ETSI EN 302 307 V1.3.1 (2013-03)



Digital Video Broadcasting (DVB);
Second generation framing structure, channel coding and modulation systems for Broadcasting,
Interactive Services, News Gathering and other broadband satellite applications (DVB-S2)





Reference

REN/JTC-DVB-319

Keywords

BSS, digital, DVB, modulation, satellite, TV

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Siret N° 348 623 562 00017 - NAF 742 C Association à but non lucratif enregistrée à la Sous-Préfecture de Grasse (06) N° 7803/88

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Foreword

This European Standard (EN) has been produced by Joint Technical Committee (JTC) Broadcast of the European Broadcasting Union (EBU), Comité Européen de Normalisation ELECtrotechnique (CENELEC) and the European Telecommunications Standards Institute (ETSI).

The work of the JTC was based on the studies carried out by the European DVB Project under the auspices of the Ad Hoc Group on DVB-S2 of the DVB Technical Module. This joint group of industry, operators and broadcasters provided the necessary information on all relevant technical matters (see bibliography).

NOTE:

The EBU/ETSI JTC Broadcast was established in 1990 to co-ordinate the drafting of standards in the specific field of broadcasting and related fields. Since 1995 the JTC Broadcast became a tripartite body by including in the Memorandum of Understanding also CENELEC, which is responsible for the standardization of radio and television receivers. The EBU is a professional association of broadcasting organizations whose work includes the co-ordination of its members' activities in the technical, legal, programme-making and programme-exchange domains. The EBU has active members in about 60 countries in the European broadcasting area; its headquarters is in Geneva.

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The Digital Video Broadcasting Project (DVB) is an industry-led consortium of broadcasters, manufacturers, network operators, software developers, regulatory bodies, content owners and others committed to designing global standards for the delivery of digital television and data services. DVB fosters market driven solutions that meet the needs and economic circumstances of broadcast industry stakeholders and consumers. DVB standards cover all aspects of digital television from transmission through interfacing, conditional access and interactivity for digital video, audio and data. The consortium came together in 1993 to provide global standardisation, interoperability and future proof specifications.

National transposition dates						
Date of adoption of this EN:	1 March 2013					
Date of latest announcement of this EN (doa):	30 June 2013					
Date of latest publication of new National Standard or endorsement of this EN (dop/e):	31 December 2013					
Date of withdrawal of any conflicting National Standard (dow):	31 December 2013					

1 Scope

DVB-S (EN 300 421 [2]) was introduced as a standard in 1994 and DVB-DSNG (EN 301 210 [3]) in 1997. The DVB-S standard specifies QPSK modulation and concatenated convolutional and Reed-Solomon channel coding, and is now used by most satellite operators worldwide for television and data broadcasting services. DVB-DSNG specifies, in addition to DVB-S format, the use of 8PSK and 16QAM modulation for satellite news gathering and contribution services.

Since 1997, digital satellite transmission technology has evolved somewhat:

- New channel coding schemes, combined with higher order modulation, promise more powerful alternatives to the DVB-S/DVB-DSNG coding and modulation schemes. The result is a capacity gain in the order of 30 % at a given transponder bandwidth and transmitted EIRP, depending on the modulation type and code rate.
- Variable Coding and Modulation (VCM) may be applied to provide different levels of error protection to different service components (e.g. SDTV and HDTV, audio, multimedia).
- In the case of interactive and point-to-point applications, the VCM functionality may be combined with the use of return channels, to achieve Adaptive Coding and Modulation (ACM). This technique provides more exact channel protection and dynamic link adaptation to propagation conditions, targeting each individual receiving terminal. ACM systems promise satellite capacity gains of up to 100 % to 200 %. In addition, service availability may be extended compared to a constant protection system (CCM) such as DVB-S or DVB-DSNG. Such gains are achieved by informing the satellite up-link station of the channel condition (e.g. C/N+I) of each receiving terminal via the satellite or terrestrial return channels.
- DVB-S and DVB-DSNG are strictly focused on a unique data format, the MPEG Transport Stream (ISO/IEC 13818-1 [1] or a reference to it). Extended flexibility to cope with other input data formats (such as multiple Transport Streams, or generic data formats) is now possible without significant complexity increase.

The present document defines a "second generation" modulation and channel coding system (denoted the "System" or "DVB-S2" for the purposes of the present document) to make use of the improvements listed above. DVB-S2 is a single, very flexible standard, covering a variety of applications by satellite, as described below. It is characterized by:

- a flexible input stream adapter, suitable for operation with single and multiple input streams of various formats (packetized or continuous);
- a powerful FEC system based on LDPC (Low-Density Parity Check) codes concatenated with BCH codes, allowing Quasi-Error-Free operation at about 0,7 dB to 1 dB from the Shannon limit, depending on the transmission mode (AWGN channel, modulation constrained Shannon limit);
- a wide range of code rates (from 1/4 up to 9/10); 4 constellations, ranging in spectrum efficiency from 2 bit/s/Hz to 5 bit/s/Hz, optimized for operation over non-linear transponders;
- a set of three spectrum shapes with roll-off factors 0,35, 0,25 and 0,20;
- Adaptive Coding and Modulation (ACM) functionality, optimizing channel coding and modulation on a frame-by-frame basis.

The System has been optimized for the following broadband satellite applications:

Broadcast Services (BS) Digital multi-programme Television (TV)/High Definition Television (HDTV)

Broadcasting services to be used for primary and secondary distribution in the Fixed Satellite Service (FSS) and the Broadcast Satellite Service (BSS) bands.

DVB-S2 is intended to provide Direct-To-Home (DTH) services for consumer Integrated Receiver Decoder (IRD), as well as collective antenna systems (Satellite Master Antenna Television - SMATV) and cable television head-end stations (possibly with remodulation, see EN 300 429 [5]). DVB-S2 may be considered a successor to the current DVB-S standard EN 300 421 [2], and may be introduced for new services and allow for a long-term migration. BS services are transported in MPEG Transport Stream format. VCM may be applied on multiple transport stream to achieve a differentiated error protection for different services (TV, HDTV, audio, multimedia). Two modes are available:

- NBC-BS (Non Backwards Compatible Broadcast Services) is not backwards-compatible with EN 300 421 [2].
- **BC-BS** (Backwards-Compatible Broadcast Services) is backwards-compatible to EN 300 421 [2] (see annex F).

In fact, with a large number of DVB-S receivers already installed, backwards compatibility may be required for a period of time, where old receivers continue to receive the same capacity as before, while the new DVB-S2 receivers could receive additional capacity broadcasts. When the complete receiver population has migrated to DVB-S2, the transmitted signal can be modified to a non-backward compatible mode, thus exploiting the full potential of DVB-S2. To facilitate the reception of DVB-S services by DVB-S2 receivers, implementation of DVB-S in DVB-S2 chips is highly recommended.

Interactive Services (IS) Interactive data services including Internet access

DVB-S2 is intended to provide interactive services to consumer IRDs and to personal computers, where DVB-S2's forward path supersedes the current DVB-S standard EN 300 421 [2] for interactive systems. The return path can be implemented using various DVB interactive systems, such as DVB-RCS (EN 301 790 [6]), DVB-RCP (ETS 300 801 [7]), DVB-RCG (EN 301 195 [8]), DVB-RCC (ES 200 800 [9]). Data services are transported in (single or multiple) Transport Stream format according to EN 301 192 [4] (e.g. using Multiprotocol Encapsulation), or in (single or multiple) generic stream format. DVB-S2 can provide Constant Coding and Modulation (CCM), or Adaptive Coding and Modulation (ACM), where each individual satellite receiving station controls the protection mode of the traffic addressed to it. Input Stream Adaptation for ACM is specified in annex D.

Digital TV Contribution and Satellite News Gathering (DTVC/DSNG)

Digital television contribution applications by satellite consist of point-to-point or point-to-multipoint transmissions, connecting fixed or transportable uplink and receiving stations. They are not intended for reception by the general public. According to ITU-R Recommendation SNG.770-1 [10], SNG is defined as "Temporary and occasional transmission with short notice of television or sound for broadcasting purposes, using highly portable or transportable uplink earth stations ...". Services are transported in single (or multiple) MPEG Transport Stream format. DVB-S2 can provide Constant Coding and Modulation (CCM), or Adaptive Coding and Modulation (ACM). In this latter case, a single satellite receiving station typically controls the protection mode of the full multiplex. Input Stream Adaptation for ACM is specified in annex D.

