/I or LM), see [Table 4.8 on page 64](#). On the LM channel, no flow control is applied and the payload flow bit is always set at one.

L_CH code b ₁ b ₀	Usage and semantics of the ACL payload header FLOW bit
00	Not defined, reserved for future use.
01 or 10	Flow control of the UA/I channels (which are used to send L2CAP messages)
11	Always set FLOW=1 on transmission and ignore the bit on reception

Table 4.8: Use of payload header flow bit on the logical channels.

The length indicator indicates the number of bytes (i.e. 8-bit words) in the payload excluding the payload header and the CRC code; i.e. the payload body only. With reference to [Figure 4.8](#) and [Figure 4.9](#), the MSB of the length field in a 1-byte header is the last (right-most) bit in the payload header; the MSB of the length field in a 2-byte header is the fourth bit (from left) of the second byte in the payload header.

2. Payload body

The payload body includes the user host information and determines the effective user throughput. The length of the payload body is indicated in the length field of the payload header.

3. CRC code generation

The 16-bit cyclic redundancy check code in the payload is generated by the CRC-CCITT polynomial 210041 (octal representation). It is generated in a way similar to the HEC. Before determining the CRC code, an 8-bit value is used to initialize the CRC generator. For the CRC code in the FHS packets sent in **master page response** state, the UAP of the slave is used. For the FHS packet sent in **inquiry response** state, the DCI (see [Section 5.4 on page 72](#)) is used. For all other packets, the UAP of the master is used.

The 8 bits are loaded into the 8 least significant (left-most) positions of the LFSR circuit, see [Figure 5.10 on page 74](#). The other 8 bits are at the same time reset to zero. Subsequently, the CRC code is calculated over the information. Then the CRC code is appended to the information; the UAP (or DCI) is disregarded. At the receive side, the CRC circuitry is in the same way initialized with the 8-bit UAP (DCI) before the received information is checked. More information can be found in [Section 5.4 on page 72](#).

4.6 PACKET SUMMARY

A summary of the packets and their characteristics is shown in [Table 4.9](#), [Table 4.10](#) and [Table 4.11](#). The user payload represents the packet payload excluding FEC, CRC, and payload header.

Type	User Payload (bytes)	FEC	CRC	Symmetric Max. Rate	Asymmetric Max. Rate
ID	na	na	na	na	na
NULL	na	na	na	na	na
POLL	na	na	na	na	na
FHS	18	2/3	yes	na	na

Table 4.9: Link control packets

Type	Payload Header (bytes)	User Payload (bytes)	FEC	CRC	Symmetric Max. Rate (kb/s)	Asymmetric Max. Rate (kb/s)	
						Forward	Reverse
DM1	1	0-17	2/3	yes	108.8	108.8	108.8
DH1	1	0-27	no	yes	172.8	172.8	172.8
DM3	2	0-121	2/3	yes	258.1	387.2	54.4
DH3	2	0-183	no	yes	390.4	585.6	86.4
DM5	2	0-224	2/3	yes	286.7	477.8	36.3
DH5	2	0-339	no	yes	433.9	723.2	57.6
AUX1	1	0-29	no	no	185.6	185.6	185.6

Table 4.10: ACL packets

Type	Payload Header (bytes)	User Payload (bytes)	FEC	CRC	Symmetric Max. Rate (kb/s)
HV1	na	10	1/3	no	64.0
HV2	na	20	2/3	no	64.0
HV3	na	30	no	no	64.0
DV*	1 D	10+(0-9) D	2/3 D	yes D	64.0+57.6 D

Table 4.11: SCO packets

*. Items followed by 'D' relate to data field only.

5 ERROR CORRECTION

There are three error correction schemes defined for Bluetooth:

- 1/3 rate FEC
- 2/3 rate FEC
- ARQ scheme for the data

The purpose of the FEC scheme on the data payload is to reduce the number of retransmissions. However, in a reasonable error-free environment, FEC gives unnecessary overhead that reduces the throughput. Therefore, the packet definitions given in [Section 4 on page 47](#) have been kept flexible to use FEC in the payload or not, resulting in the **DM** and **DH** packets for the ACL link and the **HV** packets for the SCO link. The packet header is always protected by a 1/3 rate FEC; it contains valuable link information and should be able to sustain more bit errors.

Correction measures to mask errors in the voice decoder are not included in this section. This matter is discussed in section [Section 12.3 on page 141](#).

5.1 FEC CODE: RATE 1/3

A simple 3-times repetition FEC code is used for the header. The repetition code is implemented by repeating the bit three times, see the illustration in [Figure 5.1 on page 66](#). The 3-bit repetition code is used for the entire header, and also for the voice field in the **HV1** packet.

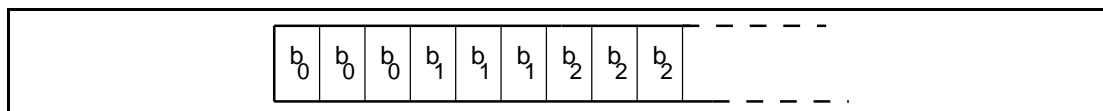


Figure 5.1: Bit-repetition encoding scheme.

5.2 FEC CODE: RATE 2/3

The other FEC scheme is a (15,10) shortened Hamming code. The generator polynomial is $g(D) = (D + 1)(D^4 + D + 1)$. This corresponds to 65 in octal notation. The LFSR generating this code is depicted in [Figure 5.2 on page 67](#). Initially all register elements are set to zero. The 10 information bits are sequentially fed into the LFSR with the switches S1 and S2 set in position 1. Then, after the final input bit, the switches S1 and S2 are set in position 2, and the five parity bits are shifted out. The parity bits are appended to the information bits. Consequently, each block of 10 information bits is encoded into a 15 bit codeword. This code can correct all single errors and detect all double errors in each codeword. This 2/3 rate FEC is used in the **DM** packets, in the data field of the **DV** packet, in the **FHS** packet, and in the **HV2** packet. Since the encoder operates with information segments of length 10, tail bits with value zero may have to be appended after the CRC bits.

The total number of bits to encode, i.e., payload header, user data, CRC, and tail bits, must be a multiple of 10. Thus, the number of tail bits to append is the least possible that achieves this (i.e., in the interval 0...9). These tail bits are not included in the payload length indicator.

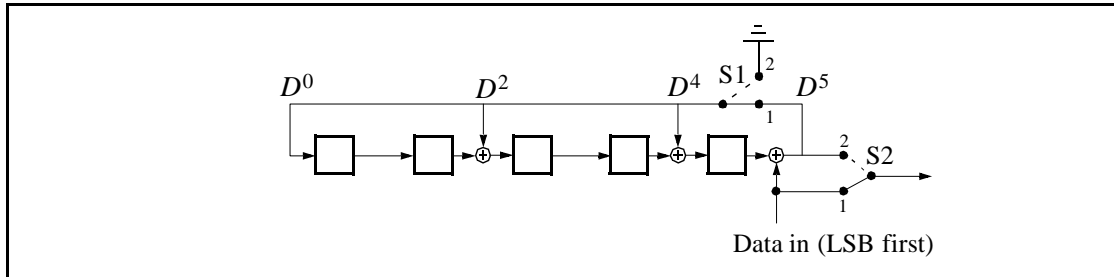


Figure 5.2: LFSR generating the (15,10) shortened Hamming code.

5.3 ARQ SCHEME

With an automatic repeat request scheme, **DM**, **DH** and the data field of **DV** packets are transmitted and retransmitted until acknowledgement of a successful reception is returned by the destination (or timeout is exceeded). The acknowledgement information is included in the header of the return packet, so-called piggy-backing. To determine whether the payload is correct or not, a cyclic redundancy check (CRC) code is added to the packet. The ARQ scheme only works on the payload in the packet (only that payload which has a CRC). The packet header and the voice payload are not protected by the ARQ scheme.

5.3.1 Unnumbered ARQ

Bluetooth uses a fast, unnumbered acknowledgment scheme: an ACK (ARQN=1) or a NAK (ARQN=0) is returned in response to the receipt of previously received packet. The slave will respond in the slave-to-master slot directly following the master-to-slave slot; the master will respond at the next event it will address the same slave (the master may have addressed other slaves between the last received packet from the considered slave and the master response to this packet). For a packet reception to be successful, at least the HEC must check. In addition, the CRC must check if present.

At the start of a new connection which may be the result of a page, page scan, master-slave switch or unpair, the master sends a POLL packet to verify the connection. In this packet the master initializes the ARQN bit to NAK. The response packet sent by the slave also has the ARQN bit set to NAK. The subsequent packets use the following rules.

The ARQ bit is affected by data packets containing CRC and empty slots only. As shown in Fig. 5.3 on page 68, upon successful reception of a CRC packet, the ARQN bit is set to ACK. If, in any receive slot in the slave, or, in a receive slot in the master following transmission of a packet, one of these events applies:



1. no access code is detected,
2. the HEC fails,
3. the CRC fails,

then the ARQN bit is set to NAK.

Packets that have correct HEC but that are addressed to other slaves, or packets other than DH, DM, or DV packets, do not affect the ARQN bit. In these cases the ARQN bit is left as it was prior to reception of the packet. If a CRC packet with a correct header has the same SEQN as the previously received CRC packet, the ARQN bit is set to ACK and the payload is disregarded without checking the CRC.

The ARQ bit in the FHS packet is not meaningful. Contents of the ARQN bit in the FHS packet should not be checked.

Broadcast packets are checked on errors using the CRC, but no ARQ scheme is applied. Broadcast packets are never acknowledged.

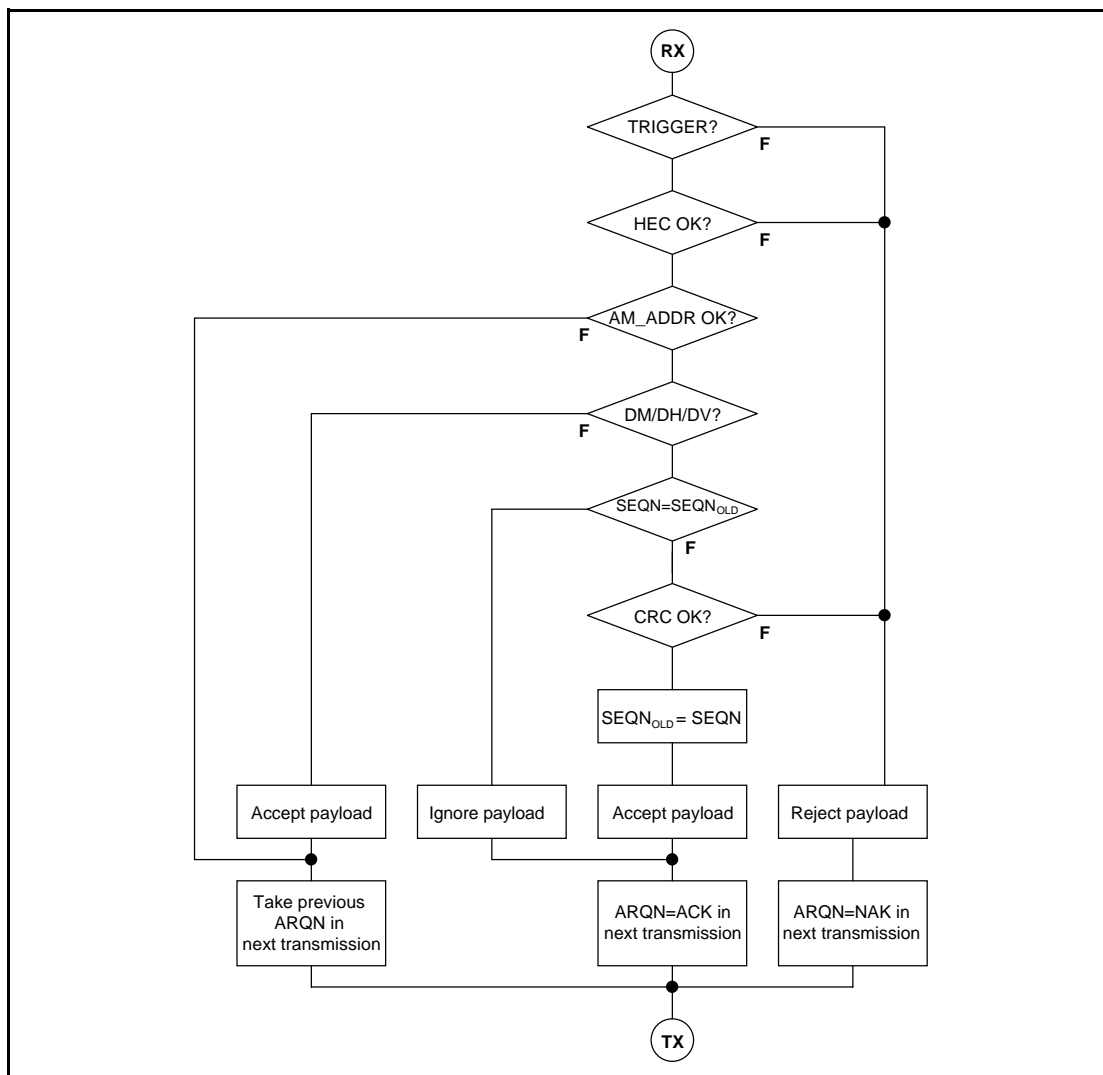


Figure 5.3: Receive protocol for determining the ARQN bit.

5.3.2 Retransmit filtering

The data payload is retransmitted until a positive acknowledgment is received (or a timeout is exceeded). A retransmission is carried out either because the packet transmission itself failed, or because the piggy-backed acknowledgment transmitted in the return packet failed (note that the latter has a lower failure probability since the header is more heavily coded). In the latter case, the destination keeps receiving the same payload over and over again. In order to filter out the retransmissions in the destination, the SEQN bit is added in the header. Normally, this bit is alternated for every new CRC data payload transmission. In case of a retransmission, this bit is not changed. So the destination can compare the SEQN bit with the previous SEQN value. If different, a new data payload has arrived; otherwise it is the same data payload and can be discarded. Only new data payloads are transferred to the link manager. Note that CRC data payloads can be carried only by **DM**, **DH** or **DV** packets.

At the start of a new connection which may be the result of a page, page scan, master slave switch or unpair, the master sends a POLL packet to verify the connection. The slave responds with a packet. The SEQN bit of the first CRC data packet, on both the master and the slave sides, is set to 1. The subsequent packets use the rules given below.

The SEQN bit is affected only by the CRC data packets as shown in [Figure 5.4](#). It is inverted every time a new CRC data packet is sent. The CRC data packet is retransmitted with the same SEQN number until an ACK is received or the packet is flushed. When an ACK is received, the SEQN bit is inverted and a new payload is sent. When the packet is flushed (see [Section 5.3.3 on page 70](#)), a new payload is sent. The SEQN bit is not necessarily inverted. However, if an ACK is received before the new packet is sent, the SEQN bit is inverted. This procedure prevents loss of the first packet of a message (after the **flush** command has been given) due to the retransmit filtering.

If a device decides to flush, and it has not received an acknowledgement for the current packet, it replaces the current packet with an ACL L2CAP continuation packet with length zero. It transmits this packet with the same sequence number as the packet it is trying to flush until it does get an ACK. Only then can it move on to the next packet. If a flush is needed for the next packet in the transmit queue before the zero-length packet has been transmitted, that next packet can be removed from the queue directly without it also being replaced by a zero-length packet. The described flushing procedure is considered optional, although strongly recommended.

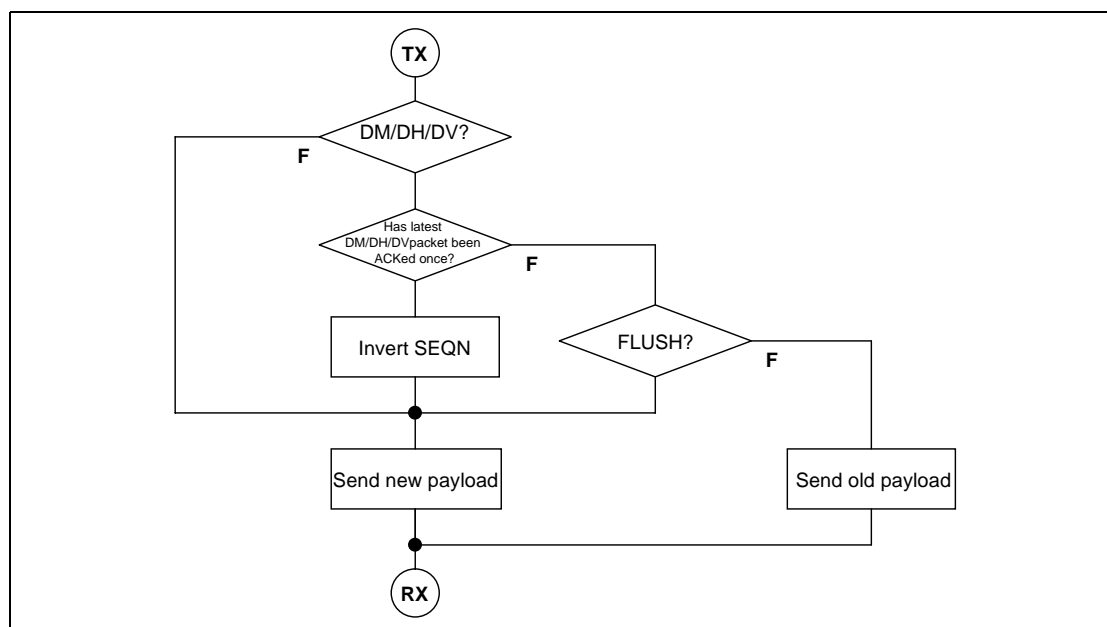


Figure 5.4: Retransmit filtering for packets with CRC.

The SEQN bit in the FHS packet is not meaningful. This bit can be set to any value. Contents of the SEQN bit in the FHS packet should not be checked. During transmission of all other packets the SEQN bit remains the same as it was in the previous packet.

5.3.3 Flushing payloads

The ARQ scheme can cause variable delay in the traffic flow since retransmissions are inserted to assure error-free data transfer. For certain communication links, only a limited amount of delay is allowed: retransmissions are allowed up to a certain limit at which the current payload must be disregarded and the next payload must be considered. This data transfer is indicated as **isochronous traffic**. This means that the retransmit process must be overruled in order to continue with the next data payload. Aborting the retransmit scheme is accomplished by *flushing* the old data and forcing the Bluetooth controller to take the next data instead.

Flushing results in loss of remaining portions of an L2CAP message. Therefore, the packet following the flush will have a start packet indication of L_CH = 10 in the payload header for the next L2CAP message. This informs the destination of the flush. (see [Section 4.5 on page 62](#)). Flushing will not necessarily result in a change in the SEQN bit value, see the previous section.

5.3.4 Multi-slave considerations

In case of a piconet with multiple slaves, the master carries out the ARQ protocol independently to each slave.

5.3.5 Broadcast packets

Broadcast packets are packets transmitted by the master to all the slaves simultaneously. A broadcast packet is indicated by the all-zero AM_ADDR (note; the FHS packet is the only packet which may have an all-zero address but is not a broadcast packet). Broadcast packets are not acknowledged (at least not at the LC level).

Since broadcast messages are not acknowledged, each broadcast packet is repeated for a fixed number of times. A broadcast packet is repeated N_{BC} times before the next broadcast packet of the same broadcast message is repeated, see Figure 5.5 on page 71. However, time - critical broadcast information may abort the ongoing broadcast repetition train. For instance, unpark messages sent at beacon instances may do this, see Section 10.8.4.5 on page 117.

Broadcast packets with a CRC have their own sequence number. The SEQN of the first broadcast packet with a CRC is set to SEQN = 1 by the master and it is inverted for each new broadcast packet with CRC thereafter. Broadcast packets without a CRC have no influence on the sequence number. The slave accepts the SEQN of the first broadcast packet it receives in a connection and checks for change in SEQN for consequent broadcast packets. Since there is no acknowledgement of broadcast messages and there is no end packet indication, it is important to receive the start packets correctly. To ensure this, repetitions of the broadcast packets that are L2CAP start packets and LMP packets will not be filtered out. These packets are indicated by L_CH=1X in the payload header as explained in section 4.5 on page 62. Only repetitions of the L2CAP continuation packets will be filtered out.

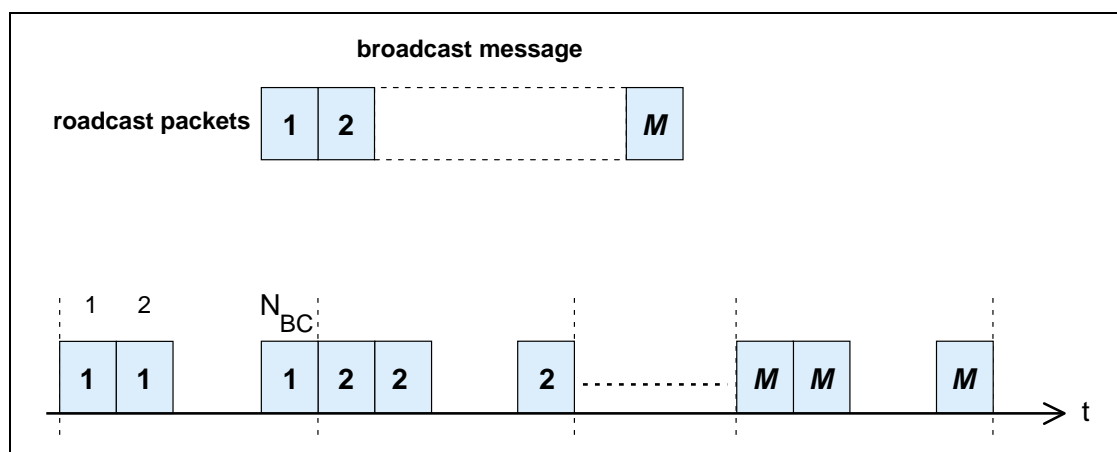


Figure 5.5: Broadcast repetition scheme

5.4 ERROR CHECKING

We can check the packet for errors or wrong delivery using the channel access code, the HEC in the header, and the CRC in the payload. At packet reception, first the access code is checked. Since the 64-bit sync word in the channel access code is derived from the 24-bit master LAP, this checks if the LAP is correct, and prevents the receiver from accepting a packet of another piconet.

The HEC and CRC are used to check both on errors and on a wrong address: to increase the address space with 8 bits, the UAP is normally included in the HEC and CRC checks. Then, even when a packet with the same access code – i.e., an access code of a device owning the same LAP but different UAP – passes the access code test, it will be discarded after the HEC and CRC tests when the UAP bits do not match. However, there is an exception where no common UAP is available in the transmitter and receiver. This is the case when the HEC and CRC are generated for the FHS packet in **inquiry response** state. In this case the default check initialization (DCI) value is used. The DCI is defined to be 0x00 (hexadecimal).

The generation and check of the HEC and CRC are summarized in [Figure 5.8 on page 73](#) and [Figure 5.11 on page 74](#). Before calculating the HEC or CRC, the shift registers in the HEC/CRC generators are initialized with the 8-bit UAP (or DCI) value. Then the header and payload information is shifted into the HEC and CRC generators, respectively (with the LSB first).

The HEC generating LFSR is depicted in [Figure 5.6 on page 72](#). The generator polynomial is $.g(D) = (D + 1)(D^7 + D^4 + D^3 + D^2 + 1) = D^8 + D^7 + D^5 + D^2 + D + 1$. Initially this circuit is pre-loaded with the 8-bit UAP such that the LSB of the UAP (denoted UAP₀) goes to the left-most shift register element, and, UAP₇ goes to the right-most element. The initial state of the HEC LFSR is depicted in [Figure 5.7 on page 73](#). Then the data is shifted in with the switch S set in position 1. When the last data bit has been clocked into the LFSR, the switch S is set in position 2, and, the HEC can be read out from the register. The LFSR bits are read out from right to left (i.e., the bit in position 7 is the first to be transmitted, followed by the bit in position 6, etc.).

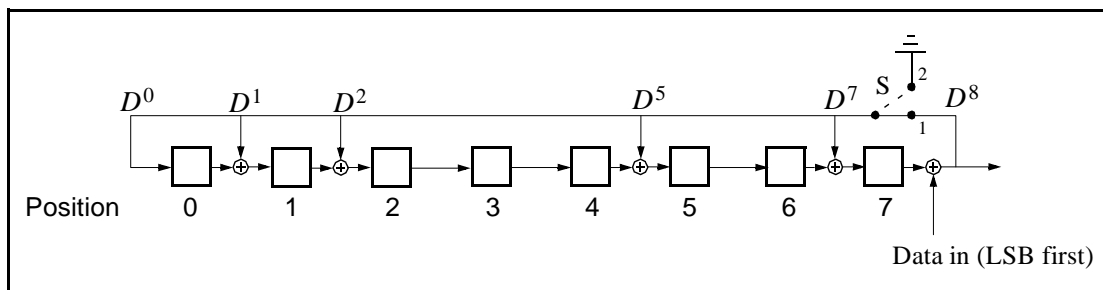


Figure 5.6: The LFSR circuit generating the HEC.

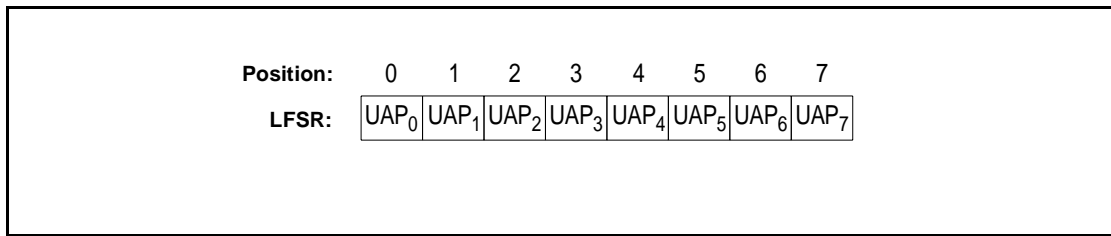


Figure 5.7: Initial state of the HEC generating circuit.

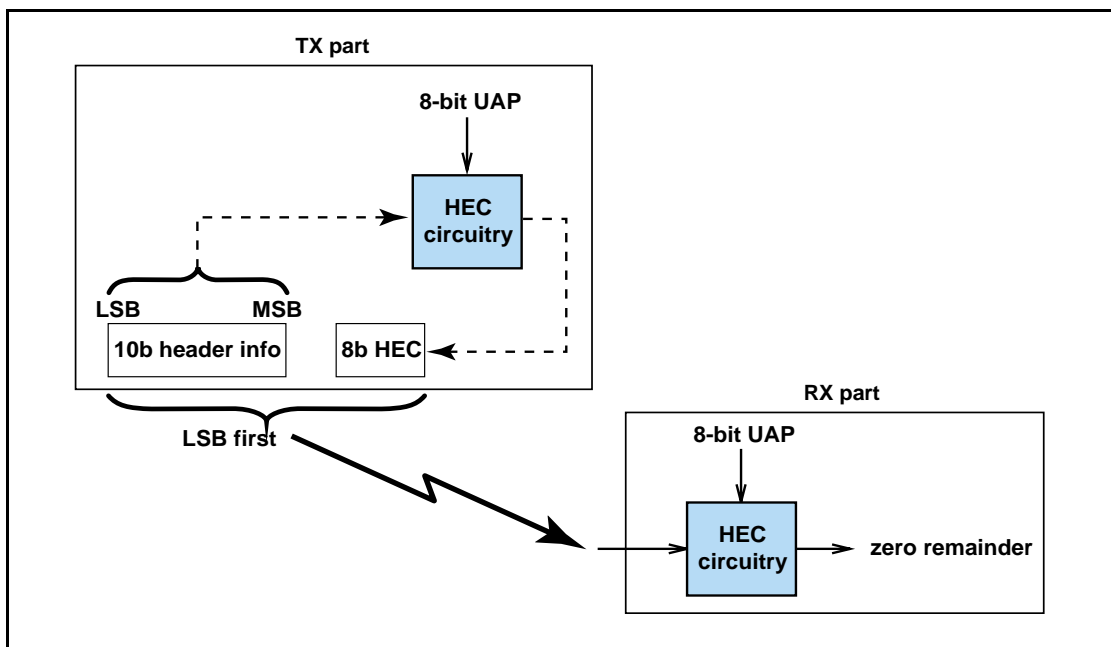


Figure 5.8: HEC generation and checking.

The 16 bit LFSR for the CRC is constructed similarly using the CRC-CCITT generator polynomial $g(D) = D^{16} + D^{12} + D^5 + 1$ (see Figure 5.9 on page 73). For this case, the 8 left-most bits are initially loaded with the 8-bit UAP (UAP₀ to the left and UAP₇ to the right) while the 8 right-most bits are reset to zero. The initial state of the 16 bit LFSR is depicted in Figure 5.10 on page 74. The switch S is set in position 1 while the data is shifted in. After the last bit has entered the LFSR, the switch is set in position 2, and, the register's contents are transmitted, from right to left (i.e., starting with position 15, then position 14, etc.).

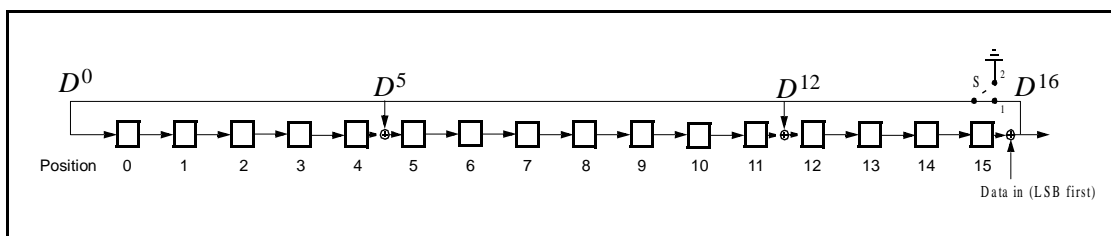


Figure 5.9: The LFSR circuit generating the CRC.



Position:	0	1	2	3	4	5	6	7	8	9	10	11	12	13	14	15
LFSR:	UAP ₀	UAP ₁	UAP ₂	UAP ₃	UAP ₄	UAP ₅	UAP ₆	UAP ₇	0	0	0	0	0	0	0	0

Figure 5.10: Initial state of the CRC generating circuit.

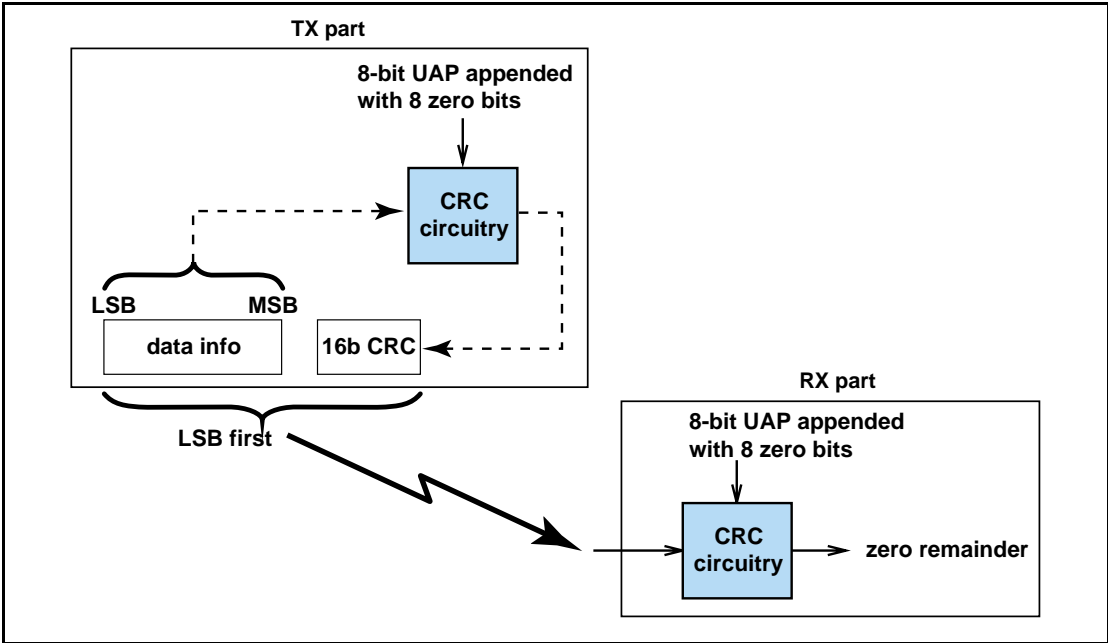


Figure 5.11: CRC generation and checking

6 LOGICAL CHANNELS

In the Bluetooth system, five logical channels are defined:

- LC control channel
- LM control channel
- UA user channel
- UI user channel
- US user channel

The control channels LC and LM are used at the link control level and link manager level, respectively. The user channels UA, UI, and US are used to carry asynchronous, isochronous, and synchronous user information, respectively. The LC channel is carried in the packet header; all other channels are carried in the packet payload. The LM, UA, and UI channels are indicated in the L_CH field in the payload header. The US channel is carried by the SCO link only; the UA and UI channels are normally carried by the ACL link; however, they can also be carried by the data in the DV packet on the SCO link. The LM channel can be carried either by the SCO or the ACL link.

6.1 LC CHANNEL (Link Control)

The LC control channel is mapped onto the packet header. This channel carries low level link control information like ARQ, flow control, and payload characterization. The LC channel is carried in every packet except in the **ID** packet which has no packet header.

6.2 LM CHANNEL (Link Manager)

The LM control channel carries control information exchanged between the link managers of the master and the slave(s). Typically, the LM channel uses protected **DM** packets. The LM channel is indicated by the L_CH code 11 in the payload header.

6.3 UA/UI CHANNEL (User Asynchronous/Isochronous Data)

The UA channel carries L2CAP transparent asynchronous user data. This data may be transmitted in one or more baseband packets. For fragmented messages, the start packet uses an L_CH code of 10 in the payload header. Remaining continuation packets use L_CH code 01. If there is no fragmentation, all packets use the L2CAP start code 10.

Isochronous data channel is supported by timing start packets properly at higher levels. At the baseband level, the L_CH code usage is the same as the UA channel.



6.4 US CHANNEL (User Synchronous Data)

The US channel carries transparent synchronous user data. This channel is carried over the SCO link.

6.5 CHANNEL MAPPING

The LC channel is mapped onto the packet header. All other channels are mapped onto the payload. The US channel can only be mapped onto the SCO packets. All other channels are mapped on the ACL packets, or possibly the SCO **DV** packet. The LM, UA, and UI channels may interrupt the US channel if it concerns information of higher priority.

7 DATA WHITENING

Before transmission, both the header and the payload are scrambled with a data whitening word in order to randomize the data from highly redundant patterns and to minimize DC bias in the packet. The scrambling is performed prior to the FEC encoding.

At the receiver, the received data is descrambled using the same whitening word generated in the recipient. The descrambling is performed after FEC decoding.

The whitening word is generated with the polynomial $g(D) = D^7 + D^4 + 1$ (i.e., 221 in octal representation) and is subsequently EXORed with the header and the payload. The whitening word is generated with the linear feedback shift register shown in [Figure 7.1 on page 77](#). Before each transmission, the shift register is initialized with a portion of the master Bluetooth clock, CLK_{6-1} , extended with an MSB of value one. This initialization is carried out with CLK_1 written to position 0, CLK_2 written to position 1, etc. An exception forms the FHS packet sent during frequency hop acquisition, where initialization of the whitening register is carried out differently. Instead of the master clock, the X-input used in the **inquiry** or **page response** (depending on current state) routine is used, see [Table 11.3](#) and [Table 11.4](#) for the 79-hop and 23-hop systems, respectively. In case of a 79-hop system, the 5-bit values is extended with two MSBs of value one. In case of a 23-hop system, the 4-bit value is extended with three bits; the two MSBs are set to one and the third most significant bit is set to zero. During register initialization, the LSB of X (i.e., X_0) is written to position 0, X_1 is written to position 1, etc.

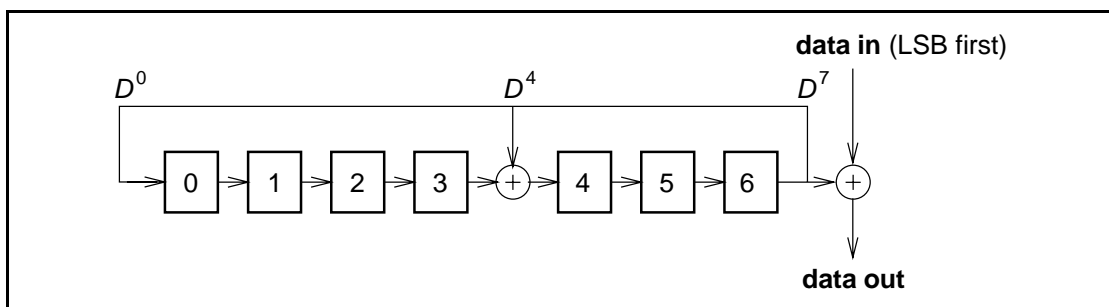


Figure 7.1: Data whitening LFSR.

After initialization, the packet header and the payload (including the CRC) are scrambled. The payload whitening continues from the state the whitening LFSR had at the end of HEC. There is no re-initialization of the shift register between packet header and payload. The first bit of the “Data In” sequence is the LSB of the packet header.

8 TRANSMIT/RECEIVE ROUTINES

This section describes the way to use the packets as defined in [Section 4 on page 47](#) in order to support the traffic on the ACL and SCO links. Both single-slave and multi-slave configurations are considered. In addition, the use of buffers for the TX and RX routines are described.

The TX and RX routines described in sections 8.1 and 8.2 are of an informative character only. The final implementation may be carried out differently.

