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Technical Specification

Broadband Radio Access Networks (BRAN);
HIPERMAN;
Physical (PHY) layer
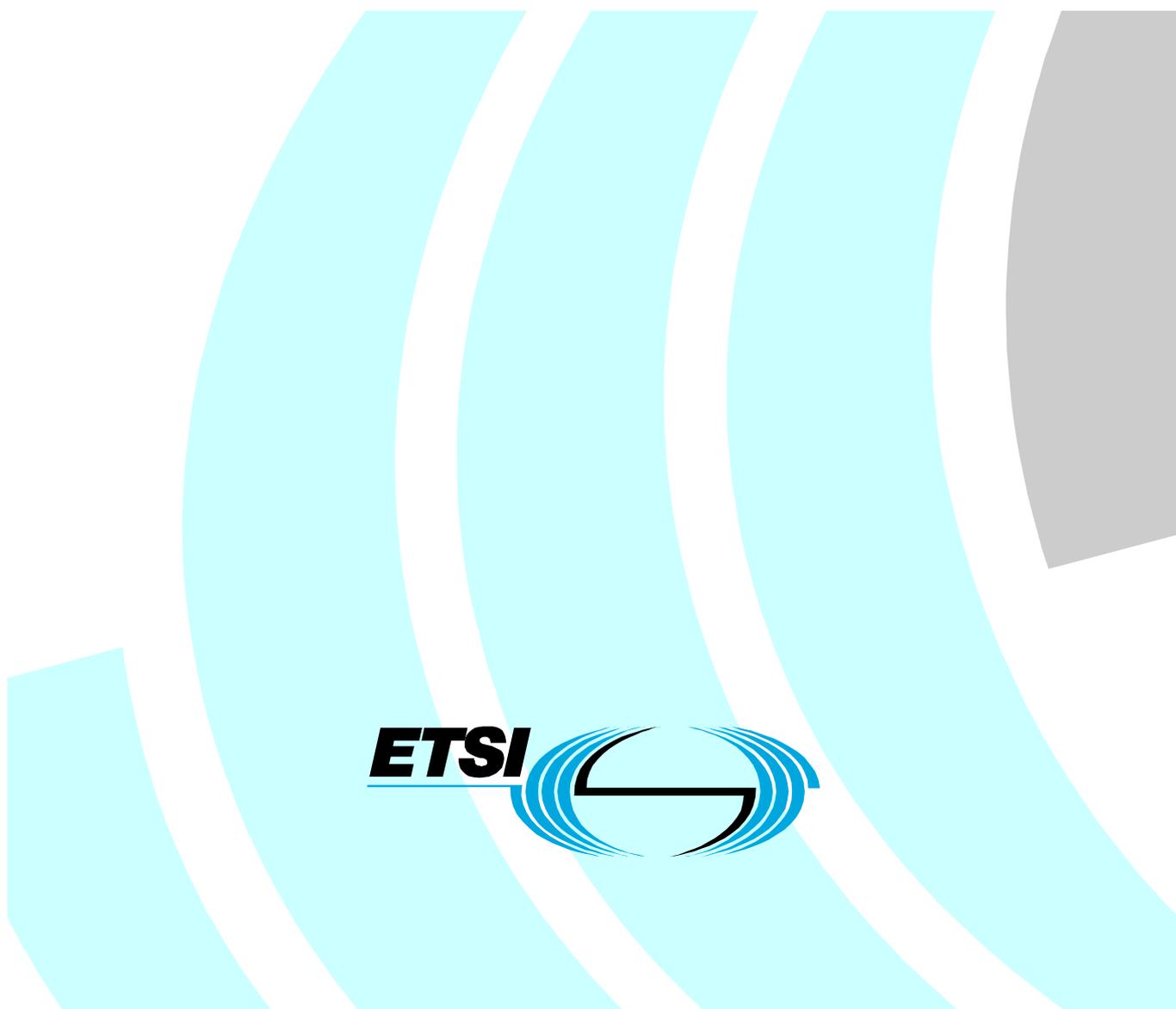
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Foreword

This Technical Specification (TS) has been produced by ETSI Project Broadband Radio Access Networks (BRAN).

The present document describes the physical layer specifications for High PERformance Radio Metropolitan Area Network (HIPERMAN), which operate on frequencies between 2 and 11 GHz. Separate ETSI documents provide details on the system overview, data link control layer (DLC), convergence layers (CL) and conformance testing requirements for HIPERMAN.

With permission of IEEE® (on file as BRAN32_5d009), portions of the present document are excerpted from IEEE Standard 802.16-2002 and IEEE Standard 802.16a-2003.

1 Scope

The present document specifies the HIPERMAN air interface with the specification layer 1 (physical layer), following the ISO-OSI model. HIPERMAN is confined only to the radio subsystems consisting of the *physical (PHY) layer* and the *DLC layer* - which are both core network independent - and the core network specific *convergence sub-layer*.

For managing radio resources and connection control, the data link control (DLC) protocol is applied, which uses the transmission services of the DLC layer. Convergence layers above the DLC layer handle the inter-working with layers at the top of the radio sub-system.

The scope of the present document is as follows:

- It gives a description of the physical layer for HIPERMAN systems.

- It specifies the transmission scheme in order to allow interoperability between equipment developed by different manufacturers. This is achieved by describing scrambling, channel coding, modulation, framing, control mechanisms, and power control to assist in radio resource management.

- The specification includes receiver requirements, transmitter requirements, frequency and timing requirements, and radio frequency channel plans necessary for radio regulatory purposes.

- Some information clauses and annexes describe parameters and system models to assist in preparing conformance, interoperability, and coexistence specifications.

2 References

The following documents contain provisions which, through reference in this text, constitute provisions of the present document.

- References are either specific (identified by date of publication and/or edition number or version number) or non-specific.

- For a specific reference, subsequent revisions do not apply.

- For a non-specific reference, the latest version applies.

Referenced documents which are not found to be publicly available in the expected location might be found at http://docbox.etsi.org/Reference.

[1]   ETSI TS 102 178: "Broadband Radio Access Networks (BRAN); HIPERMAN; Data Link Control (DLC) Layer".

[2]   IEEE P802.16-2001: "IEEE Standard for Local and Metropolitan Area Networks - Part 16: Air Interface for Fixed Broadband Wireless Access Systems".

3 Definitions, symbols and abbreviations

3.1 Definitions

For the purposes of the present document, the following terms and definitions apply:

Base Station (BS): generalized equipment consisting of one or more Base Station Controllers and one or more Base Station Transceivers

channel coding: sequence composed of three steps; randomizer, forward error correction, and interleaving

DL-MAP: structured data sequence that defined the mapping of the DL

Downlink (DL): direction from BS to SS

frequency offset index: An index number identifying a particular carrier in an OFDM signal. Frequency offset indices may be positive or negative and are counted relative to the DC carrier.

full duplex: equipment that is capable of transmitting and receiving at the same time

guard time: time at the beginning or end of each burst to allow power ramping up and down

half duplex: equipment that cannot transmit and receive at the same time

preamble: sequence of symbols with a given auto-correlation property assisting modem synchronization and channel estimation

Receive-Transmit Transition Gap (RTG): time to switch from receive to transmit at the BS

Subscriber Station (SS): generalized equipment consisting of a Subscriber Station Controller and Subscriber Station Transceiver.

Transmit-Receive Transition Gap (TTG): time to switch from transmit to receive at the BS

UL MAP: MAC message scheduling UL bursts

Uplink (UL): direction from SS to BS

3.2　Symbols

For the purposes of the present document, the following symbols apply:

BW	Nominal channel bandwidth (MHz)
F_{sa}	Sampling frenquency (MHz)
N_{cbps}	Number of coded bits per OFDM symbol (on allocated sub-channels)
N_{FFT}	Nominal size of the FFT operator
N_{used}	Number of carriers used to transport either data or pilots within a single OFDM symbol
R_{os}	BW oversampling ratio
T_b	Useful OFDM symbol time (s)
T_F	Frame duration (ms)
T_g	OFDM symbol guard time or CP time (s)
T_s	OFDM symbol time (s)
α_{avg}	Channel measurement averaging constant
Δf	Carrier spacing (Hz)

3.3　Abbreviations

For the purposes of the present document, the following abbreviations apply:

AAS	Adaptive Antenna System
BER	Bit Error Rate
BS	Base Station
BW	Bandwidth
CID	Connection Identifier
CINR	Carrier to Interference Noise Ratio
CP	Cyclic Prefix
CTC	Convolutional Turbo Code
DC	Direct Current
DCD	Downlink Channel Descriptor
DIUC	Down Link Interval Usage Code
DL	Down Link

FCH	Frame Control Header
FDD	Frequency Division Duplexing
FEC	Forward Error Correction
FFT	Fast Fourier Transform
HCS	Header Check Sequence
H-FDD	Half duplex Frequency Division Duplexing
IE	Information Element
lsb	least significant bit
MAC	Media Access Control
MAN	Metropolitan Area Network
msb	most significant bit
PDU	Protocol Data Unit
PMP	Point to Multi Point
PRBS	Pseudo Random Binary Sequence
PS	Physical Slot
QAM	Quadrature Amplitude Modulation
QPSK	Quadrature Phase Shift Keying
REQ	Request
RF	Radio Frequency
RMS	Root Mean Square
RS	Reed-Solomon
RS-CC	Reed-Solomon / Convolutional Code
RSSI	Received Signal Strength Indicator
RTG	Receive / Transmit Transition Gap
Rx	Receive
SNR	Signal to Noise Ratio
SS	Subscriber Station
STC	Space Time Coding
TDD	Time Division Duplexing
TTG	Transmit / Receive Transition Gap
Tx	Transmit
UCD	Up Link Channel Descriptor
UIUC	Up Link Interval Usage Code
UL	Up Link
XOR	Exclusive OR

4   OFDM symbol and transmitted signal

4.1   OFDM symbol description

An OFDM waveform is created by applying an Inverse-Fourier-transform to the source data. The resultant time duration is referred to as the useful symbol time T_b. A copy of the last T_g µs of the useful symbol period, termed Cyclic Prefix (CP), is prepended to enable the collection of multipath at the receiver, without loss of orthogonality between the tones. The resulting waveform is termed the symbol time T_s. Figure 1 illustrates this structure.
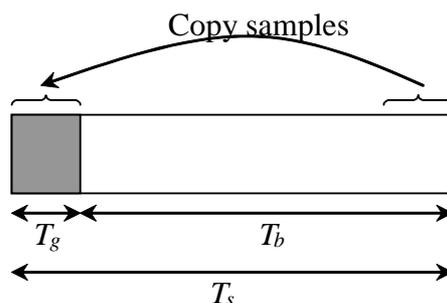
Figure 1: OFDM symbol time structure

The transmitter energy increases with the length of the CP while the receiver energy remains the same (the CP is discarded), so there is a $10\log(1 - T_g /(T_b + T_g))/\log(10)$ dB loss in SNR. Using the CP, the samples required for performing the FFT at the receiver can be taken anywhere over the length of the extended symbol. This provides multipath immunity as well as a tolerance for symbol time synchronization errors.

On system initialization, the Base Station (BS) CP fraction (T_g / T_b) shall be set to a specific value for use on the Downlink (DL). Once the BS is operational the CP value shall not be changed. On initialization, the Subscriber Station (SS) shall search all possible values of CP until it finds the CP being used by the serving BS. The SS shall use the same CP values determined in DL for the UL. Changing the CP value parameter at the BS through (re)initialization forces all SS registered on that BS to re-synchronize.

In the frequency domain, each OFDM symbol is comprised of multiple carriers (see figure 2), which belong to one of three types:

- Data carriers - for data transmission.

- Pilot carriers - for channel estimation and other purposes.

- Null carriers - for guard bands and the DC carrier.
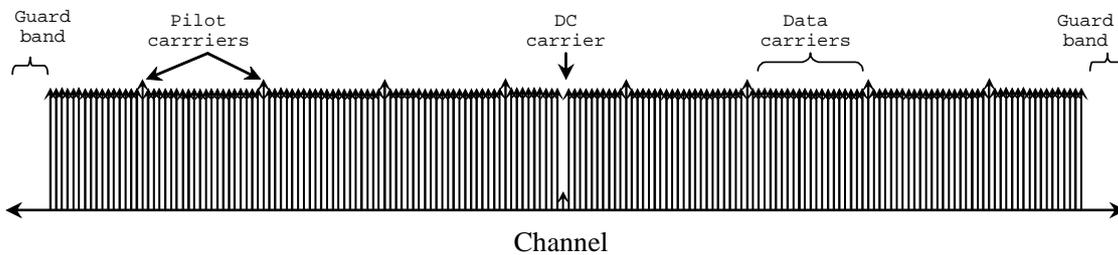
Figure 2: OFDM symbol frequency structure

4.2 Transmitted signal

Equation 1 specifies the transmitted signal voltage $s(t)$ to the antenna, as a function of time, during any OFDM symbol.

$$s(t) = \text{Re}\left\{ e^{2j\pi f_c t} \sum_{\substack{k=-N_{\text{used}}/2 \\ k \neq 0}}^{k=N_{\text{used}}/2} c_k \times e^{2j\pi k\Delta f (t - T_g)} \right\} \tag{1}$$

where: t is the time elapsed since the beginning of the subject OFDM symbol, with $0 < t < T_s$.