Data content distribution/trunking and other professional applications (PS)

These services are mainly point-to-point or point-to-multipoint, including interactive services to professional head-ends, which re-distribute services over other media. Services may be transported in (single or multiple) generic stream format. The system can provide Constant Coding and Modulation (CCM), Variable Coding and Modulation (VCM) or Adaptive Coding and Modulation (ACM). In this latter case, a single satellite receiving station typically controls the protection mode of the full TDM multiplex, or multiple receiving stations control the protection mode of the traffic addressed to each one. In either case, interactive or non-interactive, the present document is only concerned with the forward broadband channel.

DVB-S2 is suitable for use on different satellite transponder bandwidths and frequency bands. The symbol rate is matched to given transponder characteristics, and, in the case of multiple carriers per transponder (FDM), to the frequency plan adopted. Examples of possible DVB-S2 use are given in clause H.1.

Annex M specifies the implementation of a DVB-S2 profile suitable for operation in wide-band mode, without requiring a full-speed decoding of the total carrier capacity, by suitably mapping the transmitted services in time-slices.

Digital transmissions via satellite are affected by power and bandwidth limitations. Therefore DVB-S2 provides for many transmission modes (FEC coding and modulations), giving different trade-offs between power and spectrum efficiency (see clause H.1). For some specific applications (e.g. broadcasting) modes such as QPSK and 8PSK, with their quasi-constant envelope, are appropriate for operation with saturated satellite power amplifiers (in single carrier per transponder configuration). When higher power margins are available, spectrum efficiency can be further increased to reduce bit delivery cost. In these cases also 16APSK and 32APSK can operate in single carrier mode close to the satellite HPA saturation by pre-distortion techniques. All the modes are appropriate for operation in quasi-linear satellite channels, in multi-carrier Frequency Division Multiplex (FDM) type applications.

DVB-S2 is compatible with Moving Pictures Experts Group (MPEG-2 and MPEG-4) coded TV services (see ISO/IEC 13818-1 [1]), with a Transport Stream packet multiplex. Multiplex flexibility allows the use of the transmission capacity for a variety of TV service configurations, including sound and data services. All service components are Time Division Multiplexed (TDM) on a single digital carrier.

The present document:

- gives a general description of the DVB-S2 system;
- specifies the digitally modulated signal in order to allow compatibility between pieces of equipment developed
 by different manufacturers. This is achieved by describing in detail the signal processing principles at the
 modulator side, while the processing at the receive side is left open to different implementation solutions.
 However, it is necessary in the present document to refer to certain aspects of reception;
- identifies the global performance requirements and features of the System, in order to meet the service quality targets.

2 References

References are either specific (identified by date of publication and/or edition number or version number) or non-specific. For specific references, only the cited version applies. For non-specific references, the latest version of the reference document (including any amendments) applies.

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

NOTE: While any hyperlinks included in this clause were valid at the time of publication, ETSI cannot guarantee their long term validity.

2.1 Normative references

The following referenced documents are necessary for the application of the present document.

- [1] ISO/IEC 13818 (parts 1 and 2): "Information technology -- Generic coding of moving pictures and associated audio information".
- [2] ETSI EN 300 421 (V.1.1.2): "Digital Video Broadcasting (DVB); Framing structure, channel coding and modulation for 11/12 GHz satellite services".
- [3] ETSI EN 301 210: "Digital Video Broadcasting (DVB); Framing structure, channel coding and modulation for Digital Satellite News Gathering (DSNG) and other contribution applications by satellite".
- [4] ETSI EN 301 192: "Digital Video Broadcasting (DVB); DVB specification for data broadcasting".
- [5] ETSI EN 300 429: "Digital Video Broadcasting (DVB); Framing structure, channel coding and modulation for cable systems".
- [6] ETSI EN 301 790: "Digital Video Broadcasting (DVB); Interaction channel for satellite distribution systems".
- [7] ETSI ETS 300 801: "Digital Video Broadcasting (DVB); Interaction channel through Public Switched Telecommunications Network (PSTN)/ Integrated Services Digital Networks (ISDN)".

[8]	ETSI EN 301 195: "Digital Video Broadcasting (DVB); Interaction channel through the Global System for Mobile communications (GSM)".
[9]	ETSI ES 200 800: "Digital Video Broadcasting (DVB); DVB interaction channel for Cable TV distribution systems (CATV)".
[10]	ITU-R Recommendation SNG.770-1: "Uniform operational procedures for satellite news gathering (SNG)".
[11]	ETSI ETS 300 802: "Digital Video Broadcasting (DVB); Network-independent protocols for DVB interactive services".
[12]	ETSI EN 300 468: "Digital Video Broadcasting (DVB); Specification for Service Information (SI) in DVB systems".

2.2 Informative references

The following referenced documents are not necessary for the application of the present document but they assist the user with regard to a particular subject area.

[i.1]	ETSI TS 102 005: "Digital Video Broadcasting (DVB); Specification for the use of Video and Audio Coding in DVB services delivered directly over IP protocols".
[i.2]	ETSI EN 300 744: "Digital Video Broadcasting (DVB); Framing structure, channel coding and modulation for digital terrestrial television".
[i.3]	ETSI TR 101 154: "Digital Video Broadcasting (DVB); Implementation guidelines for the use of MPEG-2 Systems, Video and Audio in satellite, cable and terrestrial broadcasting applications".
[i.4]	ETSI ETR 162: "Digital Video Broadcasting (DVB); Allocation of Service Information (SI) codes for DVB systems".

3 Symbols and abbreviations

3.1 Symbols

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

α	Roll-off factor
γ	Ratio between constellation radii for 16APSK and 32APSK
c	codeword
C/N	Carrier-to-noise power ratio (N measured in a bandwidth equal to symbol rate)
C/N+I	Carrier-to-(Noise+Interference) ratio
$d_{n_{bch}-k_{bch}-1}, d_{n_{bch}-k_{bch}-2},, d_1, d_0$	BCH code redundancy bits
d(x)	BCH code remainder of the division between the generator polynomial and
	$x^{n_{bch}-k_{bch}} m(x)$
DFL	Data Field Length
dmin	LDPC code minimum distance
E_b/N_0	Ratio between the energy per information bit and single sided noise power
	spectral density
E_s/N_0	Ratio between the energy per transmitted symbol and single sided noise power
	spectral density
f_N	Nyquist frequency
f_0	Carrier frequency
G	PLS code generator matrix
g(x)	code generator polynomial
$g_1(x), g_2(x),, g_{12}(x)$	polynomials to obtain BCH code generator polynomial
01(), 07(),, 017()	polynomials to detain 2011 code generator porynomial

 $\begin{array}{ll} H(f) & RC \text{ filters frequency transfer function} \\ H_{(n-k)xn} & LDPC \text{ code parity check matrix} \end{array}$

I, Q In-phase, Quadrature phase components of the modulated signal

 $\begin{array}{cc} {\rm K_{bch}} & {\rm number\ of\ bits\ of\ BCH\ uncoded\ Block} \\ {\rm N_{bch}} & {\rm number\ of\ bits\ of\ BCH\ coded\ Block} \\ k_{\rm ldpc} & {\rm number\ of\ bits\ of\ LDPC\ uncoded\ Block} \\ n_{\rm ldnc} & {\rm number\ of\ bits\ of\ LDPC\ coded\ Block} \end{array}$

η PLFRAMING efficiency

 η_c code efficiency

 η_{MOD} number of transmitted bits per constellation symbol

 $\begin{array}{ll} \eta_{\rm tot} & {\rm System~spectral~efficiency} \\ {\rm m} & {\rm BCH~code~information~word} \\ {\rm m(x)} & {\rm BCH~code~message~polynomial} \\ m_{k_{bch}-1}, m_{k_{bch}-2}, ..., m_1, m_0 & {\rm BCH~code~information~bits} \end{array}$

M number of modulated symbols in SLOT

 p_0 , p_1 ,... $p_{n_{ldnc}-k_{ldnc}-1}$ LDPC code parity bits

 $\begin{array}{ll} P & \text{number of pilot symbols in a pilot block} \\ q & \text{code rate dependant constant for LDPC codes} \\ \theta & \text{deviation angle in hierarchical constellations} \end{array}$

 $r_{\rm m}$ In-band ripple (dB)

R_s Symbol rate corresponding to the bilateral Nyquist bandwidth of the

modulated signal

R_u Useful bit rate at the DVB-S2 system input S Number of Slots in a XFECFRAME

 T_s Symbol period

3.2 Abbreviations

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

16APSK 16-ary Amplitude and Phase Shift Keying 32APSK 32-ary Amplitude and Phase Shift Keying

8PSK 8-ary Phase Shift Keying

ACM Adaptive Coding and Modulation APSK Amplitude Phase Shift Keying ASI Asynchronous Serial Interface AWGN Additive White Gaussian Noise

BB BaseBand

BC Backwards-Compatible

NOTE: Referred to the system allowing partial stream reception by DVB-S receivers.