8.1 TX ROUTINE

The TX routine is carried out separately for each ACL link and each SCO link. [Figure 8.1 on page 78](#) shows the ACL and SCO buffers as used in the TX routine. In this figure, only a single TX ACL buffer and a single TX SCO buffer are shown. In the master, there is a separate TX ACL buffer for each slave. In addition there may be one or more TX SCO buffers for each SCO slave (different SCO links may either reuse the same TX SCO buffer, or each have their own TX SCO buffer). Each TX buffer consists of two FIFO registers: one **current** register which can be accessed and read by the Bluetooth controller in order to compose the packets, and one **next** register that can be accessed by the Bluetooth Link Manager to load new information. The positions of the switches S1 and S2 determine which register is current and which register is next; the switches are controlled by the Bluetooth Link Controller. The switches at the input and the output of the FIFO registers can never be connected to the same register simultaneously.

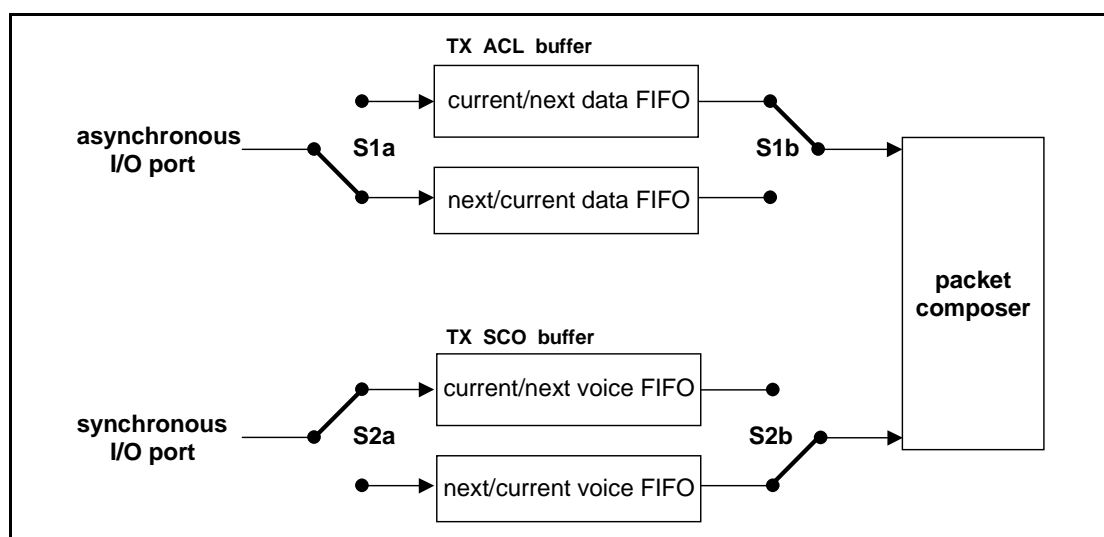


Figure 8.1: Functional diagram of TX buffering.

Of the packets common on the ACL and SCO links (**ID**, **NULL**, **POLL**, **FHS**, **DM1**) only the **DM1** packet carries a payload that is exchanged between the Link Controller and the Link Manager; this common packet makes use of the

ACL buffer. All ACL packets make use of the ACL buffer. All SCO packets make use of the SCO buffer except for the **DV** packet where the voice part is handled by the SCO buffer and the data part is handled by the ACL buffer. In the next sections, the operation for ACL traffic, SCO traffic, and combined data-voice traffic on the SCO link will be considered.

8.1.1 ACL traffic

In the case of pure (asynchronous) data, only the TX ACL buffer in [Figure 8.1 on page 78](#) has to be considered. In this case, only packet types **DM** or **DH** are used, and these can have different lengths. The length is indicated in the payload header. The selection of high-rate data or medium-rate data shall depend on the quality of the link. When the quality is good, the FEC in the data payload can be omitted, resulting in a **DH** packet. Otherwise, **DM** packets must be used.

The default TYPE in pure data traffic is **NULL**. This means that, if there is no data to be sent (the data traffic is asynchronous, and therefore pauses occur in which no data is available) or no slaves need to be polled, **NULL** packets are sent instead – in order to send link control information to the other Bluetooth unit (e.g. ACK/STOP information for received data). When no link control information is available either (no need to acknowledge and/or no need to stop the RX flow) no packet is sent at all.

The TX routine works as follows. The Bluetooth Link Manager loads new data information in the register to which the switch S1a points. Next, it gives a **flush** command to the Bluetooth Link Controller, which forces the switch S1 to change (both S1a and S1b switch in synchrony). When the payload needs to be sent, the packet composer reads the current register and, depending on the packet TYPE, builds a payload which is appended to the channel access code and the header and is subsequently transmitted. In the response packet (which arrives in the following RX slot if it concerned a master transmission, or may be postponed until some later RX slot if it concerned a slave transmission), the result of the transmission is reported back. In case of an ACK, the switch S1 changes position; if a NAK (explicit or implicit) is received instead, the switch S1 will not change position. In that case, the same payload is retransmitted at the next TX occasion.

As long as the Link Manager keeps loading the registers with new information, the Bluetooth Link Controller will automatically transmit the payload; in addition, retransmissions are performed automatically in case of errors. The Link Controller will send **NULL** or nothing when no new data is loaded. If no new data has been loaded in the **next** register, during the last transmission, the packet composer will be pointing to an empty register after the last transmission has been acknowledged and the **next** register becomes the **current** register. If new data is loaded in the **next** register, a **flush** command is required to switch the S1 switch to the proper register. As long as the Link Manager keeps loading the data and type registers before each TX slot, the data is automatically processed by the Link Controller since the S1 switch is controlled by the



ACK information received in response. However, if the traffic from the Link Manager is interrupted once and a default packet is sent instead, a **flush** command is required to continue the flow in the Link Controller.

The **flush** command can also be used in case of time-bounded (isochronous) data. In case of a bad link, many retransmission are necessary. In certain applications, the data is time-bounded: if a payload is retransmitted all the time because of link errors, it may become outdated, and the system might decide to continue with more recent data instead and skip the payload that does not come through. This is accomplished by the **flush** command as well. With the **flush**, the switch S1 is forced to change and the Link Controller is forced to consider the next data payload and overrules the ACK control.

8.1.2 SCO traffic

In case of an SCO link, we only use **HV** packet types. The synchronous port continuously loads the **next** register in the SCO buffer. The S2 switches are changed according to the T_{SCO} interval. This T_{SCO} interval is negotiated between the master and the slave at the time the SCO link is established.

For each new SCO slot, the packet composer reads the **current** register after which the S2 switch is changed. If the SCO slot has to be used to send control information with high priority concerning a control packet between the master and the considered SCO slave, or a control packet between the master and any other slave, the packet composer will discard the SCO information and use the control information instead. This control information must be sent in a DM1 packet. Data or link control information can also be exchanged between the master and the SCO slave by using the **DV** or **DM1** packets. Any ACL type of packet can be used to sent data or link control information to any other ACL slave. This is discussed next.

8.1.3 Mixed data/voice traffic

In [Section 4.4.2 on page 58](#), a **DV** packet has been defined that can support both data and voice simultaneously on a single SCO link. When the TYPE is **DV**, the Link Controller reads the data register to fill the data field and the voice register to fill the voice field. Thereafter, the switch S2 is changed. However, the position of S1 depends on the result of the transmission like on the ACL link: only if an ACK has been received will the S1 switch change its position. In each **DV** packet, the voice information is new, but the data information might be retransmitted if the previous transmission failed. If there is no data to be sent, the SCO link will automatically change from **DV** packet type to the current **HV** packet type used before the mixed data/voice transmission. Note that a **flush** command is required when the data stream has been interrupted and new data has arrived.

Combined data-voice transmission can also be accomplished by using separate ACL links in addition to the SCO link(s) if channel capacity permits this.

8.1.4 Default packet types

On the ACL links, the default type is always **NULL** both for the master and the slave. This means that if no user information needs to be send, either a **NULL** packet is sent if there is **ACK** or **STOP** information, or no packet is sent at all. The **NULL** packet can be used by the master to allocate the next slave-to-master slot to a certain slave (namely the one addressed). However, the slave is not forced to respond to the **NULL** packet from the master. If the master requires a response, it has to send a **POLL** packet.

The SCO packet type is negotiated at the LM level when the SCO link is established. The agreed packet type is also the default packet type for the SCO slots.

8.2 RX ROUTINE

The RX routine is carried out separately for the ACL link and the SCO link. However, in contrast to the master TX ACL buffer, a single RX buffer is shared among all slaves. For the SCO buffer, it depends how the different SCO links are distinguished whether extra SCO buffers are required or not. [Figure 8.2 on page 81](#) shows the ACL and SCO buffers as used in the RX routine. The RX ACL buffer consists of two FIFO registers: one register that can be accessed and loaded by the Bluetooth Link Controller with the payload of the latest RX packet, and one register that can be accessed by the Bluetooth Link Manager to read the previous payload. The RX SCO buffer also consists of two FIFO registers: one register which is filled with newly arrived voice information, and one register which can be read by the voice processing unit.

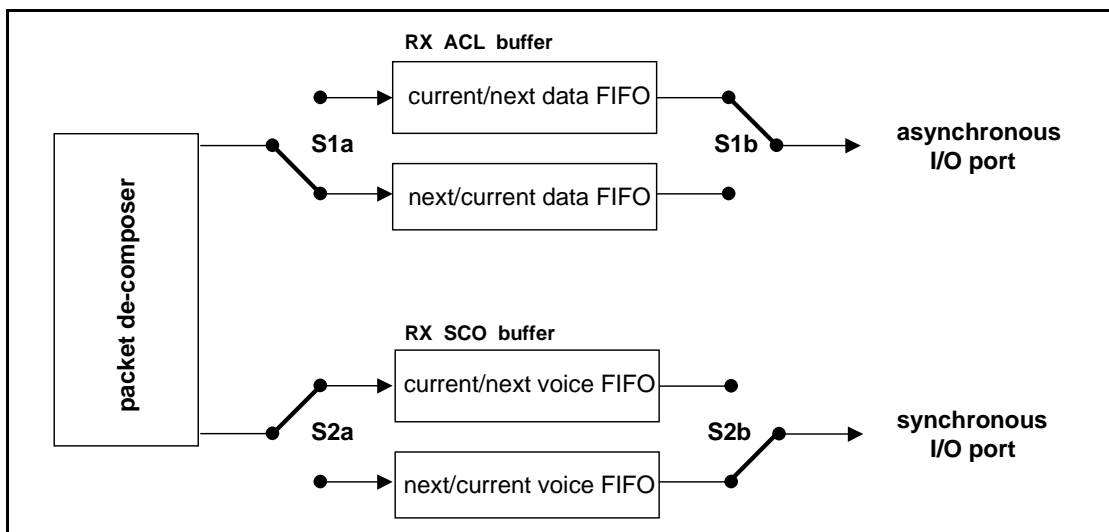


Figure 8.2: Functional diagram of RX buffering

Since the TYPE indication in the header of the received packet indicates whether the payload contains data and/or voice, the packet de-composer can automatically direct the traffic to the proper buffers. The switch S1 changes



every time the Link Manager has read the old register. If the next payload arrives before the RX register is emptied, a STOP indication must be included in the packet header of the next TX packet that is returned. The STOP indication is removed again as soon as the RX register is emptied. The SEQN field is checked before a new ACL payload is stored into the ACL register (flush indication in L_CH and broadcast messages influence the interpretation of the SEQN field see [Section 5.3 on page 67](#)).

The S2 switch is changed every T_{SCO} . If – due to errors in the header – no new voice payload arrives, the switch still changes. The voice processing unit then has to process the voice signal to account for the missing speech parts.

8.3 FLOW CONTROL

Since the RX ACL buffer can be full while a new payload arrives, flow control is required. As was mentioned earlier, the header field FLOW in the return TX packet can use STOP or GO in order to control the transmission of new data.

8.3.1 Destination control

As long as data cannot be received, a STOP indication is transmitted which is automatically inserted by the Link Controller into the header of the return packet. STOP is returned as long as the RX ACL buffer is not emptied by the Link Manager. When new data can be accepted again, the GO indication is returned. GO is the default value. Note that all packet types not including data can still be received. Voice communication for example is not affected by the flow control. Also note that although a Bluetooth unit cannot receive new information, it can still continue to transmit information: the flow control is separate for each direction.

8.3.2 Source control

On the reception of a STOP signal, the Link Controller will automatically switch to the default packet type. The ACL packet transmitted just before the reception of the STOP indication is kept until a GO signal is received. It is retransmitted as soon as a GO indication is received. Default packets are sent as long as the STOP indication is received. When no packet is received, GO is assumed implicitly. Note that the default packets contain link control information (in the header) for the receive direction (which may still be open) and may contain voice (**HV** packets). When a GO indication is received, the Link Controller resumes to transmit the data as is present in the TX ACL buffers.

In a multi-slave configuration, only the transmission to the slave that issued the STOP signal is stalled. This means that the previously described routine implemented in the master only concerns the TX ACL buffer that corresponds to the slave that cannot accept data momentarily.

8.4 BITSTREAM PROCESSES

Before the user information is sent over the air interface, several bit manipulations are performed in the transmitter to increase reliability and security. To the packet header, an HEC is added, the header bits are scrambled with a whitening word, and FEC coding is applied. In the receiver, the inverse processes are carried out. [Figure 8.3 on page 83](#) shows the processes carried out for the packet header both at the transmit and the receive side. All header bit processes are mandatory.

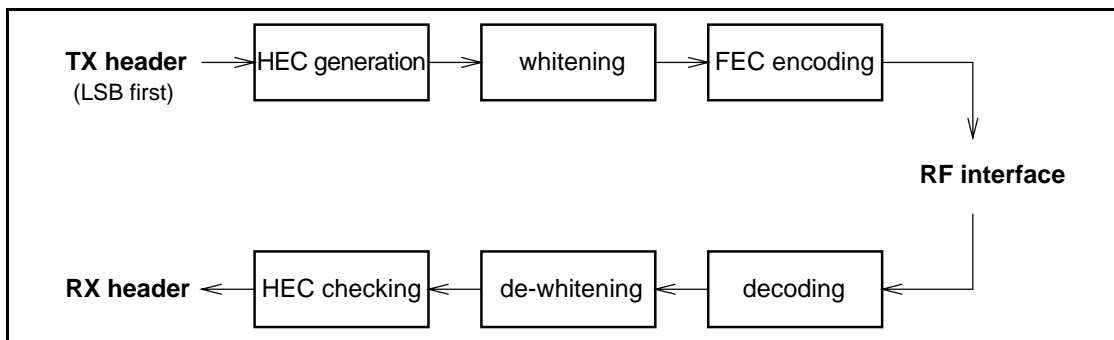


Figure 8.3: Header bit processes.

For the payload, similar processes are performed. It depends on the packet type, which processes are carried out. [Figure 8.4 on page 83](#) shows the processes that may be carried out on the payload. In addition to the processes defined for the packet header, encryption can be applied on the payload. Only whitening and de-whitening, as explained in [Section 7 on page 77](#), are mandatory for every payload; all other processes are optional and depend on the packet type and the mode enabled. In [Figure 8.4 on page 83](#), optional processes are indicated by dashed blocks.

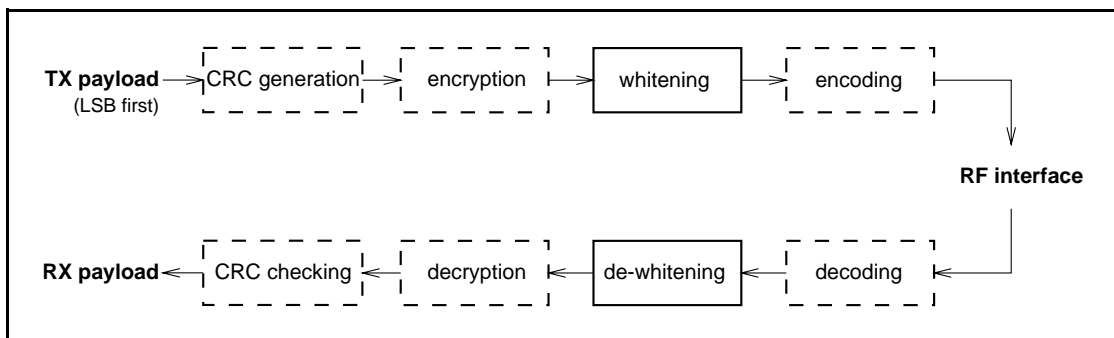


Figure 8.4: Payload bit processes.

9 TRANSMIT/RECEIVE TIMING

The Bluetooth transceiver applies a time-division duplex (TDD) scheme. This means that it alternately transmits and receives in a synchronous manner. It depends on the mode of the Bluetooth unit what the exact timing of the TDD scheme is. In the normal connection mode, *the master transmission shall always start at even numbered time slots (master CLK1=0) and the slave transmission shall always start at odd numbered time slots (master CLK1=1)*. Due to packet types that cover more than a single slot, master transmission may continue in odd numbered slots and slave transmission may continue in even numbered slots.

All timing diagrams shown in this chapter are based on the signals as present at the antenna. The term “exact” when used to describe timing refers to an ideal transmission or reception and neglects timing jitter and clock frequency imperfections.

The average timing of master packet transmission must not drift faster than 20 ppm relative to the ideal slot timing of 625 μs . The instantaneous timing must not deviate more than 1 μs from the average timing. Thus, the absolute packet transmission timing t_k of slot boundary k must fulfill the equation:

$$t_k = \left(\sum_{i=1}^k (1 + d_i) T_N \right) + j_k + \text{offset}, \quad (\text{EQ 1})$$

where T_N is the nominal slot length (625 μs), j_k denotes jitter ($|j_k| \leq 1 \mu\text{s}$) at slot boundary k , and, d_k , denotes the drift ($|d_k| \leq 20 \text{ ppm}$) within slot k . The jitter and drift may vary arbitrarily within the given limits for every slot, while “offset” is an arbitrary but fixed constant. For hold, park and sniff mode the drift and jitter parameters as described in [Link Manager Protocol Section 3.9 on page 205](#) apply.

9.1 MASTER/SLAVE TIMING SYNCHRONIZATION

The piconet is synchronized by the system clock of the master. The master never adjusts its system clock during the existence of the piconet: it keeps an exact interval of $M \times 625 \mu\text{s}$ (where M is an even, positive integer larger than 0) between consecutive transmissions. The slaves adapt their native clocks with a timing offset in order to match the master clock. This offset is updated each time a packet is received from the master: by comparing the exact RX timing of the received packet with the estimated RX timing, the slaves correct the offset for any timing misalignments. Note that the slave RX timing can be corrected with any packet sent in the master-to-slave slot, since only the channel access code is required to synchronize the slave.

The slave TX timing shall be based on the most recent slave RX timing. The RX timing is based on the latest successful trigger during a master-to-slave slot. For ACL links, this trigger must have occurred in the master-to-slave slot directly pre-

ceding the current slave transmission; for SCO links, the trigger may have occurred several master-to-slave slots before since a slave is allowed to send an SCO packet even if no packet was received in the preceding master-to-slave slot. The slave shall be able to receive the packets and adjust the RX timing as long as the timing mismatch remains within the $\pm 10 \mu\text{s}$ uncertainty window.

The master TX timing is strictly related to the master clock. The master shall keep an exact interval of $M \times 1250 \mu\text{s}$ (where M is a positive integer larger than 0) between the start of successive transmissions; the RX timing is based on this TX timing with a shift of exactly $N \times 625 \mu\text{s}$ (where N is an odd, positive integer larger than 0). During the master RX cycle, the master will also use the $\pm 10 \mu\text{s}$ uncertainty window to allow for slave misalignments. The master will adjust the RX processing of the considered packet accordingly, but will **not** adjust its RX/TX timing for the following TX and RX cycles. During periods when an active slave is not able to receive any valid channel access codes from the master, the slave may increase its receive uncertainty window and/or use predicted timing drift to increase the probability of receiving the master's bursts when reception resumes.

Timing behaviour may differ slightly depending on the current state of the unit. The different states are described in the next sections.

9.2 CONNECTION STATE

In the connection mode, the Bluetooth transceiver transmits and receives alternately, see [Figure 9.1 on page 85](#) and [Figure 9.2 on page 86](#). In the figures, only single-slot packets are shown as an example. Depending on the type and the payload length, the packet size can be up to $366 \mu\text{s}$. Each RX and TX transmission is at a different hop frequency. For multi-slot packets, several slots are covered by the same packet, and the hop frequency used in the first slot will be used throughout the transmission.

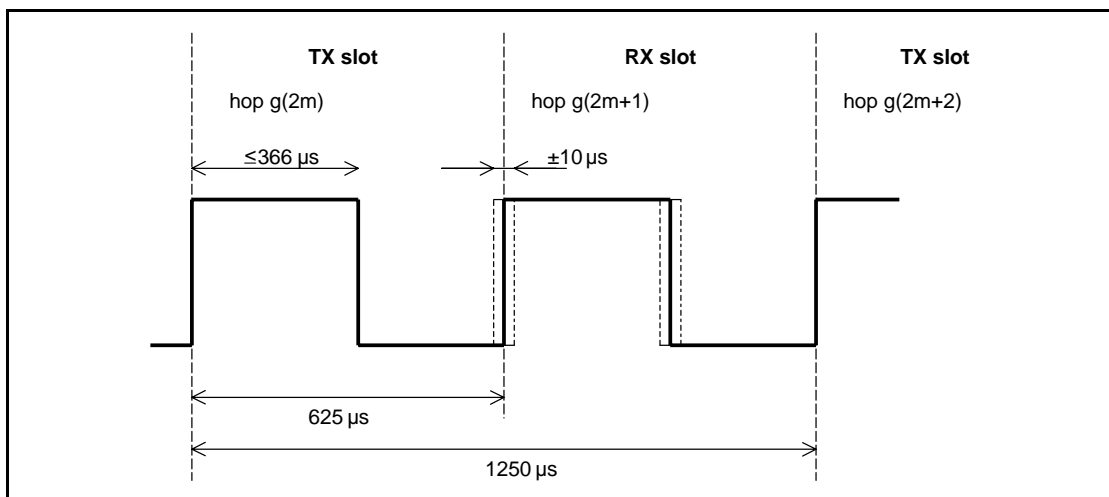


Figure 9.1: RX/TX cycle of Bluetooth master transceiver in normal mode for single-slot packets.

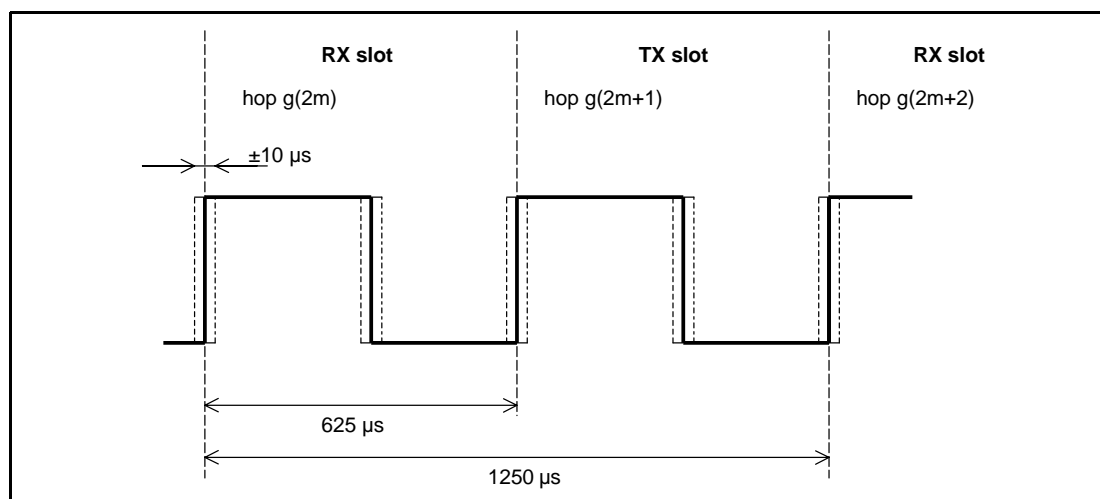


Figure 9.2: RX/TX cycle of Bluetooth slave transceiver in normal mode for single-slot packets.

The master TX/RX timing is shown in [Figure 9.1 on page 85](#). In figures 9.1 through 9.6, $f(k)$ is used for the frequencies of the page hopping sequence and $f'(k)$ denotes the corresponding page response sequence frequencies. The channel hopping frequencies are indicated by $g(m)$. After transmission, a return packet is expected $N \times 625 \mu\text{s}$ after the start of the TX burst where N is an odd, positive integer. N depends on the type of the transmitted packet. To allow for some time slipping, an uncertainty window is defined around the exact receive timing. During normal operation, the window length is $20 \mu\text{s}$, which allows the RX burst to arrive up to $10 \mu\text{s}$ too early or $10 \mu\text{s}$ too late. During the beginning of the RX cycle, the access correlator searches for the correct channel access code over the uncertainty window. If no trigger event occurs, the receiver goes to sleep until the next RX event. If in the course of the search, it becomes apparent that the correlation output will never exceed the final threshold, the receiver may go to sleep earlier. If a trigger event does occur, the receiver remains open to receive the rest of the packet.

The current master transmission is based on the previous master transmission: it is scheduled $M \times 1250 \mu\text{s}$ after the start of the previous master TX burst where M depends on the transmitted and received packet type. Note that the master TX timing is not affected by time drifts in the slave(s). If no transmission takes place during a number of consecutive slots, the master will take the TX timing of the latest TX burst as reference.

The slave's transmission is scheduled $N \times 625 \mu\text{s}$ after the start of the slave's RX burst. If the slave's RX timing drifts, so will its TX timing. If no reception takes place during a number of consecutive slots, the slave will take the RX timing of the latest RX burst as reference.

9.3 RETURN FROM HOLD MODE

In the connection state, the Bluetooth unit can be placed in a **hold** mode, see [Section 10.8 on page 110](#). In the **hold** mode, a Bluetooth transceiver neither transmits nor receives information. When returning to the normal operation after a **hold** mode in a slave Bluetooth unit, the slave must listen for the master before it may send information. In that case, the length of the search window in the slave unit may be increased from 20 μs to a larger value $X \mu\text{s}$ as illustrated in [Figure 9.3 on page 87](#). Note that only RX hop frequencies are used: the hop frequency used in the master-to-slave (RX) slot is also used in the uncertainty window extended into the preceding time interval normally used for the slave-to-master (TX) slot.

If the length of search window (X) exceeds 1250 μs , consecutive windows shall not be centered at the start of RX hops $g(2m)$, $g(2m+2)$, ... $g(2m+2i)$ (where 'i' is an integer) to avoid overlapping search windows. Consecutive windows should instead be centred at $g(2m)$, $g(2m+4)$, ... $g(2m+4i)$, which gives a maximum value $X=2500 \mu\text{s}$, or even at $g(2m)$, $g(2m+6)$, ... $g(2m+6i)$ which gives a maximum value $X=3750 \mu\text{s}$. The RX hop frequencies used shall correspond to the RX slot numbers.

It is recommended that single slot packets are used upon return from hold to minimize the synchronization time, especially after long hold periods that require search windows exceeding 625 μs .

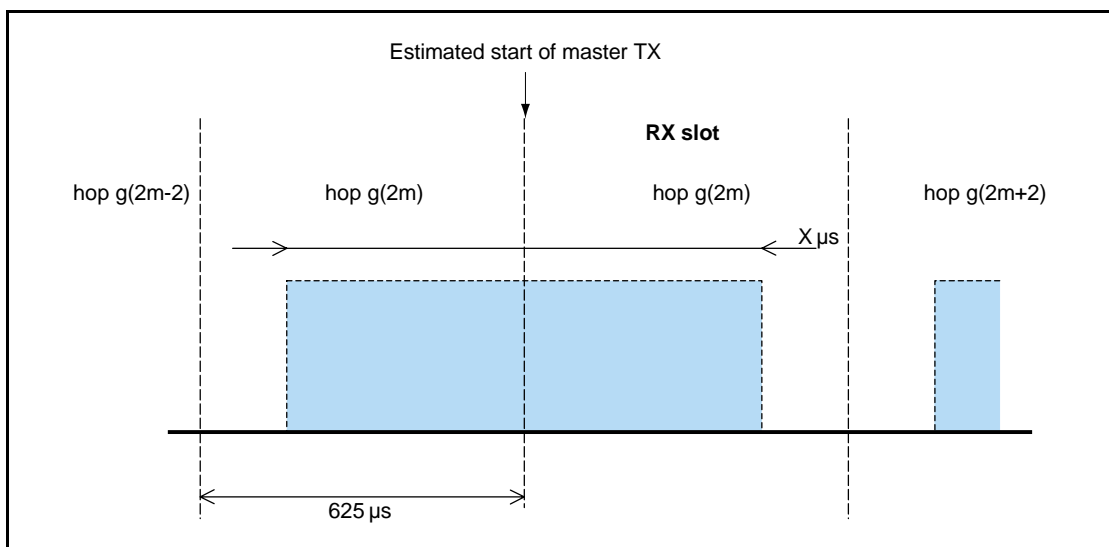


Figure 9.3: RX timing of slave returning from hold state.

9.4 PARK AND SNIFF MODES WAKE-UP

The **park** and **sniff** modes is similar to the **hold** mode. A slave in park or sniff mode periodically wakes up to listen to transmissions from the master and to re-synchronize its clock offset. As in the return from hold mode, a slave in park or sniff mode when waking up may increase the length of the search window from $20\mu\text{s}$ to a larger value $X\mu\text{s}$ as illustrated in [Figure 9.3 on page 87](#).

9.5 PAGE STATE

In the page state, the master transmits the device access code (ID packet) corresponding to the slave to be connected, rapidly on a large number of different hop frequencies. Since the ID packet is a very short packet, the hop rate can be increased from 1600 hops/s to 3200 hops/s. In a single TX slot interval, the paging master transmits on two different hop frequencies. In a single RX slot interval, the paging transceiver listens on two different hop frequencies; see [Figure 9.4 on page 88](#). During the TX slot, the paging unit sends an ID packet at the TX hop frequencies $f(k)$ and $f(k+1)$. In the RX slot, it listens for a response on the corresponding RX hop frequencies $f'(k)$ and $f'(k+1)$. The listening periods are exactly timed $625\mu\text{s}$ after the corresponding paging packets, and include a $\pm 10\mu\text{s}$ uncertainty window.

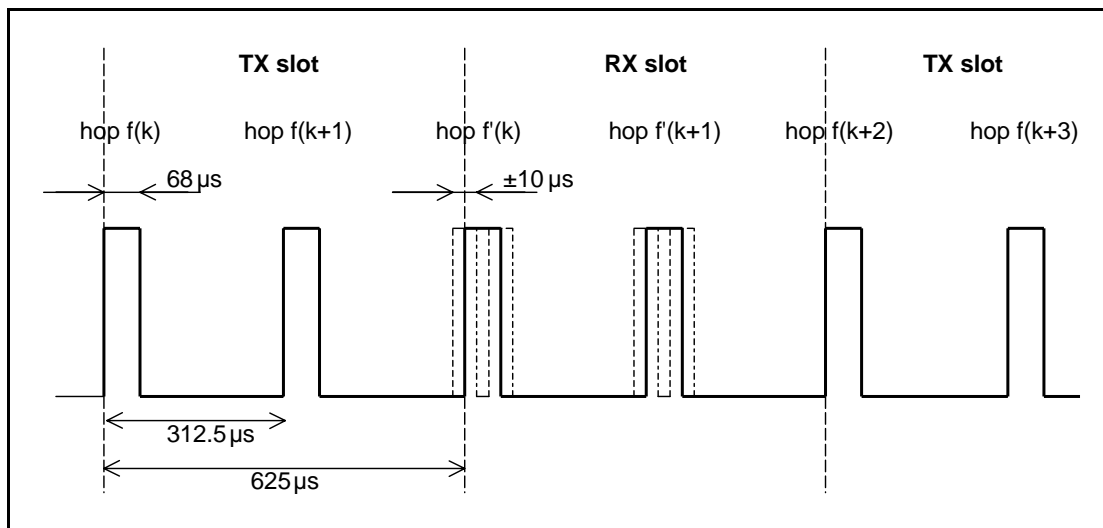


Figure 9.4: RX/TX cycle of Bluetooth transceiver in PAGE mode.

9.6 FHS PACKET

At connection setup and during a master-slave switch, an **FHS** packet is transferred from the master to the slave. This packet will establish the timing and frequency synchronization (see also [Section 4.4.1.4 on page 56](#)). After the slave unit has received the page message, it will return a response message which again consists of the ID packet and follows exactly 625 μs after the receipt of the page message. The master will send the FHS packet in the TX slot following the RX slot in which it received the slave response, according to the RX/TX timing of the master. The time difference between the response and **FHS** message will depend on the timing of the page message the slave received. In [Figure 9.5 on page 89](#), the slave receives the paging message sent **first** in the master-to-slave slot. It will then respond with an ID packet in the first half of the slave-to-master slot. The timing of the **FHS** packet is based on the timing of the page message sent first in the preceding master-to-slave slot: there is an exact 1250 μs delay between the first page message and the **FHS** packet. The packet is sent at the hop frequency $f(k+1)$ which is the hop frequency following the hop frequency $f(k)$ the page message was received in. In [Figure 9.6 on page 90](#), the slave receives the paging message sent **secondly** in the master-to-slave slot. It will then respond with an ID packet in the second half of the slave-to-master slot exactly 625 μs after the receipt of the page message. The timing of the **FHS** packet is still based on the timing of the page message sent **first** in the preceding master-to-slave slot: there is an exact 1250 μs delay between the **first** page message and the **FHS** packet. The packet is sent at the hop frequency $f(k+2)$ which is the hop frequency following the hop frequency $f(k+1)$ the page message was received in.

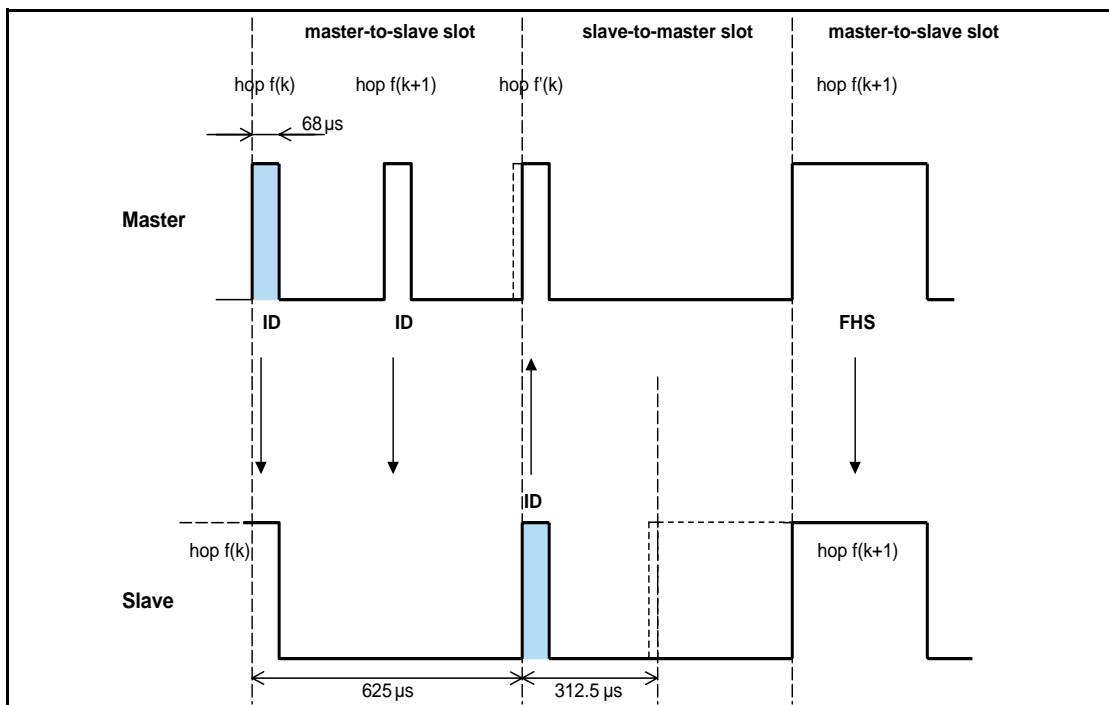


Figure 9.5: Timing of FHS packet on successful page in first half slot.

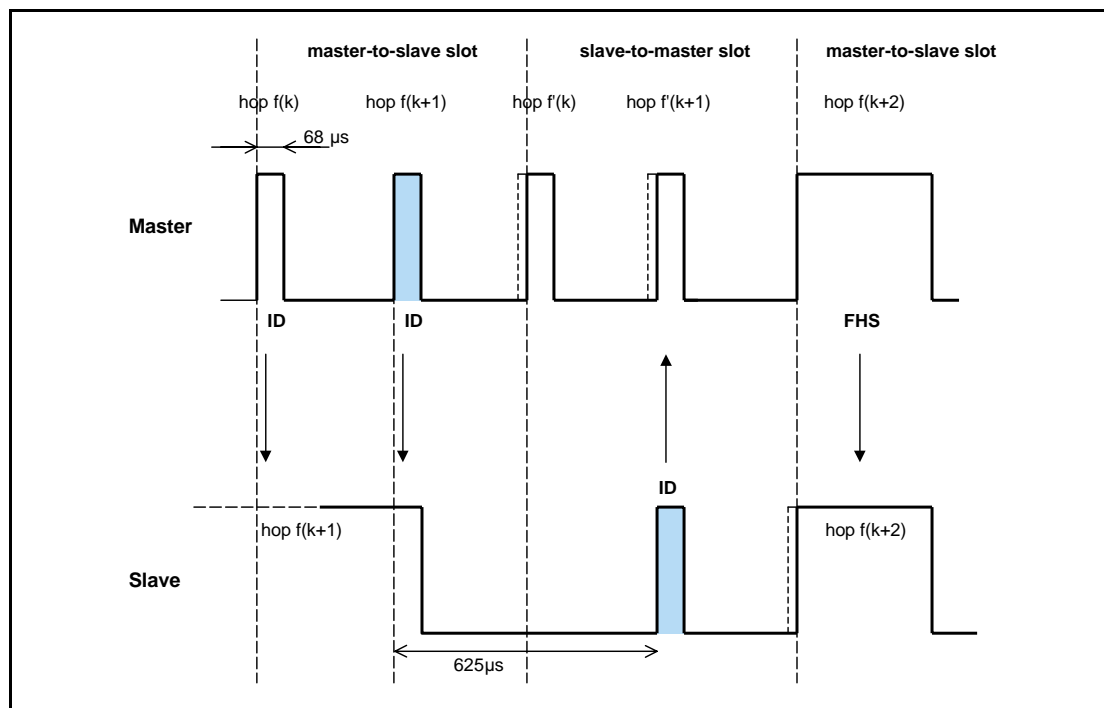


Figure 9.6: Timing of FHS packet on successful page in second half slot.