       C_k is a complex number; the data to be transmitted on the carrier whose frequency offset index is k, during the subject OFDM symbol. It specifies a point in a Quadrature Amplitude Modulation (QAM) constellation.

       f_c is the RF carrier frequency, being the centre frequency of the intended RF frequency channel.

       k is the frequency offset index.

The parameters of the transmitted OFDM signal, which shall be used, are given in table 1.

Table 1: OFDM symbol parameters

Parameter	Value
N_{FFT}	256
N_{used}	200
R_{os}	8/7
T_g / T_b	1/4, 1/8, 1/16, 1/32
Frequency offset indices of guard carriers	-128,-127...,-101 +101,+102,...,127
Frequency offset indices of Pilots	-88,-63,-38,-13,13,38,63,88
Subchannel Index: 0b01000 { 0b00100 { 0b00010 { 0b00001: 0b00011: 0b00110 { 0b00101: 0b00111: 0b01100 { 0b01010 { 0b01001: 0b01011: 0b01110 { 0b01101: 0b01111: 0b11000 { 0b10100 { 0b10010 { 0b10001: 0b10011: 0b10110 { 0b10101: 0b10111: 0b11010 { 0b11001: 0b11011: 0b11110 { 0b11101: 0b11111:	Allocated frequency offset indices of carriers: {-100:-98, -37:-35, 1:3, 64:66} {-38} { -97:-95, -34:-32, 4:6, 67:69} { -94:-92, -31:-29, 7:9, 70:72} {13} { -91:-89, -28:-26, 10:12, 73:75} { -87:-85, -50:-48, 14: 16, 51:53} {-88} { -84,-82, -47:-45, 17: 19, 54:56} { -81:-79, -44:-42, 20:22, 57:59} {63} { -78:-76, -41:-39, 23:25, 60:62} { -75:-73, -12:-10, 26:28, 89:91} {-13} { -72:-70, -9: -7, 29:31, 92:94} { -69:-67, -6: -4, 32:34, 95:97} {38} { -66:-64, -3: -1, 35:37, 98:100} { -62:-60, -25:-23, 39:41, 76:78} {-63} { -59:-57, -22:-20, 42:44, 79:81} { -56:-54, -19:-17, 45:47, 82:84} {88} { -53:-51, -16:-14, 48:50, 85:87} Note that pilot carriers are allocated only if two or more subchannels are allocated.

Using the parameters as specified in table 1, the following relationships shall hold.

$$\Delta f = \frac{R_{\text{os}} \cdot BW}{N_{\text{FFT}}}$$

$$T_b = \frac{1}{\Delta f}$$

$$T_g = \left(\frac{T_g}{T_b} \right) \times T_b$$

$$T_s = T_b + T_g$$

$$T_{\text{sa}} = \frac{1}{F_{\text{sa}}}$$

$$F_{sa} = R_{OS} \times BW$$

5 Channel Coding

Channel coding is composed of three steps: randomization, forward error correction, and interleaving. They shall be applied in this order at transmission. The complementary operations shall be applied in reverse order at reception.

5.1 Randomization

Data randomization is performed on each burst of data (i.e. not on pilots and preambles) independently. If the amount of data to transmit does not fit exactly the amount of data allocated, padding of 0xFF ("1"s only) shall be added to the end of the transmission block, up to the amount of data allocated.

The Pseudo Random Binary Sequence (PRBS) generator shall be $1 + x^{14} + x^{15}$ as shown in figure 3. Each data byte to be transmitted shall enter sequentially into the randomizer, most significant bit (msb) first. The seed value shall be used to calculate the randomization bits, which are combined in an XOR operation with the serialized bit stream of each burst. The "data out" bits from the randomizer shall be applied to the FEC.
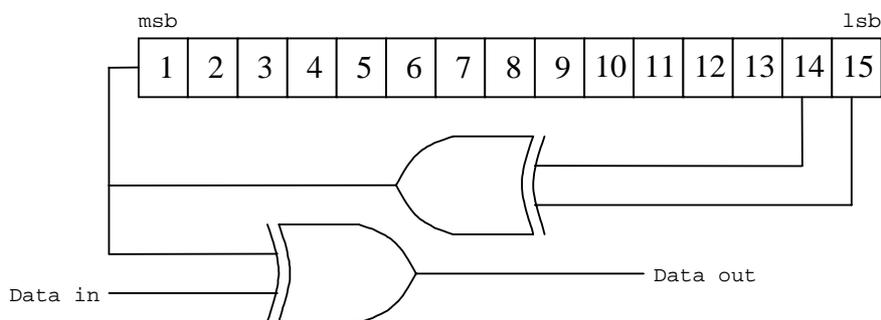
Figure 3: Data randomization PRBS

On the DL, the randomizer shall be initialized at the start of the FCH with the vector: 1 0 0 1 0 1 0 1 0 0 0 0 0 0 0. The randomizer shall not be reset at the start of burst #1. At the start of subsequent bursts the randomizer shall be initialized with the vector shown in figure 4. The OFDM symbol number (i.e. the number of the first OFDM symbol of the data burst) shall be counted from the start of the DL-subframe, the first symbol being counted as symbol #0.
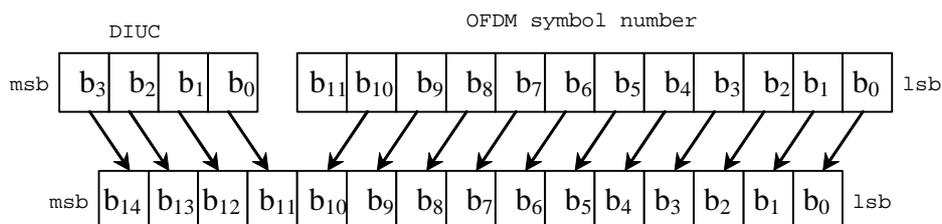
Figure 4: Scrambler DL initialization vector for bursts #2..*N*

On the UL, the randomizer shall be initialized with the vector shown in figure 5. The symbol transmitted at the time instant pointed by Allocation Start Time field of UL-MAP shall be considered symbol #0.
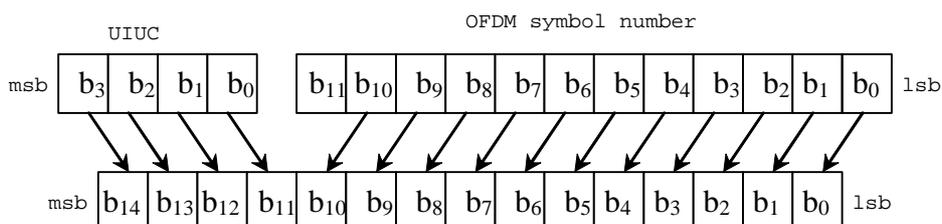
Figure 5: Scrambler UL initialization vector

5.2   Forward Error Correction (FEC)

The FEC consisting of the concatenation of a Reed-Solomon outer code and a rate-compatible convolutional inner code shall be supported on both UL and DL. Support of Convolutional Turbo Code (CTC) is optional. The Reed-Solomon-Convolutional coding rate 1/2 shall always be used as the coding mode when requesting access to the network and in the Frame Control Header (FCH) burst.

The encoding is performed by first passing the data in block format through the RS encoder and then passing it through a convolutional encoder. Eight tail bits are introduced at the end of each allocation. In the RS encoder, the redundant bits are sent before the input bits, keeping the tail bits at the end of the allocation.

5.2.1   Concatenated Reed-Solomon / convolutional code (RS-CC)

The RS encoding shall be derived from a systematic RS ($N=255$, $K=239$, $T=8$) code using GF(2^8), where:

   N is the number of overall bytes after encoding

   K is the number of data bytes before encoding

   T is the number of data bytes which can be corrected

For the systematic code, the code generator polynomial $g(x)$, shown in Equation 2, and field generator polynomial $p(x)$, shown in Equation 3, shall be used.

$$g(x) = \left(x + \lambda^0\right)\left(x + \lambda^1\right)\left(x + \lambda^2\right)\cdots\left(x + \lambda^{2T-1}\right), \qquad \lambda = 02_{\text{HEX}} \tag{2}$$

$$p(x) = x^8 + x^4 + x^3 + x^2 + 1 \tag{3}$$

This code is shortened and punctured to enable variable block sizes and variable error-correction capability. When a block is shortened to K' data bytes, add 239-K' zero bytes as a prefix. After encoding discard these 239-K' zero bytes. When a codeword is punctured to permit T' bytes to be corrected, only the first 2T' of the total 16 parity bytes shall be employed. The bit/byte conversion shall be msb first.

Each RS block is encoded by the binary convolutional encoder, which shall have native rate of 1/2, a constraint length equal to 7, and shall use the generator polynomials codes shown in Equation 4 to derive its two code bits.

$$\begin{aligned} G_1 &= 171_{\text{OCT}} &\quad \text{for } X \\ G_2 &= 133_{\text{OCT}} &\quad \text{for } Y \end{aligned} \tag{4}$$

The generator is depicted in figure 6.
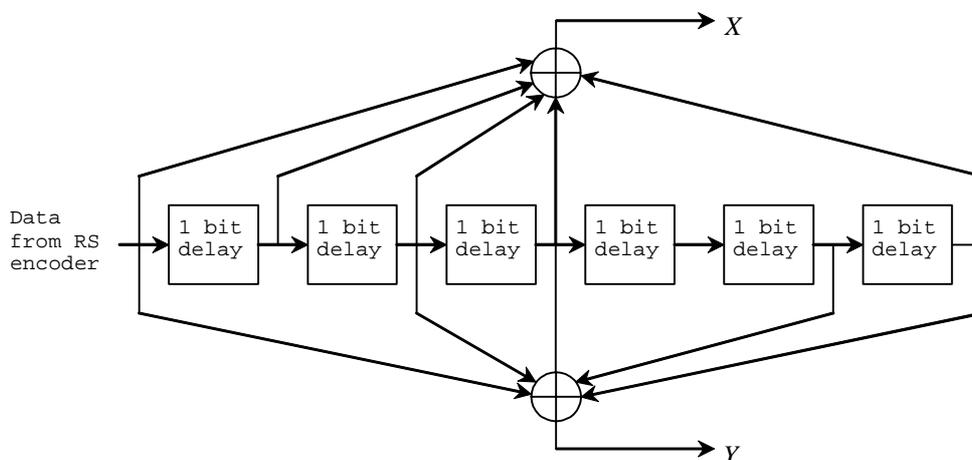
Figure 6: Convolutional encoder of rate 1/2

Puncturing patterns and serialization order which shall be used to realize different code rates are defined in table 2. Transmitted bits are denoted by "1" and removed bits are denoted by "0". *X* and *Y* are in reference to figure 6. Puncturing for Code rate 1/2 shall be implemented when sub-channelization is implemented, otherwise it shall remain unimplemented or unused.

Table 2: Convolutional code puncturing configuration

Code rate	d_{free}	*X*	*Y*	Order
1/2	10	1	1	$X_1 Y_1$
2/3	6	10	11	$X_1 Y_1 Y_2$
3/4	5	101	110	$X_1 Y_1 Y_2 X_3$
5/6	4	10101	11010	$X_1 Y_1 Y_2 X_3 Y_4 X_5$

In order to allow sharing of the error correction decoder, each of the multiple data streams subdivides its data into RS blocks. Each RS block is encoded by zero tail convolutional encoder. Eight tail bits are introduced at the end of each burst. In the RS encoder, the redundant bits are sent before the input bits, keeping the tail bits at the end of the burst.

Table 3 defines the block sizes for the different modulation levels and code rates. As 64 QAM is optional, the codes for this modulation shall only be implemented if the modulation is implemented.

Table 3: Channel encodings

Modulation	Uncoded Block Size (Bytes)	Coded Block Size (bytes)	Overall Coding Rate	RS Code	CC Code Rate
QPSK	24	48	1/2	(32,24,4)	2/3
QPSK	36	48	3/4	(40,36,2)	5/6
16 QAM	48	96	1/2	(64,48,8)	2/3
16 QAM	72	96	3/4	(80,72,4)	5/6
64 QAM	96	144	2/3	(108,96,6)	3/4
64 QAM	108	144	3/4	(120,108,6)	5/6

When sub-channelization is applied in the UL, the FEC shall bypass the RS encoder and use the Overall Coding Rate as indicated in table 3 as CC Code Rate. The Uncoded Block Size and Coded Block Size may be computed by multiplying the values listed in table 3 by the number of allocated subchannels divided by 16.

5.2.2 Convolutional Turbo Coding (Optional)

The Convolutional Turbo Code encoder, including its constituent encoder, is depicted in figure 7. It uses a double binary Circular Recursive Systematic Convolutional code. The bits of the data to be encoded are alternately fed to A and B, starting with the msb of the first byte being fed to A. The encoder is fed by blocks of k bits or N couples ($k = 2 \times N$ bits). For all the frame sizes k is a multiple of 8 and N is a multiple of 4. Further N shall be limited to: $8 \leq N/4 \leq 1024$.