BCH Bose-Chaudhuri-Hocquenghem multiple error correction binary block code

BER Bit Error Ratio

BPSK Binary Phase Shift Keying

B_S Bandwidth of the frequency Slot allocated to a service

BS Broadcast Service

BSS Broadcast Satellite Service

BW BandWidth (at -3 dB) of the transponder

CBR Constant Bit Rate

CCM Constant Coding and Modulation
CNI Carrier to Noise plus Interference ratio

CRC Cyclic Redundancy Check

D Decimal notation

Decision Directed DD **DEMUX** DEMUltipleXer Data Field DF Data Field Length DFL DNP Deleted Null Packets

DSNG Digital Satellite News Gathering

DTH Direct To Home

DTT Digital Terrestrial Television DVB Digital Video Broadcasting project DVB-S DVB System for satellite broadcasting

NOTE: As specified in EN 300 421 [2].

DVB-S2 second generation DVB System for satellite broadcasting and unicasting

European Broadcasting Union **EBU** Equivalent Isotropic Radiated Power **EIRP**

EN European Norm

FDM Frequency Division Multiplex Forward Error Correction **FEC**

FIFO First In First Out Fixed Satellite Service **FSS**

GF Galois Field Generic Stream GS

HDTV High Definition TeleVision HEXadecimal notation **HEX**

HP **High Priority**

HPA High Power Amplifier **IBO** Input Back Off

Intermediate Frequency IF

IMUX Input MUltipleXer - filter

Internet Protocol IΡ

Integrated Receiver Decoder **IRD**

Interactive Services IS

Input Stream Clock Reference **ISCR** ISI Input Stream Identifier **ISSY** Input Stream SYnchronizer

ISSYI Input Stream SYnchronizer Indicator International Telecommunications Union ITU LDPC Low Density Parity Check (codes)

Low Noise Block LNB LP Low Priority LSB Least Significant Bit

LTWTA Linearized Travelling Wave Tube Amplifier

Mode Adaptation MA Multiple Input Stream MIS **MPE** Multi-Protocol Encapsulation **MPEG** Moving Pictures Experts Group

MSB Most Significant Bit

NOTE: In DVB-S2 the MSB is always transmitted first.

MUX MUltipleX Not Applicable NA

Non-Backwards-Compatible **NBC NCR** Network Clock Reference

NP **Null Packets**

Null-Packet Deletion NPD OBO Output Back Off OCTal notation OCT

OMUX Output MUltipleXer - filter **PAT** Program Association Table **PER** (MPEG TS) Packet Error Rate

PID Packet IDentifier PL Physical Layer
PLL Phase-Locked Loop
PLS Physical Layer Signalling
PMT Program Map Table

PRBS Pseudo Random Binary Sequence

PS Professional Services
PSK Phase Shift Keying
QEF Quasi-Error-Free

QPSK Quaternary Phase Shift Keying RCS Return Channel via Satellite

RF Radio Frequency RO Roll-Off

SA Stream Adaptation

SDTV Standard Definition TeleVision

SI Service Information
SIS Single Input Stream

SMATV Satellite Master Antenna TeleVision

SNG Satellite News Gathering
SNR Signal to Noise Ratio
SOF Start Of Frame
SSA Solid State Amplifier
SSB Single SideBand
TDM Time Division Multiplex

TS Transport Stream

TSDT Transport Stream Descriptor Table
TS/GS Transport Stream/Generic Stream
TSN Time Slice Number (See Annex M)

TV TeleVision

TWT Travelling Wave Tube

TWTA Travelling Wave Tube Amplifier

UP User Packet
UPL User Packet Length
VBR Variable Bit Rate

VCM Variable Coding and Modulation

4 Transmission system description

4.1 System definition

The System is defined as the functional block of equipment performing the adaptation of the baseband digital signals, from the output of a single (or multiple) MPEG transport stream multiplexer(s) (ISO/IEC 13818-1 [1]), or from the output of a single (or multiple) generic data source(s), to the satellite channel characteristics. The System is designed to support source coding as defined in ISO/IEC 13818 [1], TR 101 154 [i.3] and TS 102 005 [i.1]. Data services may be transported in Transport Stream format according to EN 301 192 [4] (e.g. using Multi-protocol Encapsulation), or Generic Stream format.

If the received signal is above the C/N+I threshold, the Forward Error Correction (FEC) technique adopted in the System is designed to provide a "Quasi Error Free" (QEF) quality target. The definition of QEF adopted for DVB-S2 is "less than one uncorrected error-event per transmission hour at the level of a 5 Mbit/s single TV service decoder", approximately corresponding to a Transport Stream Packet Error Ratio PER< 10^{-7} before de-multiplexer.

4.2 System architecture

According to figure 1, the DVB-S2 System shall be composed of a sequence of functional blocks as described below.

Mode adaptation shall be application dependent. It shall provide input stream interfacing, Input Stream Synchronization (optional), null-packet deletion (for ACM and Transport Stream input format only), CRC-8 coding for error detection at packet level in the receiver (for packetized input streams only), merging of input streams (for Multiple Input Stream modes only) and slicing into DATA FIELDs. For Constant Coding and Modulation (CCM) and single input Transport Stream, Mode Adaptation shall consist of a "transparent" DVB-ASI (or DVB-parallel) to logical-bit conversion and CRC-8 coding. For Adaptive Coding and Modulation (ACM), Mode Adaptation shall be according to annex D.

A Base-Band Header shall be appended in front of the Data Field, to notify the receiver of the input stream format and Mode Adaptation type. To be noted that the MPEG multiplex transport packets may be asynchronously mapped to the Base-Band Frames.

For applications requiring sophisticated merging policies, in accordance with specific service requirements (e.g. Quality of Service), Mode Adaptation may optionally be performed by a separate device, respecting all the rules of the DVB-S2 specification. To allow standard interfacing between Mode and Stream Adaptation functions, an optional modulator interface (Mode Adaptation input interface) is defined, according to clauses I.1 (separate signalling circuit) or I.2 (in-band signalling).

Stream adaptation shall be applied, to provide padding to complete a Base-Band Frame and Base-Band Scrambling.

Forward Error Correction (FEC) Encoding shall be carried out by the concatenation of BCH outer codes and LDPC (Low Density Parity Check) inner codes (rates 1/4, 1/3, 2/5, 1/2, 3/5, 2/3, 3/4, 4/5, 5/6, 8/9, 9/10). Depending on the application area, the FEC coded block shall have length n_{ldpc} = 64 800 bits or 16 200 bits. When VCM and ACM is used, FEC and modulation mode may be changed in different frames, but remains constant within a frame. For Backwards Compatible modes, the bit-stream at the output of the FEC encoder shall be processed according to annex F. Bit interleaving shall be applied to FEC coded bits for 8PSK, 16APSK and 32APSK.

Mapping into QPSK, 8PSK, 16APSK and 32APSK constellations shall be applied, depending on the application area. Gray mapping of constellations shall be used for QPSK and 8PSK.

Physical layer framing shall be applied, synchronous with the FEC frames, to provide Dummy PLFRAME insertion, Physical Layer (PL) Signalling, pilot symbols insertion (optional) and Physical Layer Scrambling for energy dispersal. Dummy PLFRAMEs are transmitted when no useful data is ready to be sent on the channel. The System provides a regular physical layer framing structure, based on SLOTs of M=90 modulated symbols, allowing reliable receiver synchronization on the FEC block structure. A slot is devoted to physical layer signalling, including Start-of-Frame delimitation and transmission mode definition. This mechanism is suitable also for VCM and ACM demodulator setting. Carrier recovery in the receivers may be facilitated by the introduction of a regular raster of pilot symbols (P=36 pilot symbols every 16 SLOTs of 90 symbols), while a pilot-less transmission mode is also available, offering an additional 2,4 % useful capacity.

Base-Band Filtering and Quadrature Modulation shall be applied, to shape the signal spectrum (squared-root raised cosine, roll-off factors 0,35 or 0,25 or 0,20) and to generate the RF signal.