The slave will adjust its RX/TX timing according to the reception of the **FHS** packet (and not according to the reception of the page message). That is, the second response message that acknowledges the reception of the FHS packet is transmitted 625 μs after the start of the **FHS** packet.

9.7 MULTI-SLAVE OPERATION

As was mentioned in the beginning of this chapter, the master always starts the transmission in the even-numbered slots whereas the slaves start their transmission in the odd-numbered slots. This means that the timing of the master and the slave(s) is shifted by one slot (625 μs), see [Figure 9.7 on page 91](#).

Only the slave that is addressed by its AM_ADDR can return a packet in the next slave-to-master slot. If no valid AM_ADDR is received, the slave may only respond if it concerns its reserved SCO slave-to-master slot. In case of a broadcast message, no slave is allowed to return a packet (an exception is found in the access window for access requests in the park mode, see [Section 10.8.4 on page 112](#)).

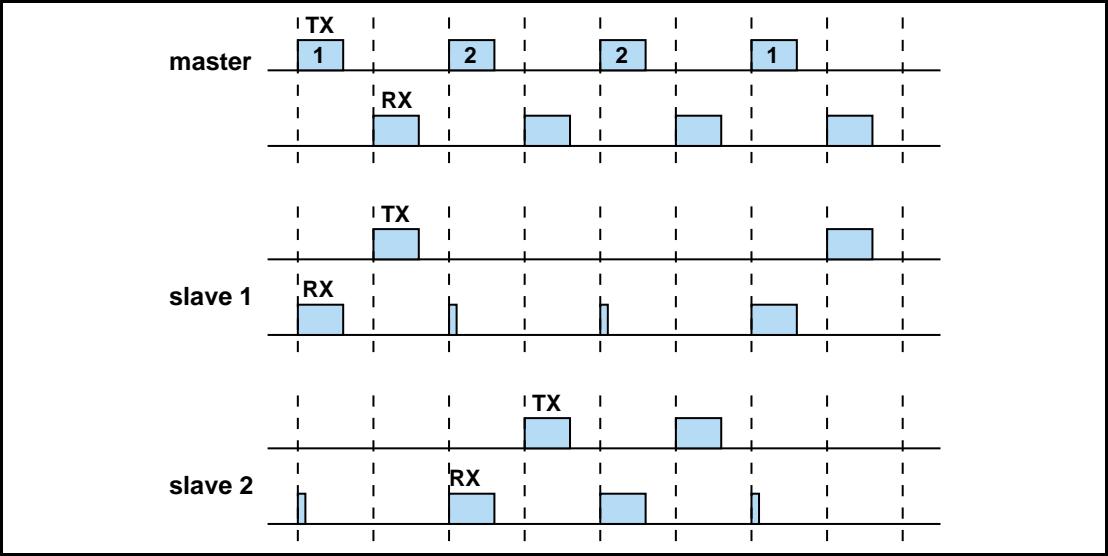


Figure 9.7: RX/TX timing in multi-slave configuration

10 CHANNEL CONTROL

10.1 SCOPE

This section describes how the channel of a piconet is established and how units can be added to and released from the piconet. Several states of operation of the Bluetooth units are defined to support these functions. In addition, the operation of several piconets sharing the same area, the so-called scatternet, is discussed. A special section is attributed to the Bluetooth clock which plays a major role in the FH synchronization.

10.2 MASTER-SLAVE DEFINITION

The channel in the piconet is characterized entirely by the master of the piconet. The Bluetooth device address (BD_ADDR) of the master determines the FH hopping sequence and the channel access code; the system clock of the master determines the phase in the hopping sequence and sets the timing. In addition, the master controls the traffic on the channel by a polling scheme.

By definition, the **master** is represented by the Bluetooth unit that initiates the connection (to one or more **slave** units). Note that the names 'master' and 'slave' only refer to the protocol on the channel: the Bluetooth units themselves are identical; that is, any unit can become a master of a piconet. Once a piconet has been established, master-slave roles can be exchanged. This is described in more detail in [Section 10.9.3 on page 121](#).

10.3 BLUETOOTH CLOCK

Every Bluetooth unit has an internal system clock which determines the timing and hopping of the transceiver. The Bluetooth clock is derived from a free running native clock which is never adjusted and is never turned off. For synchronization with other units, only offsets are used that, added to the native clock, provide temporary Bluetooth clocks which are mutually synchronized. It should be noted that the Bluetooth clock has no relation to the time of day; it can therefore be initialized at any value. The Bluetooth clock provides the heart beat of the Bluetooth transceiver. Its resolution is at least half the TX or RX slot length, or 312.5 μ s. The clock has a cycle of about a day. If the clock is implemented with a counter, a 28-bit counter is required that wraps around at $2^{28}-1$. The LSB ticks in units of 312.5 μ s, giving a clock rate of 3.2 kHz.

The timing and the frequency hopping on the channel of a piconet is determined by the Bluetooth clock of the master. When the piconet is established, the master clock is communicated to the slaves. Each slave adds an offset to its native clock to be synchronized to the master clock. Since the clocks are free-running, the offsets have to be updated regularly.

The clock determines critical periods and triggers the events in the Bluetooth receiver. Four periods are important in the Bluetooth system: 312.5 μ s, 625 μ s, 1.25 ms, and 1.28 s; these periods correspond to the timer bits CLK₀, CLK₁, CLK₂, and CLK₁₂, respectively, see [Figure 10.1 on page 93](#). Master-to-slave transmission starts at the even-numbered slots when CLK₀ and CLK₁ are both zero.

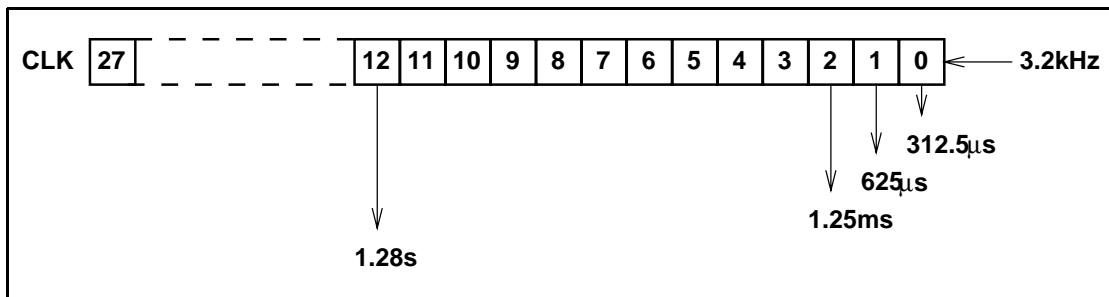


Figure 10.1: Bluetooth clock.

In the different modes and states a Bluetooth unit can reside in, the clock has different appearances:

- CLKN native clock
- CLKE estimated clock
- CLK master clock

CLKN is the free-running native clock and is the reference to all other clock appearances. In states with high activity, the native clock is driven by the reference crystal oscillator with worst case accuracy of +/-20ppm. In the low power states, like **STANDBY**, **HOLD**, **PARK** and **SNIFF**, the native clock may be driven by a low power oscillator (LPO) with relaxed accuracy (+/-250ppm).

CLKE and CLK are derived from the reference CLKN by adding an offset. CLKE is a clock estimate a paging unit makes of the native clock of the recipient; i.e. an offset is added to the CLKN of the pager to approximate the CLKN of the recipient, see [Figure 10.2 on page 94](#). By using the CLKN of the recipient, the pager speeds up the connection establishment.

CLK is the master clock of the piconet. It is used for all timing and scheduling activities in the piconet. All Bluetooth devices use the CLK to schedule their transmission and reception. The CLK is derived from the native clock CLKN by adding an offset, see [Figure 10.3 on page 94](#). The offset is zero for the master since CLK is identical to its own native clock CLKN. Each slave adds an appropriate offset to its CLKN such that the CLK corresponds to the CLKN of the master. Although all CLKNs in the Bluetooth devices run at the same nominal rate, mutual drift causes inaccuracies in CLK. Therefore, the offsets in the slaves must be regularly updated such that CLK is approximately CLKN of the master.

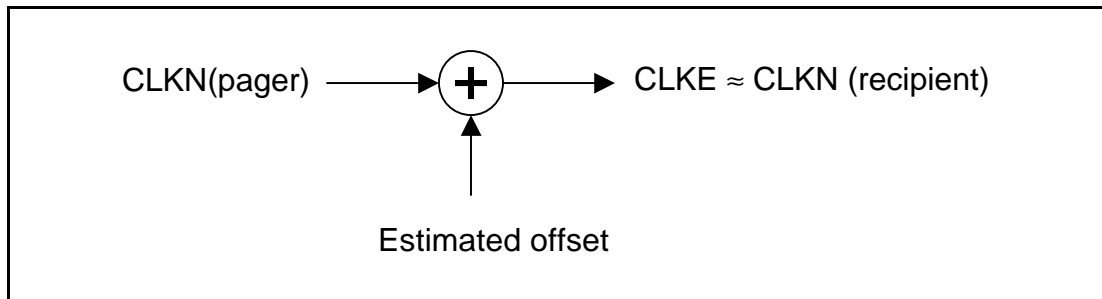


Figure 10.2: Derivation of CLKE

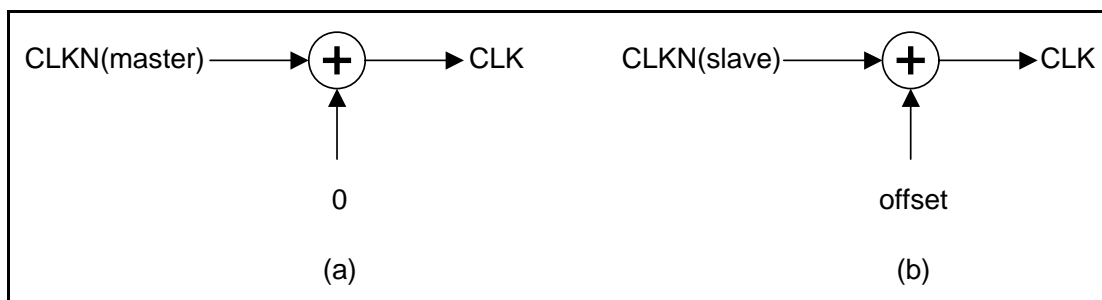


Figure 10.3: Derivation of CLK in master (a) and in slave (b).

10.4 OVERVIEW OF STATES

Figure 10.4 on page 95 shows a state diagram illustrating the different states used in the Bluetooth link controller. There are two major states: **STANDBY** and **CONNECTION**; in addition, there are seven substates, **page**, **page scan**, **inquiry**, **inquiry scan**, **master response**, **slave response**, and **inquiry response**. The substates are interim states that are used to add new slaves to a piconet. To move from one state to the other, either commands from the Bluetooth link manager are used, or internal signals in the link controller are used (such as the trigger signal from the correlator and the timeout signals).

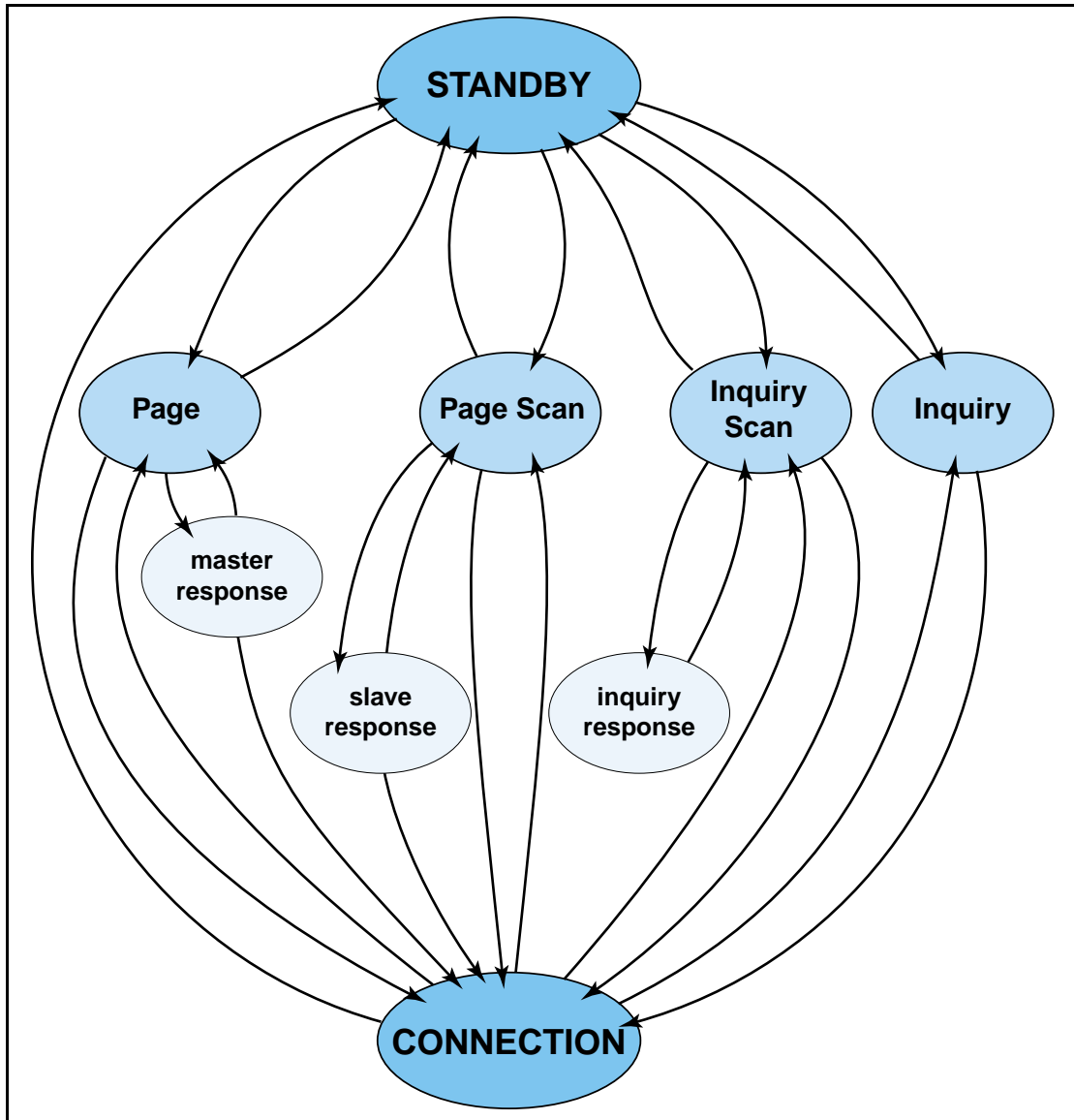


Figure 10.4: State diagram of Bluetooth link controller.

10.5 STANDBY STATE

The **STANDBY** state is the default state in the Bluetooth unit. In this state, the Bluetooth unit is in a low-power mode. Only the native clock is running at the accuracy of the LPO (or better).

The controller may leave the **STANDBY** state to scan for page or inquiry messages, or to page or inquiry itself. When responding to a page message, the unit will not return to the **STANDBY** state but enter the **CONNECTION** state as a slave. When carrying out a successful page attempt, the unit will enter the **CONNECTION** state as a master. The intervals with which scan activities can be carried out are discussed in [Section 10.6.2 on page 96](#) and [Section 10.7.2 on page 106](#).



10.6 ACCESS PROCEDURES

10.6.1 General

In order to establish new connections the procedures inquiry and paging are used. The inquiry procedure enables a unit to discover which units are in range, and what their device addresses and clocks are. With the paging procedure, an actual connection can be established. Only the Bluetooth device address is required to set up a connection. Knowledge about the clock will accelerate the setup procedure. A unit that establishes a connection will carry out a page procedure and will automatically be the master of the connection.

In the paging and inquiry procedures, the device access code (DAC) and the inquiry access code (IAC) are used, respectively. A unit in the **page scan** or **inquiry scan** substate correlates against these respective access codes with a matching correlator.

For the paging process, several paging schemes can be applied. There is one mandatory paging scheme which has to be supported by each Bluetooth device. This mandatory scheme is used when units meet for the first time, and in case the paging process directly follows the inquiry process. Two units, once connected using a mandatory paging/scanning scheme, may agree on an optional paging/scanning scheme. Optional paging schemes are discussed in “[Appendix VII](#)” on page 1025. In the current chapter, only the mandatory paging scheme is considered.

10.6.2 Page scan

In the **page scan** substate, a unit listens for its own device access code for the duration of the scan window $T_{w \text{ page scan}}$. During the scan window, the unit listens at a single hop frequency, its correlator matched to its device access code. The scan window shall be long enough to completely scan 16 page frequencies.

When a unit enters the **page scan** substate, it selects the scan frequency according to the page hopping sequence corresponding to this unit, see [Section 11.3.1 on page 134](#). This is a 32-hop sequence (or a 16-hop sequence in case of a reduced-hop system) in which each hop frequency is unique. The page hopping sequence is determined by the unit’s Bluetooth device address (BD_ADDR). The phase in the sequence is determined by $CLKN_{16-12}$ of the unit’s native clock ($CLKN_{15-12}$ in case of a reduced-hop system); that is, every 1.28s a different frequency is selected.

If the correlator exceeds the trigger threshold during the **page scan**, the unit will enter the **slave response** substate, which is described in [Section 10.6.4.1 on page 102](#).



The **page scan** substate can be entered from the **STANDBY** state or the **CONNECTION** state. In the **STANDBY** state, no connection has been established and the unit can use all the capacity to carry out the **page scan**. Before entering the **page scan** substate from the **CONNECTION** state, the unit preferably reserves as much capacity for scanning. If desired, the unit may place ACL connections in the HOLD mode or even use the PARK mode, see [Section 10.8.3 on page 112](#) and [Section 10.8.4 on page 112](#). SCO connections are preferably not interrupted by the **page scan**. In this case, the **page scan** may be interrupted by the reserved SCO slots which have higher priority than the **page scan**. SCO packets should be used requiring the least amount of capacity (**HV3** packets). The scan window shall be increased to minimize the setup delay. If one SCO link is present using **HV3** packets and $T_{SCO}=6$ slots, a total scan window $T_{w \text{ page scan}}$ of at least 36 slots (22.5ms) is recommended; if two SCO links are present using **HV3** packets and $T_{SCO}=6$ slots, a total scan window of at least 54 slots (33.75ms) is recommended.

The scan interval $T_{\text{page scan}}$ is defined as the interval between the beginnings of two consecutive page scans. A distinction is made between the case where the scan interval is equal to the scan window $T_{w \text{ page scan}}$ (continuous scan), the scan interval is maximal 1.28s, or the scan interval is maximal 2.56s. These three cases determine the behavior of the paging unit; that is, whether the paging unit shall use R0, R1 or R2, see also [Section 10.6.3 on page 98](#). [Table 10.1](#) illustrates the relationship between $T_{\text{page scan}}$ and modes R0, R1 and R2. Although scanning in the R0 mode is continuous, the scanning may be interrupted by for example reserved SCO slots. The scan interval information is included in the SR field in the FHS packet.

During page scan the Bluetooth unit may choose to use an optional scanning scheme. (An exception is the page scan after returning an inquiry response message. See [Section 10.7.4 on page 108](#) for details.)

SR mode	$T_{\text{page scan}}$	N_{page}
R0	continuous	≥ 1
R1	$\leq 1.28\text{s}$	≥ 128
R2	$\leq 2.56\text{s}$	≥ 256
Reserved	-	-

Table 10.1: Relationship between scan interval, train repetition, and paging modes R0, R1 and R2.

10.6.3 Page

The **page** substate is used by the master (source) to activate and connect to a slave (destination) which periodically wakes up in the **page scan** substate. The master tries to capture the slave by repeatedly transmitting the slave's device access code (DAC) in different hop channels. Since the Bluetooth clocks of the master and the slave are not synchronized, the master does not know exactly when the slave wakes up and on which hop frequency. Therefore, it transmits a train of identical DACs at different hop frequencies, and listens in between the transmit intervals until it receives a response from the slave.

The page procedure in the master consists of a number of steps. First, the slave's device address is used to determine the page hopping sequence, see [Section 11.3.2 on page 134](#). This is the sequence the master will use to reach the slave. For the phase in the sequence, the master uses an estimate of the slave's clock. This estimate can for example be derived from timing information that was exchanged during the last encounter with this particular device (which could have acted as a master at that time), or from an inquiry procedure. With this estimate CLKE of the slave's Bluetooth clock, the master can predict on which hop channel the slave will start page scan.

The estimate of the Bluetooth clock in the slave can be completely wrong. Although the master and the slave use the same hopping sequence, they use different phases in the sequence and will never meet each other. To compensate for the clock drifts, the master will send its page message during a short time interval on a number of wake-up frequencies. It will in fact transmit also on hop frequencies just before and after the current, predicted hop frequency. During each TX slot, the master sequentially transmits on two different hop frequencies. Since the page message is the ID packet which is only 68 bits in length, there is ample of time (224.5 μ s minimal) to switch the synthesizer. In the following RX slot, the receiver will listen sequentially to two corresponding RX hops for ID packet. The RX hops are selected according to the `page_response` hopping sequence. The `page_response` hopping sequence is strictly related to the page hopping sequence; that is: for each page hop there is a corresponding `page_response` hop. The RX/TX timing in the **page** substate has been described in [Section 9 on page 84](#), see also [Figure 9.4 on page 88](#). In the next TX slot, it will transmit on two hop frequencies different from the former ones. The synthesizer hop rate is increased to 3200 hops/s.

A distinction must be made between the 79-hop systems and the 23-hop systems. First the 79-hop systems are considered. With the increased hopping rate as described above, the transmitter can cover 16 different hop frequencies in 16 slots or 10 ms. The page hopping sequence is divided over two paging trains **A** and **B** of 16 frequencies. Train **A** includes the 16 hop frequencies surrounding the current, predicted hop frequency $f(k)$, where k is determined by the clock estimate $CLKE_{16-12}$. So the first train consists of hops

$f(k-8), f(k-7), \dots, f(k), \dots, f(k+7)$



When the difference between the Bluetooth clocks of the master and the slave is between -8×1.28 s and $+7 \times 1.28$ s, one of the frequencies used by the master will be the hop frequency the slave will listen to. However, since the master does not know when the slave will enter the **page scan** substate, he has to repeat this train **A** N_{page} times or until a response is obtained. If the slave scan interval corresponds to R1, the repetition number is at least 128; if the slave scan interval corresponds to R2, the repetition number is at least 256. Note that CLKE_{16-12} changes every 1.28 s; therefore, every 1.28 s, the trains will include different frequencies of the page hopping set.

When the difference between the Bluetooth clocks of the master and the slave is less than -8×1.28 s or larger than $+7 \times 1.28$ s, more distant hops must be probed. Since in total, there are only 32 dedicated wake-up hops, the more distant hops are the remaining hops not being probed yet. The remaining 16 hops are used to form the new 10 ms train **B**. The second train consists of hops

$f(k-16), f(k-15), \dots, f(k-9), f(k+8), \dots, f(k+15)$

Train **B** is repeated for N_{page} times. If still no response is obtained, the first train **A** is tried again N_{page} times. Alternate use of train A and train B is continued until a response is received or the timeout *pageTO* is exceeded. If during one of the listening occasions, a response is returned by the slave, the master unit enters the **master response** substate.

The description for paging and **page scan** procedures given here has been tailored towards the 79-hop systems used in the US and Europe. For the 23-hop systems as used in France, the procedure is slightly different. In the 23-hop case, the length of the page hopping sequence is reduced to 16. As a consequence, there is only a single train (train **A**) including all the page hopping frequencies. The phase to the page hopping sequence is not CLKE_{16-12} but CLKE_{15-12} . An estimate of the slave's clock does not have to be made.

The **page** substate can be entered from the **STANDBY** state or the **CONNECTION** state. In the **STANDBY** state, no connection has been established and the unit can use all the capacity to carry out the page. Before entering the page substate from the **CONNECTION** state, the unit shall free as much capacity as possible for scanning. To ensure this, it is recommended that the ACL connections are put on hold or park. However, the SCO connections shall not be disturbed by the page. This means that the page will be interrupted by the reserved SCO slots which have higher priority than the page. In order to obtain as much capacity for paging, it is recommended to use the SCO packets which use the least amount of capacity (**HV3** packets). If SCO links are present, the repetition number N_{page} of a single train shall be increased, see [Table 10.2](#). Here it has been assumed that the **HV3** packet are used with an interval $T_{\text{SCO}}=6$ slots, which would correspond to a 64 kb/s voice link.

SR mode	no SCO link	one SCO link (HV3)	two SCO links (HV3)
R0	$N_{\text{page}} \geq 1$	$N_{\text{page}} \geq 2$	$N_{\text{page}} \geq 3$
R1	$N_{\text{page}} \geq 128$	$N_{\text{page}} \geq 256$	$N_{\text{page}} \geq 384$
R2	$N_{\text{page}} \geq 256$	$N_{\text{page}} \geq 512$	$N_{\text{page}} \geq 768$

Table 10.2: Relationship between train repetition, and paging modes R0, R1 and R2 when SCO links are present.

The construction of the page train is independent on the presence of SCO links; that is, SCO packets are sent on the reserved slots but do not affect the hop frequencies used in the unreserved slots, see Figure 10.5 on page 100.

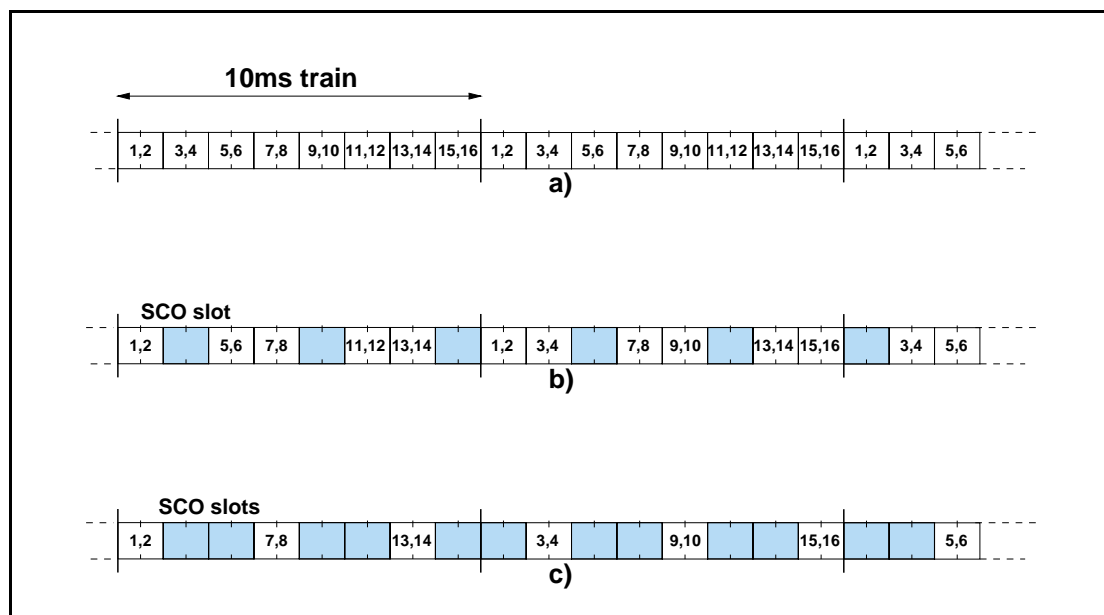


Figure 10.5: Conventional page (a), page while one SCO link present (b), page while two SCO links present (c).

For the descriptions of optional paging schemes see “Appendix VII” on page 1025.

10.6.4 Page response procedures

When a page message is successfully received by the slave, there is a coarse FH synchronization between the master and the slave. Both the master and the slave enter a response routine to exchange vital information to continue the connection setup. Important for the piconet connection is that both Bluetooth units use the same channel access code, use the same channel hopping sequence, and that their clocks are synchronized. These parameters are derived from the master unit. The unit that initializes the connection (starts paging) is defined as the master unit (which is thus only valid during the time the piconet exists). The channel access code and channel hopping sequence are



derived from the Bluetooth device address (BD_ADDR) of the master. The timing is determined by the master clock. An offset is added to the slave's native clock to temporarily synchronize the slave clock to the master clock. At start-up, the master parameters have to be transmitted from the master to the slave. The messaging between the master and the slave at start-up will be considered in this section.

The initial messaging between master and slave is shown in [Table 10.3 on page 101](#) and in [Figure 10.6 on page 102](#) and [Figure 10.7 on page 102](#). In those two figures frequencies $f(k)$, $f(k+1)$, etc. are the frequencies of the page hopping sequence determined by the slave's BD_ADDR. The frequencies $f'(k)$, $f'(k+1)$, etc. are the corresponding page_response frequencies (slave-to-master). The frequencies $g(m)$ belong to the channel hopping sequence.

Step	Message	Direction	Hopping Sequence	Access Code and Clock
1	slave ID	master to slave	page	slave
2	slave ID	slave to master	page response	slave
3	FHS	master to slave	page	slave
4	slave ID	slave to master	page response	slave
5	1st packet master	master to slave	channel	master
6	1st packet slave	slave to master	channel	master

Table 10.3: Initial messaging during start-up.

In step 1 (see [Table 10.3 on page 101](#)), the master unit is in **page** substate and the slave unit in the **page scan** substate. Assume in this step that the page message (= slave's device access code) sent by the master reaches the slave. On recognizing its device access code, the slave enters the **slave response** in step 2. The master waits for a reply from the slave and when this arrives in step 2, it will enter the **master response** in step 3. Note that during the initial message exchange, all parameters are derived from the slave's BD_ADDR, and that only the page hopping and page_response hopping sequences are used (which are also derived from the slave's BD_ADDR). Note that when the master and slave enter the response states, their clock input to the page and page_response hop selection is frozen as is described in [Section 11.3.3 on page 135](#).

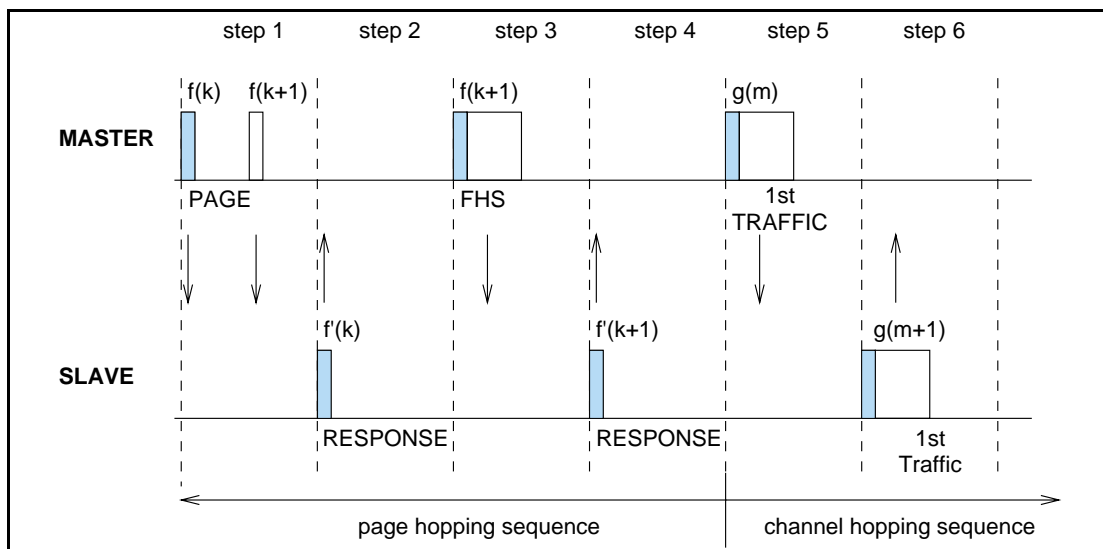


Figure 10.6: Messaging at initial connection when slave responds to first page message.

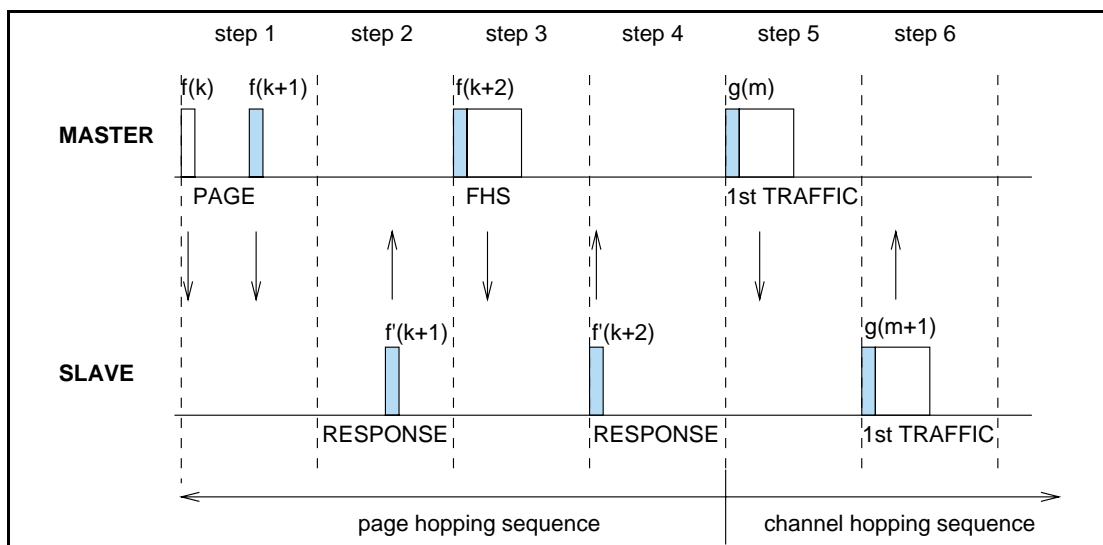


Figure 10.7: Messaging at initial connection when slave responds to second page message.

10.6.4.1 Slave response

After having received its own device access code in step 1, the slave unit transmits a response message in step 2. This response message again only consists of the slave’s device access code. The slave will transmit this response 625 μ s after the beginning of the received page message (slave ID packet) and at the response hop frequency that corresponds to the hop frequency in which the page message was received. The slave transmission is therefore time aligned to the master transmission. During initial messaging, the slave still uses the page response hopping sequence to return information to the master. The clock input $CLKN_{16-12}$ is frozen at the value it had at the time the page message was received.



After having sent the response message, the slave's receiver is activated (312.5 μ s after the start of the response message) and awaits the arrival of a **FHS** packet. Note that a **FHS** packet can already arrive 312.5 μ s after the arrival of the page message as shown in [Figure 10.7 on page 102](#), and not after 625 μ s as is usually the case in the RX/TX timing. More details about the timing can be found in [Section 9.6 on page 89](#).

If the setup fails before the **CONNECTION** state has been reached, the following procedure is carried out. The slave will keep listening as long as no **FHS** packet is received until *pagerespTO* is exceeded. Every 1.25 ms, however, it will select the next master-to-slave hop frequency according to the page hop sequence. If nothing is received after *pagerespTO*, the slave returns back to the **page scan** substate for one scan period. Length of the scan period depends on the SCO slots present. If no page message is received during this additional scan period, the slave will resume scanning at its regular scan interval and return to the state it was in prior to the first page scan state.

If a **FHS** packet is received by the slave in the **slave response** substate, the slave returns a response (slave's device access code only) in step 4 to acknowledge the reception of the **FHS** packet (still using the page response hopping sequence). The transmission of this response packet is based on the reception of the **FHS** packet. Then the slave changes to the channel (master's) access code and clock as received from the **FHS** packet. Only the 26 MSBs of the master clock are transferred: the timing is assumed such that CLK_1 and CLK_0 are both zero at the time the **FHS** packet was received as the master transmits in even slots only. From the master clock in the **FHS** packet, the offset between the master's clock and the slave's clock is determined and reported to the slave's link manager.

Finally, the slave enters the **CONNECTION** state in step 5. From then on, the slave will use the master's clock and the master *BD_ADDR* to determine the channel hopping sequence and the channel access code. The connection mode starts with a POLL packet transmitted by the master. The slave responds with any type of packet. If the POLL packet is not received by the slave, or the response packet is not received by the master, within *newconnectionTO* number of slots after FHS packet acknowledgement, the master and the slave will return to page and page scan substates, respectively. See [Section 10.8 on page 110](#)

10.6.4.2 Master response

When the master has received a response message from the slave in step 2, it will enter the **master response** routine. It freezes the current clock input to the page hop selection scheme. Then the master will transmit a **FHS** packet in step 3 containing the master's real-time Bluetooth clock, the master's 48-bit BD_ADDR address, the BCH parity bits, and the class of device. The **FHS** packet contains all information to construct the channel access code without requiring a mathematical derivation from the master device address. The **FHS** packet is transmitted at the beginning of the master-to-slave slot following the slot in which the slave has responded. So the TX timing of the **FHS** is not based on the reception of the response packet from the slave. The **FHS** packet may therefore be sent 312.5 μs after the reception of the response packet like shown in [Figure 10.7 on page 102](#) and not 625 μs after the received packet as is usual in the RX/TX timing, see also [Section 9.6 on page 89](#).