The polynomials defining the connections are described in octal and symbol notations as follows:

- for the feedback branch: 0xB, equivalently $1+D+D^3$ (in symbolic notation);

- for the Y parity bit: 0xD, equivalently $1+D^2+D^3$.
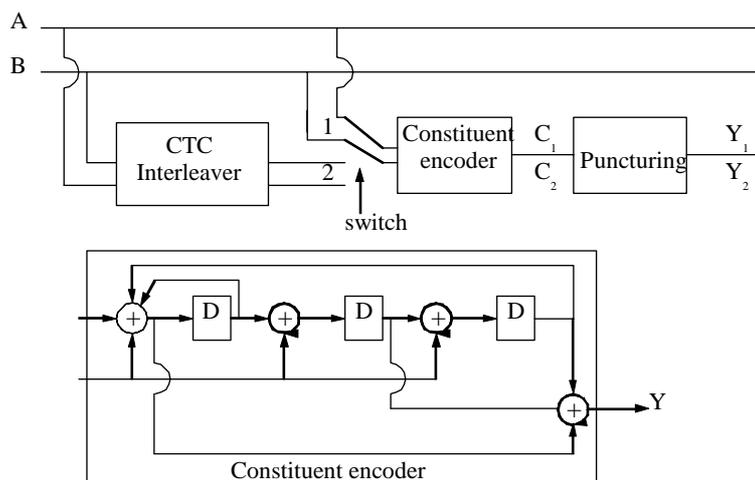
Figure 7: CTC encoder

First, the encoder (after initialization by the circulation state Sc_1, see below) is fed the sequence in the natural order (position 1) with the incremental address $i = 0 .. N-1$. This first encoding is called C_1 encoding. Then the encoder (after initialization by the circulation state Sc_2, see below) is fed by the interleaved sequence (switch in position 2) with incremental address $j = 0, … N-1$. This second encoding is called C_2 encoding.

The order in which the encoded bit shall be fed into the interleaver (see clause 5.3) is:

$A_0, B_0 .. A_{N-1}, B_{N-1}, Y_{10}, Y_{1,1} .. Y_{1,M}, Y_{20}, Y_{2,1} .. Y_{2,M},$

where M is the number of parity bits.

Table 4 gives the block sizes, code rates, channel efficiency, and code parameters for the different modulation and coding schemes. As 64 QAM is optional, the codes for this modulation shall only be implemented if the modulation is implemented.

Table 4: Optional CTC Coding per Modulation

Modulation	Data Block Size (Bytes)	Coded Block Size (Bytes)	Overall Code Rate	N	P_0	P_1
QPSK	24	48	1/2	96	7	N/4
QPSK	32	48	2/3	128	11	N/4
QPSK	36	48	3/4	144	17	N/4
16 QAM	48	96	1/2	192	11	N/4
16 QAM	72	96	3/4	288	13	N/4
64 QAM	96	144	2/3	384	17	N/4
64 QAM	108	144	3/4	432	17	N/4

5.2.2.1   CTC Interleaver

The interleaver requires the parameters P_0 and P_1, shown in table 4.

The two-step interleaver shall be performed by:

Step 1: Switch alternate couples

for $j = 1...N$

if $(j_{\mod 2} == 0)$ *let* $(B, A) = (A, B)$ *(i.e. switch the couple)*

Step 2: $P_i(j)$

The function $P_i(j)$ provides the interleaved address i of the consider couple j.

for $j = 1...N$

switch $j_{\mod_4}$:

case 0: $i = (P_0 \cdot j + 1)_{\mod_N}$

case 1: $i = (P_0 \cdot j + 1 + N/4 + P_1)_{\mod_N}$

case 2: $i = (P_0 \cdot j + 1 + P_1)_{\mod_N}$

case 3: $i = (P_0 \cdot j + 1 + N/2 + P_1)_{\mod_N}$

5.2.2.2   Determination of CTC circulation states

The state of the encoder is denoted S ($0 \leq S < 7$) with S the value read binary (left to right) out of the constituent encoder memory (see figure 7). The circulation states Sc_1 and Sc_2 are determined by the following operations:

1) Initialize the encoder with state 0. Encode the sequence in the natural order for the determination of Sc1 or in the interleaved order for determination of Sc_2. In both cases the final state of the encoder is $S0_{N-1}$;

2) according to the length N of the sequence, use table 5 to find Sc_1 or Sc_2.

Table 5: Circulation state lookup table (Sc)

$N_{\mod}$	$S0_{N-1}$							
	0	**1**	**2**	**3**	**4**	**5**	**6**	**7**
1	0	6	4	2	7	1	3	5
2	0	3	7	4	5	6	2	1
3	0	5	3	6	2	7	1	4
4	0	4	1	5	6	2	7	3
5	0	2	5	7	1	3	4	6
6	0	7	6	1	3	4	5	2

5.2.2.3   CTC puncturing

The three code-rates are achieved through selectively deleting the parity bits (puncturing). The puncturing patterns are identical for both codes C_1 and C_2.

Table 6: Circulation state lookup table (Sc)

Rate $R_n/(R_n+1)$	Y					
	0	**1**	**2**	**3**	**4**	**5**
1/2	1	1				
2/3	1	0	1	0		
3/4	1	0	0	1	0	0

5.3   Interleaving

A block interleaver shall interleave all encoded data bits with a block size corresponding to the number of coded bits per the allocated sub-channels per OFDM symbol, N_{cbps}. The interleaver is defined by a two step permutation.

The first, shown in Equation 5, ensures that adjacent coded bits are mapped onto nonadjacent carriers. The second permutation, shown in Equation 6, ensures that adjacent coded bits are mapped alternately onto less or more significant bits of the constellation, thus avoiding long runs of less reliable bits.

Let N_{cpc} be the number of coded bits per carrier, i.e. 2, 4 or 6 for QPSK, 16QAM or 64QAM, respectively. Let $s = N_{\text{cpc}}/2$. Within a block of N_{cbps} bits at transmission, let k be the index of a coded bit before the first permutation; mk be the index of that coded bit after the first and before the second permutation; and let jk be the index of that coded bit after the second permutation, just prior to modulation mapping.

The first permutation is defined by the formula:

$$mk = \left(N_{cbps}/12\right)k \bmod \left(12\right) + floor\left(k/12\right) \quad k = 0,1,...,N_{cbps}-1 \tag{5}$$

The second permutation is defined by the formula:

$$jk = s..floor\left(m.k/s\right) + \left(m.k + N_{cbps} - floor\left(12m.k/N_{cbps}\right)\right)\bmod\left(s\right) \quad k = 0,1,...,N_{cbps}-1 \tag{6}$$

The de-interleaver, which performs the inverse operation, is also defined by two permutations. Within a received block of N_{cbps} bits, let j be the index of a bit before the first permutation; let mj be the index of that bit after the first and before the second permutation; and let kj be the index of that bit after the second permutation, just prior to delivering the block to the convolutional decoder.

The first permutation is defined by the rule formula:

$$mj = s.floor\left(j/s\right) + \left(j + floor\left(12j/N_{cbps}\right)\right)\bmod\left(s\right) \quad j = 0,1,...,N_{cbps}-1 \tag{7}$$

The second permutation is defined by the rule formula:

$$kj = 12mj - \left(N_{cbps}-1\right) floor\left(12mj/N_{cbps}\right) \quad j = 0,1,...,N_{cbps}-1 \tag{8}$$

The first permutation in the de-interleaver is the inverse of the second permutation in the interleaver, and conversely. table 7 shows the bit interleaver sizes as a function of modulation and coding.

Table 7: Block sizes of bit interleaver

	N_{cbps}				
	Default (16 sub-channels)	8 sub-channels	4 sub-channels	2 sub-channels	1 sub-channels
QPSK	384	192	96	48	24
16 QAM	768	384	192	96	48
64 QAM	1152	576	288	144	72

5.4   Modulation

5.4.1   Data Modulation

After bit interleaving, the data bits are entered serially to the constellation mapper. Gray-mapped QPSK and 16QAM as shown in figure 8 shall be supported. Support of 64QAM is optional. The constellations as shown in figure 8 shall be normalized by multiplying the constellation point with the indicated factor c to achieve equal average power. For each modulation, b_0 denotes the lsb. The first bit out of the interleaver shall be mapped to the msb and so forth.

Per-allocation adaptive modulation and coding shall be supported in the DL. The UL shall support different modulation schemes for each SS based on the Media Access Control (MAC) burst configuration messages coming from the BS. The constellation-mapped data shall be subsequently modulated onto all allocated data carriers in order of increasing frequency offset index. The first symbol out of the data constellation mapping shall be modulated onto the allocated carrier with the lowest frequency offset index.
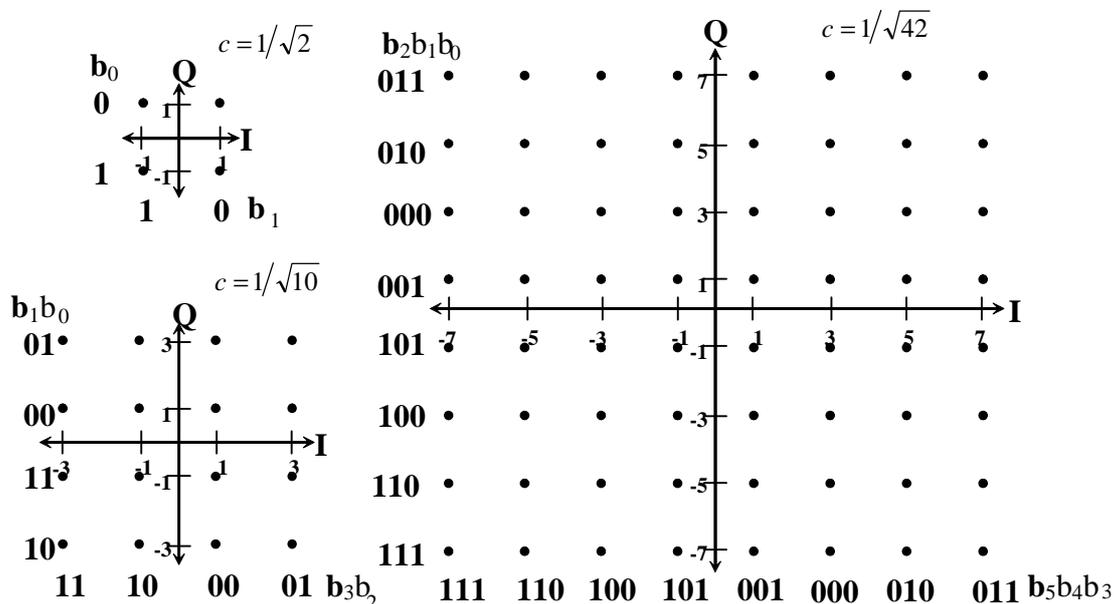

Figure 8: Modulation constellations

5.4.2   Pilot Modulation

The value of the pilot modulation for OFDM symbol k, relative to the beginning of the DL-subframe respectively the UL-subframe, shall be derived from wk, generated from PRBS $x^{11} + x^9 + 1$ as shown in figure 9. The initialization sequences that shall be used on the DL and UL are indicated in figure 9 as well. On the DL, this shall result in the sequence 11111111111000000000110... where the 3rd 1, i.e. $w_3 = 1$, shall be used in the first OFDM DL symbol following the frame preamble.
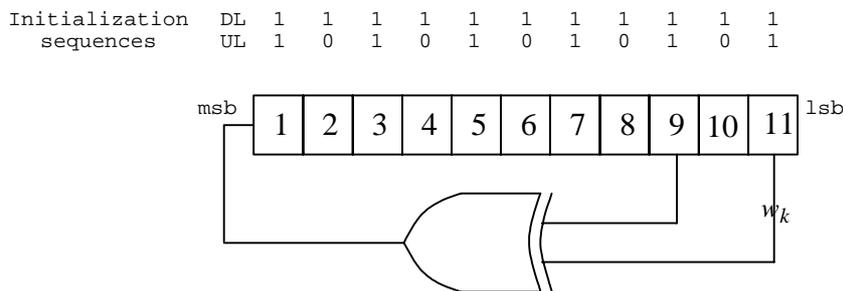

Figure 9: Pilot modulation PRBS

For each pilot (indicated by frequency offset index), the BPSK modulation shall be derived as shown in table 8.

Table 8: Pilot modulation

	c_{-88}	c_{-63}	c_{-38}	c_{-13}	c_{13}	c_{38}	c_{63}	c_{88}
DL	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - \overline{w_k})$	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - \overline{w_k})$	$2(\frac{1}{2} - \overline{w_k})$	$2(\frac{1}{2} - \overline{w_k})$	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - w_k)$
UL	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - \overline{w_k})$	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - \overline{w_k})$	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - w_k)$	$2(\frac{1}{2} - w_k)$

In an allocation of 1 subchannel, pilots shall not be modulated or transmitted.

5.4.3 Rate ID encodings

Rate_ID's, which indicate modulation and coding to be used in the first DL burst immediately following the FCH, are shown in table 9. The Rate_ID encoding is static and cannot be changed during system operation

Table 9: Rate_ID encodings

Rate_ID	Modulation RS-CC rate
0	QPSK 1/2
1	QPSK 3/4
2	16QAM 1/2
3	16QAM 3/4
4	64QAM 2/3
5	64QAM 3/4
6-15	Reserved

5.5 Example UL RS-CC Encoding

To illustrate the use of the RS-CC encoding, an example of one frame of OFDM UL data is provided, illustrating each process from randomization through carrier modulation.

Modulation Mode: QPSK, rate 3/4, Slot Offset: 14, UIUC=7 (decimal values)

From clause 5.4.2, for the UL, $w_{14} = 0$.