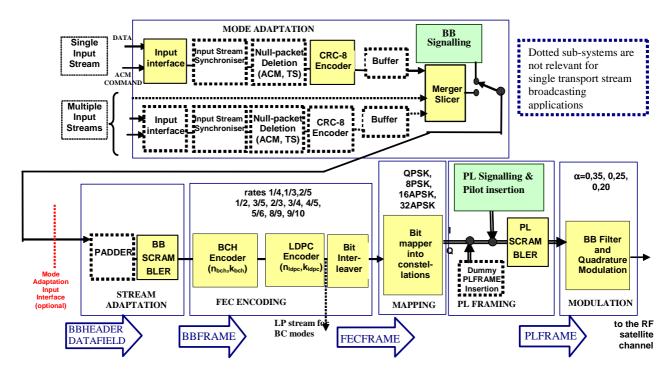


Figure 1: Functional block diagram of the DVB-S2 System

4.3 System configurations

Table 1 associates the System configurations to the applications areas. According to table 1, at least "Normative" subsystems and functionalities shall be implemented in the transmitting and receiving equipment to comply with the present document Guidelines for mode selection are given in annex H.

Table 1: System configurations and application areas

System con	Broadcast services	Interactive services	DSNG	Professional services	
QPSK	1/4,1/3, 2/5	0	N	N	N
	1/2, 3/5, 2/3, 3/4, 4/5, 5/6, 8/9, 9/10	N	N	N	N
8PSK	3/5, 2/3, 3/4, 5/6, 8/9, 9/10	N	N	N	N
16APSK	2/3, 3/4, 4/5, 5/6, 8/9, 9/10	0	N	N	N
32APSK	3/4, 4/5, 5/6, 8/9, 9/10	0	N	N	N
ССМ		N	N (see note 1)	N	N
VCM		0	0	0	0
ACM		NA	N (see note 2)	0	0
FECFRAME (normal)	64 800 (bits)	N	N	N	N
FECFRAME (short)	16 200 (bits)	NA	N	0	N
Single Transport Stream		N	N (see note 1)	N	N
Multiple Transport Streams		0	O (see note 2)	0	0
Single Generic Stream		NA	O (see note 2)	NA	0
Multiple Generic Streams		NA	O (see note 2)	NA	0
Roll-off 0,35, 0,25 and 0,20		N	N	N	N
Input Stream Synchronizer		NA except (see note 3)	O (see note 3)	O (see note 3)	O (see note 3)
Null Packet Deletion		NA except (see note 3)	O (see note 3)	O (see note 3)	O (see note 3)
Dummy Frame insertion		NA except (see note 3)	N	N	N
Wide-band mode	(see annex M)	0	0	0	0
$N = normative \Omega = optional$	NA - not applicable	•	•	•	•

N = normative, O = optional, NA = not applicable.

Within the present document, a number of configurations and mechanisms are defined as "Optional". Configurations and mechanisms explicitly indicated as "optional" within the present document, for a given application area, need not be implemented in the equipment to comply with the present document. Nevertheless, when an "optional" mode or mechanism is implemented, it shall comply with the specification as given in the present document.

5 Subsystems specification

The subsystem specification description is organized according to the functional block diagram of figure 1.

5.1 Mode adaptation

This sub-system shall perform Input Interfacing, Input Stream Synchronization (optional), Null-packet deletion (for TS input streams and ACM only), CRC-8 encoding for error detection (for packetized input streams only), input stream merging (for multiple input streams only) and input stream slicing in DATA FIELDs. Finally, base-band signalling shall be inserted, to notify the receiver of the adopted Mode Adaptation format.

According to figure 3, the input sequence(s) is (are):

- Single or multiple Transport Streams (TS).
- Single or multiple Generic Streams (packetized or continuous).

The output sequence is a BBHEADER (80 bits) followed by a DATA FIELD.

NOTE 1: Interactive service receivers shall implement CCM and Single Transport Stream.

NOTE 2: Interactive Service Receivers shall implement ACM at least in one of the two options: Multiple Transport Streams or Generic Stream (single/multiple input).

NOTE 3: Normative for single/multiple TS input stream(s) combined with ACM/VCM or for multiple TS input streams combined with CCM.

5.1.1 Input interface

The System, as defined in the present document, shall be delimited by the interfaces given in table 2.

Table 2: System interfaces

Location	Interface	Interface type	Connection	Multiplicity
Transmit station	Input	MPEG [1, 4] Transport Stream (see note 1)	from MPEG multiplexer	Single or multiple
Transmit station	Input (see note 2)	Generic Stream	From data sources	Single or multiple
Transmit station	Input (see note 3)	ACM command	From rate control unit	Single
Transmit station	Output	70 MHz/140 MHz IF, L-band IF, RF (see note 4)	to RF devices	Single or multiple
Transmit station	Input	Mode Adaptation	from Mode Adaptation block	Single

NOTE 1: For interoperability reasons, the Asynchronous Serial Interface (ASI) with 188 bytes format, data burst mode (bytes regularly spread over time) is recommended.

NOTE 2: For data services.

NOTE 3: For ACM only. Allows external setting of the ACM transmission mode.

NOTE 4: IF shall be higher than twice the symbol rate.

The input interface subsystem shall map the input electrical format into internal logical-bit format. The first received bit will be indicated as the Most Significant Bit (MSB).

A Transport Stream shall be characterized by User Packets (UP) of constant length UPL = 188×8 bits (one MPEG packet), the first byte being a Sync-byte (47_{HEX}).

A Generic Stream shall be characterized by a continuous bit-stream or a stream of constant-length User Packets (UP), with length UPL bits (maximum UPL value 64 K, UPL = 0_D means continuous stream, see clause 5.1.5). A variable length packet stream, or a constant length packet exceeding 64 kbit, shall be treated as a continuous stream.

For Generic packetized streams, if a synch-byte is the first byte of the UP, it shall be left unchanged, otherwise a sync-byte = 0_D shall be inserted before each packet, and UPL shall be increased by eight. UPL information may be derived by static modulator setting.

"ACM Command" signalling input shall allow setting, by an external "transmission mode control unit", of the transmission parameters to be adopted by the DVB-S2 modulator, for a specific portion of input data. ACM command shall be according to clause D.1.

Mode Adaptation (optional input) shall be a sequence of Data Fields (according to clause 5.1.5), where each individual Data Field is preceded by a BBHEADER, according to clause 5.1.6 and to figure 3, and Stream Adaptation Command, according to clause I.1, to allow setting, by an external 3mode adaptation unit", of the transmission parameters to be adopted by the DVB-S2 modulator, for each specific MA Packet. Mode Adaptation shall be according to clause I.1 (separate signalling circuit) or I.2 (in-band signalling).

5.1.2 Input stream synchronizer (optional, not relevant for single TS - BS)

Data processing in the DVB-S2 modulator may produce variable transmission delay on the user information. The Input Stream Synchronizer subsystem (optional) shall provide suitable means to guarantee Constant-Bit-Rate (CBR) and constant end-to-end transmission delay for packetized input streams (e.g. for Transport Streams). This process shall follow the specification given in annex D. Examples of receiver implementation are given in annex G.

5.1.3 Null-Packet Deletion (ACM and Transport Stream only)

For ACM modes and Transport Stream input data format, MPEG null-packets shall be identified (PID = 8191_D) and removed. This allows to reduce the information rate and increase the error protection in the modulator. The process is carried-out in a way that the removed null-packets can be re-inserted in the receiver in the exact place where they originally were. This process shall follow the specification given in annex D.

5.1.4 CRC-8 encoder (for packetized streams only)

If $UPL = 0_D$ (continuous generic stream) this sub-system shall pass forward the input stream without modifications.

If UPL $\neq 0_D$ the input stream is a sequence of User Packets of length UPL bits, preceded by a sync-byte (the sync-byte being = 0_D when the original stream did not contain a sync-byte).

The useful part of the UP (excluding the sync-byte) shall be processed by a systematic 8-bit CRC encoder. The generator polynomial shall be:

$$g(X) = (X^5 + X^4 + X^3 + X^2 + 1)(X^2 + X + 1)(X + 1) = X^8 + X^7 + X^6 + X^4 + X^2 + 1$$

The CRC encoder output shall be computed as:

$$CRC = remainder [X^8 u(X) : g(X)]$$

Where u(X) is the input sequence (UPL - 8 bits) to be systematically encoded. Figure 2 gives a possible implementation of the CRC generator by means of a shift register.

The register shall be initialized to all zeros before the first bit of each sequence enters the circuit.