After the master has sent its **FHS** packet, it waits for a second response from the slave in step 4 which acknowledges the reception of the **FHS** packet. Again this is only the slave's device access code. If no response is received, the master retransmits the **FHS** packet, but with an updated clock and still using the slave's parameters. It will retransmit (the clock is updated every retransmission) until a second slave response is received, or the timeout of *pagerespTO* is exceeded. In the latter case, the master turns back to the **page** substate and sends an error message to the link manager. During the retransmissions of the **FHS** packet, the master keeps using the page hopping sequence.

If the slave's response is indeed received, the master changes to the master parameters, so the channel access code and the master clock. The lower clock bits CLK₀ and CLK₁ are zero at the start of the **FHS** packet transmission and are not included in the **FHS** packet. Finally, the master enters the **CONNECTION** state in step 5. The master BD_ADDR is used to change to a new hopping sequence, the *channel hopping sequence*. The channel hopping sequence uses all 79 hop channels in a (pseudo) random fashion, see also [Section 11.3.6 on page 137](#). The master can now send its first traffic packet in a hop determined with the new (master) parameters. This first packet will be a POLL packet. See [Section 10.8 on page 110](#).

The master can now send its first traffic packet in a hop determined with the new (master) parameters. The first packet in this state is a POLL packet sent by the master. This packet will be sent within *newconnectionTO* number of slots after reception of the FHS packet acknowledgement. The slave will respond with any type of packet. If the POLL packet is not received by the slave or the POLL packet response is not received by the master within *newconnectionTO* number of slots, the master and the slave will return to page and page scan substates, respectively.

10.7 INQUIRY PROCEDURES

10.7.1 General

In the Bluetooth system, an inquiry procedure is defined which is used in applications where the destination's device address is unknown to the source. One can think of public facilities like printers or facsimile machines, or access points to a LAN. Alternatively, the inquiry procedure can be used to discover which other Bluetooth units are within range. During an **inquiry** substate, the discovering unit collects the Bluetooth device addresses and clocks of all units that respond to the inquiry message. It can then, if desired, make a connection to any one of them by means of the previously described page procedure.

The inquiry message broadcast by the source does not contain any information about the source. However, it may indicate which class of devices should respond. There is one general inquiry access code (GIAC) to inquire for any Bluetooth device, and a number of dedicated inquiry access codes (DIAC) that only inquire for a certain type of devices. The inquiry access codes are derived from reserved Bluetooth device addresses and are further described in [Section 4.2.1 on page 48](#).

A unit that wants to discover other Bluetooth units enters an **inquiry** substate. In this substate, it continuously transmits the inquiry message (which is the ID packet, see [Section 4.4.1.1 on page 55](#)) at different hop frequencies. The **inquiry** hop sequence is always derived from the LAP of the GIAC. Thus, even when DIACs are used, the applied hopping sequence is generated from the GIAC LAP. A unit that allows itself to be discovered, regularly enters the **inquiry scan** substate to respond to inquiry messages. The following sections describe the message exchange and contention resolution during inquiry response. The inquiry response is optional: a unit is not forced to respond to an inquiry message.

10.7.2 Inquiry scan

The **inquiry scan** substate is very similar to the **page scan** substate. However, instead of scanning for the unit's device access code, the receiver scans for the inquiry access code long enough to completely scan for 16 inquiry frequencies. The length of this scan period is denoted $T_{w_inquiry_scan}$. The scan is performed at a single hop frequency. As in the page procedure, the inquiry procedure uses 32 dedicated inquiry hop frequencies according to the *inquiry hopping sequence*. These frequencies are determined by the general inquiry address. The phase is determined by the native clock of the unit carrying out the **inquiry scan**; the phase changes every 1.28s.

Instead or in addition to the general inquiry access code, the unit may scan for one or more dedicated inquiry access codes. However, the scanning will follow the inquiry scan hopping sequence which is determined by the general inquiry address. If an inquiry message is recognized during an inquiry wake-up period, the Bluetooth unit either performs a backoff in **CONNECTION** or **STANDBY** state before reentering the inquiry scan substate or enters the **inquiry response** substate if a random backoff was performed before entering the inquiry scan substate..

The **inquiry scan** substate can be entered from the **STANDBY** state or the **CONNECTION** state. In the **STANDBY** state, no connection has been established and the unit can use all the capacity to carry out the **inquiry scan**. Before entering the **inquiry scan** substate from the **CONNECTION** state, the unit preferably reserves as much capacity as possible for scanning. If desired, the unit may place ACL connections in the HOLD mode or even use the PARK mode, see [Section 10.8.3 on page 112](#). SCO connections are preferably not interrupted by the **inquiry scan**. In this case, the **inquiry scan** may be interrupted by the reserved SCO slots which have higher priority than the **inquiry scan**. SCO packets should be used requiring the least amount of capacity (**HV3** packets). The scan window, $T_{w_inquiry_scan}$, shall be increased to increase the probability to respond to an inquiry message. If one SCO link is present using HV3 packets and $T_{SCO}=6$ slots, a total scan window of at least 36 slots (22.5ms) is recommended; if two SCO links are present using HV3 packets and $T_{SCO}=6$ slots, a total scan window of at least 54 slots (33.75ms) is recommended.

The scan interval $T_{inquiry_scan}$ is defined as the interval between two consecutive inquiry scans. The **inquiry scan** interval shall be at most 2.56 s.

10.7.3 Inquiry

The **inquiry** substate is used by the unit that wants to discover new devices. This substate is very similar to the **page** substate, the same TX/RX timing is used as used for paging, see [Section 9.6 on page 89](#) and [Figure 9.4 on page 88](#). The TX and RX frequencies follow the inquiry hopping sequence and the inquiry response hopping sequence, and are determined by the general inquiry access code and the native clock of the discovering device. In between inquiry transmissions, the Bluetooth receiver scans for inquiry response messages. When found, the entire response packet (which is in fact a **FHS** packet) is read, after which the unit continues with the inquiry transmissions. So the Bluetooth unit in an **inquiry** substate does not acknowledge the inquiry response messages. It keeps probing at different hop channels and in between listens for response packets. Like in the **page** substate, two 10 ms trains **A** and **B** are defined, splitting the 32 frequencies of the inquiry hopping sequence into two 16-hop parts. A single train must be repeated for at least $N_{\text{inquiry}}=256$ times before a new train is used. In order to collect all responses in an error-free environment, at least three train switches must have taken place. As a result, the **inquiry** substate may have to last for 10.24 s unless the inquirer collects enough responses and determines to abort the inquiry substate earlier. If desired, the inquirer can also prolong the inquiry substate to increase the probability of receiving all responses in an error-prone environment. If an inquiry procedure is automatically initiated periodically (say a 10 s period every minute), then the interval between two inquiry instances must be determined randomly. This is done to avoid two Bluetooth units to synchronize their inquiry procedures.

The **inquiry** substate is continued until stopped by the Bluetooth link manager (when it decides that it has sufficient number of responses), or when a timeout has been reached (*inquiryTO*).

The **inquiry** substate can be entered from the **STANDBY** state or the **CONNECTION** state. In the **STANDBY** state, no connection has been established and the unit can use all the capacity to carry out the inquiry. Before entering the inquiry substate from the **CONNECTION** state, the unit shall free as much capacity as possible for scanning. To ensure this, it is recommended that the ACL connections are put on hold or park. However, the SCO connections shall not be disturbed by the inquiry. This means that the inquiry will be interrupted by the reserved SCO slots which have higher priority than the inquiry. In order to obtain as much capacity for inquiry, it is recommended to use the SCO packets which use the least amount of capacity (**HV3** packets). If SCO links are present, the repetition number N_{inquiry} shall be increased, see [Table 10.4 on page 108](#).

Here it has been assumed that the **HV3** packet are used with an interval $T_{\text{SCO}}=6$ slots, which would correspond to a 64 kb/s voice link.

	no SCO link	one SCO link (HV3)	two SCO links (HV3)
N_{inquiry}	≥ 256	≥ 512	≥ 768

Table 10.4: Increase of train repetition when SCO links are present.

10.7.4 Inquiry response

For the inquiry operation, there is only a slave response, no master response. The master listens between inquiry messages for responses, but after reading a response, it continues to transmit inquiry messages. The slave response routine for inquiries differs completely from the slave response routine applied for pages. When the inquiry message is received in the **inquiry scan** substate, a response message containing the recipient's address must be returned. This response message is a conventional **FHS** packet carrying the unit's parameters. However, a contention problem may arise when several Bluetooth units are in close proximity to the inquiring unit and all respond to an inquiry message at the same time. First of all, every Bluetooth unit has a free running clock; therefore, it is highly unlikely that they all use the same phase of the inquiry hopping sequence. However, in order to avoid collisions between units that do wake up in the same inquiry hop channel simultaneously, the following protocol in the slave's **inquiry response** is used. If the slave receives an inquiry message, it generates a random number RAND between 0 and 1023. The slave then returns to the **CONNECTION** or **STANDBY** state for the duration of RAND time slots. Before returning to the **CONNECTION** or **STANDBY** state, the unit may go through the page scan substate; this page scan must use the mandatory page scan scheme. After at least RAND slots, the unit will return to the **inquiry scan** substate. On the first inquiry message received in this substate the slave goes into the **inquiry response** substate and returns an **FHS** response packet to the master 625 μs after the inquiry message was received. If during the scan no trigger occurs within a timeout period of *inqrespTO*, the slave returns to the **STANDBY** or **CONNECTION** state. If the unit does receive an inquiry message and returns an **FHS** packet, it adds an offset of 1 to the phase in the inquiry hop sequence (the phase has a 1.28 s resolution) and enters the **inquiry scan** substate again. If the slave is triggered again, it repeats the procedure using a new RAND. The offset to the clock accumulates each time a **FHS** packet is returned. During a 1.28 s probing window, a slave on average responds 4 times, but on different frequencies and at different times. Possible SCO slots should have priority over response packets; that is, if a response packet overlaps with an SCO slot, it is not sent but the next inquiry message is awaited.

The messaging during the inquiry routines is summarized in [Table 10.5 on page 109](#). In step 1, the master transmits an inquiry message using the inquiry access code and its own clock. The slave responds with the **FHS** packet which contains the slave's device address, native clock and other slave information. This **FHS** packet is returned at a semi-random time. The **FHS** packet is not acknowledged in the inquiry routine, but it is retransmitted at other times and frequencies as long as the master is probing with inquiry messages.



step	message	direction	hopping sequence	access code
1	ID	master to slave	inquiry	inquiry
2	FHS	slave to master	inquiry response	inquiry

Table 10.5: Messaging during inquiry routines.

If the scanning unit uses an optional scanning scheme, after responding to an inquiry with an FHS packet, it will perform page scan using the mandatory page scan scheme for $T_{\text{mandatory pscan}}$ period. Every time an inquiry response is sent the unit will start a timer with a timeout of $T_{\text{mandatory pscan}}$. The timer will be reset at each new inquiry response. Until the timer times out, when the unit performs page scan, it will use the mandatory page scanning scheme in the SR mode it uses for all its page scan intervals. Using the mandatory page scan scheme after the inquiry procedure enables all units to connect even if they do not support an optional paging scheme (yet). In addition to using the mandatory page scan scheme, an optional page scan scheme can be used in parallel for the $T_{\text{mandatory pscan}}$ period.

The $T_{\text{mandatory pscan}}$ period is included in the SP field of the FHS packet returned in the inquiry response routine, see [Section 4.4.1.4 on page 56](#). The value of the period is indicated in the [Table 10.6](#)

SP mode	$T_{\text{mandatory pscan}}$
P0	$\geq 20\text{s}$
P1	$\geq 40\text{s}$
P2	$\geq 60\text{s}$
Reserved	-

Table 10.6: Mandatory scan periods for P0, P1, P2 scan period modes.

10.8 CONNECTION STATE

In the **CONNECTION** state, the connection has been established and packets can be sent back and forth. In both units, the channel (master) access code and the master Bluetooth clock are used. The hopping scheme uses the *channel hopping sequence*. The master starts its transmission in even slots ($CLK_{1-0}=00$), the slave starts its transmission in odd slots ($CLK_{1-0}=10$)

The **CONNECTION** state starts with a POLL packet sent by the master to verify the switch to the master's timing and channel frequency hopping. The slave can respond with any type of packet. If the slave does not receive the POLL packet or the master does not receive the response packet for *newconnectionTO* number of slots, both devices will return to **page/page scan** substates.

The first information packets in the **CONNECTION** state contain control messages that characterize the link and give more details regarding the Bluetooth units. These messages are exchanged between the link managers of the units. For example, it defines the SCO links and the sniff parameters. Then the transfer of user information can start by alternately transmitting and receiving packets.

The **CONNECTION** state is left through a **detach** or **reset** command. The **detach** command is used if the link has been disconnected in the normal way. All configuration data in the Bluetooth link controller is still valid. The **reset** command is a hard reset of all controller processes. After a reset, the controller has to be reconfigured.

The Bluetooth units can be in several modes of operation during the **CONNECTION** state: active mode, sniff mode, hold mode, and park mode. These modes are now described in more detail.

10.8.1 Active mode

In the active mode, the Bluetooth unit actively participates on the channel. The master schedules the transmission based on traffic demands to and from the different slaves. In addition, it supports regular transmissions to keep slaves synchronized to the channel. Active slaves listen in the master-to-slave slots for packets. If an active slave is not addressed, it may sleep until the next new master transmission. From the type indication in the packet, the number of slots the master has reserved for its transmission can be derived; during this time, the non-addressed slaves do not have to listen on the master-to-slave slots. A periodic master transmission is required to keep the slaves synchronized to the channel. Since the slaves only need the channel access code to synchronize with, any packet type can be used for this purpose.

10.8.2 Sniff mode

In the sniff mode, the duty cycle of the slave's listen activity can be reduced. If a slave participates on an ACL link, it has to listen in every ACL slot to the master traffic. With the sniff mode, the time slots where the master can start transmission to a specific slave is reduced; that is, the master can only start transmission in specified time slots. These so-called sniff slots are spaced regularly with an interval of T_{sniff} .

The slave starts listening at the sniff slots for $N_{\text{sniff attempt}}$ consecutive receive slots unless a packet with matching AM_ADDR is received. After every reception of a packet with matching AM_ADDR, the slave continues listening at the subsequent $N_{\text{sniff timeout}}$ or remaining of the receive slots, whichever is greater. So, for $N_{\text{sniff timeout}} > 0$, the slave continues listening as long as it receives packets with matching AM_ADDR.

Note that Receive slots here are every odd-numbered slots, in which the master may start sending a packet.

Note that $N_{\text{sniff attempt}} = 1$ and $N_{\text{sniff timeout}} = 0$ cause the slave to listen only at the first sniff slot, irrespective of packets received from the master.

Note that $N_{\text{sniff attempt}} = 0$ is not allowed.

To enter the sniff mode, the master or slave shall issue a sniff command via the LM protocol. This message will contain the sniff interval T_{sniff} and an offset D_{sniff} . The timing of the sniff mode is then determined similar as for the SCO links. In addition, an initialization flag indicates whether initialization procedure 1 or 2 is being used. The device uses initialization 1 when the MSB of the current master clock (CLK_{27}) is 0; it uses initialization 2 when the MSB of the current master clock (CLK_{27}) is 1. The slave shall apply the initialization method as indicated by the initialization flag irrespective of its clock bit value CLK_{27} . The master-to-slave sniff slots determined by the master and the slave shall be initialized on the slots for which the clock satisfies the following equation

$$\text{CLK}_{27-1} \bmod T_{\text{sniff}} = D_{\text{sniff}} \quad \text{for initialization 1}$$

$$(\overline{\text{CLK}_{27}}, \text{CLK}_{26-1}) \bmod T_{\text{sniff}} = D_{\text{sniff}} \quad \text{for initialization 2}$$

The slave-to-master sniff slot determined by the master and the slave shall be initialized on the slots after the master-to-slave sniff slot defined above. After initialization, the clock value $\text{CLK}(k+1)$ for the next master-to-slave SNIFF slot is found by adding the fixed interval T_{sniff} to the clock value of the current master-to-slave sniff slot:

$$\text{CLK}(k+1) = \text{CLK}(k) + T_{\text{sniff}}$$



10.8.3 Hold mode

During the **CONNECTION** state, the ACL link to a slave can be put in a **hold** mode. This means that the slave temporarily does not support ACL packets on the channel any more (note: possible SCO links will still be supported). With the **hold** mode, capacity can be made free to do other things like scanning, paging, inquiring, or attending another piconet. The unit in **hold** mode can also enter a low-power sleep mode. During the **hold** mode, the slave unit keeps its active member address (AM_ADDR).

Prior to entering the hold mode, master and slave agree on the time duration the slave remains in the hold mode. A timer is initialized with the *holdTO* value. When the timer is expired, the slave will wake up, synchronize to the traffic on the channel and will wait for further master instructions.

10.8.4 Park mode

When a slave does not need to participate on the piconet channel, but still wants to remain synchronized to the channel, it can enter the park mode which is a low-power mode with very little activity in the slave. In the park mode, the slave gives up its active member address AM_ADDR. Instead, it receives two new addresses to be used in the park mode

- PM_ADDR: 8-bit Parked Member Address
- AR_ADDR: 8-bit Access Request Address

The PM_ADDR distinguishes a parked slave from the other parked slaves. This address is used in the master-initiated unpark procedure. In addition to the PM_ADDR, a parked slave can also be unparked by its 48-bit BD_ADDR. The all-zero PM_ADDR is a reserved address: if a parked unit has the all-zero PM_ADDR it can only be unparked by the BD_ADDR. In that case, the PM_ADDR has no meaning. The AR_ADDR is used by the slave in the slave-initiated unpark procedure. All messages sent to the parked slaves have to be carried by broadcast packets (the all-zero AM_ADDR) because of the missing AM_ADDR.

The parked slave wakes up at regular intervals to listen to the channel in order to re-synchronize and to check for broadcast messages. To support the synchronization and channel access of the parked slaves, the master supports a beacon channel described in the next section. The beacon structure is communicated to the slave when it is being parked. When the beacon structure changes, the parked slaves are updated through broadcast messages.

In addition for using it for low power consumption, the park mode is used to connect more than seven slaves to a single master. At any one time, only seven slaves can be active. However, by swapping active and parked slaves out respectively in the piconet, the number of slave virtually connected can be much larger (255 if the PM_ADDR is used, and even a larger number if the BD_ADDR is used). There is no limitation to the number of slaves that can be parked.

10.8.4.1 Beacon channel

To support parked slaves, the master establishes a beacon channel when one or more slaves are parked. The beacon channel consists of one beacon slot or a train of equidistant beacon slots which is transmitted periodically with a constant time interval. The beacon channel is illustrated in [Figure 10.8 on page 114](#). A train of N_B ($N_B \geq 1$) beacon slots is defined with an interval of T_B slots. The beacon slots in the train are separated by Δ_B . The start of the first beacon slot is referred to as the **beacon instant** and serves as the beacon timing reference. The beacon parameters N_B and T_B are chosen such that there are sufficient beacon slots for a parked slave to synchronize to during a certain time window in an error-prone environment.

When parked, the slave will receive the beacon parameters through an LMP command. In addition, the timing of the beacon instant is indicated through the offset D_B . Like for the SCO link (see [Section 3.2 on page 45](#)), two initialization procedures 1 or 2 are used. The master uses initialization 1 when the MSB of the current master clock (CLK_{27}) is 0; it uses initialization 2 when the MSB of the current master clock (CLK_{27}) is 1. The chosen initialization procedure is also carried by an initialization flag in the LMP command. The slave shall apply the initiations method as indicated by the initialization flag irrespective of its clock bit CLK_{27} . The master-to-slave slot positioned at the beacon instant shall be initialized on the slots for which the clock satisfies the following equation

$$CLK_{27-1} \bmod T_B = D_B \quad \text{for initialization 1}$$

$$(\overline{CLK_{27}}, CLK_{26-1}) \bmod T_B = D_B \quad \text{for initialization 2}$$

After initialization, the clock value $CLK(k+1)$ for the next beacon instant is found by adding the fixed interval T_B to the clock value of the current beacon instant:

$$CLK(k+1) = CLK(k) + T_B$$

The beacon channel serves four purposes:

1. transmission of master-to-slave packets which the parked slaves can use for re-synchronization
2. carrying messages to the parked slaves to change the beacon parameters
3. carrying general broadcast messages to the parked slaves
4. unparking of one or more parked slaves

Since a slave can synchronize to any packet which is preceded by the proper channel access code, the packets carried on the beacon slots do not have to contain specific broadcast packets for parked slaves to be able to synchronize; any packet can be used. The only requirement placed on the beacon slots is that there is master-to-slave transmission present. If there is no information to be sent, **NULL** packets can be transmitted by the master. If there is indeed

broadcast information to be sent to the parked slaves, the first packet of the broadcast message shall be repeated in every beacon slot of the beacon train. However, synchronous traffic like on the SCO link, may interrupt the beacon transmission.

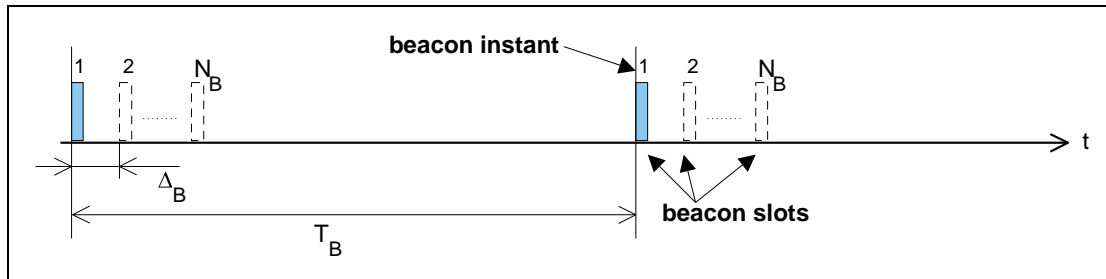


Figure 10.8: General beacon channel format

The master communicates with parked slaves using broadcast messages. Since these messages can be time - critical, an ongoing repetition train of broadcast message can be prematurely aborted by broadcast information destined to parked slaves in beacon slots and in access windows (see Section 10.8.4.2 on page 114).

10.8.4.2 Beacon access window

In addition to the beacon slots, an access window is defined where the parked slaves can send requests to be unparked. To increase reliability, the access window can be repeated M_{access} times ($M_{access} \geq 1$), see Figure 10.9 on page 114. The access window starts a fixed delay D_{access} after the beacon instant. The width of the access window is T_{access} .

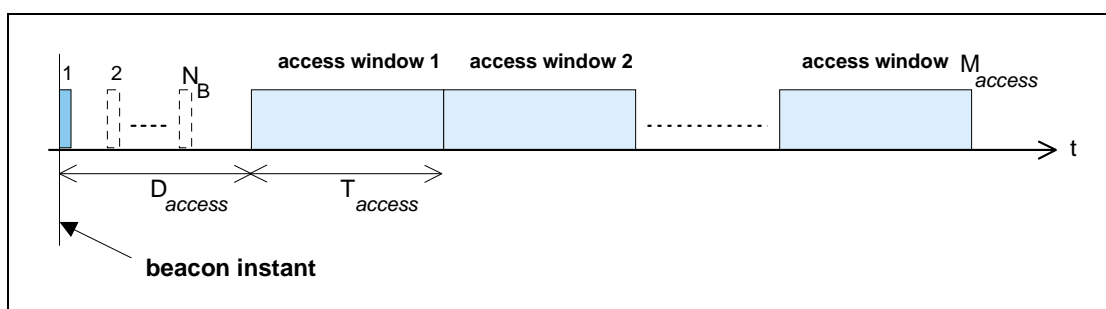


Figure 10.9: Definition of access window

The access window may support different slave access techniques, like polling, random access, or other forms of access. At this stage, only the polling technique has been defined. The format of the polling technique is shown in Figure 10.10 on page 115. The same TDD structure is used as on the piconet channel, i.e. master-to-slave transmission is alternated by slave-to-master transmission. The slave-to-master slot is divided into two half slots of 312.5 μs each. The half slot a parked slave is allowed to respond in corresponds to its access

request address (AR_ADDR), see also [section 10.8.4.6 on page 118](#). For counting the half slots to determine the access request slot, the start of the access window is used, see [Figure 10.10 on page 115](#). The slave is only allowed to send an access request in the proper slave-to-master half slot if in the preceding master-to-slave slot a broadcast packet has been received. In this way, the master polls the parked slaves.

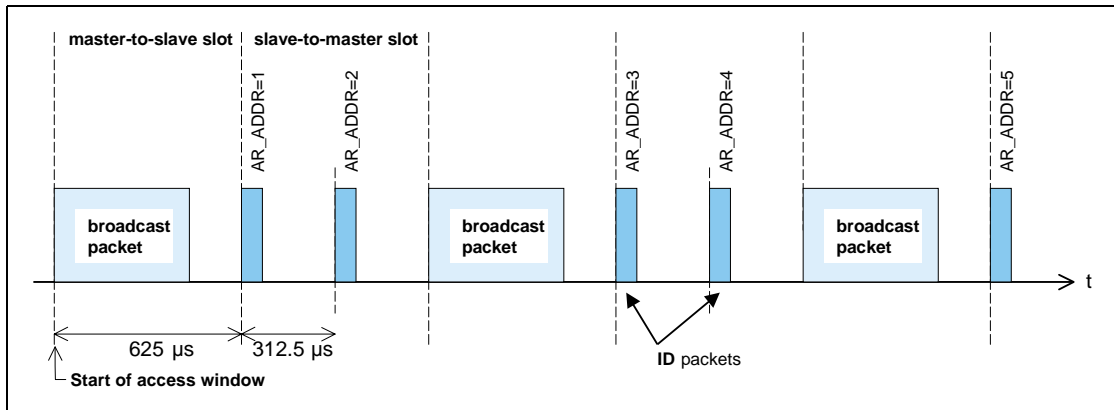


Figure 10.10: Access procedure applying the polling technique.

However, the slots of the access window can also be used for traffic on the piconet if required. For example, if an SCO connection has to be supported, the slots reserved for the SCO link may carry SCO information instead of being used for access requests, i.e. if the master-to-slave slot in the access window contains a packet different from a broadcast packet, the following slave-to-master slot cannot be used for slave access requests. Slots in the access window not affected by traffic can still be used according to the defined access structure; an example is shown in [Figure 10.11 on page 115](#): the access procedure is continued as if no interruption had taken place.

When the slave is parked, it is indicated what type of access scheme will be used. For the polling scheme, the number of slave-to-master access slots N_{acc_slot} is indicated.

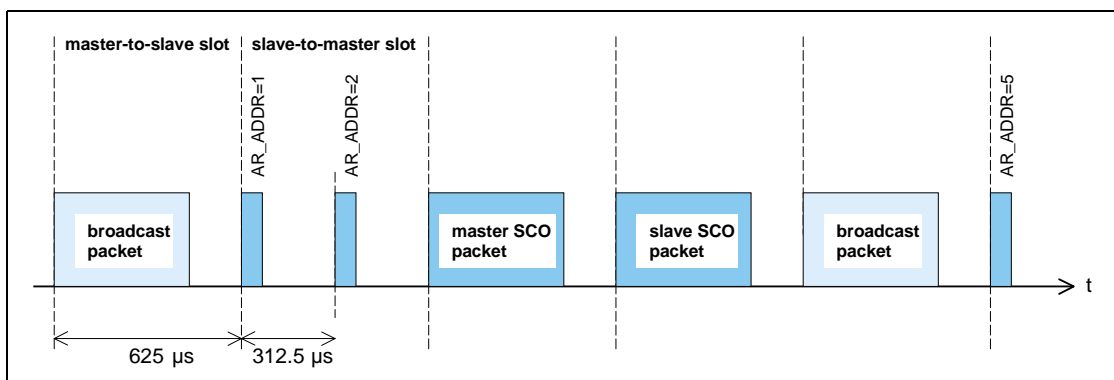


Figure 10.11: Disturbance of access window by SCO traffic



By default, the access window is always present. However, its activation depends on the master sending broadcast messages to the slave at the appropriate slots in the access window. A flag in a broadcast LMP message within the beacon slots may indicate that the access window(s) belonging to this instant will not be activated. This prevents unnecessary scanning of parked slaves that want to request access.

10.8.4.3 Parked slave synchronization

Parked slaves sleep most of the time. However, periodically they wake up to re-synchronize to the channel. Any packet exchanged on the channel can be used for synchronization. Since master transmission is mandatory on the beacon slots, parked slaves will exploit the beacon channel to re-synchronize. A parked slave will wake-up at the beacon instant to read the packet sent on the first beacon slot. If this fails, it will retry on the next beacon slot in the beacon train; in total, there are N_B opportunities per beacon instant to re-synchronize. During the search, the slave may increase its search window, see also [Section 9.4 on page 88](#). The separation between the beacon slots in the beacon train Δ_B is chosen such that consecutive search windows will not overlap.

The parked slave does not have to wake up at every beacon instant. Instead, a sleep interval can be applied which is longer than the beacon interval T_B , see [Figure 10.12 on page 117](#). The slave sleep window must be a multiple N_{B_sleep} of T_B . The precise beacon instant the slave shall wake up on is indicated by the master with D_{B_sleep} which indicates the offset (in multiples of T_B) with respect to the beacon instant ($0 < D_{B_sleep} < N_{B_sleep} - 1$). To initialize the wake-up period, the following equations are used:

$$\text{CLK}_{27-1} \bmod (N_{B_sleep} \cdot T_B) = D_B + D_{B_sleep} \cdot T_B \quad \text{for initialization 1}$$

$$(\overline{\text{CLK}_{27}}, \text{CLK}_{26-1}) \bmod (N_{B_sleep} \cdot T_B) = D_B + D_{B_sleep} \cdot T_B \quad \text{for initialization 2}$$

where initialization 1 is chosen by the master if the MSB in the current master clock is 0 and initialization 2 is chosen if the MSB in the current master clock is 1.

When the master wants to send broadcast messages to the parked slaves, it may use the beacon slots for these broadcast messages. However, if $N_B < N_{BC}$, the slots following the last beacon slot in the beacon train shall be used for the remaining $N_{BC} - N_B$ broadcast packets. If $N_B > N_{BC}$, the broadcast message is repeated on all N_B beacon slots.

A parked slave shall at least read the broadcast messages sent in the beacon slot(s) it wakes up in; the minimum wake-up activity is to read the channel access code for re-synchronization and the packet header to check for broadcast messages.

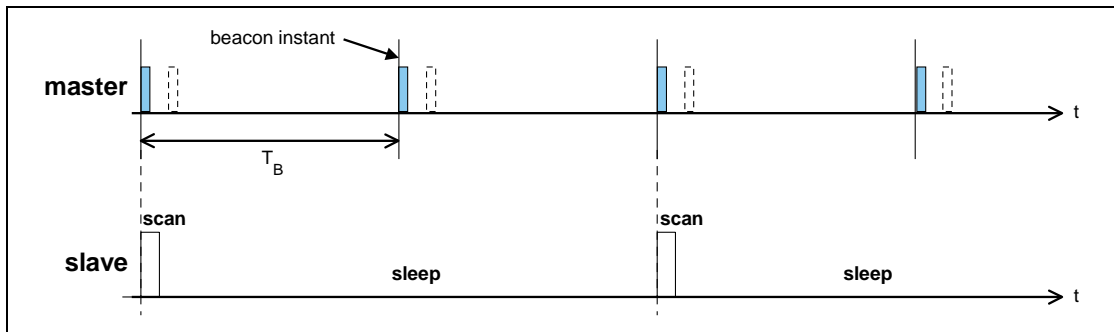


Figure 10.12: Extended sleep interval of parked slaves.

10.8.4.4 Parking

A master can park an active slave through the exchange of one or a few LMP commands. Before put into the park mode, the slave is assigned a PM_ADDR and an AR_ADDR. Every parked slave has a unique PM_ADDR; however, the AR_ADDR is not necessarily unique. Also, the beacon parameters are given by the master when the slave is parked. The slave then gives up its AM_ADDR and enters the park mode. A master can park only a single slave at a time. The park message is carried with a normal data packet and addresses the slave through its AM_ADDR.

10.8.4.5 Master-activated unparking

The master can unpark a parked slave by sending a dedicated LMP unpark command including the parked slave's address. This message is sent in a broadcast packet on the beacon slots. Either the slave's PM_ADDR is used, or its full BD_ADDR is used. The message also includes the active member address AM_ADDR the slave will use after it has re-entered the piconet. The unpark message can include a number of slave addresses so that multiple slaves can be unparked simultaneously. For each slave, a different AM_ADDR is assigned.

After having received the unpark message, the parked slave matching the PM_ADDR or BD_ADDR will leave the park mode and enter the active mode. It will keep listening to the master until it is addressed by the master through its AM_ADDR. The first packet sent by the master must be a POLL packet. The return packet in response to the POLL packet confirms that the slave has been unparked. If no response packets from the slave is received for *newconnectionTO* number of slots after the end of beacon repetition period, the master will unpark the slave again. If the slave does not receive the POLL packet for *newconnectionTO* number of slots after the end of beacon repetition period, it will return to park, with the same beacon parameters. After confirming that the slave is active, the master decides in which mode the slave will continue.



10.8.4.6 Slave-activated unparking

A slave can request access to the channel through the access window defined in [section 10.8.4.2 on page 114](#). As shown in [Figure 10.10 on page 115](#), the access window includes several slave-to-master half slots where the slave can send an access request message. The specific half slot the slave is allowed to respond in, corresponds to its access request address (AR_ADDR) which it has received when it was parked. The order of the half slots (in [Figure 10.10](#) the AR_ADDR numbers linearly increase from 1 to 5) is not fixed: an LMP command sent in the beacon slots may reconfigure the access window. When a slave desires access to the channel, it sends an access request message in the proper slave-to-master half slot. The access request message of the slave is the **ID** packet containing the device access code (DAC) of the master (which is in this case the channel access code without the trailer). The parked slave is only allowed to transmit an access request message in the half slot when in the preceding master-to-slave slot, a broadcast packet has been received. This broadcast message can contain any kind of broadcast information not necessarily related to the parked slave(s). If no broadcast information is available, a broadcast **NULL** or broadcast **POLL** packet shall be sent.

After having sent an access request, the parked slave will listen for an unpark message from the master. As long as no unpark message is received, the slave will repeat the access requests in the subsequent access windows. After the last access window (there are M_{access} windows in total, see [Section 10.8.4.2 on page 114](#)), the parked slave shall listen for an additional N_{poll} time slots for an unpark message. If no unpark message is received within N_{poll} slots after the end of the last access window, the slave may return to sleep and retry an access attempt after the next beacon instant.

After having received the unpark message, the parked slave matching the PM_ADDR or BD_ADDR will leave the park mode and enter the active mode. It will keep listening to the master until it is addressed by the master through its AM_ADDR. The first packet sent by the master must be a POLL packet. The return packet in response to the POLL packet confirms that the slave has been unparked. If no response packet from the slave is received for *newconnectionTO* number of slots after N_{poll} slots after the end of the last access window, the master will send the unpark message to the slave again. If the slave does not receive the POLL packet for *newconnectionTO* number of slots after N_{poll} slots after the end of the last access window, it will return to park, with the same beacon parameters. After confirming that the slave is active, the master decides in which mode the slave will continue.

10.8.4.7 Broadcast scan window

In the beacon train, the master can support broadcast messages to the parked slaves. However, it may extend its broadcast capacity by indicating to the parked slaves that more broadcast information is following after the beacon train. This is achieved by a special LMP command ordering the parked slaves

(as well as the active slaves) to listen to the channel for broadcast messages during a limited time window. This time window starts at the beacon instant and continues for the period as indicated in the LMP command sent in the beacon train.

10.8.5 Polling schemes

10.8.5.1 Polling in active mode

The master always has full control over the piconet. Due to the stringent TDD scheme, slaves can only communicate with the master and not to other slaves. In order to avoid collisions on the ACL link, a slave is only allowed to transmit in the slave-to-master slot when addressed by the AM_ADDR in the packet header in the preceding master-to-slave slot. If the AM_ADDR in the preceding slot does not match, or an AM_ADDR cannot be derived from the preceding slot, the slave is not allowed to transmit.

On the SCO links, the polling rule is slightly modified. The slave is allowed to transmit in the slot reserved for his SCO link unless the (valid) AM_ADDR in the preceding slot indicates a different slave. If no valid AM_ADDR can be derived in the preceding slot, the slave is still allowed to transmit in the reserved SCO slot.

10.8.5.2 Polling in park mode

In the park mode, parked slaves are allowed to send access requests in the access window provided a broadcast packet is received in the preceding master-to-slave slot. Slaves in active mode will not send in the slave-to-master slots following the broadcast packet since they are only allowed to send if addressed specifically.

10.8.6 Slot reservation scheme

The SCO link is established by negotiations between the link managers which involves the exchange of important SCO timing parameters like T_{SCO} and D_{SCO} through LMP messages.

10.8.7 Broadcast scheme

The master of the piconet can broadcast messages which will reach all slaves. A broadcast packet is characterized by the all-zero AM_ADDR. Each new broadcast message (which may be carried by a number of packets) shall start with the flush indication (L_CH=10).

A broadcast packet is never acknowledged. In an error-prone environment, the master may carry out a number of retransmissions to increase the probability for error-free delivery, see also [Section 5.3.5 on page 71](#).



In order to support the **park** mode (as described in [Section 10.8.4 on page 112](#)), a master transmission shall take place at fixed intervals. This master transmission will act as a beacon to which slaves can synchronize. If no traffic takes place at the beacon event, broadcast packets shall be sent. More information is given in [Section 10.8.4 on page 112](#).

10.9 SCATTERNET

10.9.1 General

Multiple piconets may cover the same area. Since each piconet has a different master, the piconets hop independently, each with their own channel hopping sequence and phase as determined by the respective master. In addition, the packets carried on the channels are preceded by different channel access codes as determined by the master device addresses. As more piconets are added, the probability of collisions increases; a graceful degradation of performance results as is common in frequency-hopping spread spectrum systems.