Input Data (Hex)

45 29 C4 79 AD 0F 55 28 AD 87 B5 76 1A 9C 80 50 45 1B 9F D9 2A 88 95 EB AE B5 2E 03 4F 09 14 69 58 0A 5D

Randomized Data (Hex)

D5 0E A4 AA EF E4 DB 51 88 91 6B 00 DF AA 1E E7 02 A8 0E 70 4F 7F C9 D8 66 1D 9D F0 E7 20 E4 9D 7A 32 91

Reed-Solomon encoded Data (Hex)

95 CE 22 76 D5 0E A4 AA EF E4 DB 51 88 91 6B 00 DF AA 1E E7 02 A8 0E 70 4F 7F C9 D8 66 1D 9D F0 E7 20 E4 9D 7A 32 91 00

Convolutionally Encoded Data (Hex)

D5 2E 96 38 FE 93 1E 6A AF 17 D3 44 E4 8B 45 8F 13 D6 AF 27 E3 B2 D5 0A 57 C1 2A F1 A9 73 86 71 3F F1 03 95 F8 ED 2D 30 A2 E0 DB D2 F8 8E B3 4C

Interleaved Data (Hex)

D4 EA 3A CA D0 85 A2 C8 75 DE 23 B5 AB 57 F7 E9 7C 3C 01 9E 8B FC 98 17 0E 2E DC 5E D2 38 B7 0C 26 83 E4 EF 6F 52 8D C9 FC 1D 7C A2 36 66 1D 4A

Carrier Mapping (frequency offset index: I value Q value)

    -100: -1 -1, -99: 1 -1, -98: 1 -1, -97: 1 1, -96: -1 -1, -95: -1 1, -94: -1 1, -93: -1 1, -92: 1 1, -91: -1 -1,
    -90: -1 1, -89: -1 1, -88:pilot= 1 0, -87: -1 -1, -86: 1 1, -85: -1 1, -84: -1 1, -83: -1 -1, -82: 1 -1, -81: 1 1,
    -80: 1 1, -79: -1 1, -78: 1 1, -77: 1 -1, -76: 1 -1, -75: -1 1, -74: -1 1, -73: 1 1, -72: -1 1, -71: -1 -1,
    -70: 1 1, -69: -1 1, -68: 1 1, -67: 1 -1, -66: -1 -1, -65: 1 -1, -64: 1 -1, -63:pilot= -1 0, -62: -1 -1, -61: 1 -1,
    -60: -1 -1, -59: -1 1, -58: 1 1, -57: -1 1, -56: 1 1, -55: -1 -1, -54: -1 1, -53: -1 -1, -52: 1 -1, -51: 1 -1,
    -50: -1 1, -49: -1 1, -48: -1 1, -47: -1 -1, -46: 1 -1, -45: 1 -1, -44: 1 -1, -43: -1 -1, -42: -1 -1, -41: -1 -1,
    -40: 1 -1, -39: -1 -1, -38:pilot= 1 0, -37: -1 -1, -36: -1 1, -35: -1 1, -34: 1 -1, -33: 1 -1, -32: -1 -1, -31: -1 -1,
    -30: 1 1, -29: 1 1, -28: -1 -1, -27: -1 -1, -26: 1 1, -25: 1 1, -24: 1 1, -23: 1 1, -22: 1 -1, -21: -1 1,
    -20: 1 -1, -19: -1 -1, -18: -1 1, -17: -1 1, -16: 1 1, -15: -1 1, -14: -1 -1, -13:pilot= -1 0, -12: -1 -1, -11: -1 -1,
    -10: -1 -1, -9: 1 1, -8: -1 1, -7: 1 -1, -6: -1 1, -5: 1 1, -4: 1 1, -3: 1 -1, -2: 1 -1, -1: -1 -1,
    0: 0 0, 1: 1 1, 2: 1 1, 3: -1 -1, 4: -1 1, 5: 1 1, 6: -1 1, 7: -1 -1, 8: -1 1, 9: -1 -1,
    10: 1 -1, 11: -1 -1, 12: 1 1, 13:pilot= 1 0, 14: 1 -1, 15: 1 -1, 16: -1 -1, 17: -1 1, 18: -1 -1, 19: 1 -1,
    20: 1 1, 21: -1 1, 22: 1 1, 23: -1 -1, 24: -1 1, 25: 1 1, 26: -1 1, 27: -1 -1, 28: 1 -1, 29: -1 -1,
    30: 1 1, 31: 1 1, 32: -1 -1, 33: 1 1, 34: 1 1, 35: -1 1, 36: 1 -1, 37: -1 1, 38:pilot= 1 0, 39: -1 1,
    40: 1 1, 41: 1 1, 42: -1 -1, 43: -1 -1, 44: -1 1, 45: 1 -1, 46: 1 1, 47: -1 1, 48: -1 1, 49: -1 -1,
    50: -1 -1, 51: 1 -1, 52: -1 1, 53: -1 -1, 54: -1 -1, 55: 1 -1, 56: 1 -1, 57: 1 1, 58: -1 1, 59: -1 1,
    60: 1 1, 61: -1 -1, 62: 1 -1, 63:pilot= 1 0, 64: -1 -1, 65: 1 1, 66: -1 1, 67: 1 -1, 68: -1 -1, 69: -1 -1,
    70: -1 -1, 71: 1 1, 72: 1 1, 73: 1 -1, 74: -1 -1, 75: 1 -1, 76: 1 -1, 77: -1 -1, 78: -1 -1, 79: 1 1,
    80: -1 1, 81: -1 1, 82: 1 1, 83: -1 1, 84: 1 1, 85: -1 -1, 86: 1 -1, 87: -1 1, 88:pilot= 1 0, 89: 1 -1,
    90: -1 1, 91: 1 -1, 92: -1 1, 93: 1 1, 94: 1 -1, 95: -1 -1, 96: 1 -1, 97: 1 -1, 98: 1 1, 99: -1 1, 100: -1 1

Note that the above QPSK values (all values with exception of the BPSK pilots) are to be normalized with a factor $1/\sqrt{2}$ as indicated in figure 8.

5.6   Preamble structure and modulation

All preambles are structured as either one or two OFDM symbols. The OFDM symbols are defined by the values of the composing subcarriers. Each of those OFDM symbols contains a cyclic prefix, which length is the same as the CP for data OFDM symbols.

The first preamble in the DL PHY PDU, as well as the initial ranging preamble, consists of two consecutive OFDM symbols. The first OFDM symbol uses only subcarriers the indices of which are a multiple of 4. As a result, the time domain waveform of the first symbol consists of 4 repetitions of 64-sample fragment, preceded by a CP. The second OFDM symbol utilizes only even subcarriers, resulting in time domain structure composed of 2 repetitions of a 128-sample fragment, preceded by a CP. The time domain structure is exemplified in figure 10. This combination of the two OFDM symbols is referred to as the long preamble.

| CP | 64 | 64 | 64 | 64 | CP | 128 | 128 |

T_g ◄———► T_b ◄———► T_g ◄———► T_b
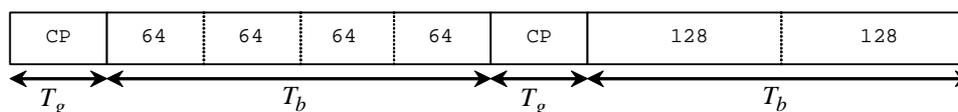
Figure 10: DL and network entry preamble structure

The frequency domain sequences for all full-bandwidth preambles are derived from the sequence:

$P_{ALL}(-100:100)=$
 { 1-j, 1-j, -1-j, 1+j, 1-j, 1-j, -1+j, 1-j, 1-j, 1-j, 1+j, -1-j, 1+j, 1+j, -1-j, 1+j, -1-j,
 -1-j, 1-j, -1+j, 1-j, 1-j, -1-j, 1+j, 1-j, 1-j, -1+j, 1-j, 1-j, 1-j, 1+j, -1-j, 1+j, 1+j,
 -1-j, 1+j, -1-j, -1-j, 1-j, -1+j, 1-j, 1-j, -1-j, 1+j, 1-j, 1-j, -1+j, 1-j, 1-j, 1-j, 1+j,
 -1-j, 1+j, 1+j, -1-j, 1+j, -1-j, -1-j, 1-j, -1+j, 1+j, 1+j, 1-j, -1+j, 1+j, 1+j, -1-j,
 1+j, 1+j, 1+j, -1+j, 1-j, -1+j, -1+j, 1-j, -1+j, 1-j, 1-j, 1+j, -1-j, -1-j, -1-j, -1+j,
 1-j, -1-j, -1-j, 1+j, -1-j, -1-j, -1-j, 1-j, -1+j, 1-j, 1-j, -1+j, 1-j, -1+j, -1+j, -1-j,
 1+j, 0, -1-j, 1+j, -1+j, -1+j, -1-j, 1+j, 1+j, 1+j, -1-j, 1+j, 1-j, 1-j, 1-j, -1+j, -1+j,
 -1+j, -1+j, 1-j, -1-j, -1-j, -1+j, 1-j, 1+j, 1+j, -1+j, 1-j, 1-j, 1-j, -1+j, 1-j, -1-j, -1-j,
 -1-j, 1+j, 1+j, 1+j, 1+j, -1-j, -1+j, -1+j, 1+j, -1-j, 1-j, 1-j, 1+j, -1-j, -1-j, -1-j, 1+j,
 -1-j, -1+j, -1+j, -1+j, 1-j, 1-j, 1-j, 1-j, -1+j, 1+j, 1+j, -1-j, 1+j, -1+j, -1+j, -1-j,
 1+j, 1+j, 1+j, -1-j, 1+j, 1-j, 1-j, 1-j, -1+j, -1+j, -1+j, -1+j, 1-j, -1-j, -1-j, 1-j, -1+j,
 -1-j, -1-j, 1-j, -1+j, -1+j, -1+j, 1-j, -1+j, 1+j, 1+j, 1+j, -1-j, -1-j, -1-j, -1-j, 1+j,
 1-j, 1-j}   (9)

The frequency domain sequence for the 4 times 64 sequence P_{4x64} is defined by:

$$P_{4x64}(k) = \begin{cases} \sqrt{2} \cdot \sqrt{2} \cdot \text{conj}(P_{ALL}(k)) & k_{\text{mod}4} = 0 \\ 0 & k_{\text{mod}4} \neq 0 \end{cases} \qquad (10)$$

In Equation 10, the factor of sqrt(2) equates the Root-Mean-Square (RMS) power with that of the data section. The additional factor of sqrt(2) is related to the 3 dB boost.

The frequency domain sequence for the 2 times 128 sequence P_{EVEN} is defined by:

$$P_{EVEN}(k) = \begin{cases} \sqrt{2} \cdot P_{ALL}(k) & k_{\text{mod}2} = 0 \\ 0 & k_{\text{mod}2} \neq 0 \end{cases} \qquad (11)$$

In P_{EVEN}, the factor of sqrt(2) is related to the 3 dB boost.

In the UL, when the entire 16 subchannels are used, the data preamble, as shown in figure 11 consists of one OFDM symbol utilizing only even subcarriers. The time domain waveform consists of 2 times 128 samples preceded by a CP. The subcarrier values shall be set according to the sequence P_{EVEN}. This preamble is referred to as the short preamble.

This preamble shall also precede all allocations during the AAS portion of a frame and shall be used as burst preamble on the DL bursts when indicated in the DL-MAP_IE.

CP	128	128

T_g T_b
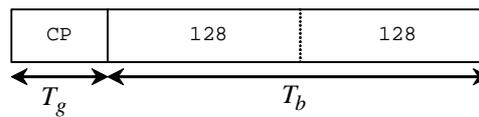
Figure 11: UL data and DL AAS preamble structure

In the DL bursts which start with a preamble and which fall within the STC-encoded region, the preamble shall be transmitted from both transmit antennas simultaneously and shall consist of a single OFDM symbol. The preamble transmitted from the first antenna shall use only even subcarriers, the values of which are set according to the sequence P_EVEN. The preamble transmitted from the second antenna shall use only odd subcarriers, the values of which shall be set according to the sequence P_{ODD}.

$$P_{ODD}(k) = \begin{cases} 0 & k_{\text{mod}2} = 0 \\ \sqrt{2} \cdot P_{ALL}(k) & k_{\text{mod}2} \neq 0 \end{cases} \qquad (12)$$

The AAS network entry preamble shall use odd subcarriers only, using the sequence of values P_AAS. The time domain waveform consists of two groups of 128 samples preceded by a CP, the sign of the samples in the second group being reversed from those in the first group, as shown in figure 128ak. The sequence P_AAS is defined by:

$$P_{\text{AAS}}(k) = \begin{cases} 0 & k_{\text{mod}2} = 0 \\ \sqrt{2} \cdot \text{conj}(P_{\text{ALL}}(k)) & k_{\text{mod}2} \neq 0 \end{cases} \tag{13}$$

Using mesh, bursts sent in the control subframe shall start with the long preamble. In the data subframe, the bursts shall by default start with the long preamble, but neighbours may negotiate to use the short preamble by setting the preamble flag in the Neighbour Link Info field.

In mesh mode, bursts sent in the control subframe shall start with the long preamble. In the mesh data sub-frame, the bursts shall be default start with the long preamble, but neighbours may negotiate to use the short preamble by setting the preamble flag in the Neighbour Link Info field.