The computed CRC-8 shall replace the sync-byte of the following UP. As described in clause 5.1.6, the sync-byte is copied into the SYNC field of the BBHEADER for transmission.

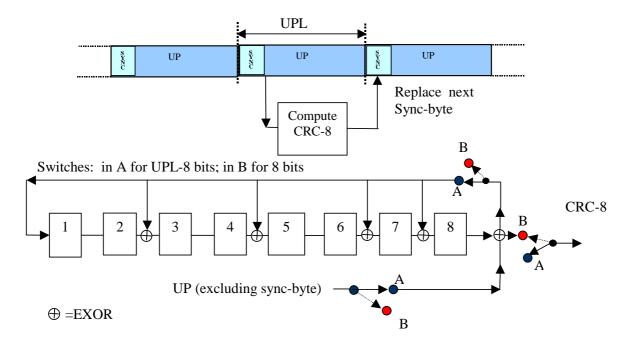


Figure 2: Implementation of the CRC-8 encoder

5.1.5 Merger/Slicer

According to figure 3, the Merger/Slicer input stream(s) is (are) organized as Generic continuous Stream(s) or Packetized Input Stream(s). The UP length is UPL bits (where UPL = 0 means continuous sequence). The input stream(s) shall be buffered until the Merger/Slicer may read them.

The Slicer shall read (i.e. slice) from its input (single input stream), or from one of its inputs (multiple input streams) a DATA FIELD, composed of DFL bits (Data Field Length), where:

 K_{bch} -(10 × 8) \geq DFL \geq 0 (K_{bch} as per table 5, 80 bits are dedicated to the BBHEADER, see clause 5.1.6).

The Merger shall concatenate, in a single output, different data fields read and sliced from one of its inputs. In presence of a single stream, only the slicing functionality applies.

A DATA FIELD shall be composed of bits taken from a single input port and shall be transmitted in a homogeneous transmission mode (FEC code and modulation). The Merger/Slicer prioritization policies are application dependent and shall follow the strategies described in table 4 (Single Transport Stream Broadcast services) and in table D.2 (for other application areas).

Depending on the applications, the Merger/Slicer shall either allocate a number of input bits equal to the maximum DATAFIELD capacity (DFL = K_{bch} -80), thus breaking UPs in subsequent DATAFIELDs, or shall allocate an integer number of UPs within the DATAFIELD, making the DFL variable within the above specified boundaries.

When a DATA FIELD is not available at the merger/slicer request on any input port, the Physical Layer Framing sub-system shall generate and transmit a DUMMY PLFRAME (see clause 5.5.1 and table 12).

After Sync-byte replacing by CRC-8 (see clause 5.1.4), it is necessary to provide the receiver a method to recover UP synchronization (when the receiver is already synchronized to the DATA FIELD). Therefore the number of bits from the beginning of the DATA FIELD and the beginning of the first complete UP (first bit of the CRC-8) (see figure 3) shall be detected by the Merger/Slicer and stored in SYNCD field (i.e. SYNC Distance) of the Base-Band Header (see clause 5.1.6). For example, SYNCD = 0_D means that the first USER PACKET is aligned to the DATA FIELD.

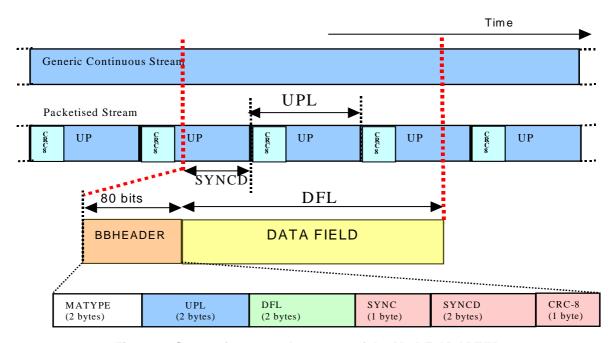


Figure 3: Stream format at the output of the MODE ADAPTER

5.1.6 Base-Band Header insertion

A fixed length base-band Header (BBHEADER) of 10 bytes shall be inserted in front of the DATA FIELD, describing its format (the maximum efficiency loss introduced by the BBHEADER is 0,25 % for n_{ldpc} = 64 800 and 1 % for n_{ldpc} = 16 200 assuming inner code rate 1/2).

MATYPE (2 bytes): describes the input stream(s) format, the type of Mode Adaptation and the transmission Roll-off factor, as explained in table 3.

First byte (MATYPE-1):

- TS/GS field (2 bits): Transport Stream Input or Generic Stream Input (packetized or continuous).
- SIS/MIS field (1 bit): Single Input Stream or Multiple Input Stream.
- CCM/ACM field (1 bit): Constant Coding and Modulation or Adaptive Coding and Modulation (VCM is signalled as ACM).
- ISSYI (1 bit), (Input Stream Synchronization Indicator): If ISSYI = 1 = active, the ISSY field is inserted after UPs (see annex D).

- NPD (1 bit): Null-packet deletion active/not active.
- RO (2 bits): Transmission Roll-off factor (α).

Second byte (MATYPE-2):

If SIS/MIS = Multiple Input Stream, then second byte = Input Stream Identifier (ISI); else second byte
reserved.

UPL (2 bytes): User Packet Length in bits, in the range 0 to 65 535.

EXAMPLE 1: 0000_{HEX} = continuous stream.

EXAMPLE 2: $000A_{HEX} = UP \text{ length of } 10 \text{ bits.}$

EXAMPLE 3: $UPL = 188x8_D$ for MPEG transport stream packets.

DFL (2 bytes): Data Field Length in bits, in the range 0 to 58 112.

EXAMPLE 4: $000A_{HEX}$ = Data Field length of 10 bits.

SYNC (1 byte): copy of the User Packet Sync-byte:

- for packetized Transport or Generic Streams: copy of the User Packet Sync byte;
- for Continuous Generic Streams: SYNC= 00 B8 reserved for transport layer protocol signalling according to Reference [i.4]; SYNC= B9-FF user private).

EXAMPLE 5: SYNC = 47_{HEX} for MPEG transport stream packets.

EXAMPLE 6: SYNC = 00_{HEX} when the input Generic packetized stream did not contain a sync-byte (therefore the receiver, after CRC-8 decoding, shall remove the CRC-8 field without reinserting the Sync-byte).

SYNCD (2 bytes):

- for packetized Transport or Generic Streams: distance in bits from the beginning of the DATA FIELD and the first UP from this frame (first bit of the CRC-8). SYNCD = 65535_D means that no UP starts in the DATA FIELD;
- for Continuous Generic Streams: SYNCD= 0000 FFFF reserved for future uses.

CRC-8 (1 byte): error detection code applied to the first 9 bytes of the BBHEADER.

CRC-8 shall be computed using the encoding circuit of figure 2 (switch in A for 72 bits, in B for 8 bits).

The BBHEADER transmission order is from the MSB of the TS/GS field.

Table 4 shows the BBHEADER and the slicing policy for a Single Transport Stream Broadcast Service. For other application areas, BBHEADERs and merging/slicing policies are defined in table D.2.

Table 3: MATYPE-1 field mapping

TS/GS	SIS/MIS	CCM/ACM	ISSYI	NPD	RO
11 = Transport	1 = single	1 = CCM	1 = active	1 = active	00 = 0.35
00 = Generic Packetized	0 = multiple	0 = ACM	0 = not-active	0 = not-active	01 = 0,25
01 = Generic continuous					10 = 0,20
10 = reserved					11 = reserved

Table 4: BBHeader (Mode Adaptation characteristics) and Slicing Policy for Single Transport Stream Broadcast services

Application area/configuration	MATYPE-1	MATYPE-2	UPL	DFL	SYNC	SYNCD	CRC-8	Slicing policy
Broadcasting services / CCM, single TS	11-1-1-0-0-Y	XXXXXXX	188 _D x8	K _{bch} -80 _D	47 _{HEX}	Υ		Break No timeout No Padding No Dummy frame

X = not defined; Y = according to configuration/computation.

Break = break packets in subsequent DATAFIELDs; Timeout: maximum delay in merger/slicer buffer.

5.2 Stream adaptation

Stream adaptation (see figures 1 and 4) provides padding to complete a constant length (K_{bch} bits) BBFRAME and scrambling. K_{bch} depends on the FEC rate, as reported in table 5. Padding may be applied in circumstances when the user data available for transmission are not sufficient to completely fill a BBFRAME, or when an integer number of UPs has to be allocated in a BBFRAME.

The input stream shall be a BBHEADER followed by a DATA FIELD. The output stream shall be a BBFRAME.