If multiple piconets cover the same area, a unit can participate in two or more overlaying piconets by applying time multiplexing. To participate on the proper channel, it should use the associated master device address and proper clock offset to obtain the correct phase. A Bluetooth unit can act as a slave in several piconets, but only as a master in a single piconet: since two piconets with the same master are synchronized and use the same hopping sequence, they are one and the same piconet. A group of piconets in which connections consists between different piconets is called a **scatternet**.

A master or slave can become a slave in another piconet by being paged by the master of this other piconet. On the other hand, a unit participating in one piconet can page the master or slave of another piconet. Since the paging unit always starts out as master, a master-slave role exchange is required if a slave role is desired. This is described in the [section 10.9.3 on page 121](#).

10.9.2 Inter-piconet communications

Time multiplexing must be used to switch between piconets. In case of ACL links only, a unit can request to enter the **hold** or **park** mode in the current piconet during which time it may join another piconet by just changing the channel parameters. Units in the **sniff** mode may have sufficient time to visit another piconet in between the sniff slots. If SCO links are established, other piconets can only be visited in the non-reserved slots in between. This is only possible if there is a single SCO link using **HV3** packets. In the four slots in between, one other piconet can be visited. Since the multiple piconets are not synchronized, guard time must be left to account for misalignment. This means that only 2 slots can effectively be used to visit another piconet in between the **HV3** packets.

Since the clocks of two masters of different piconets are not synchronized, a slave unit participating in two piconets has to take care of two offsets that, added to its own native clock, create one or the other master clock. Since the two master clocks drift independently, regular updates of the offsets are required in order for the slave unit to keep synchronization to both masters.

10.9.3 Master-slave switch

There are several occasions when a master-slave (MS) switch is desirable. Firstly, a MS switch is necessary when a unit paging the master of an existing piconet wants to join this piconet, since, by definition, the paging unit initially is master of a "small" piconet only involving the pager (master) and the paged (slave) unit. Secondly, when a slave in an existing piconet wants to set up a new piconet, involving itself as master and the current piconet master as slave. The latter case implies a double role of the original piconet master; it becomes a slave in the new piconet while still maintaining the original piconet as master. Thirdly, a much more complicated example is when a slave wants to fully take over an existing piconet, i.e., the switch also involves transfer of other slaves of the existing piconet to the new piconet. Clearly, this can be achieved by letting the new master setup a completely new piconet through the conventional paging scheme. However, that would require individual paging of the old slaves, and, thus, take unnecessarily long time. Instead, letting the new master utilize timing knowledge of the old master is more efficient. As a consequence of the MS switch, the slaves in the piconet have to be transferred to the new piconet, changing their timing and their hopping scheme.

The MS switch is described in step 1 through step 3 below. Prior to the MS switch, encryption if present, must be stopped in the old piconet. For the third example involving the transfer, new piconet parameters have to be communicated to each slave. The process of this is described in step 4 below. Unfortunately, even though all the hooks are defined for an efficient transfer at baseband level, there are still many issues that lack sufficient support in the higher layers of the Bluetooth specification (such as how to handle security and transfer all kind of slave information from old to new master). Until all levels of the specification fully supports this kind of transfer, this functionality will have to be taken care of at application layer. These transfer procedures are outside the scope of the baseband specification.

The MS switch procedure will now be described in more detail. For the master and slave involved in the role switch, the MS switch results in a reversal of their TX and RX timing: a TDD switch. Moreover, since the piconet parameters are derived from the device address and clock of the master, an MS switch inherently involves a redefinition of the piconet as well: a piconet switch. The new piconet's parameters are derived from the former slave's device address and clock.

Assume unit A wants to become master; unit B was the former master. Then there are basically two alternative scenarios: either the slave takes the MS



switch initiative or the master takes the MS switch initiative. These scenarios are described in [Link Manager Protocol, section 3.12 on page 208](#).

Both slave A and master B do the TDD switch but keep the former hopping scheme (still using the device address and clock of unit B), so there is no piconet switch yet. The slot offset information sent by slave A is not used yet but is used in step 3. Unit A now becomes the master, unit B the slave. The AM_ADDR formerly used by unit A in its slave role, is now used by slave B.

At the moment of the TDD switch, both units A and B will start a timer with a time out of `newconnectionTO`. The timer is stopped in slave B as soon as it receives an FHS packet from master A on the TDD-switched channel, the timer is stopped in master A as soon as it receives an ID packet from slave B. If the `newconnectionTO` expires, the master and slave will return to the old piconet timing and take their old role of master and slave. The FHS packet is sent by master A using the "old" piconet parameters. The AM_ADDR in the FHS packet header is the former AM_ADDR used by unit A. The AM_ADDR carried in the FHS payload is the new AM_ADDR intended for unit B when operating on the new piconet. After the FHS acknowledgment, which consists of the ID packet and is sent by the slave on the old hopping sequence, both master A and slave B turn to the new channel parameters of the new piconet as indicated by the FHS.

Since the old and new masters' clocks are synchronous, the clock information sent in the FHS payload should indicate the new master's clock at the beginning of the FHS packet transmission. Furthermore, the 1.25 ms resolution of the clock information given in the FHS packet is not sufficient for aligning the slot boundaries of the two piconets. The slot-offset information in the LMP message previously sent by unit A is used to provide more accurate timing information. The slot offset indicates the delay between the start of the master-to-slave slots of the old and new piconet channels. This timing information ranges from 0 to 1249 μ s with a resolution of 1 μ s. It is used together with the clock information in the FHS packet to accurately position the correlation window when switching to the new master's timing after acknowledgment of the FHS packet. After reception of the FHS packet acknowledgment, the new master A switches to its own timing and sends a POLL packet to verify the switch. Both the master and the slave will start a new timer with a time out of `newconnectionTO` on FHS packet.acknowledgment. The start of this timer shall be aligned with the beginning of the first master TX slot boundary of the new piconet, following the FHS packet acknowledgment. The slave stops the timer when the POLL packet is received; the master stops the timer when the POLL packet is acknowledged. The slave uses a NULL packet to acknowledge the POLL. If no response is received, the master re-sends the POLL packet until `newconnectionTO` is reached. Should this timer expire, both the slave and the master return to the old piconet timing with the old master and slave roles. The procedure may then start again beginning at step 1. Aligning the timer with TX boundaries of the new piconet ensures that no unit returns to the old piconet timing in the middle of a master RX slot.

If the new master wishes to take over slaves from the old piconet (which were slaves to the old master B), a piconet switch is enforced on each slave separately. Since the existing slaves already have the correct TDD timing, a TDD switch is not required. Master A sends a slot-offset LMP message to indicate the difference in slot timing of the old and new piconet channel. Thereafter, master A sends an FHS packets and waits for an acknowledgment in the form of an ID packet. When sending the FHS packet, the master starts a timer with a time out of `newconnectionTO`. The timer is stopped in master A as soon as it receives an ID packet from the slave. Transmission of the FHS packet and the acknowledgment is carried out with the "old" piconet parameters of unit B (compare this to the page hopping scheme used during connection establishment, see Section 10.6.4 on page 104). Should the timer in master A expire, it may restart the transfer operation of step 4. After FHS acknowledgment by the slave, the communication to this slave continues with the new device address and clock of unit A. The FHS packet sent to each slave has the old `AM_ADDR` in the FHS packet header and their new `AM_ADDR` in the FHS packet payload (the new `AM_ADDR` may be identical to the old `AM_ADDR`).

After reception of the FHS packet acknowledgment, the new master A switches to its own timing and hopping sequence and sends a POLL packet to verify the switch. Both the master and the slave will start a timer with a time out of `newconnectionTO` on FHS packet acknowledgment. The start of this timer shall be aligned with the beginning of the first master TX slot boundary of the new piconet, following the FHS packet acknowledgment. The slave stops the timer when the POLL packet is received; the master stops the timer when the POLL packet is acknowledged. The slave uses a NULL packet to acknowledge the POLL. If no response is received, the master re-sends the POLL packet until `newconnectionTO` is reached. If the timer expires, both the slave and the master return to the old piconet parameters. The procedure of step 4 may then be repeated. If an existing slave is out of the range of the new master, master A cannot switch the slave to the new piconet. In that case, the slave loses the connection with the existing piconet after the TDD switch when master B is replaced with master A. As a result, the first message sent by master A being the slot-offset LMP message is never acknowledged by this slave. Due to the link supervision mechanism, the slave will eventually be detached from the old piconet.

Summarized, the MS-switch takes place in two steps: first a TDD switch of the considered master and slave, followed by a piconet switch of the both participants. Then, if so desired, other slaves of the old piconet can be transferred to the new piconet. When a unit have acknowledged the reception of the FHS packet, this unit uses the new piconet parameters defined by the new master and the piconet switch is completed. Note that the SEQN of the first data packet containing a CRC on the new piconet channel is set to 1, see [section 5.3.2 on page 69](#)

A parked slave must be unparked before it can participate in a MS switch. Parked slaves that are transferred to a new piconet shall be activated using the old park parameters, changed to the new piconet parameters, and then returned to the park mode using the new park parameters.

10.10 POWER MANAGEMENT

Features are included into Bluetooth to ensure a low-power operation. These features are both at the microscopic level when handling the packets, and at the macroscopic level using certain operation modes.

10.10.1 Packet handling

In order to minimize power consumption, packet handling is minimized both at TX and RX sides. At the TX side, power is minimized by only sending useful data. This means that if only link control information needs to be exchanged, **NULL** packets will be used. No transmission is carried out at all if there is no link control information or involves a NAK only (NAK is implicit on no reply). If there is data to be sent, the payload length is adapted in order to send only the valid data bytes. At the RX side, packet processing takes place in different steps. If no valid access code is found in the search window, the transceiver returns to sleep. If an access code is found, the receiver unit is woken up and starts to process the packet header. If the HEC fails, the unit will return to sleep after the packet header. A valid header will indicate if a payload will follow and how many slots are involved.

10.10.2 Slot occupancy

As was described in [Section 4.4 on page 54](#), the packet type indicates how many slots a packet may occupy. A slave not addressed in the first slot can go to sleep for the remaining slots the packet may occupy. This can be read from the TYPE code.

10.10.3 Low-power modes

In [Section 10.8 on page 110](#), three modes were described during the **CONNECTION** state which reduce power consumption. If we list the modes in increasing order of power efficiency then the **sniff** mode has the higher duty cycle, followed by the **hold** mode with a lower duty cycle, and finishing with the **park** mode with the lowest duty cycle.

10.11 LINK SUPERVISION

A connection may break down due to various reasons such as a device moving out of range or a power failure condition. Since this may happen without any prior warning, it is important to monitor the link on both the master and the slave side to avoid possible collisions when the AM_ADDR is reassigned to another slave.

To be able to supervise link loss, both the master and the slave use link supervision timers, $T_{\text{supervision}}$. Upon reception of a packet that passes the HEC check and has the correct AM_ADDR, the timer is reset. If at any time in con-



nection state, the timer reaches the *supervisionTO* value, the connection is reset. The same timeout value is used for both SCO and ACL connections.

The timeout period, *supervisionTO*, is negotiated at the LM level. Its value is chosen so that the supervision timeout will be longer than hold and sniff periods. Link supervision of a parked slave will be done by unparking and re-parking the slave.

11 HOP SELECTION

In total, 10 types of hopping sequences are defined – five for the 79-hop and five for the 23-hop system, respectively. Using the notation of parentheses () for figures related to the 23-hop system, these sequences are:

- A **page hopping sequence** with 32 (16) unique wake-up frequencies distributed equally over the 79 (23) MHz, with a period length of 32 (16);
- A **page response sequence** covering 32 (16) unique response frequencies that all are in an one-to-one correspondence to the current page hopping sequence. The master and slave use different rules to obtain the same sequence;
- An **inquiry sequence** with 32 (16) unique wake-up frequencies distributed equally over the 79 (23) MHz, with a period length of 32 (16);
- A **inquiry response sequence** covering 32 (16) unique response frequencies that all are in an one-to-one correspondence to the current inquiry hopping sequence.
- A **channel hopping sequence** which has a very long period length, which does not show repetitive patterns over a short time interval, but which distributes the hop frequencies equally over the 79 (23) MHz during a short time interval;

For the page hopping sequence, it is important that we can easily shift the phase forward or backward, so we need a 1-1 mapping from a counter to the hop frequencies. For each case, both a hop sequence from master to slave and from slave to master are required.

The inquiry and inquiry response sequences always utilizes the GIAC LAP as lower address part and the DCI ([Section 5.4 on page 72](#)) as upper address part in deriving the hopping sequence, even if it concerns a DIAC inquiry.

11.1 GENERAL SELECTION SCHEME

The selection scheme consists of two parts:

- selecting a sequence;
- mapping this sequence on the hop frequencies;

The general block diagram of the hop selection scheme is shown in [Figure 11.1 on page 127](#). The mapping from the input to a particular hop frequency is performed in the selection box. Basically, the input is the native clock and the current address. In **CONNECTION** state, the native clock (CLKN) is modified by an offset to equal the master clock (CLK). Only the 27 MSBs of the clock are used. In the **page** and **inquiry** substates, all 28 bits of the clock are used. However, in **page** substate the native clock will be modified to the master's estimate of the paged unit.

The address input consists of 28 bits, i.e., the entire LAP and the 4 LSBs of the UAP. In **CONNECTION** state, the address of the master is used. In **page** substate the address of the paged unit is used. When in **inquiry** substate, the UAP/LAP corresponding to the GIAC is used. The output constitutes a pseudo-random sequence, either covering 79 hop or 23 hops, depending on the state.

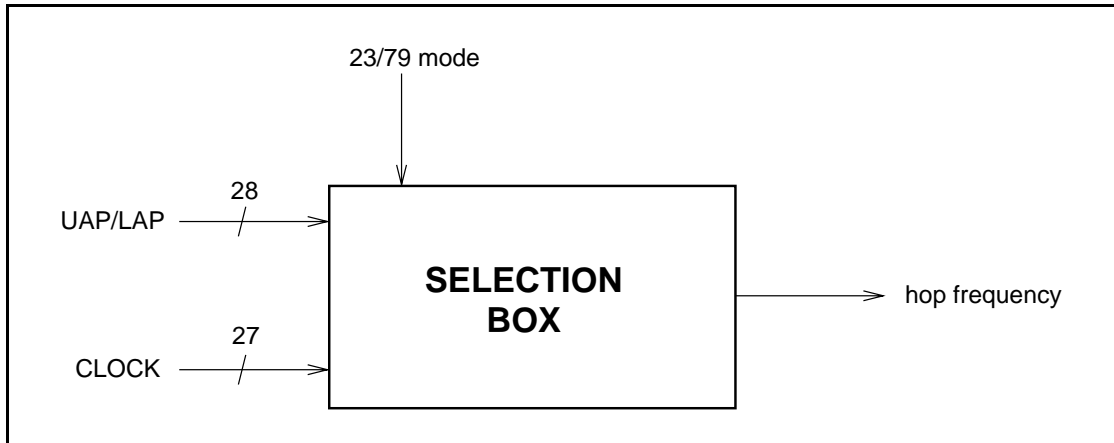


Figure 11.1: General block diagram of hop selection scheme.

For the 79-hop system, the selection scheme chooses a segment of 32 hop frequencies spanning about 64 MHz and visits these hops once in a random order. Next, a different 32-hop segment is chosen, etc. In case of the **page**, **page scan**, or **page response** substates, the same 32-hop segment is used all the time (the segment is selected by the address; different units will have different paging segments). In connection state, the output constitutes a pseudo-random sequence that slides through the 79 hops or 23 hops, depending on the selected hop system. For the 23-hop systems, the segment size is 16. The principle is depicted in Figure 11.2

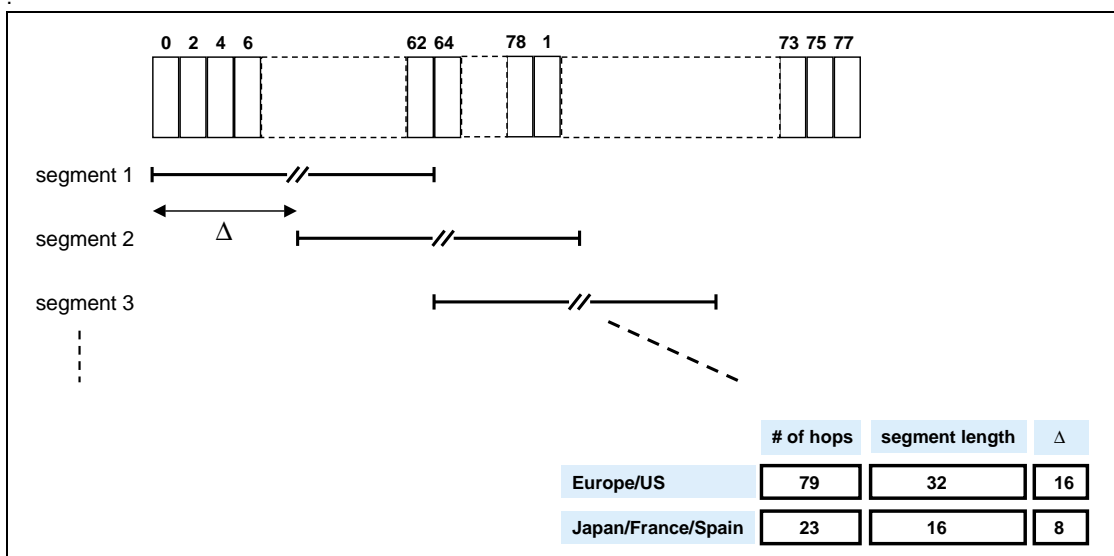


Figure 11.2: Hop selection scheme in CONNECTION state.

11.2 SELECTION KERNEL

The hop selection kernels for the 79 hop system and the 23 hop system are shown in [Figure 11.3 on page 128](#) and [Figure 11.4 on page 128](#), respectively. The X input determines the phase in the 32-hop segment, whereas Y1 and Y2 selects between master-to-slave and slave-to-master transmission. The inputs A to D determine the ordering within the segment, the inputs E and F determine the mapping onto the hop frequencies. The kernel addresses a register containing the hop frequencies. This list should be created such that first all even hop frequencies are listed and then all odd hop frequencies. In this way, a 32-hop segment spans about 64 MHz, whereas a 16-hop segment spans the entire 23-MHz.

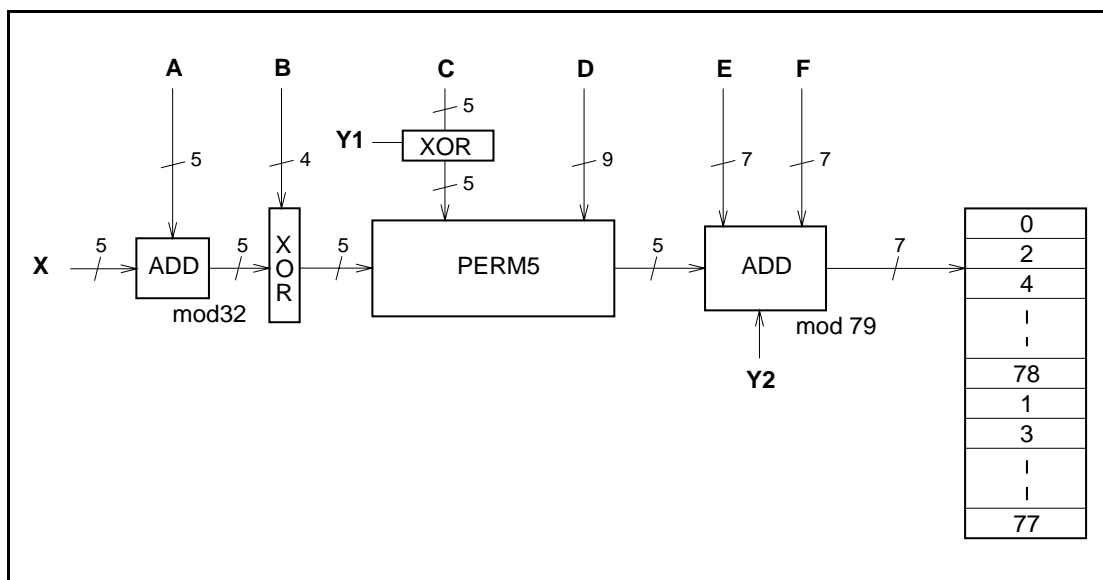


Figure 11.3: Block diagram of hop selection kernel for the 79-hop system.

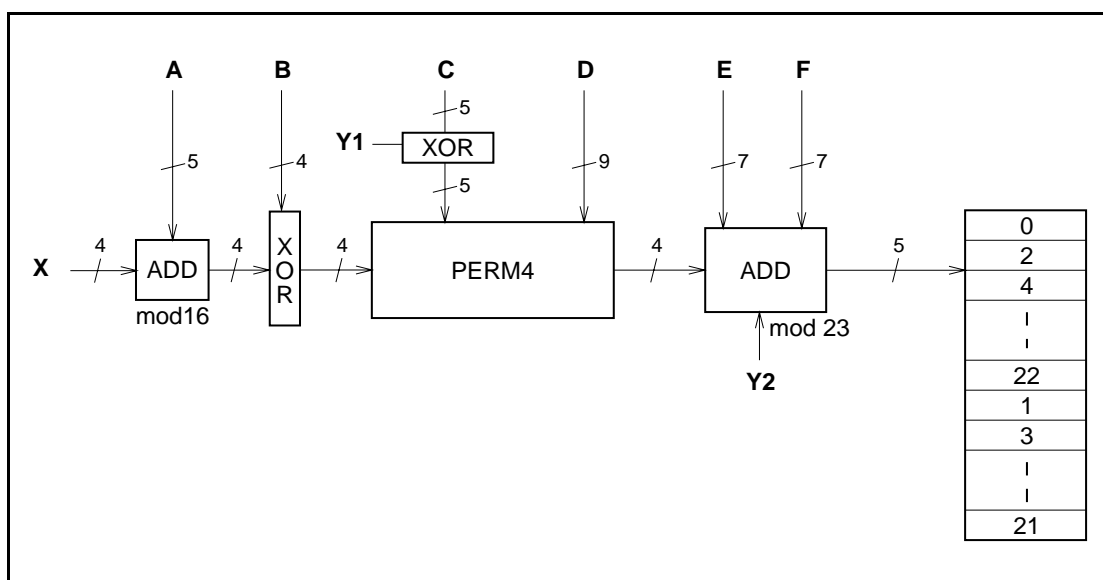


Figure 11.4: Block diagram of hop selection kernel for the 23-hop system.

The selection procedure consists of an addition, an XOR operation, a permutation operation, an addition, and finally a register selection. In the remainder of this chapter, the notation A_i is used for bit i of the BD_ADDR.

11.2.1 First addition operation

The first addition operation only adds a constant to the phase and applies a modulo 32 or a modulo 16 operation. For the page hopping sequence, the first addition is redundant since it only changes the phase within the segment. However, when different segments are concatenated (as in the channel hopping sequence), the first addition operation will have an impact on the resulting sequence.

11.2.2 XOR operation

Let Z' denote the output of the first addition. In the XOR operation, the four LSBs of Z' are modulo-2 added to the address bits A_{22-19} . The operation is illustrated in [Figure 11.5 on page 129](#).

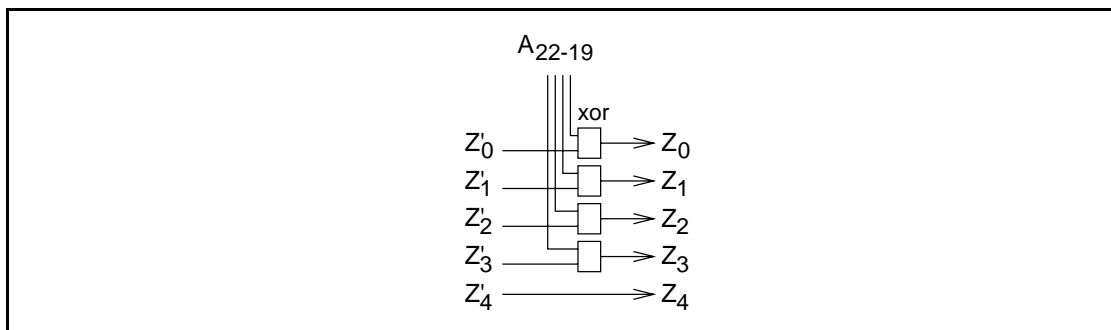


Figure 11.5: XOR operation for the 79-hop system. The 23-hop system is the same except for the Z'_4/Z_4 wire that does not exist.



11.2.3 Permutation operation

The permutation operation involves the switching from 5 inputs to 5 outputs for the 79 hop system and from 4 inputs to 4 outputs for 23 hop system, in a manner controlled by the control word. In [Figure 11.6 on page 131](#) and [Figure 11.7 on page 131](#) the permutation or switching box is shown. It consists of 7 stages of butterfly operations. [Table 11.1](#) and [Table 11.2](#) shows the control of the butterflies by the control signals P. Note that P_{0-8} corresponds to D_{0-8} , and, P_{i+9} corresponds to $C_i \oplus Y1$ for $i = 0 \dots 4$ in [Figure 11.3](#) and [Figure 11.4](#).

Control signal	Butterfly	Control signal	Butterfly
P ₀	{Z ₀ ,Z ₁ }	P ₈	{Z ₁ ,Z ₄ }
P ₁	{Z ₂ ,Z ₃ }	P ₉	{Z ₀ ,Z ₃ }
P ₂	{Z ₁ ,Z ₂ }	P ₁₀	{Z ₂ ,Z ₄ }
P ₃	{Z ₃ ,Z ₄ }	P ₁₁	{Z ₁ ,Z ₃ }
P ₄	{Z ₀ ,Z ₄ }	P ₁₂	{Z ₀ ,Z ₃ }
P ₅	{Z ₁ ,Z ₃ }	P ₁₃	{Z ₁ ,Z ₂ }
P ₆	{Z ₀ ,Z ₂ }		
P ₇	{Z ₃ ,Z ₄ }		

Table 11.1: Control of the butterflies for the 79 hop system

Control signal	Butterfly	Control signal	Butterfly
P ₀	{Z ₀ ,Z ₁ }	P ₈	{Z ₀ ,Z ₂ }
P ₁	{Z ₂ ,Z ₃ }	P ₉	{Z ₁ ,Z ₃ }
P ₂	{Z ₀ ,Z ₃ }	P ₁₀	{Z ₀ ,Z ₃ }
P ₃	{Z ₁ ,Z ₂ }	P ₁₁	{Z ₁ ,Z ₂ }
P ₄	{Z ₀ ,Z ₂ }	P ₁₂	{Z ₀ ,Z ₁ }
P ₅	{Z ₁ ,Z ₃ }	P ₁₃	{Z ₂ ,Z ₃ }
P ₆	{Z ₀ ,Z ₁ }		
P ₇	{Z ₂ ,Z ₃ }		

Table 11.2: Control of the butterflies for the 23 hop system

The Z input is the output of the XOR operation as described in the previous section. The butterfly operation can be implemented with multiplexers as depicted in [Figure 11.8 on page 131](#).

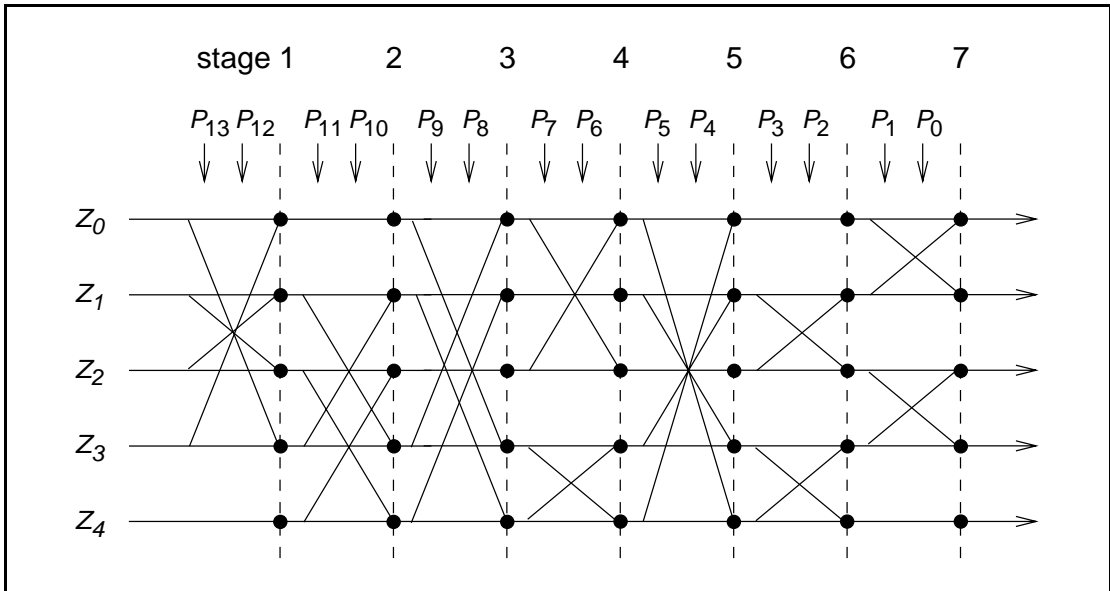


Figure 11.6: Permutation operation for the 79 hop system.

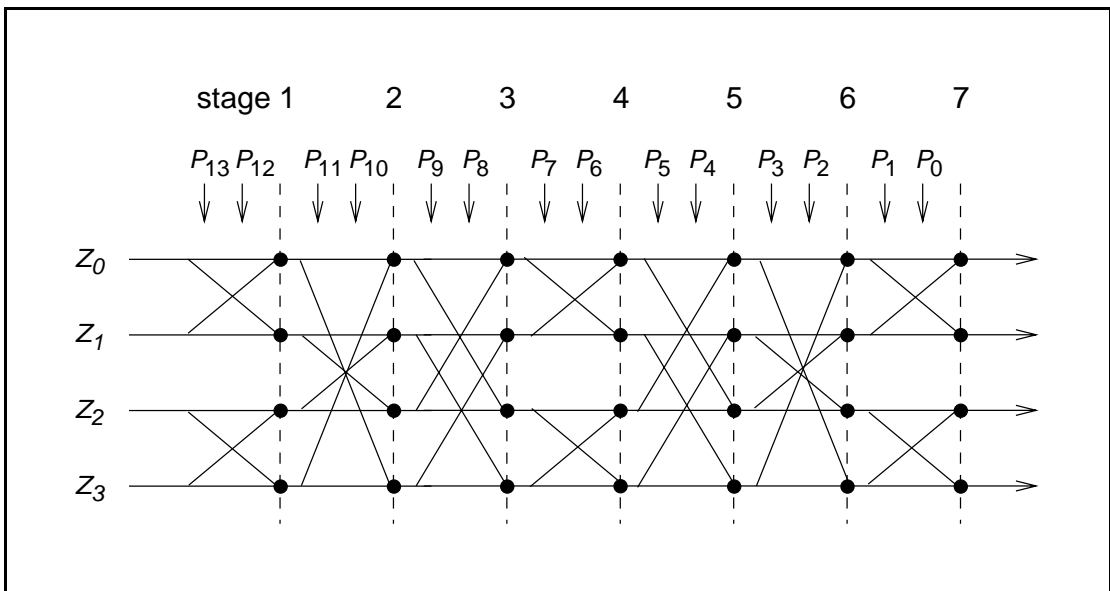


Figure 11.7: Permutation operation for the 23 hop system.

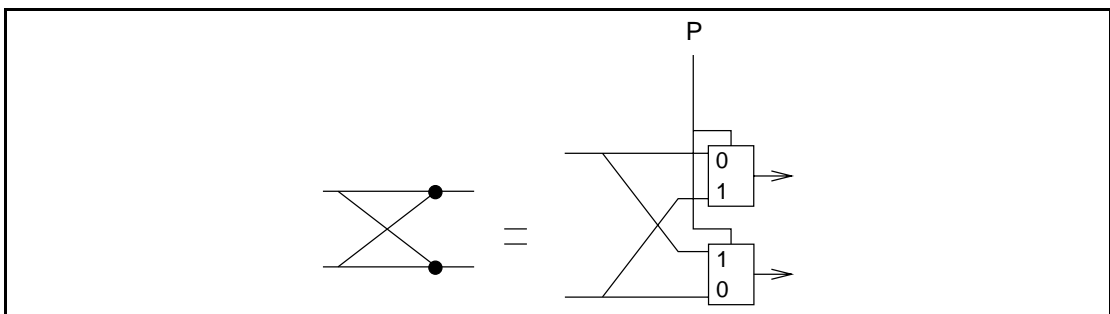


Figure 11.8: Butterfly implementation.



11.2.4 Second addition operation

The addition operation only adds a constant to the output of the permutation operation. As a result, the 16-hop or 32-hop segment is mapped differently on the hop frequencies. The addition is applied modulo 79 or modulo 23 depending on the system type (Europe/US vs. others).

11.2.5 Register bank

The output of the adder addresses a bank of 79 or 23 registers. The registers are loaded with the synthesizer code words corresponding to the hop frequencies 0 to 78 or 0 to 22. Note that the upper half of the bank contains the even hop frequencies, whereas the lower half of the bank contains the odd hop frequencies.

11.3 CONTROL WORD

In the following section $X_{i:j}$, $i < j$, will denote bits $i, i+1, \dots, j$ of the bit vector X . By convention, X_0 is the least significant bit of the vector X .

The control word P of the kernel is controlled by the overall control signals X , $Y1$, $Y2$, and A to F as illustrated in [Figure 11.3 on page 128](#) and [Figure 11.4 on page 128](#). During paging and inquiry, the inputs A to E use the address values as given in the corresponding columns of [Table 11.3 on page 133](#) and [Table 11.4 on page 133](#). In addition, the inputs X , $Y1$ and $Y2$ are used. The F input is unused. In the 79-hop system, the clock bits $CLK_{6:2}$ (i.e., input X) specifies the phase within the length 32 sequence, while for the 23-hop system, $CLK_{5:2}$ specifies the phase within the length 16 sequence. For both systems, CLK_1 (i.e., inputs $Y1$ and $Y2$) is used to select between TX and RX. The address inputs determine the sequence order within segments. The final mapping onto the hop frequencies is determined by the register contents.

In the following we will distinguish between three types of clocks: the piconet's master clock, the Bluetooth unit's native clock, and the clock estimate of a paged Bluetooth unit. These types are marked in the following way:

1. $CLK_{27:0}$: Master clock of the current piconet.
2. $CLKN_{27:0}$: Native clock of the unit.
3. $CLKE_{27:0}$: The paging unit's estimate of the paged unit's native clock.

During the **CONNECTION** state, the inputs A , C and D result from the address bits being bit-wise XORed with the clock bits as shown in the "Connection state" column of [Table 11.3 on page 133](#) and [Table 11.4 on page 133](#) (the two MSBs are XORed together, the two second MSBs are XORed together, etc.). Consequently, after every 32 (16) time slots, a new length 32 (16) segment is selected in the 79-hop (23-hop) system. The sequence order within a specific



segment will not be repeated for a very long period. Thus, the overall hopping sequence consists of concatenated segments of 32-hops each. Since each 32-hop sequence spans more than 80% of the 79 MHz band, the desired frequency spreading over a short time interval is obtained.

	Page scan/ Inquiry scan	Page/Inquiry	Page response (master/slave) and Inquiry response	Connection state
X	$CLKN_{16-12} / Xir_{4-0}^{(79)}$	$Xp_{4-0}^{(79)} / Xi_{4-0}^{(79)}$	$Xprm_{4-0}^{(79)} / Xprs_{4-0}^{(79)} / Xir_{4-0}^{(79)}$	CLK_{6-2}
Y1	0	$CLKE_1 / CLKN_1$	$CLKE_1 / CLKN_1 / 1$	CLK_1
Y2	0	$32 \times CLKE_1 / 32 \times CLKN_1$	$32 \times CLKE_1 / 32 \times CLKE_1 \quad 32 \times 1$	$32 \times CLK_1$
A	A_{27-23}	A_{27-23}	A_{27-23}	$A_{27-23} \oplus CLK_{25-21}$
B	A_{22-19}	A_{22-19}	A_{22-19}	A_{22-19}
C	$A_{8,6,4,2,0}$	$A_{8,6,4,2,0}$	$A_{8,6,4,2,0}$	$A_{8,6,4,2,0} \oplus CLK_{20-16}$
D	A_{18-10}	A_{18-10}	A_{18-10}	$A_{18-10} \oplus CLK_{15-7}$
E	$A_{13,11,9,7,5,3,1}$	$A_{13,11,9,7,5,3,1}$	$A_{13,11,9,7,5,3,1}$	$A_{13,11,9,7,5,3,1}$
F	0	0	0	$16 \times CLK_{27-7} \bmod \mathcal{D}$

Table 11.3: Control for 79-hop system.

	Page scan/ Inquiry scan	Page/Inquiry	Page response (master/slave) and Inquiry response	Connection state
X	$CLKN_{15-12} / Xir_{3-0}^{(23)}$	$Xp_{3-0}^{(23)} / Xi_{3-0}^{(23)}$	$Xprm_{3-0}^{(23)} / Xprs_{3-0}^{(23)} / Xir_{3-0}^{(23)}$	CLK_{5-2}
Y1	0	$CLKE_1 / CLKN_1$	$CLKE_1 / CLKN_1 / 1$	CLK_1
Y2	0	$16 \times CLKE_1 / 16 \times CLKN_1$	$16 \times CLKE_1 / 16 \times CLKE_1 \quad 16 \times 1$	$16 \times CLK_1$
A	A_{27-23}	A_{27-23}	A_{27-23}	$A_{27-23} \oplus CLK_{25-21}$
B	A_{22-19}	A_{22-19}	A_{22-19}	A_{22-19}
C	$A_{8,6,4,2,0}$	$A_{8,6,4,2,0}$	$A_{8,6,4,2,0}$	$A_{8,6,4,2,0} \oplus CLK_{20-16}$

Table 11.4: Control for 23-hop system.