In the UL, when subchannelization transmissions are employed, the data preamble consists of a 256 sample sequence preceded by a CP whose length is the same as the cyclic prefix for data OFDM symbols. This preamble is referred to as the subchannelization preamble. The frequency domain sequence for the 256 samples is defined by P_{SUB}. Preamble carriers that do not fall within the allocated subchannels shall be set to zero.

P_{SUB} (-100:100) =
{ 1+j, 1+j, -1-j, 1+j, -1+j, 1+j, 1+j, 1+j, -1-j, -1-j, 1-j, -1-j, 1-j, 1+j, 1-j, 1+j, 1+j, -1-j, -1-j,
1+j, 1-j, 1+j, -1-j, 1+j, 1+j, 1+j, 1+j, -1-j, 1+j, -1+j, 1+j, 1+j, 1+j, -1-j, -1-j, 1-j, -1-j, -1-j,
1+j, 1-j, 1+j, 1+j, -1-j, -1-j, 1+j, 1-j, 1+j, -1-j, 1+j, 1+j, 1+j, 1+j, -1-j, 1+j, -1+j, 1+j, 1+j,
1+j, -1-j, -1-j, 1-j, -1-j, 1-j, -1-j, -1+j, -1-j, -1-j, 1+j, 1+j, -1-j, -1+j, -1-j, 1+j, -1-j, -1-j,
1+j, 1+j, -1-j, 1+j, -1+j, 1+j, 1+j, 1+j, -1-j, -1-j, 1-j, -1-j, 1-j, -1-j, -1+j, -1-j, -1-j, 1+j, 1+j,
-1-j, -1+j, -1-j, 1+j, -1-j, -1-j, 0,1+j, 1+j, -1-j, 1+j, -1+j, 1+j, 1+j, 1+j, -1-j, -1-j, 1-j, -1-j,
1-j, 1+j, 1-j, 1+j, 1+j, -1-j, -1-j, 1+j, 1-j, 1+j, -1-j, 1+j, 1+j, -1-j, -1-j, 1+j, -1-j, 1-j, -1-j, -
1-j, -1-j, 1+j, 1+j, -1+j, 1+j, 1-j, -1-j, -1+j, -1-j, -1-j, 1+j, 1+j, -1-j, -1+j, -1-j, 1+j, -1-j, -1-
j, -1-j, -1-j, 1+j, -1-j, 1-j, -1-j, -1-j, -1-j, 1+j, 1+j, -1+j, 1+j, -1+j, 1+j, 1-j, 1+j, 1+j, -1-j, -
1-j, 1+j, 1-j, 1+j, -1-j, 1+j, 1+j, 1+j, 1+j, -1-j, 1+j, -1+j, 1+j, 1+j, 1+j, -1-j, -1-j, 1-j, -1-j, -
1-j, -1-j, -1+j, -1-j, -1-j, 1+j, 1+j, -1-j, -1+j, -1-j, 1+j, -1-j, -1-j}   (14)

In the case that the UL allocation contains midambles, the midambles will consist of one OFDM symbol and shall be identical to the preamble used with the allocation.

6   Frame structures

6.1   PMP

6.1.1   Duplexing modes

The PHY shall support Frequency Division Duplexing (FDD) or Time Division Duplexing (TDD). FDD SSs may be Half Duplex FDD (H-FDD).

The OFDM PHY supports a frame-based transmission. The frame interval contains BS and SS transmissions comprised of PHY PDUs, Transmit/receive Transition Gap (TTG) and Receive/transmit Transition Gap (RTG) gaps, and guard intervals. A frame consists of a DL sub-frame and an UL sub-frame. A DL sub-frame consists of only one DL PHY PDU. A UL sub-frame consists of contention intervals scheduled for initial ranging and bandwidth request purposes and one or multiple UL PHY PDUs, each transmitted from a different SS.

A UL PHY PDU consists of only one burst, which is made up of a short preamble and an integer number of OFDM symbols. The burst PHY parameters of an UL PHY PDU are specified by a 4-bit UIUC in the UL-MAP. The UIUC encoding is defined in the Uplink Channel Descriptor (UCD) messages. Note the difference between a PHY PDU and a Burst.

A DL PHY PDU starts with a long preamble, which is used for PHY synchronization. The preamble is followed by a FCH burst. The FCH burst is one OFDM symbol long and is transmitted using QPSK rate 1/2 with the mandatory coding scheme. The FCH contains DL_Frame_Prefix to specify the burst profile and length of DL burst #1. The Rate_ID encoding is defined in table 9. A DL-MAP message shall immediately follow the DL_Frame:Prefix. An UL-MAP message shall immediately follow the DL-MAP message. Note that in the case of the remainder of the FCH being smaller than the size of the two messages combined they will "spill" over into DL Burst #1. UCD and DCD messages may be transmitted following the DL-MAP and UL-MAP messages. Although DL burst #1 contains broadcast MAC control messages, it is not necessary to use the most robust well-know modulation/coding. A more efficient modulation/coding may be used if it is supported and applicable to all the SSs of the serving BS. With exception of the maps, no MAC PDUs shall be split over multiple consecutive bursts with different burst profiles.

The FCH is followed by one or multiple DL bursts, each transmitted with different burst profiles. Each DL burst consists of an integer number of OFDM symbols, and its burst profiles are specified by a 4-bit Downlink Interval Usage Code (DIUC) in the DL-MAP. The DIUC encoding is defined in the DCD messages.

With the OFDM PHY, a PHY burst, either a DL PHY burst or an UL PHY burst, consists of an integer number of OFDM symbols, carrying MAC messages, i.e., MAC PDUs. To form an integer number of OFDM symbols, a burst payload may be padded by the bytes 0xFF. Then the payload should be scrambled, encoded, and modulated using the burst PHY parameters specified by the present document.

In each TDD frame (see figure 12), the Tx/Rx transition gap (TTG) and Rx/Tx transition gap (RTG) shall be inserted between the DL and UL sub-frame and at the end of each frame respectively to allow the BS to turn around.

In TDD and H-FDD systems subscriber station allowances must be made by a transmit-receive turnaround gap SSRTG and by a receive-transmit turnaround gap SSTTG. The BS shall not transmit DL information to a station later than (SSRTG+RTD) before its scheduled UL allocation, and shall not transmit DL information to it earlier than (SSTTG-RTD) after the end of scheduled UL allocation, where RTD denotes Round-Trip Delay. The parameters SSRTG and SSTTG are capabilities provided by the SS to BS upon request during network entry.

For TDD mode SSRTG and SSTTG shall be no more than 50 µs. For H-FDD mode SSRTG and SSTTG shall be no more than 100 µs.
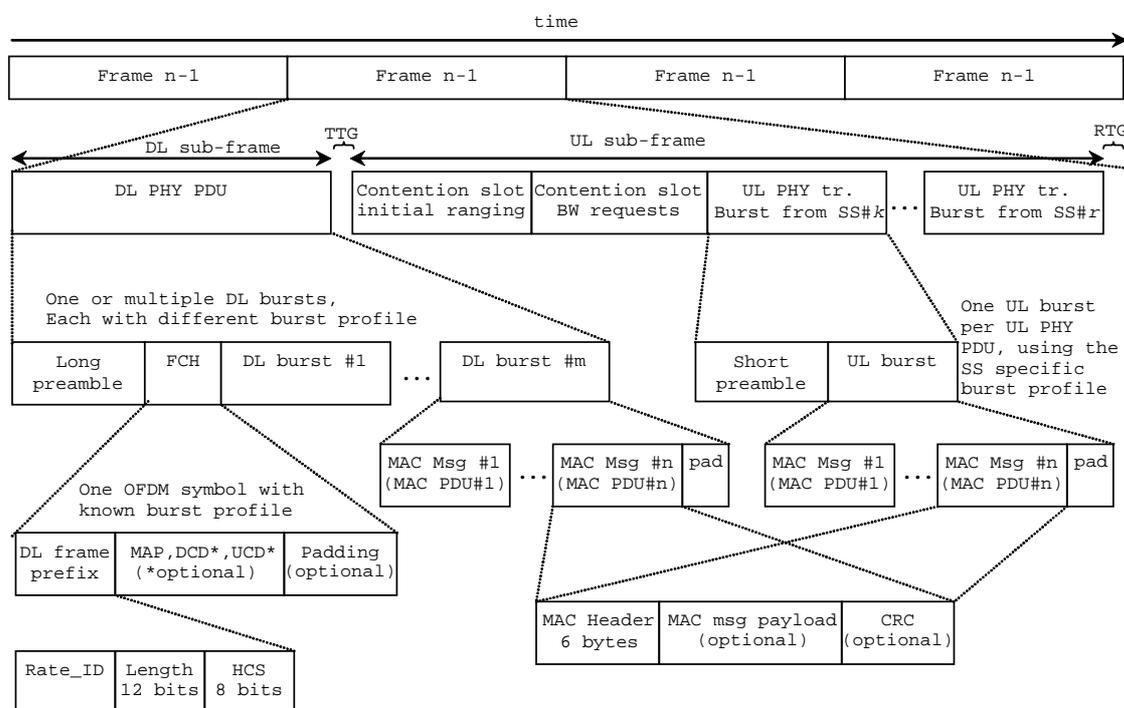
Figure 12: Example of OFDM PMP frame structure with TDD
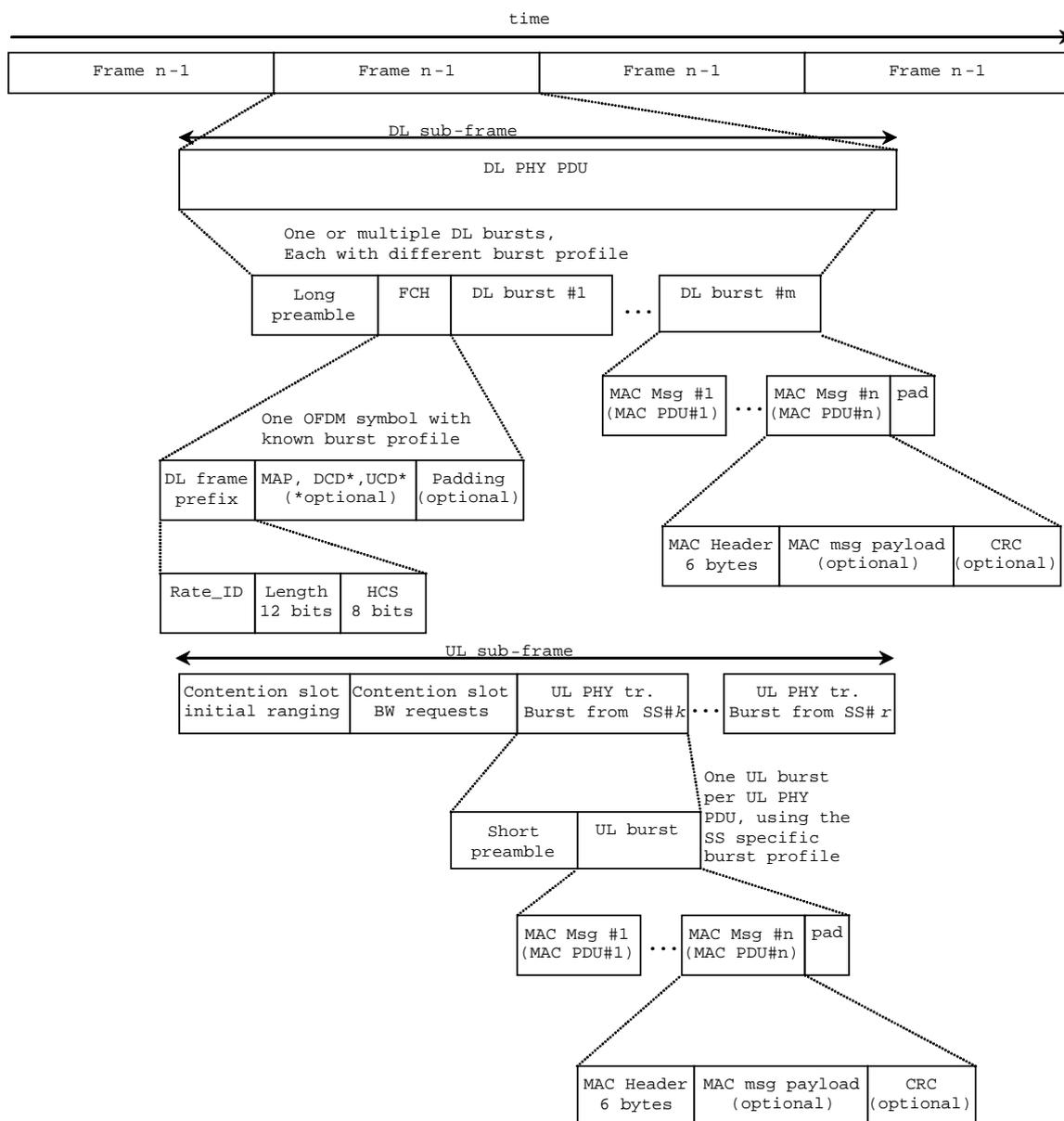
Figure 13: Example of OFDM PMP frame structure with FDD

6.1.2 DL Frame Prefix

Table 10: OFDM DL Frame Prefix format

Syntax	Size	Notes
DL_Frame_Prefix_Format() {		
Rate_ID	4 bits	
Length	12 bits	
HCS	8 bits	
}		

Rate_ID
  Field that defines the burst profile of the following burst. Encoding is specified in table 9.