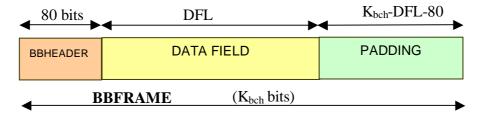


Figure 4: BBFRAME format at the output of the STREAM ADAPTER

5.2.1 Padding

 $(K_{bch}\text{-DFL-80})$ zero bits shall be appended after the DATA FIELD. The resulting BBFRAME shall have a constant length of K_{bch} bits. For Broadcast Service applications, DFL = K_{bch} -80, therefore no padding shall be applied.

5.2.2 BB scrambling

The complete BBFRAME shall be randomized. The randomization sequence shall be synchronous with the BBFRAME, starting from the MSB and ending after K_{hch} bits.

The scrambling sequence shall be generated by the feed-back shift register of figure 5. The polynomial for the Pseudo Random Binary Sequence (PRBS) generator shall be:

$$1 + X^{14} + X^{15}$$

Loading of the sequence (100101010000000) into the PRBS register, as indicated in figure 5, shall be initiated at the start of every BBFRAME.

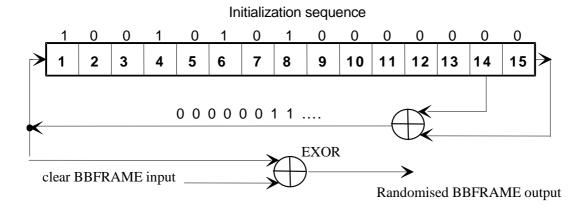


Figure 5: Possible implementation of the PRBS encoder

5.3 FEC encoding

This sub-system shall perform outer coding (BCH), Inner Coding (LDPC) and Bit interleaving. The input stream shall be composed of BBFRAMEs and the output stream of FECFRAMEs.

Each BBFRAME (K_{bch} bits) shall be processed by the FEC coding subsystem, to generate a FECFRAME (n_{ldpc} bits). The parity check bits (BCHFEC) of the systematic BCH outer code shall be appended after the BBFRAME, and the parity check bits (LDPCFEC) of the inner LDPC encoder shall be appended after the BCHFEC field, as shown in figure 6.

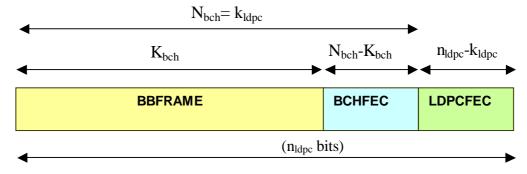


Figure 6: Format of data before bit interleaving $(n_{ldpc} = 64\ 800\ bits$ for normal FECFRAME, $n_{ldpc} = 16\ 200\ bits$ for short FECFRAME)

Table 5a gives the FEC coding parameters for the normal FECFRAME (n_{ldpc} = 64 800 bits) and table 5b for the short FECFRAME (n_{ldpc} = 16 200 bits).

Table 5a: Coding parameters (for normal FECFRAME n_{ldpc} = 64 800)

LDPC code	BCH Uncoded Block K _{bch}	BCH coded block N_{bch} LDPC Uncoded Block k_{ldpc}	BCH t-error correction	LDPC Coded Block n _{ldpc}
1/4	16 008	16 200	12	64 800
1/3	21 408	21 600	12	64 800
2/5	25 728	25 920	12	64 800
1/2	32 208	32 400	12	64 800
3/5	38 688	38 880	12	64 800
2/3	43 040	43 200	10	64 800
3/4	48 408	48 600	12	64 800
4/5	51 648	51 840	12	64 800
5/6	53 840	54 000	10	64 800
8/9	57 472	57 600	8	64 800
9/10	58 192	58 320	8	64 800

Table 5b: Coding parameters (for short FECFRAME n_{ldpc} = 16 200)

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LDPC Code	BCH Uncoded Block K _{bch}	BCH coded block N _{bch} LDPC Uncoded Block <i>k_{ldpc}</i>	BCH t-error	Effective LDPC Rate	LDPC Coded Block
identifier		,	correction	k _{ldpc} /16 200	n _{ldpc}
1/4	3 072	3 240	12	1/5	16 200
1/3	5 232	5 400	12	1/3	16 200
2/5	6 312	6 480	12	2/5	16 200
1/2	7 032	7 200	12	4/9	16 200
3/5	9 552	9 720	12	3/5	16 200
2/3	10 632	10 800	12	2/3	16 200
3/4	11 712	11 880	12	11/15	16 200
4/5	12 432	12 600	12	7/9	16 200
5/6	13 152	13 320	12	37/45	16 200
8/9	14 232	14 400	12	8/9	16 200
9/10	NA	NA	NA	NA	NA

5.3.1 Outer encoding (BCH)

A t-error correcting BCH (N_{bch} , K_{bch}) code shall be applied to each BBFRAME (K_{bch}) to generate an error protected packet. The BCH code parameters for $n_{ldpc}=64\,800$ are given in table 5a and for $n_{ldpc}=16\,200$ in table 5b.

The generator polynomial of the t error correcting BCH encoder is obtained by multiplying the first t polynomials in table 6a for $n_{\rm ldpc} = 64\,800$ and in table 5b for $n_{\rm ldpc} = 16\,200$.

Table 6a: BCH polynomials (for normal FECFRAME n_{ldpc} = 64 800)

g ₁ (x)	$1+x^2+x^3+x^5+x^{16}$
g ₂ (x)	$1+x+x^4+x^5+x^6+x^8+x^{16}$
g ₃ (x)	$1+x^2+x^3+x^4+x^5+x^7+x^8+x^9+x^{10}+x^{11}+x^{16}$
g ₄ (x)	$1+x^2+x^4+x^6+x^9+x^{11}+x^{12}+x^{14}+x^{16}$
g ₅ (x)	$1+x+x^2+x^3+x^5+x^8+x^9+x^{10}+x^{11}+x^{12}+x^{16}$
g ₆ (x)	$1+x^2+x^4+x^5+x^7+x^8+x^9+x^{10}+x^{12}+x^{13}+x^{14}+x^{15}+x^{16}$
g ₇ (x)	$1+x^2+x^5+x^6+x^8+x^9+x^{10}+x^{11}+x^{13}+x^{15}+x^{16}$
g ₈ (x)	$1+x+x^2+x^5+x^6+x^8+x^9+x^{12}+x^{13}+x^{14}+x^{16}$
g ₉ (x)	$1+x^5+x^7+x^9+x^{10}+x^{11}+x^{16}$
g ₁₀ (x)	$1+x+x^2+x^5+x^7+x^8+x^{10}+x^{12}+x^{13}+x^{14}+x^{16}$
g ₁₁ (x)	$1+x^2+x^3+x^5+x^9+x^{11}+x^{12}+x^{13}+x^{16}$
g ₁₂ (x)	$1+x+x^5+x^6+x^7+x^9+x^{11}+x^{12}+x^{16}$

Table 6b: BCH polynomials (for short FECFRAME n_{ldpc} = 16 200)

g ₁ (x)	$1+x+x^3+x^5+x^{14}$
g ₂ (x)	$1+x^6+x^8+x^{11}+x^{14}$
g ₃ (x)	$1+x+x^2+x^6+x^9+x^{10}+x^{14}$
g ₄ (x)	$1+x^4+x^7+x^8+x^{10}+x^{12}+x^{14}$
g ₅ (x)	$1+x^2+x^4+x^6+x^8+x^9+x^{11}+x^{13}+x^{14}$
g ₆ (x)	$1+x^3+x^7+x^8+x^9+x^{13}+x^{14}$
g ₇ (x)	$1+x^2+x^5+x^6+x^7+x^{10}+x^{11}+x^{13}+x^{14}$
g ₈ (x)	$1+x^5+x^8+x^9+x^{10}+x^{11}+x^{14}$
g ₉ (x)	$1+x+x^2+x^3+x^9+x^{10}+x^{14}$
g ₁₀ (x)	$1+x^3+x^6+x^9+x^{11}+x^{12}+x^{14}$
g ₁₁ (x)	$1+x^4+x^{11}+x^{12}+x^{14}$
g ₁₂ (x)	$1+x+x^2+x^3+x^5+x^6+x^7+x^8+x^{10}+x^{13}+x^{14}$

BCH encoding of information bits $\mathbf{m} = (m_{k_{bch}-1}, m_{k_{bch}-2}, ..., m_1, m_0)$ onto a codeword:

$$c = (m_{k_{hch}-1}, m_{k_{hch}-2}, ..., m_1, m_0, d_{n_{hch}-k_{hch}-1}, d_{n_{hch}-k_{hch}-2}, ..., d_1, d_0)$$
 is achieved as follows:

- Multiply the message polynomial $m(x) = m_{k_{bch}-1} x^{k_{bch}-1} + m_{k_{bch}-2} x^{k_{bch}-2} + ... + m_1 x + m_0$ by $x^{n_{bch}-k_{bch}}$.
- Divide $x^{n_{bch}-k_{bch}} m(x)$ by g(x), the generator polynomial. Let $d(x) = d_{n_{bch}-k_{bch}-1} x^{n_{bch}-k_{bch}-1} + ... + d_1 x + d_0$ be the remainder.
- Set the codeword polynomial $c(x) = x^{n_{bch} k_{bch}} m(x) + d(x)$.