	Page scan/ Inquiry scan	Page/Inquiry	Page response (master/slave) and Inquiry response	Connection state
D	A_{18-10}	A_{18-10}	A_{18-10}	$A_{18-10} \oplus \text{CLK}_{15-7}$
E	$A_{13, 11, 9, 7, 5, 3, 1}$	$A_{13, 11, 9, 7, 5, 3, 1}$	$A_{13, 11, 9, 7, 5, 3, 1}$	$A_{13, 11, 9, 7, 5, 3, 1}$
F	0	0	0	$6 \times \text{CLK}_{27-6} \bmod 23$

Table 11.4: Control for 23-hop system.

11.3.1 Page scan and Inquiry scan substates

In **page scan**, the Bluetooth device address of the scanning unit is used as address input. In **inquiry scan**, the GIAC LAP and the four LSBs of the DCI (as A_{27-24}), are used as address input for the hopping sequence. Naturally, for the transmitted access code and in the receiver correlator, the appropriate GIAC or DIAC is used. The application decides which inquiry access code to use depending on the purpose of the inquiry.

The five X input bits vary depending on the current state of the unit. In the **page scan** and **inquiry scan** substates, the native clock (CLKN) is used. In **CONNECTION** state the master clock (CLK) is used as input. The situation is somewhat more complicated for the other states.

11.3.2 Page substate

In the **page** substate of the 79-hop system, the paging unit shall start using the **A**-train, i.e., $\{f(k-8), \dots, f(k), \dots, f(k+7)\}$, where $f(k)$ is the source's estimate of the current receiver frequency in the paged unit. Clearly, the index k is a function of all the inputs in Figure 11.3. There are 32 possible paging frequencies within each 1.28 second interval. Half of these frequencies belongs to the **A**-train, the rest (i.e., $\{f(k+8), \dots, f(k+15), f(k-16), \dots, f(k-9)\}$) belongs to the **B**-train. In order to achieve the -8 offset of the **A**-train, a constant of 24 can be added to the clock bits (which is equivalent to -8 due to the modulo 32 operation). Clearly, the **B**-train may be accomplished by setting the offset to 8. A cyclic shift of the order within the trains is also necessary in order to avoid a possible repetitive mismatch between the paging and scanning units. Thus,

$$Xp^{(79)} = [\text{CLKE}_{16-12} + k_{offset} + (\text{CLKE}_{4-2,0} - \text{CLKE}_{16-12}) \bmod 16] \bmod 32, \quad (\text{EQ } 2)$$

where

$$k_{offset} = \begin{cases} 24 & \text{A-train,} \\ 8 & \text{B-train.} \end{cases} \quad (\text{EQ } 3)$$

Alternatively, each switch between the **A**- and **B**-trains may be accomplished by adding 16 to the current value of k_{offset} (originally initialized with 24).

In the **page** substate of the 23-hop system, the paging unit makes use of the **A**-train only. A constant offset of 8 is used in order to start with $f(k - 8)$. Moreover, only four bits are needed since the additions are modulo 16. Consequently,

$$X_p^{(23)} = [\text{CLKE}_{15-12} + 8 + \text{CLKE}_{4-2,0}] \bmod 16, \tag{EQ 4}$$

11.3.3 Page response

11.3.3.1 Slave response

A unit in the **page scan** substate recognizing its own access code enters the **slave response** substate. In order to eliminate the possibility of losing the link due to discrepancies of the native clock CLKN and the master's clock estimate CLKE, the four bits CLKN_{16-12} must be frozen at their current value. The value is frozen to the content it has in the slot where the recipient's access code is detected. Note that the actual native clock is *not* stopped; it is merely the values of the bits used for creating the X-input that are kept fixed for a while. In the sequel, a frozen value is marked by an asterisk (*).

For each response slot the paged unit will use an X-input value one larger (modulo 32 or 16) than in the preceding response slot. However, the first response is made with the X-input kept at the same value as it was when the access code was recognized. Let N be a counter starting at zero. Then, the X-input in the $(N + 1)$ -th response slot (the first response slot being the one immediately following the page slot now responding to) of the **slave response** substate becomes

$$X_{prs}^{(79)} = [\text{CLKN}^*_{16-12} + N] \bmod 32, \tag{EQ 5}$$

and

$$X_{prs}^{(23)} = [\text{CLKN}^*_{15-12} + N] \bmod 16, \tag{EQ 6}$$

for the 79-hop and 23-hop systems, respectively. The counter N is set to zero in the slot where the slave acknowledges the page (see [Figure 10.6 on page 102](#) and [Figure 10.7 on page 102](#)). Then, the value of N is increased by one each time CLKN_1 is set to zero, which corresponds to the start of a master TX slot. The X-input is constructed this way until the first accepted **FHS** packet is received *and* the immediately following response packet has been transmitted. After this the slave enters the **CONNECTION** state using the parameters received in the **FHS** packet.



11.3.3.2 Master response

The paging unit enters **master response** substate upon receiving a slave response. Clearly, also the master must freeze its estimated slave clock to the value that triggered a response from the paged unit. It is equivalent to using the values of the clock estimate when receiving the slave response (since only $CLKE_1$ will differ from the corresponding page transmission). Thus, the values are frozen when the slave **ID** packet is received. In addition to the used clock bits, also the current value of k_{offset} must be frozen. The master will adjust its X-input in the same way the paged unit does, i.e., by incrementing this value by one for each time $CLKE_1$ is set to zero. The first increment shall be done before sending the **FHS** packet to the paged unit. Let N be a counter starting at one. The rules for forming the X-inputs become

$$X_{prm}^{(79)} = [CLKE_{16-12}^* + k_{offset}^* + (CLKE_{4-2,0}^* - CLKE_{16-12}^*) \bmod 16 + N] \bmod 32, \quad (EQ 7)$$

and

$$X_{prm}^{(23)} = [CLKE_{15-12}^* + 8 + CLKE_{4-2,0}^* + N] \bmod 16, \quad (EQ 8)$$

for the 79-hop and 23-hop systems, respectively. The value of N is increased each time $CLKE_1$ is set to zero, which corresponds to the start of a master TX slot.

11.3.4 Inquiry substate

The X-input of the **inquiry** substate is quite similar to what is used in the **page** substate. Since no particular unit is addressed, the native clock $CLKN$ of the inquirer is used. Moreover, which of the two train offsets to start with is of no real concern in this state. Consequently,

$$X_i^{(79)} = [CLKN_{16-12} + k_{offset} + (CLKN_{4-2,0} - CLKN_{16-12}) \bmod 16] \bmod 32, \quad (EQ 9)$$

where k_{offset} is defined by (EQ 3) on page 134. The initial choice of the offset is arbitrary. For the 23-hop system,

$$X_i^{(23)} = [CLKN_{15-12} + 8 + CLKN_{4-2,0}] \bmod 16, \quad (EQ 10)$$

The GIAC LAP and the four LSBs of the DCI (as A_{27-24}) are used as address input for the hopping sequence generator. (EQ 11)

11.3.5 Inquiry response

The **inquiry response** substate is similar to the **slave response** substate with respect to the X-input. However, there is no need to freeze the clock input, thus

$$X_{ir}^{(79)} = [\text{CLKN}_{16-12} + N] \bmod 32, \quad (\text{EQ 12})$$

and

$$X_{ir}^{(23)} = [\text{CLKN}_{15-12} + N] \bmod 16, \quad (\text{EQ 13})$$

for the 79-hop and 23-hop systems, respectively. Furthermore, the counter N is increased not on **clocks**, but rather after each **FHS** packet has been transmitted in response to the inquiry. There is no restriction on the initial value of N as it is independent of the corresponding value in the inquiring unit.

The GIAC LAP and the four LSBs of the DCI (as A_{27-24}) are used as address input for the hopping sequence generator. The other input bits to the generator are the same as in the case of page response.

11.3.6 Connection state

In **CONNECTION** state, the clock bits to use in the channel hopping sequence generation are always according to the master clock, CLK. The address bits are taken from the Bluetooth device address of the master.

12 BLUETOOTH AUDIO

On the Bluetooth air-interface, either a 64 kb/s log PCM format (A-law or μ -law) is used, or a 64 kb/s CVSD (Continuous Variable Slope Delta Modulation) is used. The latter format applies an adaptive delta modulation algorithm with syllabic companding.

The voice coding on the line interface should have a quality equal to or better than the quality of 64 kb/s log PCM.

[Table 12.1 on page 138](#) summarizes the voice coding schemes supported on the air interface. The appropriate voice coding scheme is selected after negotiations between the Link Managers.

Voice Codecs	
linear	CVSD
8-bit logarithmic	A-law
	μ -law

Table 12.1: Voice coding schemes supported on the air interface.

12.1 LOG PCM CODEC

Since the voice channels on the air-interface can support a 64 kb/s information stream, a 64 kb/s log PCM traffic can be used for transmission. Either A-law or μ -law compression can be applied. In the event that the line interface uses A-law and the air interface uses μ -law or vice versa, a conversion from A-law to μ -law is performed. The compression method follows ITU-T recommendations G. 711.

12.2 CVSD CODEC

A more robust format for voice over the air interface is a delta modulation. This modulation scheme follows the waveform where the output bits indicate whether the prediction value is smaller or larger than the input waveform. To reduce slope overload effects, syllabic companding is applied: the step size is adapted according to the average signal slope. The input to the CVSD encoder is 64 ksamples/s linear PCM. Block diagrams of the CVSD encoder and CVSD decoder are shown in [Figure 12.1 on page 139](#), [Figure 12.2 on page 139](#) and [Figure 12.3 on page 139](#). The system is clocked at 64 kHz.

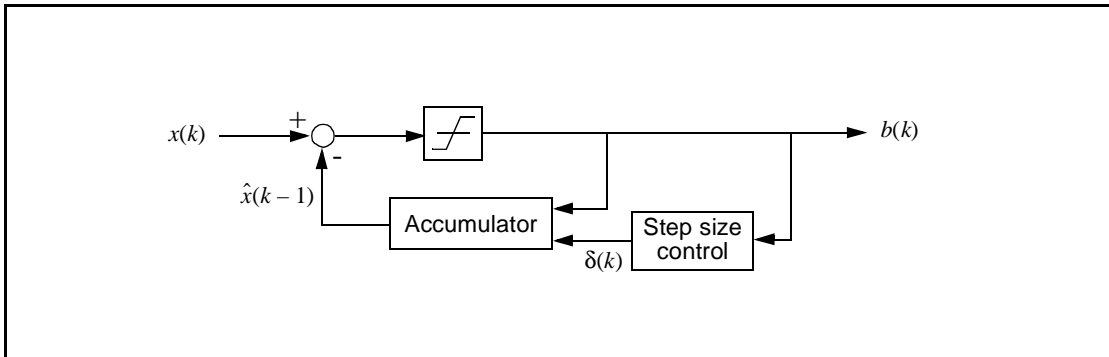


Figure 12.1: Block diagram of CVSD encoder with syllabic companding.

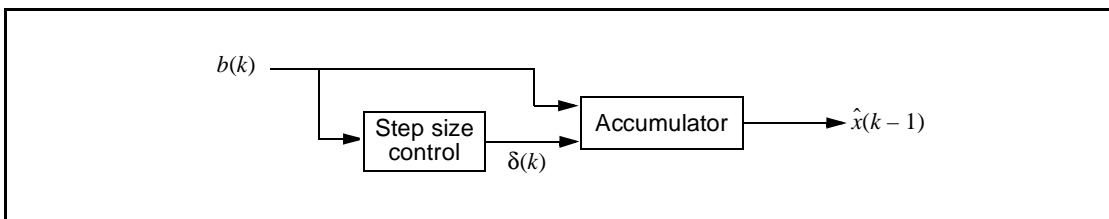


Figure 12.2: Block diagram of CVSD decoder with syllabic companding.

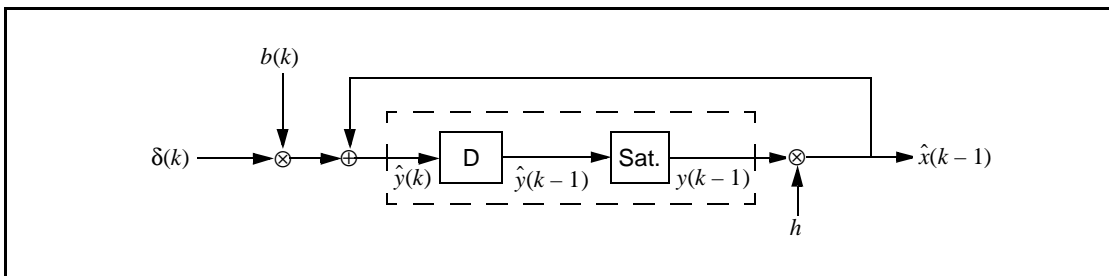


Figure 12.3: Accumulator procedure

Let $\text{sgn}(x) = 1$ for $x \geq 0$, otherwise $\text{sgn}(x) = -1$. On air these numbers are represented by the sign bit; i.e. negative numbers are mapped on “1” and positive numbers are mapped on “0”. Denote the CVSD encoder output bit $b(k)$, the accumulator contents $y(k)$, and the step size $\delta(k)$. Furthermore, let h be the decay factor for the accumulator, let β denote the decay factor for the step size, and, let α be the syllabic companding parameter. The latter parameter monitors the slope by considering the K most recent output bits

Let

$$\hat{x}(k) = hy(k). \tag{EQ 14}$$

Then, the CVSD encoder internal state is updated according to the following set of equations:

$$b(k) = \text{sgn}\{x(k) - \hat{x}(k-1)\}, \tag{EQ 15}$$



$$\alpha = \begin{cases} 1, & \text{if } J \text{ bits in the last } K \text{ output bits are equal,} \\ 0, & \text{otherwise,} \end{cases} \quad (\text{EQ 16})$$

$$\delta(k) = \begin{cases} \min\{\delta(k-1) + \delta_{min}, \delta_{max}\}, & \alpha = 1, \\ \max\{\beta\delta(k-1), \delta_{min}\}, & \alpha = 0, \end{cases} \quad (\text{EQ 17})$$

$$y(k) = \begin{cases} \min\{\hat{y}(k), y_{max}\}, & \hat{y}(k) \geq 0. \\ \max\{\hat{y}(k), y_{min}\}, & \hat{y}(k) < 0. \end{cases} \quad (\text{EQ 18})$$

where

$$\hat{y}(k) = \hat{x}(k-1) + b(k)\delta(k). \quad (\text{EQ 19})$$

In these equations, δ_{min} and δ_{max} are the minimum and maximum step sizes, and, y_{min} and y_{max} are the accumulator’s negative and positive saturation values, respectively. Over air, the bits are sent in the same order they are generated by the CVSD encoder.

For a 64 kb/s CVSD, the parameters as shown in [Table 12.2](#) must be used. The numbers are based on a 16 bit signed number output from the accumulator. These values result in a time constant of 0.5 ms for the accumulator decay, and a time constant of 16 ms for the step size decay

Parameter	Value
h	$1 - \frac{1}{32}$
β	$1 - \frac{1}{1024}$
J	4
K	4
δ_{min}	10
δ_{max}	1280
y_{min}	-2^{15} or $-2^{15} + 1$
y_{max}	$2^{15} - 1$

Table 12.2: CVSD parameter values. The values are based on a 16 bit signed number output from the accumulator.

12.3 ERROR HANDLING

In the **DV** and **HV3** packet, the voice is not protected by FEC. The quality of the voice in an error-prone environment then depends on the robustness of the voice coding scheme. CVSD, in particular, is rather insensitive to random bit errors, which are experienced as white background noise. However, when a packet is rejected because either the channel access code or the HEC test was unsuccessful, measures have to be taken to “fill” in the lost speech segment.

The voice payload in the **HV2** packet is protected by a 2/3 rate FEC. For errors that are detected but cannot be corrected, the receiver should try to minimize the audible effects. For instance, from the 15-bit FEC segment with uncorrected errors, the 10-bit information part as found before the FEC decoder should be used. The **HV1** packet is protected by a 3 bit repetition FEC. For this code, the decoding scheme will always assume zero or one-bit errors. Thus, there exist no detectable but uncorrectable error events for **HV1** packets.

12.4 GENERAL AUDIO REQUIREMENTS

These specifications are tentative and will be fixed within 18 months after the release of the Bluetooth Specification version 1.0 Draft Foundation.

12.4.1 Signal levels

For A-law and μ -law log-PCM encoded signals the requirements on signal levels follows ITU-T G.711.

Full swing at the 16 bit linear PCM interface to the CVSD encoder is defined to be 3 dBm0. A digital CVSD encoded test signal is provided in a Test Signal file available on the [website](#). This signal is generated by a software implementation of a reference CVSD encoder. The digital encoder input signal (1020 Hz, sine-wave) generating the test signal has a nominal power of -15 dBm0. When the CVSD encoded test signal is fed through the CVSD receiver chain, the nominal output power should be -15 ± 1.0 dBm0.

12.4.2 CVSD audio quality

For Bluetooth audio quality the requirements are put on the transmitter side. The 64 ksamples/s linear PCM input signal must have negligible spectral power density above 4 kHz. A set of reference input signals are encoded by the transmitter and sent through a reference decoder (available on the [website](#)). The power spectral density in the 4-32 kHz band of the decoded signal at the 64 ksamples/s linear PCM output, should be more than 20 dB below the maximum in the 0-4 kHz range.

13 BLUETOOTH ADDRESSING

13.1 BLUETOOTH DEVICE ADDRESS (BD_ADDR)

Each Bluetooth transceiver is allocated a unique 48-bit Bluetooth device address (BD_ADDR). This address is derived from the IEEE802 standard. This 48-bit address is divided into three fields:

- LAP field: lower address part consisting of 24 bits
- UAP field: upper address part consisting of 8 bits
- NAP field: non-significant address part consisting of 16 bits

The LAP and UAP form the significant part of the BD_ADDR. The total address space obtained is 2^{32} .

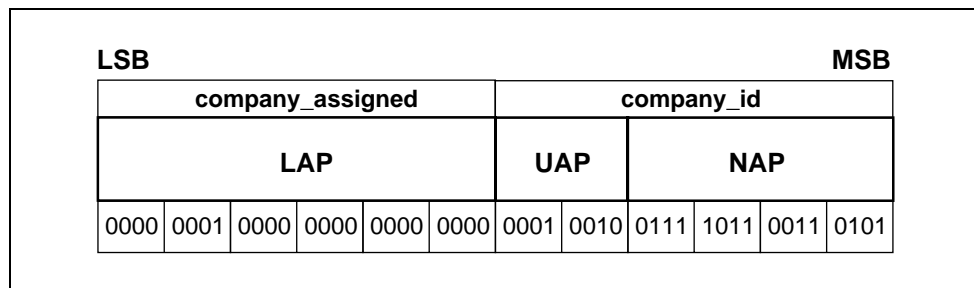


Figure 13.1: Format of BD_ADDR

13.2 ACCESS CODES

In the Bluetooth system, 72-bit and 68-bit access codes are used for signalling purposes. Three different access codes are defined, see also [Section 4.2.1 on page 48](#):

- device access code (DAC)
- channel access code (CAC)
- inquiry access code (IAC)

There is one general IAC (GIAC) for general inquiry operations and there are 63 dedicated IACs (DIACs) for dedicated inquiry operations. All codes are derived from a LAP of the BD_ADDR. The device access code is used during page, page scan and page response substates. It is a code derived from the unit's BD_ADDR. The channel access code characterizes the channel of the piconet and forms the preamble of all packets exchanged on the channel. The channel access code is derived from the LAP of the master BD_ADDR. Finally, the inquiry access code is used in inquiry operations. A general inquiry access code is common to all Bluetooth units; a set of dedicated inquiry access codes is used to inquire for classes of devices.

The access code is also used to indicate to the receiver the arrival of a packet. It is used for timing synchronization and offset compensation. The receiver correlates against the entire sync word in the access code, providing a very robust signalling. During channel setup, the code itself is used as an ID packet to support the acquisition process. In addition, it is used during random access procedures in the PARK state.

The access code consists of preamble, sync word and a trailer, see [Section 4.2 on page 48](#). The next two sections describe the generation of the sync word.

13.2.1 Synchronization word definition

The sync words are based on a (64,30) expurgated block code with an overlay (bit-wise XOR) of an 64 bit full length PN-sequence. The expurgated code guarantees large Hamming distance ($d_{min} = 14$) between sync words based on different addresses. The PN sequence improves the auto correlation properties of the access code. The following steps describe how to generate the sync word:

1. Generate information sequence;
2. XOR this with the “information covering” part of the PN overlay sequence;
3. Generate the codeword;
4. XOR the codeword with all 64 bits of the PN overlay sequence;

The information sequence is generated by appending 6 bits to the 24 bit LAP (step 1). The appended bits are 001101 if the MSB of the LAP equals 0. If the MSB of the LAP is 1 the appended bits are 110010. The LAP MSB together with the appended bits constitute a length-seven Barker sequence. The purpose of including a Barker sequence is to further improve the auto correlation properties. In step 2 the information is pre-scrambled by XORing it with the bits $p_{34} \dots p_{63}$ of the *pseudo-random noise* (PN) sequence (defined in [section 13.2.2 on page 145](#)). After generating the codeword (step 3), the complete PN sequence is XORed to the codeword (step 4). This step de-scrambles the information part of the codeword. At the same time the parity bits of the codeword are scrambled. Consequently, the original LAP and Barker sequence are ensured a role as a part of the access code sync word, and the cyclic properties of the underlying code is removed. The principle is depicted in [Figure 13.2 on page 144](#)

In the sequel, binary sequences will be denoted by their corresponding D-transform (in which D^i represents a delay of i time units). Let

$p'(D) = p'_0 + p'_1 D + \dots + p'_{62} D^{62}$ be the 63 bit pseudo-random sequence, where p'_0 is the first bit (LSB) leaving the PRNG (see [Figure 13.3 on page 146](#)), and, p'_{62} is the last bit (MSB). To obtain 64 bits, an extra zero is appended at the *end* of this sequence (thus, $p'(D)$ is unchanged). For notational convenience, the

reciprocal of this extended polynomial, $p(D) = D^{63}p'(1/D)$, will be used in the sequel. This is the sequence $p'(D)$ in reverse order. We denote the 24 bit lower address part (LAP) of the Bluetooth address by $a(D) = a_0 + a_1D + \dots + a_{23}D^{23}$ (a_0 is the LSB of the Bluetooth address).

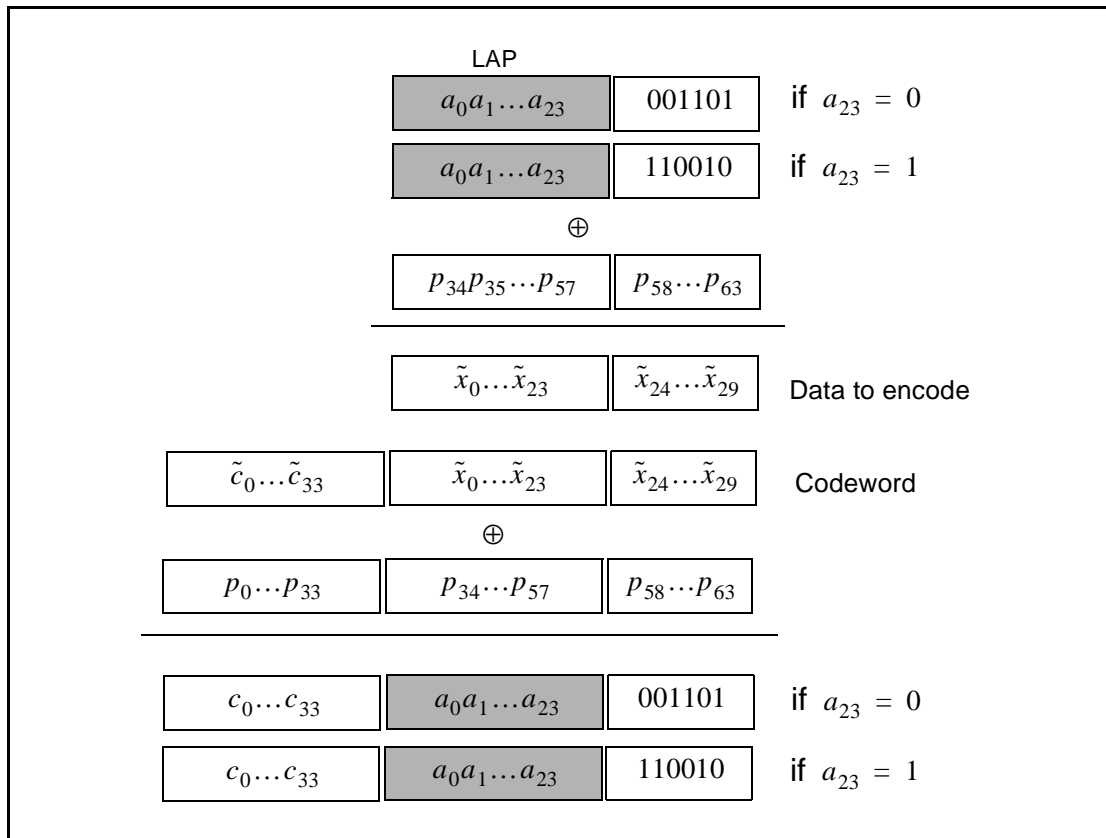


Figure 13.2: Construction of the sync word.

The (64,30) block code generator polynomial is denoted $g(D) = (1 + D)g'(D)$, where $g'(D)$ is the generator polynomial 157464165547 (octal notation) of a primitive binary (63,30) BCH code. Thus, in octal notation we have

$$g(D) = 260534236651, \tag{EQ 20}$$

the left-most bit corresponds to the high-order (g_{34}) coefficient. The DC-free four bit sequences 0101 and 1010 can be written

$$\begin{cases} F_0(D) = D + D^3, \\ F_1(D) = 1 + D^2, \end{cases} \tag{EQ 21}$$

respectively. Furthermore, we define

$$\begin{cases} B_0(D) = D^2 + D^3 + D^5, \\ B_1(D) = 1 + D + D^4, \end{cases} \quad (\text{EQ 22})$$

which are used to create the length seven Barker sequences. Then, the access code is generated by the following procedure:

1. Format the 30 information bits to encode:

$$x(D) = a(D) + D^{24}B_{a_{23}}(D).$$

2. Add the information covering part of the PN overlay sequence:

$$\tilde{x}(D) = x(D) + p_{34} + p_{35}D + \dots + p_{63}D^{29}.$$

3. Generate parity bits of the (64,30) expurgated block code:¹

$$\tilde{c}(D) = D^{34}\tilde{x}(D) \bmod g(D).$$

4. Create the codeword:

$$\tilde{s}(D) = D^{34}\tilde{x}(D) + \tilde{c}(D).$$

5. Add the PN sequence:

$$s(D) = \tilde{s}(D) + p(D).$$

6. Append the (DC-free) preamble and trailer:

$$y(D) = F_{c_0}(D) + D^4s(D) + D^{68}F_{a_{23}}(D).$$

13.2.2 Pseudo-random noise sequence generation

To generate the pseudo-random noise sequence we use the primitive polynomial $h(D) = 1 + D + D^3 + D^4 + D^6$. The LFSR and its starting state are shown in [Figure 13.3 on page 146](#). The PN sequence generated (including the extra terminating zero) becomes (hexadecimal notation) 83848D96BBCC54FC. The LFSR output starts with the left-most bit of this PN sequence. This corresponds to $p'(D)$ of the previous section. Thus, using the reciprocal $p(D)$ as overlay gives the 64 bit sequence

$$p = 3F2A33DD69B121C1, \quad (\text{EQ 23})$$

1. $x(D) \bmod y(D)$ denotes the rest when $x(D)$ is divided by $y(D)$.

where the left-most bit is p_0 (there are two initial zeros in the binary representation of the hexadecimal digit 3), and $p_{63} = 1$ is the right-most bit.

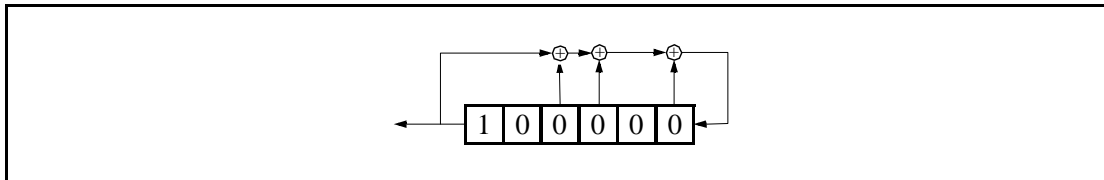


Figure 13.3: LFSR and the starting state to generate $p'(D)$

13.2.3 Reserved addresses for GIAC and DIAC

There is a block of 64 contiguous LAPs reserved for Bluetooth inquiry operations; one LAP common to all Bluetooth devices is reserved for general inquiry, the remaining 63 LAPs are reserved for dedicated inquiry of specific classes of Bluetooth devices. The same 64-block is used regardless of the contents of UAP and NAP. Consequently, none of these LAPs can be part of a user BD_ADDR.

When initializing HEC and CRC for the FHS packet of **inquiry response**, the UAP is replaced by DCI. Likewise, whenever one of the reserved BD_ADDRs is used for generating a frequency hop sequence, the UAP will be replaced by the DCI.

The reserved LAP addresses are tentatively chosen as 0x9E8B00–0x9E8B3F. The general inquiry LAP is tentatively chosen to 0x9E8B33. All addresses have the LSB at the rightmost position, hexadecimal notation.

13.3 ACTIVE MEMBER ADDRESS (AM_ADDR)

Each slave active in a piconet is assigned a 3-bit active member address (AM_ADDR). The all-zero AM_ADDR is reserved for broadcast messages. The master does not have an AM_ADDR. Its timing relative to the slaves distinguishes it from the slaves. A slave only accepts a packet with a matching AM_ADDR and broadcast packets. The AM_ADDR is carried in the packet header. The AM_ADDR is only valid as long as a slave is active on the channel. As soon as it is disconnected or parked, it loses the AM_ADDR.

The AM_ADDR is assigned by the master to the slave when the slave is activated. This is either at connection establishment or when the slave is unparked. At connection establishment, the AM_ADDR is carried in the **FHS** payload (the **FHS** header itself carries the all-zero AM_ADDR). When unparking, the AM_ADDR is carried in the unpark message.

13.4 PARKED MEMBER ADDRESS (PM_ADDR)

A slave in park mode can be identified by its BD_ADDR or by a dedicated parked member address (PM_ADDR). This latter address is a 8-bit member address that separates the parked slaves. The PM_ADDR is only valid as long as the slave is parked. When the slave is activated it is assigned an AM_ADDR but loses the PM_ADDR. The PM_ADDR is assigned to the slave the moment it is parked.

The all-zero PM_ADDR is reserved for parked slaves that only use their BD_ADDR to be unparked.

13.5 ACCESS REQUEST ADDRESS (AR_ADDR)

The access request address is used by the parked slave to determine the slave-to-master half slot in the access window it is allowed to send access request messages in, see also [Section 10.8.4.6 on page 118](#). The AR_ADDR is assigned to the slave when it enters the park mode and is only valid as long as the slave is parked. The AR_ADDR is not necessarily unique; i.e. different parked slaves may have the same AR_ADDR.

14 BLUETOOTH SECURITY

The Bluetooth technology provides peer-to-peer communications over short distances. In order to provide usage protection and information confidentiality, the system has to provide security measures both at the application layer and the link layer. These measures shall be appropriate for a peer environment. This means that in each Bluetooth unit, the authentication and encryption routines are implemented in the same way. Four different entities are used for maintaining security at the link layer: a public address which is unique for each user¹, two secret keys, and a random number which is different for each new transaction. The four entities and their sizes as used in Bluetooth are summarized in [Table 14.1](#).

Entity	Size
BD_ADDR	48 bits
Private user key, authentication	128 bits
Private user key, encryption configurable length (byte-wise)	8-128 bits
RAND	128 bits

Table 14.1: Entities used in authentication and encryption procedures.

The Bluetooth device address (BD_ADDR) is the 48-bit IEEE address which is unique for each Bluetooth unit. The Bluetooth addresses are publicly known, and can be obtained via MMI interactions, or, automatically, via an inquiry routine by a Bluetooth unit.

The secret keys are derived during initialization and are further never disclosed. Normally, the encryption key is derived from the authentication key during the authentication process. For the authentication algorithm, the size of the key used is always 128 bits. For the encryption algorithm, the key size may vary between 1 and 16 octets (8 - 128 bits). The size of the encryption key shall be configurable for two reasons. The first has to do with the many different requirements imposed on cryptographic algorithms in different countries – both w.r.t. export regulations and official attitudes towards privacy in general. The second reason is to facilitate a future upgrade path for the security without the need of a costly redesign of the algorithms and encryption hardware; increasing the effective key size is the simplest way to combat increased computing power at the opponent side. Currently (1999) it seems that an encryption key size of 64 bits gives satisfying protection for most applications.

The encryption key is entirely different from the authentication key (even though the latter is used when creating the former, as is described in [Section 14.5.4 on page 176](#)). Each time encryption is activated, a new encryption key

1. The BD_ADDR is not a secured identity.

shall be generated. Thus, the lifetime of the encryption key does not necessarily correspond to the lifetime of the authentication key.

It is anticipated that the authentication key will be more static in its nature than the encryption key – once established, the particular application running on the Bluetooth device decides when, or if, to change it. To underline the fundamental importance of the authentication key to a specific Bluetooth link, it will often be referred to as the link key.

The RAND is a random number which can be derived from a random or pseudo-random process in the Bluetooth unit. This is not a static parameter, it will change frequently.

In the remainder of this chapter, the terms user and application will be used interchangeably to designate the entity that is at the originating or receiving side.

14.1 RANDOM NUMBER GENERATION

Each Bluetooth unit has a random number generator. Random numbers are used for many purposes within the security functions – for instance, for the challenge-response scheme, for generating authentication and encryption keys, etc. Ideally, a true random generator based on some physical process with inherent randomness is used. Examples of such processes are thermal noise from a semiconductor or resistor and the frequency instability of a free running oscillator. For practical reasons, a software based solution with a pseudo-random generator is probably preferable. In general, it is quite difficult to classify the randomness of a pseudo-random sequence. Within Bluetooth, the requirements placed on the random numbers used are that they be non-repeating and randomly generated.

The expression ‘non-repeating’ means that it shall be highly unlikely that the value should repeat itself within the lifetime of the authentication key. For example, a non-repeating value could be the output of a counter that is unlikely to repeat during the lifetime of the authentication key, or a date/time stamp.

The expression ‘randomly generated’ means that it shall not be possible to predict its value with a chance that is significantly larger than 0 (e.g., greater than $1/2^L$ for a key length of L bits).

Clearly, the LM can use such a generator for various purposes; i.e. whenever a random number is needed (such as the RANDs, the unit keys, K_{init} , K_{master} , and random back-off or waiting intervals).

14.2 KEY MANAGEMENT

It is important that the encryption key size within a specific unit cannot be set by the user – this must be a factory preset entity. In order to prevent the user from over-riding the permitted key size, the Bluetooth baseband processing does not accept an encryption key given from higher software layers. Whenever a new encryption key is required, it must be created as defined in [Section 14.5.4 on page 176](#).

Changing a link key should also be done through the defined baseband procedures. Depending on what kind of link key it is, different approaches are required. The details are found in [Section 14.2.2.7 on page 156](#).

14.2.1 Key types

The link key is a 128-bit random number which is shared between two or more parties and is the base for all security transactions between these parties. The link key itself is used in the authentication routine. Moreover, the link key is used as one of the parameters when the encryption key is derived.

In the following, a session is defined as the time interval for which the unit is a member of a particular piconet. Thus, the session terminates when the unit disconnects from the piconet.

The link keys are either semi-permanent or temporary. A semi-permanent link key is stored in non-volatile memory and may be used after the current session is terminated. Consequently, once a semi-permanent link key is defined, it may be used in the authentication of several subsequent connections between the Bluetooth units sharing it. The designation semi-permanent is justified by the possibility to change it. How to do this is described in [Section 14.2.2.7 on page 156](#).

The lifetime of a temporary link key is limited by the lifetime of the current session – it cannot be reused in a later session. Typically, in a point-to-multipoint configuration where the same information is to be distributed securely to several recipients, a common encryption key is useful. To achieve this, a special link key (denoted master key) can temporarily replace the current link keys. The details of this procedure are found in [Section 14.2.2.6 on page 156](#).

In the sequel we sometimes refer to what is denoted as the current link key. This is simply the link key in use at the current moment. It can be semi-permanent or temporary. Thus, the current link key is used for all authentications and all generation of encryption keys in the on-going connection (session).

In order to accommodate for different types of applications, four types of link keys have been defined:

- the combination key K_{AB}
- the unit key K_A
- the temporary key K_{master}
- the initialization key K_{init}

In addition to these keys there is an encryption key, denoted K_C . This key is derived from the current link key. Whenever the encryption is activated by a LM command, the encryption key shall be changed automatically. The purpose of separating the authentication key and encryption key is to facilitate the use of a shorter encryption key without weakening the strength of the authentication procedure. There are no governmental restrictions on the strength of authentication algorithms. However, in some countries, such restrictions exist on the strength of encryption algorithms.

For a Bluetooth unit, the combination key K_{AB} and the unit key K_A are functionally indistinguishable; the difference is in the way they are generated. The unit key K_A is generated in, and therefore dependent on, a single unit A. The unit key is generated once at installation of the Bluetooth unit; thereafter, it is very rarely changed. The combination key is derived from information in both units A and B, and is therefore always dependent on two units. The combination key is derived for each new combination of two Bluetooth units.