Length
  Number of OFDM symbols (PHY payload) in the burst immediately following the FCH burst. The minimum value shall be 6.

HCS

Header Check Sequence used to detect errors in the DL Frame Prefix. The generator polynomial shall be

$$g(x) = x^8 + x^2 + x + 1.$$

The transmitter shall take the RateID and Length bytes and divide them by $g(x)$ and use the remainder as HCS code. At the receiver dividing the DL_Frame_Prefix by $g(x)$ then gives remainder 0 if correct. (Example: RateID=1 and Length=204 symbols: Encode the byte sequence [0x10 0xCC] and obtain 0x3D as the HCS byte.)

6.2 Mesh

In addition to the PMP frame structure, an optional frame structure (see figure 14) is defined to facilitate mesh networks.
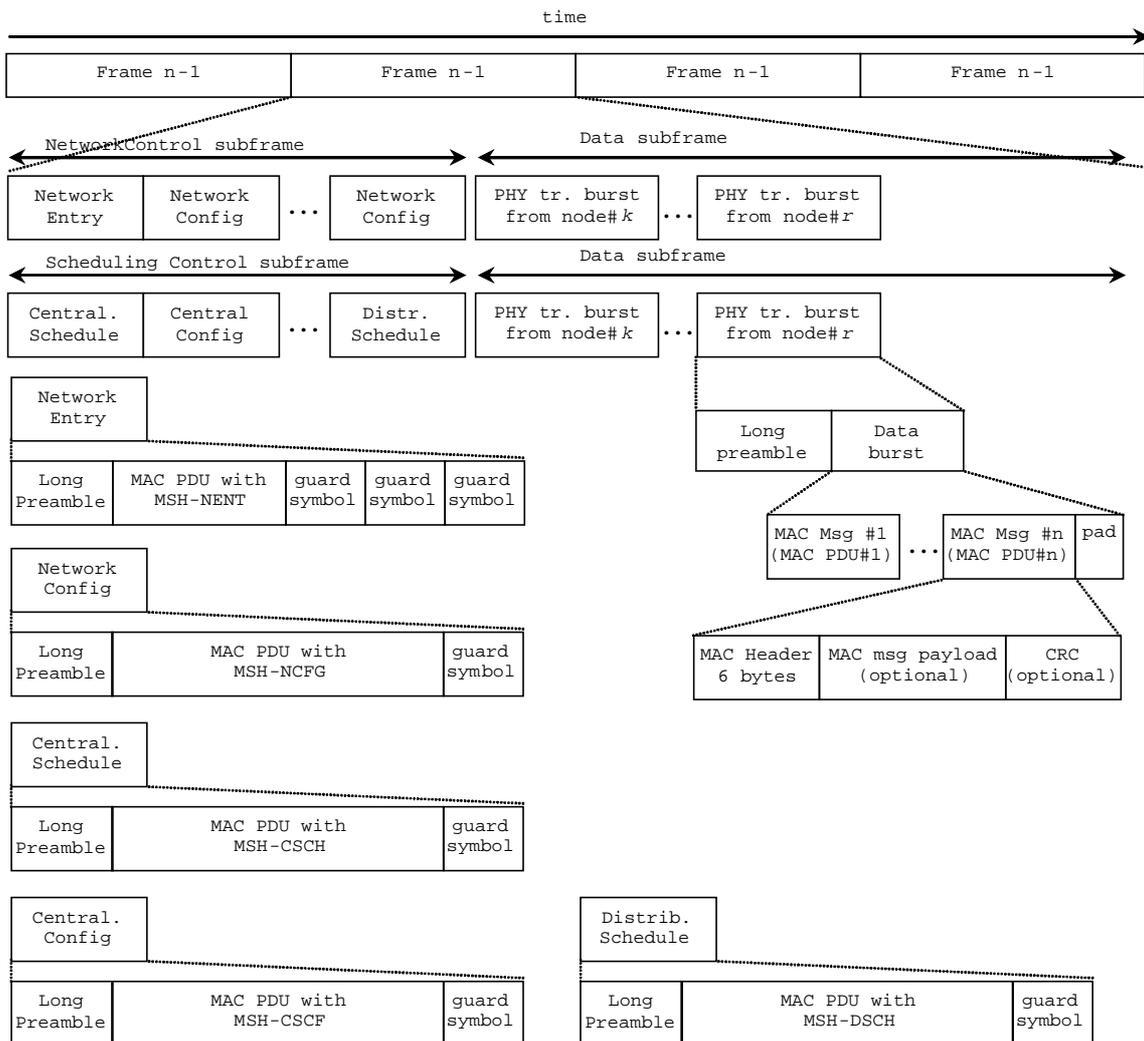
Figure 14: Mesh frame structure

A mesh frame consists of a control and data subframe. The control subframe has two basic functionalities. One is the creation and maintenance of cohesion between the different systems, termed "network control" in figure 14. The other is the co-ordinated scheduling of data-transfers between systems, termed "schedule control" figure 14. Frames with a network control subframe occur periodically, as indicated in the network descriptor. All other frames have a schedule control subframe. The length of the control subframe is fixed and of length MSH-CTRL-LEN x 7 OFDM symbols, with MSH-CRTL-LEN indicated in the network descriptor.

During a network control subframe, the first 7 symbols are allocated for network entry, followed by MSH-CTRL-LEN - 1 sets of 7 symbols for network configuration. During a schedule control subframe, the network descriptor indicates how many (MSH-DSCH-NUM) distributed scheduling messages may occur in the control subframe. The first (MSH-CRTL-LEN - MSH-DSCH-NUM) x 7 symbols are allocated to transmission bursts containing MSH-CSCH and MSH-CSCF PDUs, whereas the remainder is allocated to transmission bursts containing MSH-DSCH PDUs.

Distributed scheduling messages (using the long preamble) may further occur in the data subframe if not in conflict with the scheduling dictated in the control subframe.

All transmissions in the control subframe are sent using the most robust mandatory burst profile. The data subframe is divided into minislots, which are, with possible exception of the last minislot in the frame, of size ceiling((OFDM symbols per frame-MSH-CTRL-LEN x 7)/256). A scheduled allocation consists of one or more minislots.

6.3   Frame duration codes

Table 11 indicates the nominal frame durations that are allowed. The frame duration used can be determined by the periodicity of the frame start preambles. Once a specific frame duration has been selected by the BS, it should not be changed. Changing the frame duration shall force all SSs to resynchronize to the BS.

Table 11: Frame duration Codes

Code	Frame Duration (ms)	Frames/s
0	2,5	400
1	4	250
2	5	200
3	8	125
4	10	100
5	15,625	64
6	20	50

7   Control Mechanisms

7.1   Synchronization

7.1.1   Network synchronization

It is recommended (but not required) that all BSs be time synchronized to a common timing signal. In the event of the loss of the network timing signal, BSs may continue to operate and shall automatically resynchronize to the network timing signal when it is recovered. The synchronizing reference shall be a 1pps timing pulse based on a 10 MHz frequency reference. These signals are typically provided by a GPS receiver.

Frequency references derived from the timing reference may be used to control the frequency accuracy of Base-Stations provided that they meet the frequency accuracy requirements of clause 12. This applies during normal operation and during loss of timing reference.

7.2   Ranging

There are two types of ranging processes, initial ranging and periodic ranging. Initial ranging (coarse synchronization) and power control is performed during (re)registration, after synchronization is lost; and on a periodic basis. Initial ranging uses the initial ranging contention-based interval, which requires a long preamble and the use of the most robust mandatory burst profile. Periodic ranging uses the regular UL burst.

During registration, a new subscriber registers during the random access channel and if successful is entered into a ranging process under control of the BS. The ranging process is cyclic in nature where default time and power parameters are used to initiate the process followed by cycles where (re)calculated parameters are used in succession until parameters meet acceptance criteria for the new subscriber. These parameters are monitored, measured and stored at the BS and transmitted to the SS for use during normal exchange of data. The stored parameters are updated in a periodic manner based on configurable update intervals to ensure changes in the channel can be accommodated. The update intervals shall vary in a controlled manner on a SS by SS basis.

Ranging on re-registration follows the same process as new registration.

One of the parameters that limit cell radius is the round trip propagation time. Round trip propagation time shall be taken into account when determining the time open for initial ranging. This time should be at least equal to the maximum tolerable round trip delay plus the number of OFDM symbols necessary to transmit the ranging burst.

A BS supporting the AAS option may allocate in the UL subframe an 8 OFDM symbol initial ranging slot for AAS SSs that have to initially alert the BS of their presence. This period shall be marked in the UL-MAP as Initial-Maintenance (UIUC=1), but shall be marked by a non-used CID such that no non-AAS subscriber (or AAS subscriber that can decode the UL-MAP message) uses this interval for initial maintenance. During the first OFDM symbol of this AAS initial ranging slot, the BS shall transmit the AAS network entry preamble. In TDD mode the BS can use the last OFDM symbol of the DL subframe to transmit the AAS network entry preamble and mark this symbol as Gap (DIUC=13) in de DL-MAP. The AAS initial ranging slot shall then be at the beginning of the UL subframe. This eliminates unnecessary TX-RX switching.

SSs that compute their $P_{TX\_IR\_max}$ to exceed their maximum power level and SSs that have attempted initial ranging with the maximum power level using RNG-REQ may, if the BS supports sub-channelization, attempt initial ranging in an initial ranging slot using the following burst format:

The SS shall transmit the long preamble as defined in clause 5.6. This shall be followed by a single randomly selected subchannel with duration of one OFDM symbol, containing the preamble for this subchannel as defined in clause 5.6.

The BS need only detect that energy is sent on a single subchannel and may respond by allocating a single subchannel identifying the SS by the Transmit Opportunity and Frame Number in which the transmission was received.

A SS attempting subchannelized initial ranging shall use its maximum power setting for the initial ranging burst.

7.3 Bandwidth requesting

There may be two types of Request (REQ) Regions in a frame. These two types are REQ Region-Full and REQ Region-Focused.

In a REQ Region-Full, a SS may send a message containing a Bandwidth Request MAC Header. Each Transmit Opportunity shall consist of a short preamble and one OFDM symbol using the most robust mandatory burst profile.

In a REQ Region-Focused, a station shall send a short code over a Transmit Opportunity which consists of 4 subcarriers by two OFDM symbols. Each Transmit Opportunity within a MAC frame shall be indexed by consecutive Transmit Opportunity Indices.

All SSs shall be capable of the Full Contention Transmission. Capability of the Focused Contention Transmission is optional. The SS shall follow the backoff procedure as described in [2], clause 6.2.8.

7.3.1 Parameter Selection

The SS shall examine the UL_MAP message for a future frame and select (in accordance with clause 6.2.8 of [2]) a future REQ Region during which to make its request. If Focused Contention Supported = 1 is returned by the BS in the SBC-RSP message during SS initialization and if the SS is capable of focused contention, it may choose either a REQ Region-Full or REQ Region-Focused. Otherwise, it shall choose a REQ Region-Full.

If the chosen REQ Region is a REQ Region-Focused, the SS shall also select, at random with equal probability, a contention code from table 12 and similarly a contention channel from table 13. The BS may split the Contention Channel Indexes using the Subchannelization Focused Contention Code TLV in the UCD to allow indication of whether the SS desires a full symbol allocation or a single subchannel allocation.

Table 12: OFDM contention codes

Contention Code Index	bit 0	bit 1	bit 2	bit 3
0	1	1	1	1
1	1	-1	1	-1
2	1	1	-1	-1
3	1	-1	-1	1
4	-1	-1	-1	-1
5	-1	1	-1	1
6	-1	-1	1	1
7	-1	1	1	-1

Table 13: OFDM contention channels

Contention Channel Index	frequency offset index 0	frequency offset index 1	frequency offset index 2	frequency offset index 3
0	-100	-50	+1	+51
1	-99	-49	+2	+52
2	-98	-48	+3	+53
...	...	...	...	...
k	k-100	k-50	k+1	k+51
...	...	...	...	...
48	-52	-2	+49	+99
49	-51	-1	+50	+100

7.3.2   Full Contention Transmission

If the chosen REQ Region is a REQ Region-Full, the SS shall transmit the short preamble as defined in clause 5.6, followed by a single OFDM symbol using the most robust mandatory burst profile which may contain a Bandwidth Request MAC Header.

If the Full Contention allocation appears in subchannelized region, the allocation is partitioned into Transmission Opportunities (TOs) both in frequency and in time. The width (in subchannels) and length (in OFDM symbols) of each transmission opportunity is defined in the UCD message defining UIUC=2. The transmission of an SS shall contain a subchannelized preamble corresponding to the TO chosen, followed by data OFDM symbols using the most robust mandatory burst profile.

7.3.3   Focused Contention Transmission

The REQ Region-Focused bandwidth requesting mechanism consists of two phases as defined in [1]. In the first phase, a SS requesting bandwidth shall send a signal to the BS in the UL interval of REQ Region Focused identified by UIUC=3. One REQ Region Focused UL TO with UIUC=3 shall be 4 subcarriers by two OFDM symbols. This bandwidth requesting signal transmission is described subsequently. In the second phase, allocated with UIUC=4, if the content of Focused_Contention_IE matches the TO used by the SS in the first phase, the SS may send a Bandwidth Request MAC Header using the most robust mandatory burst profile.