5.3.2 Inner encoding (LDPC)

LDPC encoder systematically encodes an information block of size k_{ldpc} , $i = (i_0, i_1, ..., i_{k_{ldpc}-1})$ onto a codeword of size n_{ldpc} , $c = (i_0, i_1, ..., i_{k_{ldpc}-1}, p_0, p_1, ..., p_{n_{ldpc}-k_{ldpc}-1})$. The transmission of the codeword starts in the given order from i_0 and ends with $p_{n_{ldpc}-k_{ldpc}-1}$.

LDPC code parameters (n_{ldvc}, k_{ldvc}) are given in tables 5a and 5b.

For Backwards Compatible modes, the output of the inner encoder is processed according to annex F.

5.3.2.1 Inner coding for normal FECFRAME

The task of the encoder is to determine $n_{ldpc} - k_{ldpc}$ parity bits $(p_0, p_1, ..., p_{n_{ldpc} - k_{ldpc} - 1})$ for every block of k_{ldpc} information bits, $(i_0, i_1, ..., i_{k_{ldpc} - 1})$. The procedure is as follows:

- Initialize $p_0 = p_1 = p_2 = ... = p_{n_{ldnc} k_{ldnc} 1} = 0$.
- Accumulate the first information bit, i_0 , at parity bit addresses specified in the first row of tables B.1 through B.11 in annex B. For example, for rate 2/3 (table B.6), (all additions are in GF(2)):

$$p_{0} = p_{0} \oplus i_{0} \qquad p_{2767} = p_{2767} \oplus i_{0}$$

$$p_{10491} = p_{10491} \oplus i_{0} \qquad p_{240} = p_{240} \oplus i_{0}$$

$$p_{16043} = p_{16043} \oplus i_{0} \qquad p_{18673} = p_{18673} \oplus i_{0}$$

$$p_{506} = p_{506} \oplus i_{0} \qquad p_{9279} = p_{9279} \oplus i_{0}$$

$$p_{12826} = p_{12826} \oplus i_{0} \qquad p_{10579} = p_{10579} \oplus i_{0}$$

$$p_{8065} = p_{8065} \oplus i_{0} \qquad p_{20928} = p_{20928} \oplus i_{0}$$

For the next 359 information bits, i_m , m = 1, 2, ..., 359 accumulate i_m at parity bit addresses $\{x + m \mod 360 \times q\} \mod (n_{ldpc} - k_{ldpc})$ where x denotes the address of the parity bit accumulator corresponding to the first bit i_0 , and q is a code rate dependent constant specified in table 7a. Continuing with the example, q = 60 for rate 2/3. So for example for information bit i_1 , the following operations are performed,

$$p_{60} = p_{60} \oplus i_1$$
 $p_{2827} = p_{2827} \oplus i_1$ $p_{10551} = p_{10551} \oplus i_1$ $p_{300} = p_{300} \oplus i_1$

$$p_{16103} = p_{16103} \oplus i_1$$

$$p_{18733} = p_{18733} \oplus i_1$$

$$p_{566} = p_{566} \oplus i_1$$

$$p_{9339} = p_{9339} \oplus i_1$$

$$p_{12886} = p_{12886} \oplus i_1$$

$$p_{10639} = p_{10639} \oplus i_1$$

$$p_{8125} = p_{8125} \oplus i_1$$

$$p_{20988} = p_{20988} \oplus i_1$$

$$p_{8286} = p_{8286} \oplus i_1$$

- For the 361^{st} information bit i_{360} , the addresses of the parity bit accumulators are given in the second row of the tables B.1 through B.11. In a similar manner the addresses of the parity bit accumulators for the following 359 information bits i_m , m = 361, 362, ..., 719 are obtained using the formula $\{x + (m \mod 360) \times q\} \mod (n_{ldpc} k_{ldpc})$ where x denotes the address of the parity bit accumulator corresponding to the information bit i_{360} , i.e. the entries in the second row of the tables B.1 through B.11.
- In a similar manner, for every group of 360 new information bits, a new row from tables B.1 through B.11 are used to find the addresses of the parity bit accumulators.

After all of the information bits are exhausted, the final parity bits are obtained as follows:

• Sequentially perform the following operations starting with i = 1.

$$p_i = p_i \oplus p_{i-1}, \quad i = 1, 2, \dots, n_{ldpc} - k_{ldpc} - 1$$

• Final content of p_i , $i = 0,1,...,n_{ldpc} - k_{ldpc} - 1$ is equal to the parity bit p_i .

Table 7a: q values for normal frames

Code Rate	q
1/4	135
1/3	120
2/5	108
1/2	90
3/5	72
2/3	60
3/4	45
4/5	36
5/6	30
8/9	20
9/10	18

5.3.2.2 Inner coding for short FECFRAME

 k_{ldpc} BCH encoded bits shall be systematically encoded to generate n_{ldpc} bits as described in clause 5.3.2.1, replacing table 7a with table 7b, the tables of annex B with the tables of annex C.

Table 7b: q values for short frames

Code Rate	q
1/4	36
1/3	30
2/5	27
1/2	25
3/5	18
2/3	15
3/4	12
4/5	10
5/6	8
8/9	5

5.3.3 Bit Interleaver (for 8PSK, 16APSK and 32APSK only)

For 8PSK, 16APSK, and 32APSK modulation formats, the output of the LDPC encoder shall be bit interleaved using a block interleaver. Data is serially written into the interleaver column-wise, and serially read out row-wise (the MSB of BBHEADER is read out first, except 8PSK rate 3/5 case where MSB of BBHEADER is read out third) as shown in figures 7 and 8.

The configuration of the block interleaver for each modulation format is specified in table 8.

Table 8: Bit Interleaver structure

Modulation	Rows (for $n_{ldpc} = 64800$)	Rows (for n _{ldpc} = 16 200)	Columns
8PSK	21 600	5 400	3
16APSK	16 200	4 050	4
32APSK	12 960	3 240	5

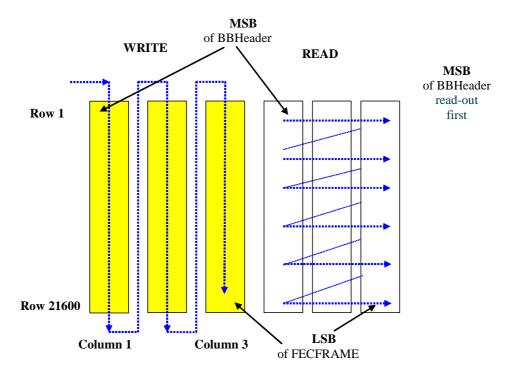


Figure 7: Bit Interleaving scheme for 8PSK and normal FECFRAME length (all rates except 3/5)

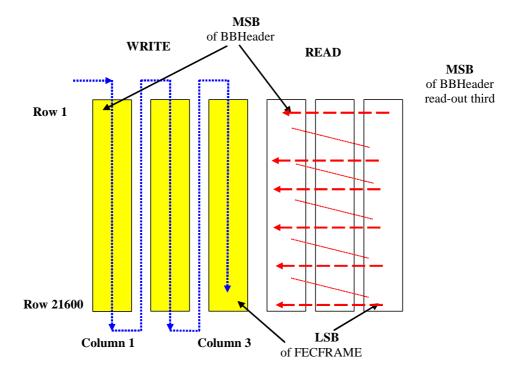


Figure 8: Bit Interleaving scheme for 8PSK and normal FECFRAME length (rate 3/5 only)

5.4 Bit mapping into constellation

Each FECFRAME (which is a sequence of 64 800 bits for normal FECFRAME, or 16 200 bits for short FECFRAME), shall be serial-to-parallel converted (parallelism level = η_{MOD} 2 for QPSK, 3 for 8PSK, 4 for 16APSK, 5 for 32APSK) in figures 9 to 12, the MSB of the FECFRAME is mapped into the MSB of the first parallel sequence. Each parallel sequence shall be mapped into constellation, generating a (I,Q) sequence of variable length depending on the selected modulation efficiency η_{MOD} .