It depends on the application or the device whether a unit key or a combination key is used. Bluetooth units which have little memory to store keys, or, when installed in equipment that must be accessible to a large group of users, will preferably use their own unit key. In that case, they only have to store a single key. Applications that require a higher security level preferably use the combination keys. These applications will require more memory since a combination key for each link to a different Bluetooth unit has to be stored.

The master key, K_{master} , is a link key only used during the current session. It will replace the original link key only temporarily. For example, this may be utilized when a master wants to reach more than two Bluetooth units simultaneously using the same encryption key, see [Section 14.2.2.6 on page 156](#).

The initialization key, K_{init} , is used as link key during the initialization process when no combination or unit keys have been defined and exchanged yet or when a link key has been lost. The initialization key protects the transfer of initialization parameters. The key is derived from a random number, an L-octet PIN code, and a BD_ADDR. This key is only to be used during initialization.

The PIN can be a fixed number provided with the Bluetooth unit (for example when there is no MMI as in a PSTN plug). Alternatively, the PIN can be selected arbitrarily by the user, and then entered in both units that have to be



matched. The latter procedure is used when both units have an MMI, for example a phone and a laptop. Entering a PIN in both units is more secure than using a fixed PIN in one of the units, and should be used whenever possible. Even if a fixed PIN is used, it shall be possible to change the PIN; this in order to prevent re-initialization by users who once got hold of the PIN. If no PIN is available, a default value of zero is to be used.

For many applications the PIN code will be a relatively short string of numbers. Typically, it may consist of only four decimal digits. Even though this gives sufficient security in many cases, there exist countless other, more sensitive, situations where this is not reliable enough. Therefore, the PIN code can be chosen to be any length from 1 to 16 octets. For the longer lengths, we envision the units exchanging PIN codes not through mechanical (i.e. human) interaction, but rather through means supported by software at the application layer. For example, this can be a Diffie-Hellman key agreement, where the exchanged key is passed on to the K_{init} generation process in both units, just as in the case of a shorter PIN code.

14.2.2 Key generation and initialization

The link keys have to be generated and distributed among the Bluetooth units in order to be used in the authentication procedure. Since the link keys must be secret, they cannot be obtained through an inquiry routine in the same way as the Bluetooth addresses. The exchange of the keys takes place during an initialization phase which has to be carried out separately for each two units that want to implement authentication and encryption. All initialization procedures consist of the following five parts:

- generation of an initialization key
- generation of link key
- link key exchange
- authentication
- generating of encryption key in each unit (optional)

After the initialization procedure, the units can proceed to communicate, or the link can be disconnected. If encryption is implemented, the E_0 algorithm is used with the proper encryption key derived from the current link key. For any new connection established between units A and B, they will use the common link key for authentication, instead of once more deriving K_{init} from the PIN. A new encryption key derived from that particular link key will be created next time encryption is activated.

If no link key is available, the LM shall automatically start an initialization procedure.

14.2.2.1 Generation of initialization key, K_{init}

A link key used temporarily during initialization is derived – the initialization key K_{init} . This key is derived by the E_{22} algorithm from a BD_ADDR, a PIN code, the length of the PIN (in octets), and a random number IN_RAND. The principle is depicted in [Figure 14.15 on page 176](#). The 128-bit output from E_{22} will be used for key exchange during the generation of a linkkey. When the units have performed the link key exchange, the initialization key shall be discarded.

When the initialization key is generated, the PIN is augmented with the BD_ADDR. If one unit has a fixed PIN the BD_ADDR of the other unit is used. If both units have a variable PIN the BD_ADDR of the device that received IN_RAND is used. If both units have a fixed PIN they cannot be paired. Since the maximum length of the PIN used in the algorithm cannot exceed 16 octets, it is possible that not all octets of BD_ADDR will be used. This procedure ensures that K_{init} depends on the identity of the unit with a variable PIN. A fraudulent Bluetooth unit may try to test a large number of PINs by each time claiming another BD_ADDR. It is the application's responsibility to take countermeasures against this threat. If the device address is kept fixed, the waiting interval until next try is permitted is increased exponentially (see [Section 14.4.1 on page 169](#)).

The details of the E_{22} algorithm can be found in [Section 14.5.3 on page 174](#).

14.2.2.2 Authentication

The authentication procedure is carried out as described in [Section 14.4 on page 168](#). Note that during each authentication, a new AU_RAND_A is issued.

Mutual authentication is achieved by first performing the authentication procedure in one direction and immediately followed by performing the authentication procedure in the opposite direction.

As a side effect of a successful authentication procedure an auxiliary parameter, the Authenticated Ciphering Offset (ACO), will be computed. The ACO is used for ciphering key generation as described in [Section 14.2.2.5 on page 155](#).

The claimant/verifier status is determined by the LM.

14.2.2.3 Generation of a unit key

A unit key K_A is generated when the Bluetooth unit is for the first time in operation; i.e. not during each initialization! The unit key is generated by the E_{21} algorithm as described in [Section 14.5.3 on page 174](#). Once created, the unit key is stored in non-volatile memory and (almost) never changed. If after initial-

ization the unit key is changed, the previously initialized units will possess a wrong link key. At initialization, the application has to determine which of the two parties will provide the unit key as link key. Typically, this will be the unit with restricted memory capabilities, since this unit only has to remember its own unit key. The unit key is transferred to the other party and then stored as link key for that particular party. So, for example in Figure 14.1 on page 154, the unit key of unit A, K_A , is being used as link key for the connection A-B; unit A sends the unit key K_A to unit B; unit B will store K_A as the link key K_{BA} . For another initialization, for example with unit C, unit A will reuse its unit key K_A , whereas unit C stores it as K_{CA}

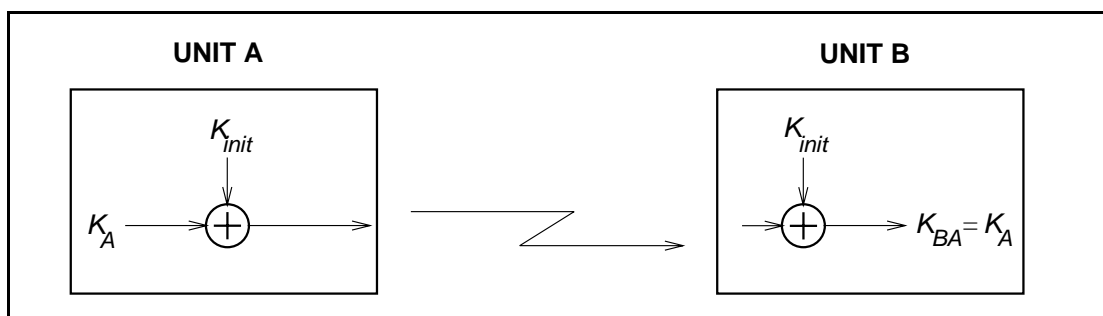


Figure 14.1: Generation of unit key. When the unit key has been exchanged, the initialization key shall be discarded in both units.

14.2.2.4 Generation of a combination key

If it is desired to use a combination key, this key is first generated during the initialization procedure. The combination key is the combination of two numbers generated in unit A and B, respectively. First, each unit generates a random number, say LK_RAND_A and LK_RAND_B . Then, utilizing E_{21} with the random number and the own BD_ADDR , the two random numbers

$$LK_K_A = E_{21}(LK_RAND_A, BD_ADDR_A), \tag{EQ 24}$$

and

$$LK_K_B = E_{21}(LK_RAND_B, BD_ADDR_B), \tag{EQ 25}$$

are created in unit A and unit B, respectively. These numbers constitute the units' contribution to the combination key that is to be created. Then, the two random numbers LK_RAND_A and LK_RAND_B are exchanged securely by XORing with the current link key, say K . Thus, unit A sends $K \oplus LK_RAND_A$ to unit B, while unit B sends $K \oplus LK_RAND_B$ to unit A. Clearly, if this is done during the initialization phase the link key $K = K_{init}$

When the random numbers LK_RAND_A and LK_RAND_B have been mutually exchanged, each unit recalculates the other units contribution to the combina-

tion key. This is possible since each unit knows the Bluetooth device address of the other unit. Thus, unit A calculates (EQ 25) on page 154 and unit B calculates (EQ 24) on page 154. After this, both units combine the two numbers to generate the 128-bit link key. The combining operation is a simple bitwise modulo-2 addition (i.e. XOR). The result is stored in unit A as the link key K_{AB} and in unit B as the link key K_{BA} . When both units have derived the new combination key, a mutual authentication procedure shall be initiated to confirm the success of the transaction. The old link key shall be discarded after a successful exchange of a new combination key. The message flow between master and slave and the principle for creating the combination key is depicted in Figure 14.2 on page 155.

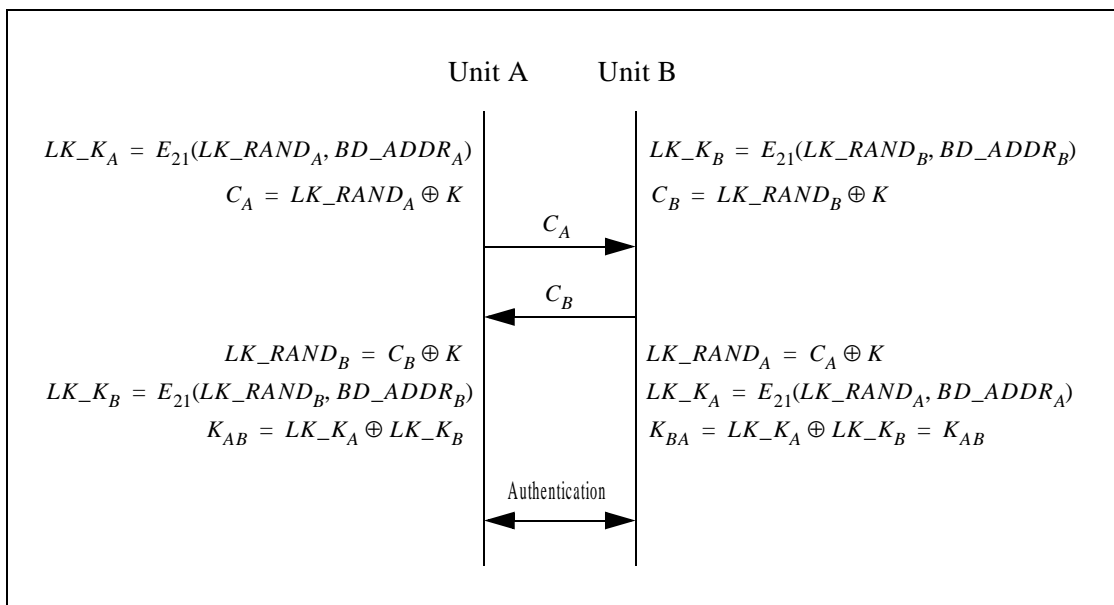


Figure 14.2: Generating a combination key. The old link key (K) shall be discarded after the exchange of a new combination key has succeeded

14.2.2.5 Generating the encryption key

The encryption key, K_C , is derived by algorithm E_3 from the current link key, a 96-bit Ciphering Offset number (COF), and a 128-bit random number. The COF is determined in one of two ways. If the current link key is a master key, then COF is derived from the master BD_ADDR. Otherwise the value of COF is set to the value of ACO as computed during the authentication procedure. More precisely, we have¹

$$\text{COF} = \begin{cases} \text{BD_ADDR} \cup \text{BD_ADDR}, & \text{if link key is a master key} \\ \text{ACO}, & \text{otherwise.} \end{cases} \quad (\text{EQ 26})$$

1. $x \cup y$ denotes the concatenation of the octet strings x and y .



There is an explicit call of E_3 when the LM activates encryption. Consequently, the encryption key is automatically changed each time the unit enters the encryption mode. The details of the key generating function E_3 can be found in [Section 14.5.4 on page 176](#).

14.2.2.6 Point-to-multipoint configuration

It is quite possible for the master to use separate encryption keys for each slave in a point-to-multipoint configuration with ciphering activated. Then, if the application requires more than one slave to listen to the same payload, each slave must be addressed individually. This may cause unwanted capacity loss for the piconet. Moreover, a Bluetooth unit (slave) is not capable of switching between two or more encryption keys in real time (e.g., after looking at the AM_ADDR in the header). Thus, the master cannot use different encryption keys for broadcast messages and individually addressed traffic. Alternatively, the master may tell several slave units to use a common link key (and, hence, indirectly also to use a common encryption key) and broadcast the information encrypted. For many applications, this key is only of temporary interest. In the sequel, this key is denoted K_{master} .

The transfer of necessary parameters is protected by the routine described in [Section 14.2.2.8 on page 157](#). After the confirmation of successful reception in each slave, the master shall issue a command to the slaves to replace their respective current link key by the new (temporary) master key. Before encryption can be activated, the master also has to generate and distribute a common EN_RANDOM to all participating slaves. Using this random number and the newly derived master key, each slave generates a new encryption key.

Note that the master must negotiate what encryption key length to use individually with each slave who wants to use the master key. In case the master already has negotiated with some of these slaves, it has knowledge of what sizes can be accepted. Clearly, there might be situations where the permitted key lengths of some units are incompatible. In that case, the master must have the limiting unit excluded from the group.

When all slaves have received the necessary data, the master can communicate information on the piconet securely using the encryption key derived from the new temporary link key. Clearly, each slave in possession of the master key can eavesdrop on all encrypted traffic, not only the traffic intended for itself. If so desired, the master can tell all participants to fall back to their old link keys simultaneously.

14.2.2.7 Modifying the link keys

In certain circumstances, it is desirable to be able to modify the link keys. A link key based on a unit key can be changed, but not very easily. The unit key is created once during the first use. Changing the unit key is a less desirable

alternative, as several units may share the same unit key as link key (think of a printer whose unit key is distributed among all users using the printer's unit key as link key). Changing the unit key will require re-initialization of all units trying to connect. In certain cases, this might be desirable; for example to deny access to previously allowed units.

If the key change concerns combination keys, then the procedure is rather straightforward. The change procedure is identical to the procedure illustrated in [Figure 14.2 on page 155](#), using the current value of the combination key as link key. This procedure can be carried out at any time after the authentication and encryption start. In fact, since the combination key corresponds to a single link, it can be modified each time this link is established. This will improve the security of the system since then old keys lose their validity after each session.

Of course, starting up an entirely new initialization procedure is also a possibility. In that case, user interaction is necessary since a PIN is required in the authentication and encryption procedures.

14.2.2.8 Generating a master key

The key-change routines described so far are semi-permanent. To create the master link key, which can replace the current link key during an initiated session (see [Section 14.2.2.6 on page 156](#)), other means are needed. First, the master creates a new link key from two 128-bit random numbers, RAND1 and RAND2. This is done by

$$K_{master} = E_{22}(\text{RAND1}, \text{RAND2}, 16). \quad (\text{EQ 27})$$

Clearly, this key is a 128-bit random number. The reason to use the output of E_{22} and not directly chose a random number as the key, is to avoid possible problems with degraded randomness due to a poor implementation of the random number generator within the Bluetooth unit.

Then, a third random number, say RAND, is transmitted to the slave. Using E_{22} with the current link key and RAND as inputs, both the master and slave computes a 128-bit overlay. The master sends the bitwise XOR of the overlay and the new link key to the slave. The slave, who knows the overlay, recalculates K_{master} . To confirm the success of this transaction, the units shall perform a mutual authentication procedure using the new link key. This procedure is then repeated for each slave who shall receive the new link key. The ACO values from the involved authentications should not replace the current existing ACO as this ACO is needed to (re)compute a ciphering key when the master wants to fall back to the previous link (non-temporary) key.

When so required – and potentially long after the actual distribution of the master key – the master activates encryption by an LM command. Before doing that, the master must ensure that all slaves receive the same random number,

say EN_RAND, since the encryption key is derived through the means of E_3 individually in all participating units. Then, each slave computes a new encryption key,

$$K_C = E_3(K_{master}, EN_RAND, COF), \tag{EQ 28}$$

where the value of COF is derived from the master's BD_ADDR as specified by equation (EQ 26) on page 155. The details on the encryption key generating function can be found in Section 14.5.4 on page 176. The principle of the message flow between the master and slave when generating the master key is depicted in Figure 14.3. Note that in this case the ACO produced during the authentication is not used when computing the ciphering key.

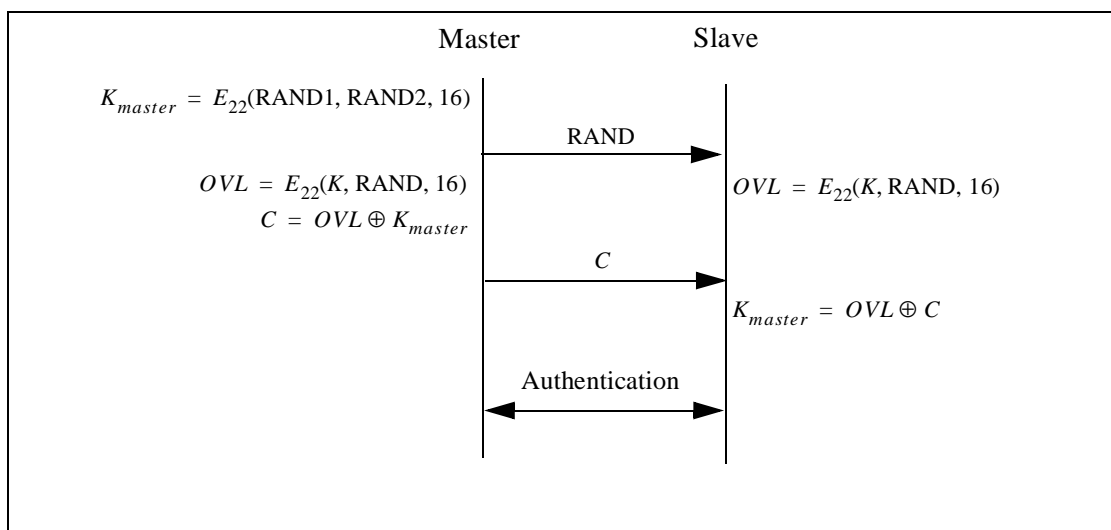


Figure 14.3: Master link key distribution and computation of the corresponding encryption key.

14.3 ENCRYPTION

User information can be protected by encryption of the packet payload; the access code and the packet header are never encrypted. The encryption of the payloads is carried out with a stream cipher called E_0 that is re-synchronized for every payload. The overall principle is shown in Figure 14.4 on page 159.

The stream cipher system E_0 consists of three parts. One part performs the initialization (generation of the payload key), the second part generates the key stream bits, and the third part performs the encryption and decryption. The payload key generator is very simple – it merely combines the input bits in an appropriate order and shift them into the four LFSRs used in the key stream generator. The main part of the cipher system is the second, as it also will be used for the initialization. The key stream bits are generated by a method derived from the summation stream cipher generator attributable to Massey and Rueppel. The method has been thoroughly investigated, and there exist good estimates of its strength with respect to presently known methods for cryptanalysis. Although the summation generator has weaknesses that can be

used in so-called correlation attacks, the high re-synchronization frequency will disrupt such attacks.

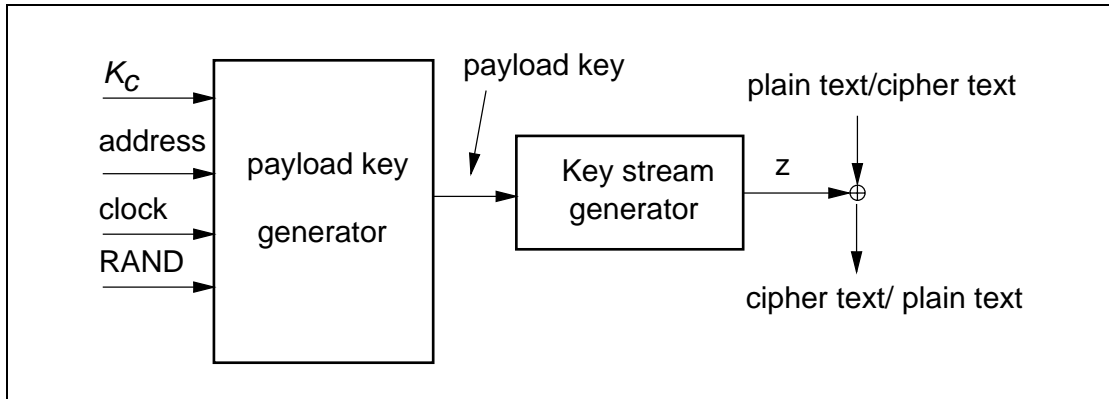


Figure 14.4: Stream ciphering for Bluetooth with E_0 .

14.3.1 Encryption key size negotiation

Each Bluetooth device implementing the baseband specification needs a parameter defining the maximal allowed key length, L_{max} , $1 \leq L_{max} \leq 16$ (number of octets in the key). For each application, a number L_{min} is defined indicating the smallest acceptable key size for that particular application. Before generating the encryption key, the involved units must negotiate to decide what key size to actually use.

The master sends a suggested value, $L_{sug}^{(M)}$, to the slave. Initially, the suggested value is set $L_{sug}^{(M)} = L_{max}^{(M)}$. If $L_{sug}^{(M)} \leq L_{min}^{(S)}$, and the slave supports the suggested length, the slave acknowledges and this value will be the length of the encryption key for this link. However, if both conditions are not fulfilled, the slave sends a new proposal, $L_{sug}^{(S)}$, to the master. This value should be the largest among all supported lengths less than the previous master suggestion. Then, the master performs the corresponding test on the slave suggestion. This procedure is repeated until a key length agreement is reached, or, one unit aborts the negotiation. An abortion may be caused by lack of support for $L_{sug}^{(M)}$ and all smaller key lengths, or if $L_{sug}^{(S)} < L_{min}^{(M)}$. In case of abortion Bluetooth link encryption can not be employed.

The possibility of a failure in setting up a secure link is an unavoidable consequence of letting the application decide whether to accept or reject a suggested key size. However, this is a necessary precaution. Otherwise a fraudulent unit could enforce a weak protection on a link by claiming a small maximum key size.



14.3.2 Encryption modes

If a slave has a semi-permanent link key (i.e. a combination key or a unit key), it can only accept encryption on slots individually addressed to itself (and, of course, in the reverse direction to the master). In particular, it will assume that broadcast messages are not encrypted. The possible traffic modes are described in Table 14.2. When an entry in the table refers to a link key, it means that the encryption/decryption engine uses the encryption key derived from that link key.

Broadcast traffic	Individually addressed traffic
No encryption	No encryption
No encryption	Encryption, Semi-permanent link key

Table 14.2: Possible traffic modes for a slave using a semi-permanent link key.

If the slave has received a master key, there are three possible combinations as defined in Table 14.3 on page 160. In this case, all units in the piconet uses a common link key, K_{master} . Since the master uses encryption keys derived from this link key for all secure traffic on the piconet, it is possible to avoid ambiguity in the participating slaves on which encryption key to use. Also in this case the default mode is that broadcast messages are not encrypted. A specific LM-command is required to activate encryption – both for broadcast and for individually addressed traffic.

Broadcast traffic	Individually addressed traffic
No encryption	No encryption
No encryption	Encryption, K_{master}
Encryption, K_{master}	Encryption, K_{master}

Table 14.3: Possible encryption modes for a slave in possession of a master key.

The master can issue an LM-command to the slaves telling them to fall back to their previous semi-permanent link key. Then, regardless of the previous mode they were in, they will end up in the first row of Table 14.2 on page 160; i.e. no encryption.

14.3.3 Encryption concept

For the encryption routine, a stream cipher algorithm will be used in which ciphering bits are bit-wise modulo-2 added to the data stream to be sent over the air interface. The payload is ciphered after the CRC bits are appended, but, prior to the FEC encoding.

Each packet payload is ciphered separately. The cipher algorithm E_0 uses the master Bluetooth address, 26 bits of the master real-time clock (CLK_{26-1}) and the encryption key K_C as input, see [Figure 14.5 on page 161](#) (where it is assumed that unit A is the master).

The encryption key K_C is derived from the current link key, COF, and a random number, EN_RAND_A (see [Section 14.5.4 on page 176](#)). The random number is issued by the master before entering encryption mode. Note that EN_RAND_A is publicly known since it is transmitted as plain text over the air.

Within the E_0 algorithm, the encryption key K_C is modified into another key denoted K'_C . The maximum effective size of this key is factory preset and may be set to any multiple of eight between one and sixteen (8-128 bits). The procedure for deriving the key is described in [Section 14.3.5 on page 164](#).

The real-time clock is incremented for each slot. The E_0 algorithm is re-initialized at the start of each new packet (i.e. for Master-to-Slave as well as for Slave-to-Master transmission). By using CLK_{26-1} at least one bit is changed between two transmissions. Thus, a new keystream is generated after each re-initialization. For packets covering more than a single slot, the Bluetooth clock as found in the first slot is being used for the entire packet.

The encryption algorithm E_0 generates a binary keystream, K_{cipher} , which is modulo-2 added to the data to be encrypted. The cipher is symmetric; decryption is performed in exactly the same way using the same key as used for encryption.

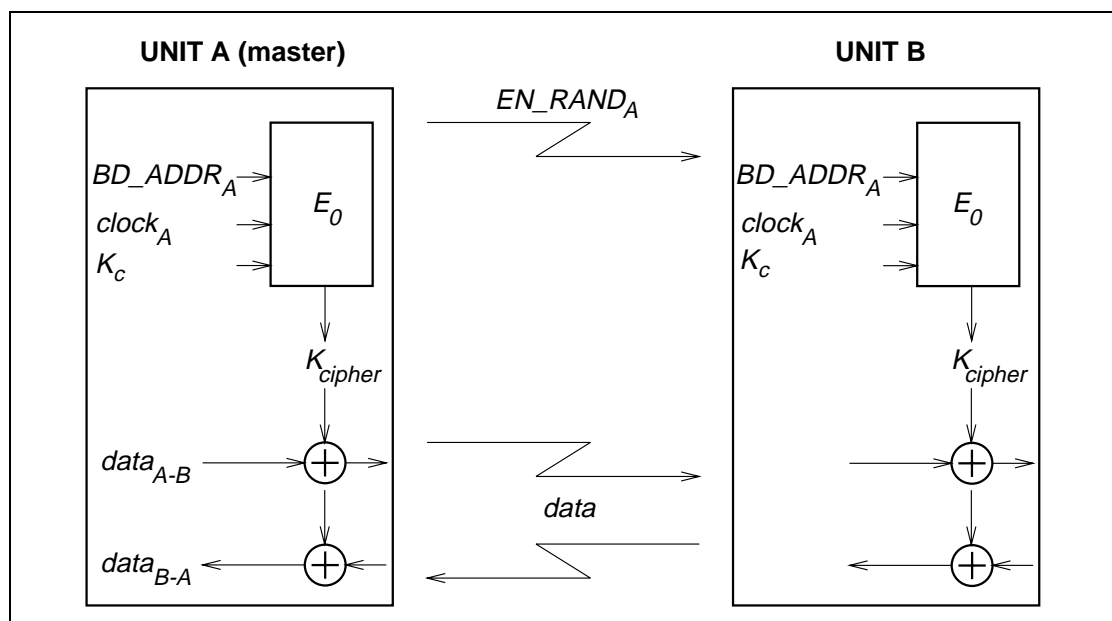


Figure 14.5: Functional description of the encryption procedure

14.3.4 Encryption algorithm

The system uses linear feedback shift registers (LFSRs) whose output is combined by a simple finite state machine (called the summation combiner) with 16 states. The output of this state machine is the key stream sequence, or, during initialization phase, the randomized initial start value. The algorithm is presented with an encryption key K_C , an 48-bit Bluetooth address, the master clock bits CLK_{26-1} , and a 128-bit RAND value. Figure 14.6 on page 162 shows the setup.

There are four LFSRs ($LFSR_1, \dots, LFSR_4$) of lengths $L_1 = 25$, $L_2 = 31$, $L_3 = 33$, and, $L_4 = 39$, with feedback polynomials as specified in Table 14.4 on page 163. The total length of the registers is 128. These polynomials are all primitive. The Hamming weight of all the feedback polynomials is chosen to be five – a reasonable trade-off between reducing the number of required XOR gates in the hardware realization and obtaining good statistical properties of the generated sequences.

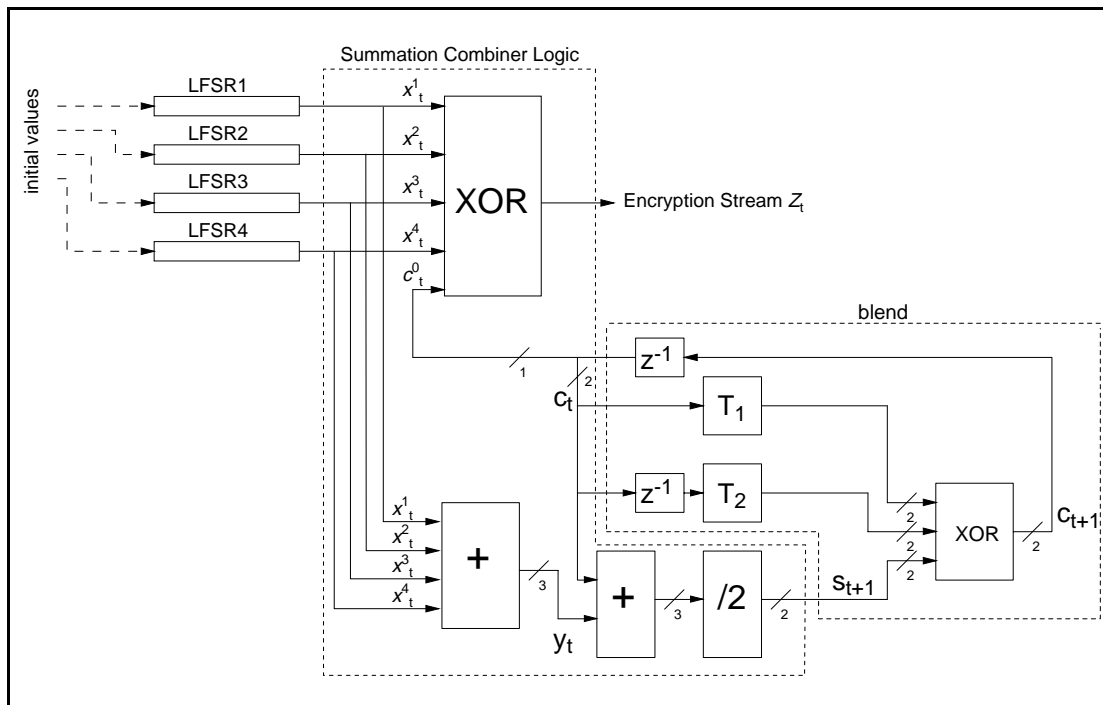


Figure 14.6: Concept of the encryption engine.

i	L_i	feedback $f_i(t)$	weight
1	25	$t^{25} + t^{20} + t^{12} + t^8 + 1$	5
2	31	$t^{31} + t^{24} + t^{16} + t^{12} + 1$	5
3	33	$t^{33} + t^{28} + t^{24} + t^4 + 1$	5
4	39	$t^{39} + t^{36} + t^{28} + t^4 + 1$	5

Table 14.4: The four primitive feedback polynomials.

Let x_t^i denote the t^{th} symbol of LSFR _{i} . From the four-tuple x_t^1, \dots, x_t^4 we derive the value y_t as

$$y_t = \sum_{i=1}^4 x_t^i, \tag{EQ 29}$$

where the sum is over the integers. Thus y_t can take the values 0,1,2,3, or 4. The output of the summation generator is now given by the following equations

$$z_t = x_t^1 \oplus x_t^2 \oplus x_t^3 \oplus x_t^4 \oplus c_t^0 \in \{0, 1\}, \tag{EQ 30}$$

$$s_{t+1} = (s_{t+1}^1, s_{t+1}^0) = \left\lfloor \frac{y_t + c_t}{2} \right\rfloor \in \{0, 1, 2, 3\}, \tag{EQ 31}$$

$$c_{t+1} = (c_{t+1}^1, c_{t+1}^0) = s_{t+1} \oplus T_1[c_t] \oplus T_2[c_{t-1}], \tag{EQ 32}$$

where $T_1[.]$ and $T_2[.]$ are two different linear bijections over GF(4). Suppose GF(4) is generated by the irreducible polynomial $x^2 + x + 1$, and let α be a zero of this polynomial in GF(4). The mappings T_1 and T_2 are now defined as

$$T_1: \text{GF}(4) \rightarrow \text{GF}(4)$$

$$x \mapsto x$$

$$T_2: \text{GF}(4) \rightarrow \text{GF}(4)$$

$$x \mapsto (\alpha + 1)x.$$

We can write the elements of GF(4) as binary vectors. This is summarized in [Table 14.5](#).

Since the mappings are linear, we can realize them using XOR gates; i.e.

x	$T_1[x]$	$T_2[x]$
00	00	00
01	01	11
10	10	01
11	11	10

Table 14.5: The mappings T_1 and T_2 .

$$T_1: (x_1, x_0) \mapsto (x_1, x_0),$$

$$T_2: (x_1, x_0) \mapsto (x_0, x_1 \oplus x_0).$$

14.3.4.1 The operation of the cipher

Figure 14.7 on page 164 gives an overview of the operation in time. The encryption algorithm shall run through the initialization phase before the start of transmitting or receiving a new packet. Thus, for multislot packets the cipher is initialized using the clock value of the first slot in the multislot sequence.

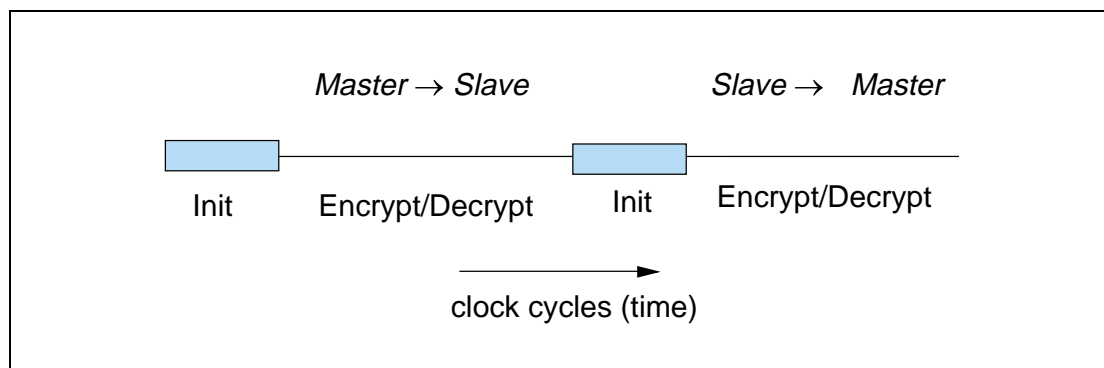


Figure 14.7: Overview of the operation of the encryption engine. Between each start of a packet (TX or RX), the LFSRs are re-initialized.

14.3.5 LFSR initialization

The key stream generator needs to be loaded with an initial value for the four LFSRs (in total 128 bits) and the 4 bits that specify the values of c_0 and c_{-1} . The 132 bit initial value is derived from four inputs by using the key stream generator itself. The input parameters are the key K_C , a 128-bit random number RAND, a 48-bit Bluetooth address, and the 26 master clock bits CLK_{26-1} .

The effective length of the encryption key can vary between 8 and 128 bits. Note that the actual key length as obtained from E_3 is 128 bits. Then, within E_0 , the key length is reduced by a modulo operation between K_C and a polynomial of desired degree. After reduction, the result is encoded with a block code in

order to distribute the starting states more uniformly. The operation is defined in (EQ 33) on page 165.

When the encryption key has been created the LFSRs are loaded with their initial values. Then, 200 stream cipher bits are created by operating the generator. Of these bits, the last 128 are fed back into the key stream generator as an initial value of the four LFSRs. The values of c_t and c_{t-1} are kept. From this point on, when clocked the generator produces the encryption (decryption) sequence which is bitwise XORed to the transmitted (received) payload data.

In the following, we will denote octet i of a binary sequence X by the notation $X[i]$. We define bit 0 of X to be the LSB. Then, the LSB of $X[i]$ corresponds to bit $8i$ of the sequence X the MSB of $X[i]$ is bit $8i + 7$ of X . For instance, bit 24 of the Bluetooth address is the LSB of $ADR[3]$.

The details of the initialization are as follows:

1. Create the encryption key to use from the 128-bit secret key K and the 128-bit publicly known EN_RAND . Let $L, 1 \leq L \leq 16$, be the effective key length in number of octets. The resulting encryption key will be denoted K'_C :

$$K'_C(x) = g_2^{(L)}(x)(K_C(x) \bmod g_1^{(L)}(x)), \tag{EQ 33}$$

where $\deg(g_1^{(L)}(x)) = 8L$ and $\deg(g_2^{(L)}(x)) \leq 128 - 8L$. The polynomials are defined in Table 14.6.

2. Shift in the 3 inputs K'_C , the Bluetooth address, the clock, and the six-bit constant 111001 into the LFSRs. In total 208 bits are shifted in.
 - a) Open all switches shown in Figure 14.8 on page 167;
 - b) Arrange inputs bits as shown in Figure 14.8; Set the content of all shift register elements to zero. Set $t = 0$.
 - c) Start shifting bits into the LFSRs. The rightmost bit at each level of Figure 14.8 is the first bit to enter the corresponding LFSR.
 - d) When the first input bit at level i reaches the rightmost position of $LFSR_i$, close the switch of this LFSR.
 - e) At $t = 39$ (when the switch of $LFSR_4$ is closed), reset both blend registers $c_{39} = c_{39-1} = 0$; Up to this point, the content of c_t and c_{t-1} has been of no concern. However, from this moment forward their content will be used in computing the output sequence.
 - f) From now on output symbols are generated. The remaining input bits are continuously shifted into their corresponding shift register. When the last bit has been shifted in, the shift register is clocked with input = 0;

Note: When finished, $LFSR_1$ has effectively clocked 30 times with feedback closed, $LFSR_2$ has clocked 24 times, $LFSR_3$ has



clocked 22 times, and LFSR₄ has effectively clocked 16 times with feedback closed.