If the chosen REQ Region is a REQ Region-Focused, after choosing its four parameters, the SS shall transmit, during the chosen Transmit Opportunity in the chosen frame, four carriers which comprise the chosen contention channel. The amplitude of all other carriers shall be zero.

During both OFDM symbols, the amplitude of each of the four carriers shall be boosted somewhat above its normal amplitude, i.e. that used during a non-contention OFDM symbol, including the current power-control correction. The boost in dB shall equal the value of the Focused Contention Power Boost parameter in the current Uplink Channel Descriptor (UCD).

During the first OFDM symbol of the Transmit Opportunity, the phase of the four carriers is not specified.

During the second OFDM symbol of the Transmit Opportunity, the phases shall depend on the corresponding bit in the chosen contention code, and the phase transmitted during the first OFDM symbol on the same carrier. If the code bit is +1, the phase shall be the same as that transmitted during the first OFDM symbol. If the code bit is -1, the phase shall be inverted, 180 degrees with respect to the phase transmitted during the first OFDM symbol.

7.4   Power control

As with frequency control, a power control algorithm shall be supported for the uplink channel with both an initial calibration and periodic adjustment procedure without loss of data. The objective of the power control algorithm is to bring the received power density from a given subscriber to a desired level. The received power density is defined as total power received from a given subscriber divided by the number of active subcarriers. When subchannelization is not employed, the number of active subcarriers is equal for all the subscribers and the power control algorithm shall bring the total received power from a given subscriber to the desired level. The base station shall be capable of providing accurate power measurements of the received burst signal. This value can then be compared against a reference level, and the resulting error can be fed back to the subscriber station in a calibration message coming from the MAC sublayer. The power control algorithm shall be designed to support power attenuation due to distance loss or power fluctuations at rates of at most 30 dB/s with depths of at least 10 dB. The exact algorithm implementation is vendor-specific.

The total power control range consists of both a fixed portion and a portion that is automatically controlled by feedback. The power control algorithm shall take into account the interaction of the Radio Frequency (RF) power amplifier with different burst profiles. For example, when changing from one burst profile to another, margins should be maintained to prevent saturation of the amplifier and to prevent violation of emissions masks.

When subchannelization is employed the SS shall maintain the same transmitted power density unless the maximum power level is reached. That is, when the number of active subchannels allocated to a user is reduced, the total transmitted power shall be reduced proportionally by the SS, without additional power control messages. When the number of subchannels is increased the total transmitted power shall also be increased proportionally. However, the transmitted power level shall not exceed the maximum levels dictated by signal integrity considerations and regulatory requirements.

When subchannelization is employed, SS shall interpret power control messages as the required changes to the transmitted power density.

Subscriber stations shall report the maximum available power, and the normalized transmitted power. These parameters may be used by the Base station for optimal assignment of coding schemes and modulations and also for optimal allocation of subchannels. The algorithm is vendor specific. These parameters are reported in the REG-RSP message. The normalized transmitted power shall also reported in the REP-RSP message if the relevant flag in the REP-REQ message has been set. The current transmitted power is the power of the burst which carries the message. The maximum available power is reported for QPSK QAM16 and QAM64 constellations.

The current transmitted power and the maximum power parameters are reported in dBm. The parameters are quantized in 0,5 dBm steps ranging from -64 dBm (encoded 0x00) to 63,5 dBm (encoded 0xFF). Values outside this range shall be assigned the closest extreme. SSs that do not support QAM64 shall report the value of 0x00 in the maximum QAM64 power field.

8   Space-Time Coding (optional)

STC (see for example IEEE journal on select areas in communications (see bibliography)), may used on the downlink to provide 2nd order (Space) transmit diversity.

There are two transmit antennas on the BS side and one reception antenna on the SS side. This scheme requires Multiple Input Single Output channel estimation. Decoding is very similar to maximum ratio combining.

Figure 15 shows STC insertion into the OFDM chain. Each Tx antenna has its own OFDM chain, but they have the same Local Oscillator for synchronization purposes.
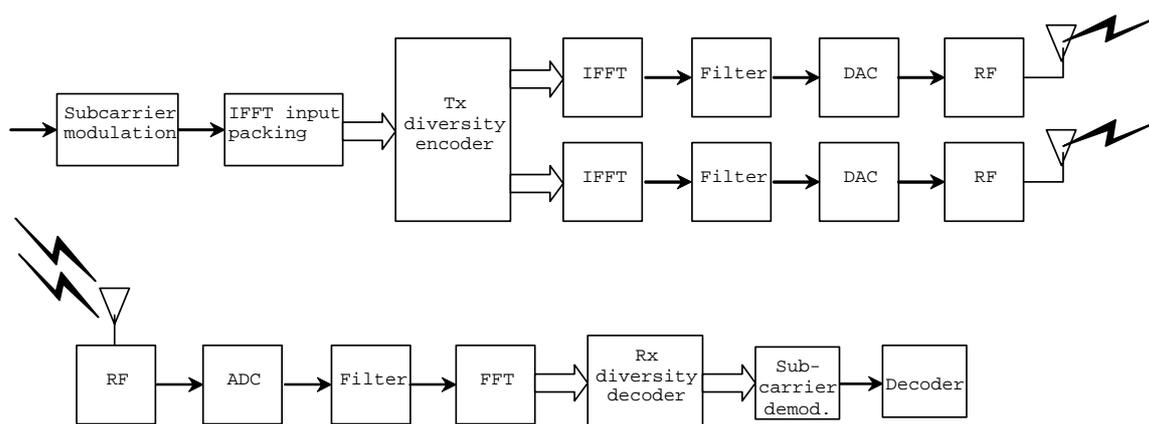
Figure 15: Illustration of STC

Both antennas transmit in the same time two different OFDM data symbols. Transmission is performed twice so as to decode and get 2nd order diversity. Time domain (Space-Time) repetition is used.

Both antennas transmit at the same time, and they share the same Local Oscillator. Thus, the received signal has exactly the same auto-correlation properties as for a single antenna. So, time and frequency coarse and fine estimation can be performed in the same way as for a single antenna. The scheme requires Multiple Input Single Output channel estimation, which is provisioned by starting the PHY burst following the STC_IE with a preamble transmitted from both antennas (see clause 5.6) to estimate the channel from both transmit antennas.

The basic scheme transmits two complex symbols s_0 and s_1, using the channel twice with channel vector values h_0 (for antenna 0) and h_1 (for antenna 1).

First channel use (i.e. first OFDM symbol): Antenna 0 transmits s_0, antenna 1 transmits s_1

Second channel use (i.e. second OFDM symbol): Antenna 0 transmits $-s_1^*$, antenna 1 transmits s_0^*.

Receiver gets r_0 (first channel use) and r_1 (second channel use) and computes s_0 and s_1 estimates:

$$\hat{s}_0 = h_0^* \times r_0 + h_1 \times r_1^* \tag{15}$$

$$\hat{s}_1 = h_1^* \times r_0 - h_0 \times r_1^* \tag{16}$$

These estimates benefit from 2nd order diversity as in the 1Tx-2Rx Maximum Ratio Combining scheme. OFDM symbols are processed by pairs. The precoding operation, and consecutively the receive decoding (as described in Equations 15 and 16), is applied independently to same-numbered subcarriers in two consecutive OFDM data symbols. Note that the two OFDM symbols may belong to different PHY bursts and even use different constellations. An individual PHY burst may contain any integer number of symbols. The aggregate duration of all PHY bursts following the last STC preamble or between any two STC preambles shall be a multiple of 2.

On a given pilot subcarrier, the same pilot symbol is used for the STC block. If the STC block consists of OFDM symbol k and $k+1$ and p_s is the pilot symbol for pilot subcarrier s as derived for OFDM symbol k from 5.4.2, then the modulation on pilot subcarrier s during OFDM symbol k shall be p_s on both antenna 0 and 1. During OFDM symbol $k+1$, it shall be $-p_s$ on antenna 0 and p_s on antenna 1.

9   Channel quality measurements

9.1   Introduction

Receive Signal Strength Indicator (RSSI) and Carrier to Interference Noise Ratio (CINR) signal quality measurements and associated statistics can aid in such processes as BS selection/assignment and burst adaptive profile selection. As channel behaviour is time-variant, both mean and standard deviation are defined. Implementation of the RSSI and CINR statistics and their reports is mandatory.

The process by which RSSI measurements are taken does not necessarily require receiver demodulation lock; for this reason, RSSI measurements offer reasonably reliable channel strength assessments even at low signal levels. On the other hand, although CINR measurements require receiver lock, they provide information on the actual operating condition of the receiver, including interference, noise levels, and signal strength.

9.2   RSSI mean and standard deviation

When the BS mandates collection of RSSI measurements, a SS shall obtain an RSSI measurement from the OFDM DL preambles. From a succession of RSSI measurements, the SS shall derive and update estimates of the mean and the standard deviation of the RSSI, and report them when solicited via REP-RSP messages.

Mean and standard deviation statistics shall be reported in units of dBm. To prepare such reports, statistics shall be quantized in 1 dB increments, ranging from -40 dBm (encoded 0x53) to -123 dBm (encoded 0x00). Values outside this range shall be assigned the closest extreme value within the scale.

The method used to estimate the RSSI of a single message is left to individual implementation, but the relative accuracy of a single signal strength measurement, taken from a single message, shall be ±2 dB, with an absolute accuracy of ±4 dB. The specified accuracy shall apply to the range of RSSI values starting from 6 dB below the sensitivity level of the most robust mode or -123 dBm (whichever is higher) up to -40 dBm. In addition, the range over which these single-message measurements are measured should extend 3 dB on each side beyond the -40 dBm to -123 dBm limits for the final averaged statistics that are reported. All measurements are taken at the antenna connector.

The (linear) mean RSSI statistics (in mW), derived from a multiplicity of single messages, shall be updated using

$$\hat{\mu}_{\text{RSSI}}[k] = \begin{cases} R[0] & k = 0 \\ (1 - \alpha_{\text{avg}})\hat{\mu}_{\text{RSSI}}[k\text{-}1] + \alpha_{\text{avg}}R[k] & k > 0 \end{cases} \quad \text{mW} \tag{17}$$

where k is the time index for the message (with the initial message being indexed by $k = 0$, the next message by $k = 1$, etc.), $R[k]$ is the RSSI in mW measured during message k, and α_{avg} is an averaging parameter specified by the BS. The mean estimate in dBm shall then be derived from:

$$\hat{\mu}_{\text{RSSI dBm}}[k] = 10 \log(\hat{\mu}_{\text{RSSI}}[k]) \quad \text{dBm} \tag{18}$$

To solve for the standard deviation in dB, the expectation-squared statistic shall be updated using:

$$\hat{x}^2_{\text{RSSI}}[k] = \begin{cases} |R[0]|^2 & k = 0 \\ (1 - \alpha_{\text{avg}})\hat{x}^2_{\text{RSSI}}[k\text{-}1] + \alpha_{\text{avg}}|R[k]|^2 & k > 0 \end{cases} \tag{19}$$

and the result applied to:

$$\hat{\sigma}_{\text{RSSI dB}} = 5 \log\left(\left|\hat{x}^2_{\text{RSSI}}[k] - \hat{\mu}^2_{\text{RSSI}}\right|\right) \quad \text{dB} \tag{20}$$

9.3   CINR mean and standard deviation

When Carrier to Interference Noise Ratio (CINR) measurements are mandated by the BS, a SS shall obtain a CINR measurement from the OFDM DL preamble. From a succession of these measurements, the SS shall derive and update estimates of the mean and the standard deviation of the CINR, and report them when solicited via REP-RSP messages

Mean and standard deviation statistics for CINR shall be reported in units of dB. To prepare such reports, statistics shall be quantized in 1 dB increments, ranging from a minimum of -10 dB (encoded 0x00) to a maximum of 53 dB (encoded 0x3F). Values outside this range shall be assigned the closest extreme value within the scale.

The method used to estimate the CINR of a single message is left to individual implementation, but the relative and absolute accuracy of a CINR measurement derived from a single message shall be ±1 dB and ±2 dB, respectively, for all input CINRs above 0 dB. In addition, the range over which these single-packet measurements are measured should extend 3 dB on each side beyond the -10 dB to 53 dB limits for the final reported, averaged statistics.

One possible method to estimate the CINR of a single message is to compute the ratio of signal power to residual error for each data sample, and then average the results from each data sample, using:

$$CINR[k] = \sum_{n=0}^{n-1} \frac{|s[k,n]|^2}{|r[k,n] - s[k,n]|^2} \qquad (21)$$

where $r[k,n]$ received sample n within message k; $s[k,n]$ the corresponding detected or pilot sample (with channel state weighting) corresponding to received symbol n.

The mean CINR statistic (in dB) shall be derived from a multiplicity of single messages using

$$\hat{\mu}_{\text{CINR dB}}[k] = 10\log(\hat{\mu}_{\text{CINR}}[k]) \qquad \text{dB} \qquad (22)$$

where

$$\hat{\mu}_{\text{CINR}}[k] = \begin{cases} CINR[0] & k = 0 \\ (1-\alpha_{\text{avg}})\hat{\mu}_{\text{CINR}}[k-1] + \alpha_{\text{avg}} CINR[k] & k > 0 \end{cases} \qquad (23)$$

k is the time index for the message (with the initial message being indexed by $k=0$, the next message by $k=1$, etc.); $CINR[k]$ is a linear measurement of CINR (derived by any mechanism which delivers the prescribed accuracy) for message k; and α_{avg} is an averaging parameter specified by the BS.