3. To mix initial data, continue to clock until 200 symbols have been produced with all switches closed $t = 239$ ();
4. Keep blend registers and c_{t-1} make a parallel load of the last 128 generated bits into the LFSRs according to Figure 14.9 at $t = 240$;

After the parallel load in item 4, the blend register contents will be updated for each subsequent clock.

L	deg	$g_1^{(L)}$	deg	$g_2^{(L)}$
1	[8]	00000000 00000000 00000000 0000011d	[119]	00e275a0 abd218d4 cf928b9b bf6cb08f
2	[16]	00000000 00000000 00000000 0001003f	[112]	0001e3f6 3d7659b3 7f18c258 cff6efef
3	[24]	00000000 00000000 00000000 010000db	[104]	000001be f66c6c3a b1030a5a 1919808b
4	[32]	00000000 00000000 00000001 000000af	[96]	00000001 6ab89969 de17467f d3736ad9
5	[40]	00000000 00000000 00000100 00000039	[88]	00000000 01630632 91da50ec 55715247
6	[48]	00000000 00000000 00010000 00000291	[77]	00000000 00002c93 52aa6cc0 54468311
7	[56]	00000000 00000000 01000000 00000095	[71]	00000000 000000b3 f7fffce2 79f3a073
8	[64]	00000000 00000001 00000000 0000001b	[63]	00000000 00000000 a1ab815b c7ec8025
9	[72]	00000000 00000100 00000000 00000609	[49]	00000000 00000000 0002c980 11d8b04d
10	[80]	00000000 00010000 00000000 00000215	[42]	00000000 00000000 0000058e 24f9a4bb
11	[88]	00000000 01000000 00000000 0000013b	[35]	00000000 00000000 0000000c a76024d7
12	[96]	00000001 00000000 00000000 000000dd	[28]	00000000 00000000 00000000 1c9c26b9
13	[104]	00000100 00000000 00000000 0000049d	[21]	00000000 00000000 00000000 0026d9e3
14	[112]	00010000 00000000 00000000 0000014f	[14]	00000000 00000000 00000000 00004377
15	[120]	01000000 00000000 00000000 000000e7	[7]	00000000 00000000 00000000 00000089
16	[128]	1 00000000 00000000 00000000 00000000	[0]	00000000 00000000 00000000 00000001

Table 14.6: Polynomials used when creating K'_C .

All polynomials are in hexadecimal notation. The LSB is in the rightmost position.

In Figure 14.8, all bits are shifted into the LFSRs, starting with the least significant bit (LSB). For instance, from the third octet of the address, ADR[2], first ADR₁₆ is entered, followed by ADR₁₇, etc. Furthermore, CL₀ corresponds to CLK_{1,...}, CL₂₅ corresponds to CLK₂₆.

Note that the output symbols $x_p^i, i = 1, \dots, 4$ are taken from the positions 24, 24, 32, and 32 for LFSR₁, LFSR₂, LFSR₃, and LFSR₄, respectively (counting the leftmost position as number 1).

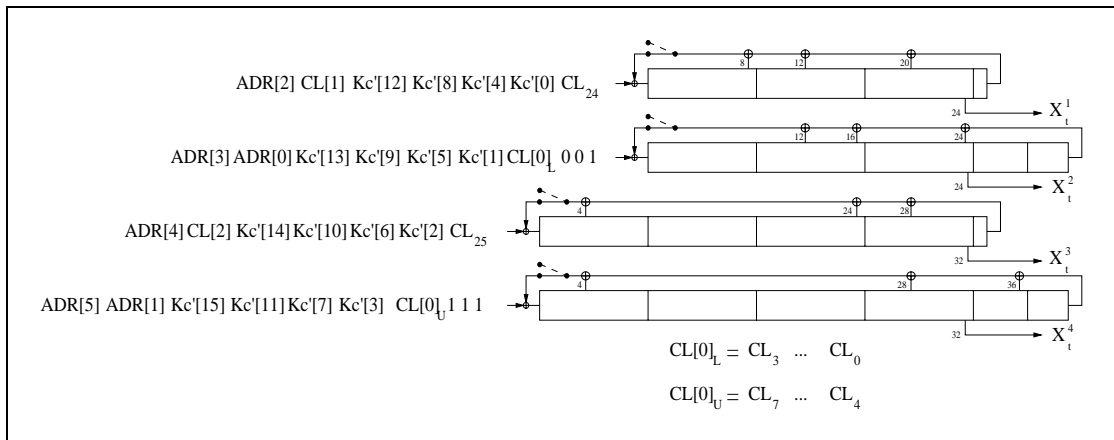


Figure 14.8: Arranging the input to the LFSRs.

In Figure 14.9, the 128 binary output symbols Z_0, \dots, Z_{127} are arranged in octets denoted $Z[0], \dots, Z[15]$. The LSB of $Z[0]$ corresponds to the first of these symbols, the MSB of $Z[15]$ is the latest output from the generator. These bits shall be loaded into the LFSRs according to the figure. It is a parallel load and no update of the blend registers is done. The first output symbol is generated at the same time. The octets are written into the registers with the LSB in the left-most position (i.e. the opposite of before). For example, Z_{24} is loaded into position 1 of LFSR₄.

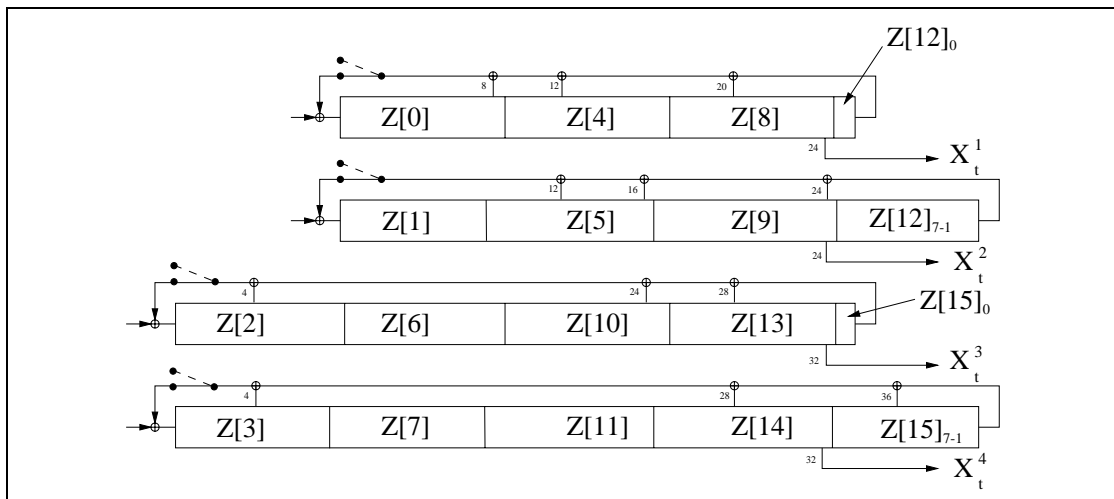


Figure 14.9: Distribution of the 128 last generated output symbols within the LFSRs.

14.3.6 Key stream sequence

When the initialization is finished, the output from the summation combiner is used for encryption/decryption. The first bit to use is the one produced at the parallel load, i.e. at $t = 240$. The circuit is run for the entire length of the current payload. Then, before the reverse direction is started, the entire initialization process is repeated with updated values on the input parameters.

Sample data of the encryption output sequence can be found in “Appendix IV” on page 923, Encryption Sample Data. A necessary, but not sufficient, condition for all Bluetooth-compliant implementations is to produce these encryption streams for identical initialization values.

14.4 AUTHENTICATION

The entity authentication used in Bluetooth uses a challenge-response scheme in which a claimant’s knowledge of a secret key is checked through a 2-move protocol using symmetric secret keys. The latter implies that a correct claimant/verifier pair share the same secret key, for example K . In the challenge-response scheme the verifier challenges the claimant to authenticate a random input (the challenge), denoted by AU_RAND_A , with an authentication code, denoted by E_1 , and return the result $SRES$ to the verifier, see Figure 14.10 on page 168. This figure shows also that in Bluetooth the input to E_1 consists of the tuple AU_RAND_A and the Bluetooth device address (BD_ADDR) of the claimant. The use of this address prevents a simple reflection attack¹. The secret K shared by units A and B is the current link key.

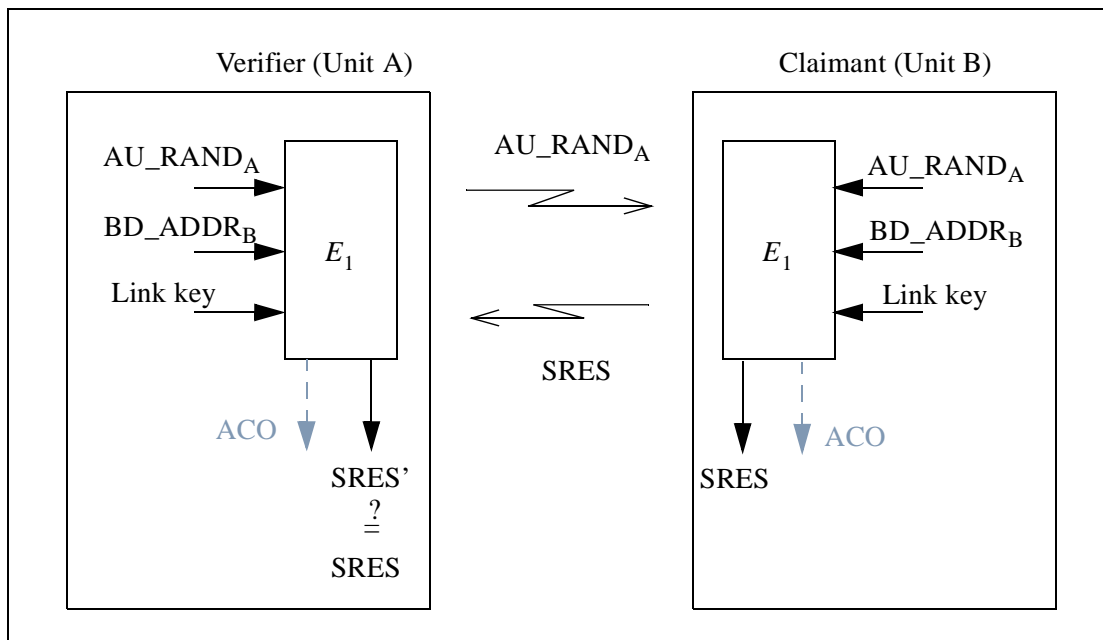


Figure 14.10: Challenge-response for the Bluetooth.

The challenge-response scheme for symmetric keys used in the Bluetooth is depicted in Figure 14.11 on page 169.

1. The reflection attack actually forms no threat in Bluetooth because all service requests are dealt with on a FIFO bases. When præemption is introduced, this attack is potentially dangerous.

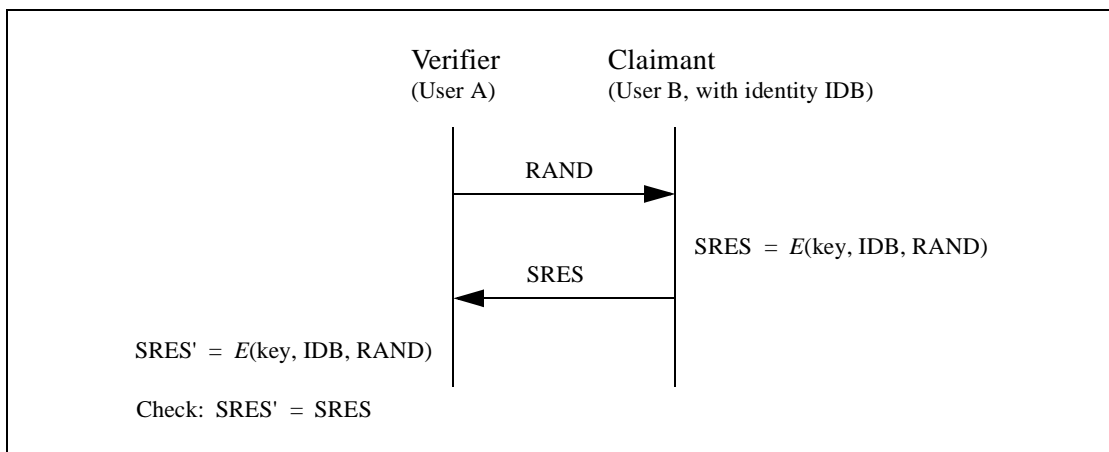


Figure 14.11: Challenge-response for symmetric key systems.

In the Bluetooth, the verifier is not necessarily the master. The application indicates who has to be authenticated by whom. Certain applications only require a one-way authentication. However, in some peer-to-peer communications, one might prefer a mutual authentication in which each unit is subsequently the challenger (verifier) in two authentication procedures. The LM coordinates the indicated authentication preferences by the application to determine in which direction(s) the authentication(s) has to take place. For mutual authentication with the units of [Figure 14.10 on page 168](#), after unit A has successfully authenticated unit B, unit B could authenticate unit A by sending a AU_RAND_B (different from the AU_RAND_A that unit A issued) to unit A, and deriving the SRES and SRES' from the new AU_RAND_B , the address of unit A, and the link key K_{AB} .

If an authentication is successful the value of ACO as produced by E_1 should be retained.

14.4.1 Repeated attempts

When the authentication attempt fails, a certain waiting interval must pass before the verifier will initiate a new authentication attempt to the same claimant, or before it will respond to an authentication attempt initiated by a unit claiming the same identity as the suspicious unit. For each subsequent authentication failure with the same Bluetooth address, the waiting interval shall be increased exponentially. That is, after each failure, the waiting interval before a new attempt can be made, for example, twice as long as the waiting interval prior to the previous attempt¹. The waiting interval shall be limited to a maximum. The maximum waiting interval depends on the implementation. The waiting time shall exponentially decrease to a minimum when no new failed attempts are being made during a certain time period. This procedure prevents

1. An other appropriate value larger than 1 may be used.

an intruder to repeat the authentication procedure with a large number of different keys.

To make the system somewhat less vulnerable to denial-of-service attacks, the Bluetooth units should keep a list of individual waiting intervals for each unit it has established contact with. Clearly, the size of this list must be restricted only to contain the N units with which the most recent contact has been made. The number N can vary for different units depending on available memory size and user environment.

14.5 THE AUTHENTICATION AND KEY-GENERATING FUNCTIONS

This section describes the algorithmic means for supporting the Bluetooth security requirements on authentication and key generation.

14.5.1 The authentication function E_1

The authentication function proposed for the Bluetooth is a computationally secure authentication code, or often called a MAC. E_1 uses the encryption function called SAFER+. The algorithm is an enhanced version¹ of an existing 64-bit block cipher SAFER-SK128, and it is freely available. In the sequel the block cipher will be denoted as the function A_r , which maps under a 128-bit key, a 128-bit input to a 128-bit output, i.e.

$$A_r: \{0, 1\}^{128} \times \{0, 1\}^{128} \rightarrow \{0, 1\}^{128} \quad (\text{EQ 34})$$

$$(k \times x) \mapsto t.$$

The details of A_r are given in the next section. The function E_1 is constructed using A_r as follows

$$E_1: \{0, 1\}^{128} \times \{0, 1\}^{128} \times \{0, 1\}^{48} \rightarrow \{0, 1\}^{32} \times \{0, 1\}^{96} \quad (\text{EQ 35})$$

$$(K, \text{RAND}, \text{address}) \mapsto (\text{SRES}, \text{ACO}),$$

where $\text{SRES} = \text{Hash}(K, \text{RAND}, \text{address}, 6)[0, \dots, 3]$, where *Hash* is a keyed hash function defined as²,

$$\text{Hash}: \{0, 1\}^{128} \times \{0, 1\}^{128} \times \{0, 1\}^{8 \times L} \times \{6, 12\} \rightarrow \{0, 1\}^{128} \quad (\text{EQ 36})$$

$$(K, I_1, I_2, L) \mapsto A'_r([\tilde{K}], [E(I_2, L) +_{16} (A_r(K, I_1) \oplus_{16} I_1)]),$$

1. It is presently one of the contenders for the Advanced Encryption Standard (AES) submitted by Cylink, Corp, Sunnyvale, USA
2. The operator $+_{16}$ denotes bitwise addition mod 256 of the 16 octets, and the operator \oplus_{16} denotes bitwise XORing of the 16 octets.

and where

$$E: \{0, 1\}^{8 \times L} \times \{6, 12\} \rightarrow \{0, 1\}^{8 \times 16} \quad (\text{EQ 37})$$

$$(X[0, \dots, L-1], L) \mapsto (X[i(\text{mod } L)] \text{ for } i = 0 \dots 15),$$

is an expansion of the L octet word X into a 128-bit word. Thus we see that we have to evaluate the function A_r twice for each evaluation of E_1 . The key \tilde{K} for the second use of A_r (actually A'_r) is offseted from K as follows¹

$$\begin{aligned} \tilde{K}[0] &= (K[0] + 233) \pmod{256} & \tilde{K}[1] &= K[1] \oplus 229, \\ \tilde{K}[2] &= (K[2] + 223) \pmod{256} & \tilde{K}[3] &= K[3] \oplus 193, \\ \tilde{K}[4] &= (K[4] + 179) \pmod{256} & \tilde{K}[5] &= K[5] \oplus 167, \\ \tilde{K}[6] &= (K[6] + 149) \pmod{256} & \tilde{K}[7] &= K[7] \oplus 131, \\ \tilde{K}[8] &= K[8] \oplus 233, & \tilde{K}[9] &= (K[9] + 229) \pmod{256}, \\ \tilde{K}[10] &= K[10] \oplus 223, & \tilde{K}[11] &= (K[11] + 193) \pmod{256}, \\ \tilde{K}[12] &= K[12] \oplus 179, & \tilde{K}[13] &= (K[13] + 167) \pmod{256}, \\ \tilde{K}[14] &= K[14] \oplus 149, & \tilde{K}[15] &= (K[15] + 131) \pmod{256}. \end{aligned} \quad (\text{EQ 38})$$

A data flowchart of the computation of E_1 is depicted in [Figure 14.12 on page 172](#). E_1 is also used to deliver the parameter ACO (Authenticated Ciphering Offset) that is used in the generation of the ciphering key by E_3 see equations [\(EQ 26\) on page 155](#) and [\(EQ 46\) on page 176](#). The value of ACO is formed by the octets 4 through 15 of the output of the hash function defined in [\(EQ 36\) on page 170](#), i.e.

$$\text{ACO} = \text{Hash}(K, \text{RAND}, \text{address}, 6)[4, \dots, 15]. \quad (\text{EQ 39})$$

1. The constants are the first largest primes below 257 for which 10 is a primitive root.

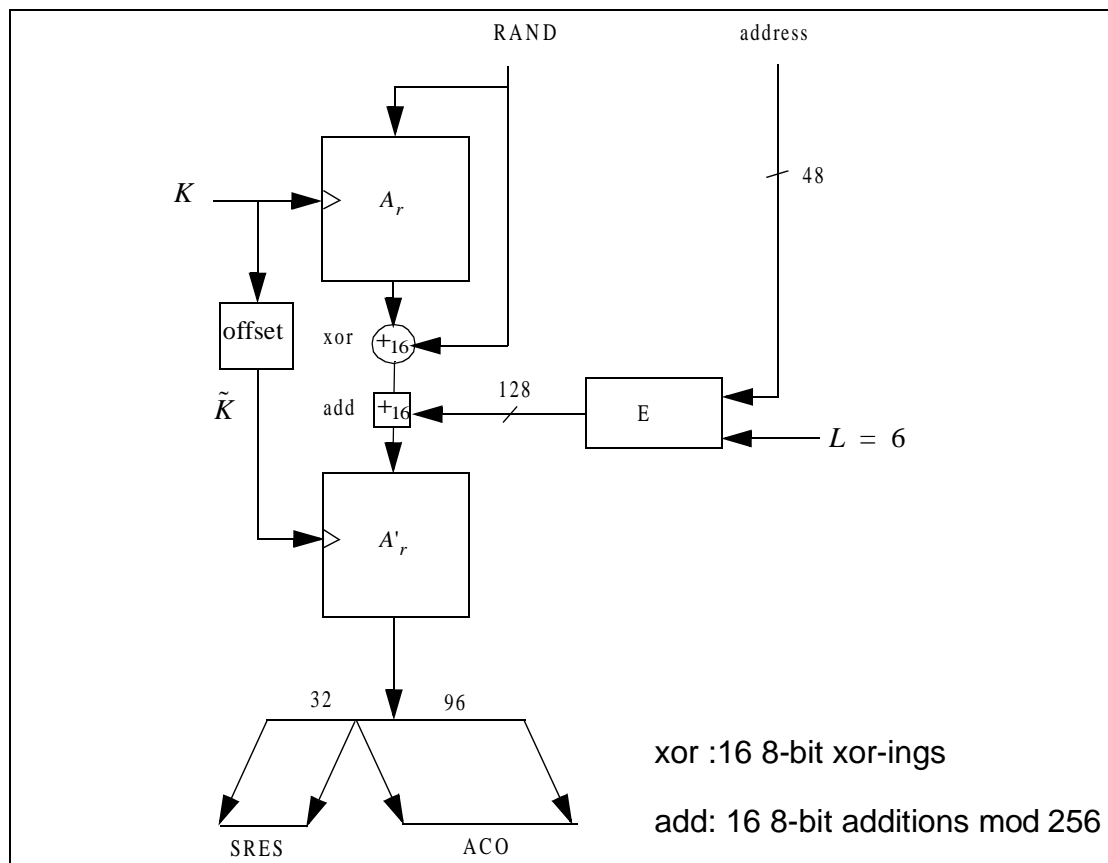


Figure 14.12: Flow of data for the computation of E_1 .

14.5.2 The functions A_r and A'_r

The function A_r is identical to SAFER+. It consists of a set of 8 layers, (each layer is called a round) and a parallel mechanism for generating the sub keys $K_p[j]$, $p = 1, 2, \dots, 17$, the so-called round keys to be used in each round. The function will produce a 128-bit result from a 128-bit “random” input string and a 128-bit “key”. Besides the function A_r , a slightly modified version referred to as A'_r is used in which the input of round 1 is added to the input of the 3rd round. This is done to make the modified version non-invertible and prevents the use of A'_r (especially in E_{2x}) as an encryption function. See [Figure 14.13 on page 173](#) for details.

14.5.2.1 The round computations

The computations in each round are a composition of encryption with a round key, substitution, encryption with the next round key, and, finally, a Pseudo Hadamard Transform (PHT). The computations in a round are shown in [Figure 14.13 on page 173](#). The sub keys for round r , $r = 1, 2, \dots, 8$ are denoted

$K_{2r-1}[j], K_{2r}[j], j = 0, 1, \dots, 15$. After the last round $k_{17}[j]$ is applied in a similar fashion as all previous odd numbered keys.

14.5.2.2 The substitution boxes “e” and “l”

In Figure 14.13 on page 173 two boxes occur, marked “e” and “l”. These boxes implement the same substitutions as used in SAFER+; i.e. they implement

$$\begin{aligned}
 e, l & : \quad \{0, \dots, 255\} \rightarrow \{0, \dots, 255\}, \\
 e & : \quad i \mapsto (45^i \pmod{257}) \pmod{256}, \\
 l & : \quad i \mapsto j \text{ s.t. } i = e(j).
 \end{aligned}$$

Their role, as in the SAFER+ algorithm, is to introduce non-linearity.

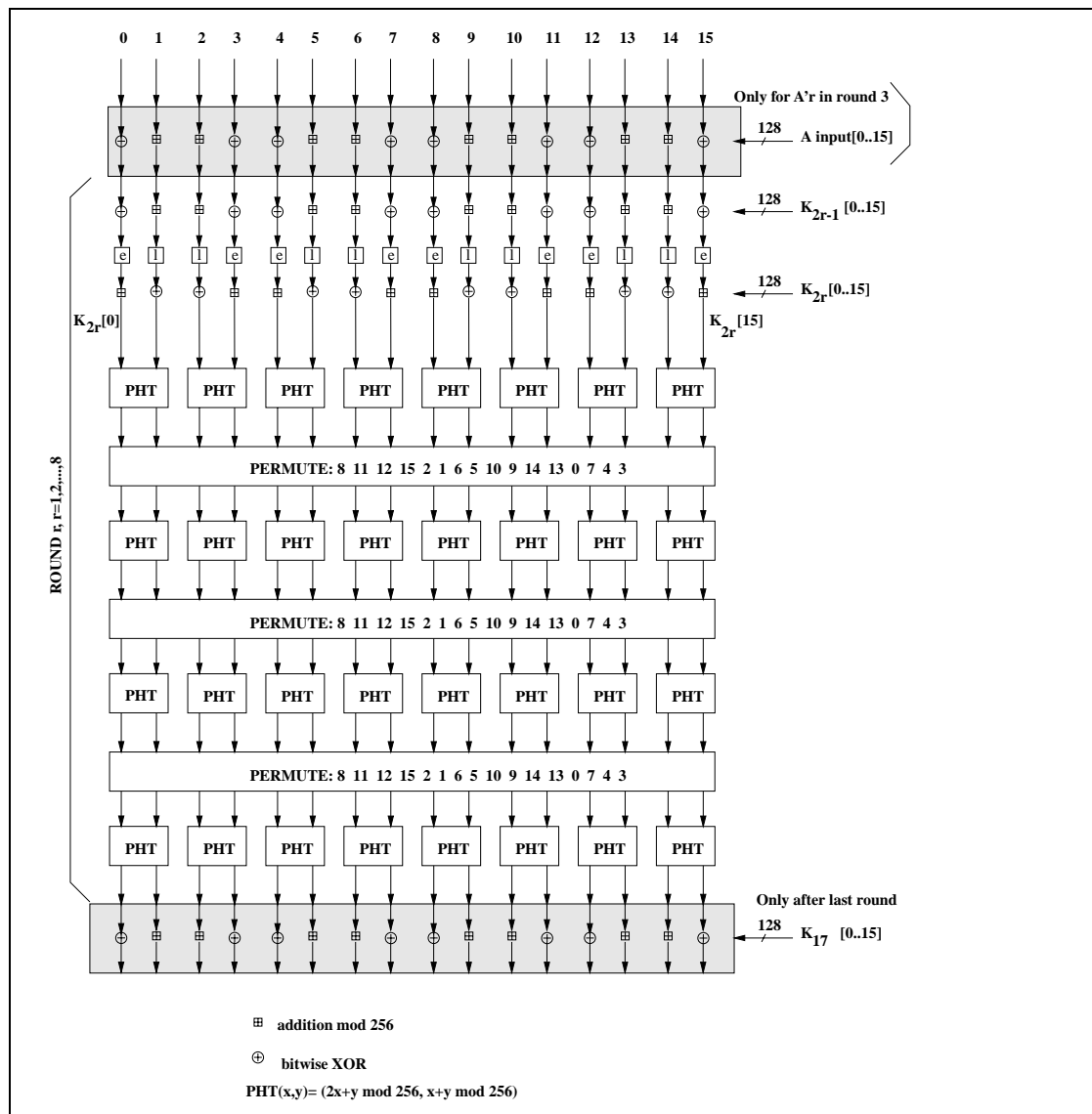


Figure 14.13: One round in A_r and A'_r . The permutation boxes show how input byte indices are mapped onto output byte indices. Thus, position 0 (leftmost) is mapped on position 8, position 1 is mapped on position 11, et cetera.

14.5.2.3 Key scheduling

In each round, 2 batches of 16 octet-wide keys are needed. These so-called round keys are derived as specified by the key scheduling in SAFER+. Figure 14.14 on page 174 gives an overview of how the round keys $K_p[j]$ are determined. The bias vectors B_2, B_3, \dots, B_{17} are computed according to following equation:

$$p[i] = \left(\left(45^{(45^{17p+i+1} \bmod 27)} \bmod 27 \right) \bmod 26 \right), \text{ for } i = 0, \dots, 15. \quad (\text{EQ 40})$$

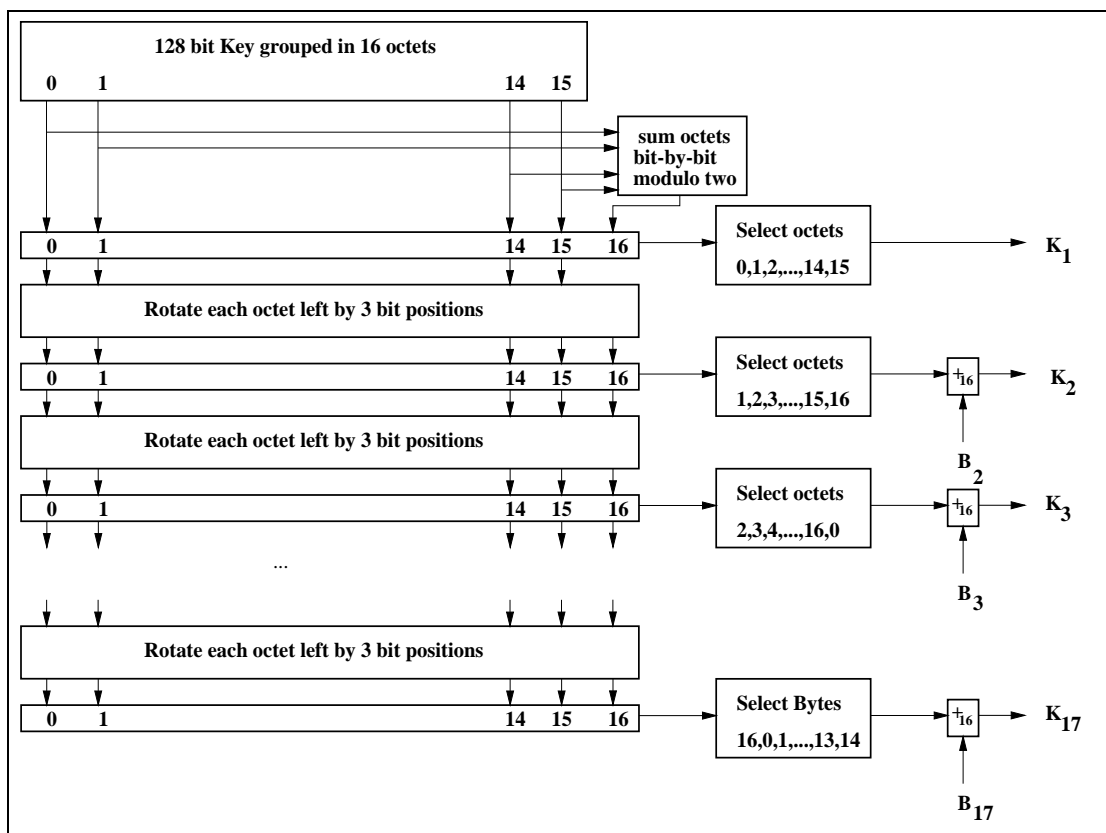


Figure 14.14: Key scheduling in A_r .

14.5.3 E_2 -Key generation function for authentication

The key used for authentication is derived through a procedure that is shown in Figure 14.15 on page 176. The figure shows two different modes of operation for the algorithm. In the first mode, the function E_2 should produce on input of a 128-bit RAND value and a 48-bit address, a 128-bit link key K . This mode is utilized when creating unit keys and combination keys. In the second mode the function E_2 should produce, on input of a 128-bit RAND value and an 1 octet user PIN, a 128-bit link key K . The second mode is used to create the initialization key, and also whenever a master key is to be generated.

When the initialization key is generated, the PIN is augmented with the BD_ADDR, see [Section 14.2.2.1 on page 153](#) for which address to use. The augmentation always starts with the least significant octet of the address immediately following the most significant octet of the PIN. Since the maximum length of the PIN used in the algorithm cannot exceed 16 octets, it is possible that not all octets of BD_ADDR will be used.

This key generating algorithm again exploits the cryptographic function. Formally E_{21} can be expressed for mode 1 (denoted A'_{1r}) as

$$E_{21}: \{0, 1\}^{128} \times \{0, 1\}^{48} \rightarrow \{0, 1\}^{128} \quad (\text{EQ 41})$$

$$(\text{RAND}, \text{address}) \mapsto A'_{1r}(X, Y)$$

where (for mode 1)

$$\begin{cases} X = \text{RAND}[0 \dots 14] \cup (\text{RAND}[15] \oplus 6) \\ Y = \bigcup_{i=0}^{15} \text{address}[i \pmod{6}] \end{cases} \quad (\text{EQ 42})$$

Let L be the number of octets in the user PIN. The augmenting is defined by

$$\text{PIN}' = \begin{cases} \text{PIN}[0 \dots L - 1] \cup \text{BD_ADDR}[0 \dots \min\{5, 15 - L\}], & L < 16, \\ \text{PIN}[0 \dots L - 1], & L = 16, \end{cases} \quad (\text{EQ 43})$$

Then, in mode 2, E_2 (denoted E_{22}) can be expressed as

$$E_{22}: \{0, 1\}^{8L'} \times \{0, 1\}^{128} \times \{1, 2, \dots, 16\} \rightarrow \{0, 1\}^{128} \quad (\text{EQ 44})$$

$$(\text{PIN}', \text{RAND}, L') \mapsto A'_{2r}(X, Y)$$

where

$$\begin{cases} X = \bigcup_{i=0}^{15} \text{PIN}'[i \pmod{L'}], \\ Y = \text{RAND}[0 \dots 14] \cup (\text{RAND}[15] \oplus L'), \end{cases} \quad (\text{EQ 45})$$

and $L' = \min\{16, L + 6\}$ is the number of octets in PIN'.

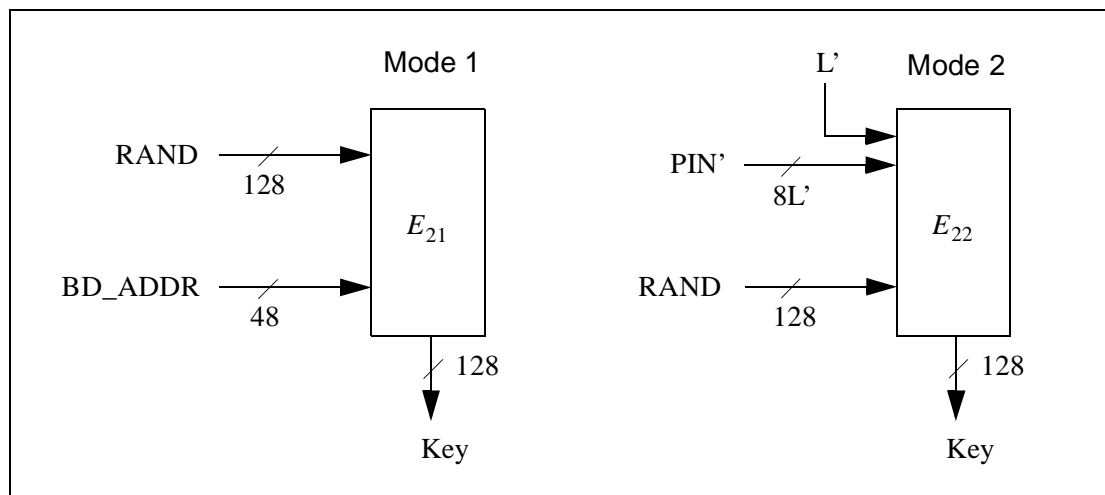


Figure 14.15: Key generating algorithm E_2 and its two modes. Mode 1 is used for unit and combination keys, while mode 2 is used for K_{init} and K_{master} .

14.5.4 E_3 -Key generation function for encryption

The ciphering key K_C used by E_0 is generated by E_3 . The function is constructed using A' , as follows

$$E_3: \{0, 1\}^{128} \times \{0, 1\}^{128} \times \{0, 1\}^{96} \rightarrow \{0, 1\}^{128} \tag{EQ 46}$$

$$(K, RAND, COF) \mapsto Hash(K, RAND, COF, 12)$$

where $Hash$ is the hash function as defined by (EQ 36) on page 170. Note that the produced key length is 128 bits. However, before use within E_0 , the encryption key K_C will be shortened to the correct encryption key length, as described in Section 14.3.5 on page 164. A block scheme of E_3 is depicted in Figure 14.16.

The value of COF is determined as specified by equation (EQ 26) on page 155.

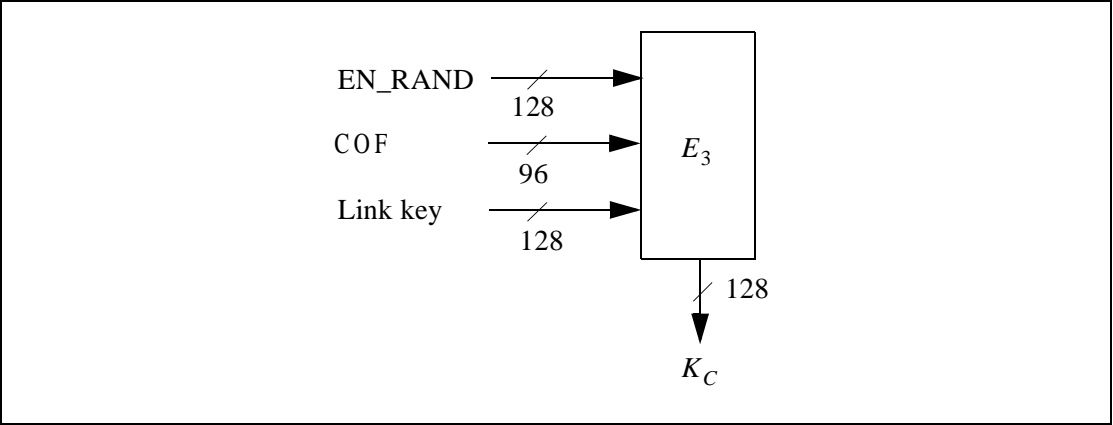


Figure 14.16: Generation of the encryption key.



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