To solve for the standard deviation, the expectation-squared statistic shall be updated using

$$\hat{x}^2_{\text{CINR}}[k] = \begin{cases} |CINR[0]|^2 & k = 0 \\ (1-\alpha_{\text{avg}})\hat{x}^2_{\text{RSSI}}[k-1] + \alpha_{\text{avg}}|CINR[k]|^2 & k > 0 \end{cases} \qquad (24)$$

and the result applied to

$$\hat{\sigma}_{\text{CINR dB}} = 5\log\left(\left|\hat{x}^2_{\text{RSSI}}[k] - \hat{\mu}^2_{\text{RSSI}}\right|\right) \qquad \text{dB} \qquad (25)$$

10   Transmitter requirements

10.1   Transmitter channel bandwidth

Transmitter channel bandwidths allowed shall be limited to the regulatory provisioned bandwidth divided by any integer rounded down to the nearest multiple of 250 kHz, resulting in channel bandwidths no less than 1,25 MHz.

If the resulting bandwidth is an odd multiple of 250 kHz, then for any band for which support is claimed, the RF carrier shall only be tunable to every odd multiple of 125 kHz within that band. If the resulting channel bandwidth is an even multiple of 250 kHz, then for any band for which support is claimed, the RF carrier shall only be tunable to every even multiple of 125 kHz within that band. For FDD systems, support shall be claimed separately for UL and DL.

10.2 Transmit power level control

The transmitter shall support monotonic power level control of 30 dB minimum with a minimum step size of 1 dB and a relative accuracy of $\pm$ 0,5 dB for a SS and 10 dB minimum with a minimum step size of 1 dB and a relative accuracy of $\pm$ 0,5 dB for a BS.

10.2.1 Transmitter spectral flatness

The average energy of the constellations in each of the spectral lines shall deviate no more than indicated in table 14. The absolute difference between adjacent carriers shall not exceed 0,06 dB. This data shall be taken from the channel estimation step.

Table 14: OFDM Spectral Flatness

Spectral Lines	Spectral Flatness
Spectral lines from -50 to -1 and +1 to +50	$\pm$2dB from the measured energy averaged over all 200 active tones
Spectral lines from -100 to -50 and +50 to +100	+2/-4dB from the measured energy averaged over all 200 active tones

10.2.2 Transmitter constellation error and test method

To ensure that the receiver SNR does not degrade more than 0,5 dB due to the transmitter SNR, the relative constellation RMS error, averaged over carriers, OFDM frames, and packets, shall not exceed a burst profile dependent value according to table 15.

Table 15: Allowed relative constellation error versus data rate

Burst type	Relative constellation error (dB)
QPSK-1/2	-16
QPSK-3/4	-18,5
16QAM-1/2	-21,5
16QAM-3/4	-25
64QAM-2/3	-28,5
64QAM-3/4	-31

All measurement errors taken together shall be 10 dB less than the required noise level, i.e. if a specification is TX S/N = 10 dB, the measurement S/N should be at least 20 dB. Measurements shall be taken with all non-guard carriers active.

The sampled signal shall be processed in a manner similar to an actual receiver, according to the following steps, or an equivalent procedure:

a)   Start of frame shall be detected.

b)   Transition from short sequences to channel estimation sequences shall be detected, and fine timing (with one sample resolution) shall be established.

c)   Coarse and fine frequency offsets shall be estimated.

d)   The packet shall be de-rotated according to estimated frequency offset.

e)   The complex channel response coefficients shall be estimated for each of the carriers.

f) For each of the data OFDM symbols: transform the symbol into carrier received values, estimate the phase from the pilot carriers, de-rotate the carrier values according to estimated phase, and divide each carrier value with a complex estimated channel response coefficient.

g) For each data-carrying carrier, find the closest constellation point and compute the Euclidean distance from it.

h) Compute the RMS average of all errors in a packet. It is given by:

$$\text{Error}_{\text{RMS}} = \frac{\displaystyle\sum_{i=1}^{N_{\text{f}}} \frac{\displaystyle\sum_{j=1}^{L_{\text{P}}}\left[\sum_{k=1}^{N_{\text{FFT}}}\left\{[I(i,j,k)-I_0(i,j,k)]^2 + [Q(i,j,k)-Q_0(i,j,k)]^2\right\}\right]}{P_0 \times L_{\text{P}} \times N_{\text{FFT}}}}{N_{\text{f}}} \qquad (26)$$

where

L_{P} is the length of the packet;

N_{f} is the number of frames for the measurement;

$\{I_0(i,j,k), Q_0(i,j,k)\}$ denotes the ideal symbol point of the ith frame, j^{th} OFDM symbol of the

frame, k^{th} carrier of the OFDM symbol in the complex plane;

$\{I(i,j,k), Q(i,j,k)\}$ denotes the point of the ith frame, j^{th} OFDM symbol of the frame, k^{th} carrier of the OFDM symbol in the complex plane;

P_0 is the average power of the constellation.

11 Receiver requirements

11.1 Receiver Sensitivity

The bit error rate (BER) shall be less than 10^{-6} at the power levels indicated by Equation 27 for standard message and test conditions. The minimum input levels are measured as follows:

- at the antenna connector or· through a calibrated radiated test environment;

- using the defined standardized message packet formats; and

- using an AWGN channel.

The receiver minimum input level sensitivity (*RSS*) shall be better than (assuming 5 dB implementation margin and 7 dB Noise figure):

$$RSS = -102 + SNR_{\text{Rx}} + 10 \cdot \log_{10}\left(BW \cdot R_{os} \cdot \frac{N_{\text{used}}}{N_{\text{FFT}}} \cdot \frac{N_{\text{subchannels}}}{16}\right) \qquad (27)$$

where:

SNR_{Rx} the assumed receiver SNR as per table 16 in dB

$N_{subchannels}$ the number of allocated subchannels (default 16 if no subchannelization is used).

Table 16: Receiver SNR assumptions

Modulation	Coding rate	Receiver SNR (dB)
QPSK	1/2	6
	3/4	8.5
16QAM	1/2	11,5
	3/4	15
64QAM	2/3	18,5
	3/4	21

Test messages for measuring Receiver Sensitivity shall be based on a continuous stream of MAC PDUs, each with a payload consisting of an R times repeated sequence $S_{modulation}$. For each modulation, a different sequence applies:

$S_{QPSK} =$ [0xE4, 0xB1, 0xE1, 0xB4]

$S_{16QAM} =$ [0xA8, 0x20, 0xB9, 0x31, 0xEC, 0x64, 0xFD, 0x75]

$S_{64QAM} =$ [0xB6, 0x93, 0x49, 0xB2, 0x83, 0x08, 0x96, 0x11, 0x41, 0x92, 0x01, 0x00, 0xBA, 0xA3, 0x8A, 0x9A, 0x21, 0x82, 0xD7, 0x15, 0x51, 0xD3, 0x05, 0x10, 0xDB, 0x25, 0x92, 0xF7, 0x97, 0x59, 0xF3, 0x87, 0x18, 0xBE, 0xB3, 0xCB, 0x9E, 0x31, 0xC3, 0xDF, 0x35, 0xD3, 0xFB, 0xA7, 0x9A, 0xFF, 0xB7, 0xDB]

For each mandatory test message, the (R,$S_{modulation}$) tuples that shall apply are:

   Short length test message payload (288 data bytes): (72,S_{QPSK}), (36,S_{16QAM}), (6,S_{64QAM})

   Mid length test message payload (864 data bytes): (216,S_{QPSK}), (108,S_{16QAM}), (18,S_{64QAM})

   Long length test message payload (1536 data bytes): (384,S_{QPSK}), (192,S_{16QAM}), (32,S_{64QAM})

The test condition requirements are: ambient room temperature, shielded room, conducted measurement at the RF port if available, radiated measurement in a calibrated test environment if the antenna is integrated, and RS FEC is enabled. The test shall be repeated for each test message length and for each (R,$S_{modulation}$) tuple as identified above, using the mandatory FEC scheme. The results shall meet or exceed the sensitivity requirements indicated by Equation 27.

11.2   Receiver adjacent and alternate channel rejection

The adjacent channel rejection and alternate channel rejection shall be measured by setting the desired signal's strength n×3dB (n being a positive non-zero integer) above the rate dependent receiver sensitivity (see Equation 27) until the maximum input signal level is reached, and raising the power level of the interfering signal until the specified error rate of 1×10^{-6} is obtained. The power difference between the interfering signal and the desired signal is the corresponding adjacent channel rejection. The interfering signal in the adjacent channel shall be a conforming OFDM signal, unsynchronized with the signal in the channel under test. For non-adjacent channel testing the test method is identical except the interfering channel shall be any channel other than the adjacent channel or the co-channel.

The minimum rejection shall exceed the levels shown in table 17.

Table 17: Adjacent and Non-Adjacent Channel rejection

Modulation/coding	Adjacent channel rejection (dB)	Non-adjacent channel rejection (dB)
16QAM-3/4	+11 dBr or -30 dBm (see note)	+30 dBr or -30 dBm (see note)
64QAM-2/3	+4 dBr or -30 dBm (see note)	+23 dBr or -30 dBm (see note)
NOTE: Whichever is less. "dBr" indicates an interferer level relative to the desired signal's strength, whereas "dBm" indicates an absolute level. All levels are measured at the antenna connector.		

11.3   Receiver maximum input signal

The receiver shall be capable of receiving a maximum on-channel signal of -30 dBm, and shall tolerate a maximum signal of 0 dBm without damage. Power levels are measured at the antenna connector.

11.4   Receiver linearity

The receiver shall have a minimum Input Intercept Point (IIP3) of -10 dBm.

11.5   Out-of-Band signal rejection

The receiver's immunity for out-of-band signals shall be measured by setting the desired signal's strength to a given level, and raising the power level of the out-of-band signal until the specified BER of 10^{-6} is obtained. The out-of-band signal shall be a CW signal.

For the PHY to be compliant, the minimum out-of-band immunity level shall exceed the values of table 18, for the levels of the desired input signal: receiver minimum input sensitivity level+ n×3dB (n being a positive, non-zero integer) until the maximum input signal level is reached.

Table 18: Out-of-Band Signal Immunity Levels

Out-of-Band frequency	Minimum Out-of-Band immunity level
$|f_{blocker} - f_{signal}| < 100$ MHz	Idem to the interference levels of 12.2.
100 MHz $< |f_{blocker} - f_{signal}| < 1$ GHz	-30 dBm or +20 dBr (see notes 1 and 2)
1GHz $< |f_{blocker} - f_{signal}|$	-10 dBm or +30 dBr (see note 1)
NOTE 1: Whichever of the two is larger. "dBr" indicates a level relative to the desired input signal's level, whereas "dBm" indicates an absolute level.	
NOTE 2: Up to two out-of-band frequency bands with a bandwidth of 30 MHz are tolerated, where this specification is relaxed by 10 dB.	

All power levels are measured at the antenna connector.

11.6   Spurious emissions

Emissions of HiperMan equipment made outside the time of transmitted bursts shall not exceed the following values.

Table 19: Spurious emissions

Frequency Range	Maximum power	Measurement bandwidth
30 MHz to 1 GHz	-57 dBm	100 kHz
1 GHz to 26,5 GHz	-50 dBm	1 MHz

The indicated bandwidths are nominal -3 dB bandwidths. All power levels measured at the antenna connector.

12   Frequency and timing requirements

The BS transmitted centre frequency, receive centre frequency, and the symbol clock frequency shall be derived from the same reference oscillator. The reference frequency tolerance shall be ±4ppm.

At the BS, the transmitted centre frequency, receive centre frequency and the symbol clock frequency shall be derived from the same reference oscillator. At the BS the reference frequency tolerance shall be ±4 ppm.

The SS shall synchronize its transmitted centre frequency and symbol clock frequency to the BS with a maximum tolerance of 2 % of the FFT carrier spacing.

At the SS, both the transmitted centre frequency and the symbol clock frequency shall be synchronized to the BS with a tolerance of maximum 2 % of the carrier spacing.

For mesh capable devices, all devices shall have a ±6 ppm maximum frequency tolerance and achieve synchronization to its neighbouring nodes with a tolerance of maximum 3 % of the carrier spacing.

During the synchronization period, the SS shall acquire frequency synchronization within the specified tolerance before attempting any uplink transmission. During normal operation, the SS shall track the frequency changes and shall defer any transmission if synchronization is lost.

All SSs shall acquire and adjust their timing such that all uplink OFDM symbols arrive time coincident at the Base-Station to an accuracy of $\pm$ 50 % of the minimum guard-interval or better.

13   Parameters and constants

A number of parameters and constants are defined, which are used by the DLC.

- Physical Slot (PS)

$$PS = \frac{4}{\Delta f \cdot N_{FFT}} \qquad (28)$$

- Timing Adjust Units

$$\text{Timing Adjust Unit} = \frac{1}{\Delta f \cdot N_{FFT}} \qquad (29)$$

Annex A (informative):
Bibliography

Alamouti, S.M., "A Simple Transmit Diversity Technique for Wireless Communications", IEEE journal on select areas in communications, Vol.16, No. 8, pages 1451-1458, October 1998.

History

Document history		
V1.1.1	November 2003	Publication