The input sequence shall be a FECFRAME, the output sequence shall be a XFECFRAME (compleX FECFRAME), composed of 64 $800/\eta_{MOD}$ (normal XFECFRAME) or 16 $200/\eta_{MOD}$ (short XFECFRAME) modulation symbols. Each modulation symbol shall be a complex vector in the format (I,Q) (I being the in-phase component and Q the quadrature component) or in the equivalent format ρ exp(j ϕ) (ρ being the modulus of the vector and ϕ being its phase).

5.4.1 Bit mapping into QPSK constellation

For QPSK, the System shall employ conventional Gray-coded QPSK modulation with absolute mapping (no differential coding). Bit mapping into the QPSK constellation shall follow figure 9. The normalized average energy per symbol shall be equal to $\rho^2 = 1$.

Two FECFRAME bits are mapped to a QPSK symbol i.e. bits 2i and 2i+1 determines the ith QPSK symbol, where i = 0, 1, 2, ..., (N/2)-1 and N is the coded LDPC block size.

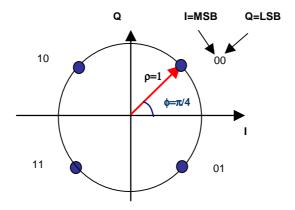


Figure 9: Bit mapping into QPSK constellation

5.4.2 Bit mapping into 8PSK constellation

For 8PSK, the System shall employ conventional Gray-coded 8PSK modulation with absolute mapping (no differential coding). Bit mapping into the 8PSK constellation shall follow figure 10. The normalized average energy per symbol shall be equal to $\rho^2 = 1$.

Bits 3i, 3i+1, 3i+2 of the interleaver output determine the i^{th} 8PSK symbol where i = 0, 1, 2, ... (N/3)-1 and N is the coded LDPC block size.

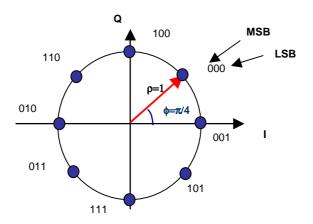


Figure 10: Bit mapping into 8PSK constellation

5.4.3 Bit mapping into 16APSK constellation

The 16APSK modulation constellation (figure 11) shall be composed of two concentric rings of uniformly spaced 4 and 12 PSK points, respectively in the inner ring of radius R_1 and outer ring of radius R_2 .

The ratio of the outer circle radius to the inner circle radius ($\gamma = R_2/R_1$) shall comply with table 9.

Two are the admitted values for the constellation amplitudes, allowing performance optimization according to the channel characteristics (e.g. single or multiple carriers per transponder, use of non-linear predistortion):

- E=1 (E=unit average symbol energy) corresponding to $[R_1]^2 + 3[R_2]^2 = 4$.
- $R_2=1$.

Bits 4i, 4i+1, 4i+2 and 4i+3 of the interleaver output determine the i^{th} 16APSK symbol, where i = 0, 1, 2, ..., (N/4)-1 and N is the coded LDPC block size.

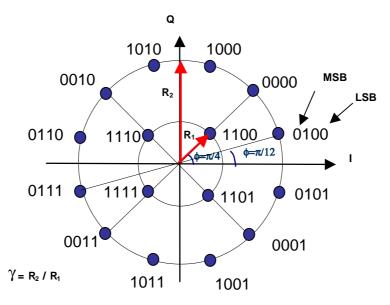


Figure 11: 16APSK signal constellation

Table 9: Optimum constellation radius ratio γ (linear channel) for 16APSK

Code rate	Modulation/coding spectral efficiency	γ
2/3	2,66	3,15
3/4	2,99	2,85
4/5	3,19	2,75
5/6	3,32	2,70
8/9	3,55	2,60
9/10	3,59	2,57

5.4.4 Bit mapping into 32APSK

The 32APSK modulation constellation (see figure 12) shall be composed of three concentric rings of uniformly spaced 4, 12 and 16 PSK points, respectively in the inner ring of radius R_1 , the intermediate ring of radius R_2 and the outer ring or radius R_3 . Table 10 defines the values of $Y_1 = R_2/R_1$ and $Y_2 = R_3/R_1$.

Two are the admitted values for the constellation amplitudes, allowing performance optimization according to the channel characteristics (e.g., single or multiple carriers per transponder, use of non-linear predistortion):

- E=1 (E=unit average symbol energy) corresponding to $[R_1]^2 + 3[R_2]^2 + 4[R_3]^2 = 8$.
- $R_3=1$.

Bits 5i, 5i+1, 5i+2, 5i+3 and 5i+4 of the interleaver output determine the i^{th} 32APSK symbol, where i = 0, 1, 2, (N/5)-1.

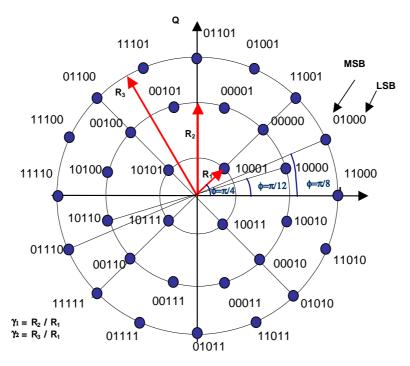


Figure 12: 32APSK signal constellation

Table 10: optimum constellation radius ratios γ_1 and γ_2 (linear channel) for 32 APSK

Code rate	Modulation/coding spectral efficiency	γ ₁	γ_2
3/4	3,74	2,84	5,27
4/5	3,99	2,72	4,87
5/6	4,15	2,64	4,64
8/9	4,43	2,54	4,33
9/10	4,49	2,53	4,30

5.5 Physical Layer (PL) framing

The PLFraming sub-system shall generate a physical layer frame (named PLFRAME) by performing the following processes (see figures 1 and 13):

- Dummy PLFRAME generation when no XFECFRAME is ready to be processed and transmitted.
- XFECFRAME slicing into an integer number S of constant length SLOTs (length: M = 90 symbols each); S shall be according to table 11.
- PLHEADER generation and insertion before the XFECFRAME for receiver configuration. PLHEADER shall occupy exactly one SLOT (length: M = 90 Symbols).
- Pilot Block insertion (for modes requiring pilots) every 16 SLOTS, to help receiver synchronization. The Pilot Block shall be composed of P = 36 pilot symbols.
- Randomization of the (I, Q) modulated symbols by means of a physical layer scrambler.

The input stream of the sub-system shall be a XFECFRAME and the output a scrambled PLFRAME.

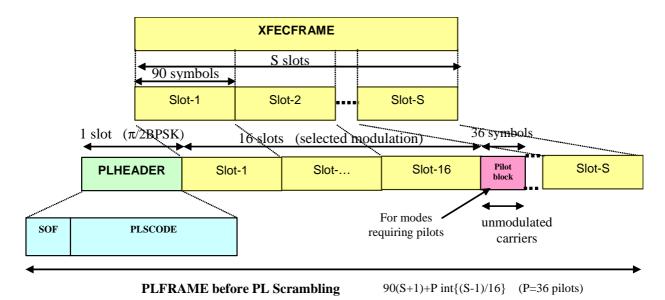


Figure 13: Format of a "Physical Layer Frame" PLFRAME

Table 11: S = number of SLOTs (M = 90 symbols) per XFECFRAME

	n _{ldpc} = 64 800 (normal frame)			_{pc} = 16 200 nort frame)
η _{MOD} (bit/s/Hz)	S	η % no-pilot	S	η % no-pilot
2	360	99,72	90	98,90
3	240	99,59	60	98,36
4	180	99,45	45	97,83
5	144	99,31	36	97,30

The PLFRAMING efficiency is $\eta = 90S/[90(S+1) + P \text{ int}\{(S-1)/16\}]$, where P = 36 and int $\{.\}$ is the integer function.

5.5.1 Dummy PLFRAME insertion

A Dummy PLFRAME shall be composed of a PLHEADER (see clause 5.5.2) and of 36 SLOTS of un-modulated carriers ($I = (1/\sqrt{2})$, $Q = (1/\sqrt{2})$).

5.5.2 PL signalling

The PLHEADER is intended for receiver synchronization and physical layer signalling.

NOTE: After decoding the PLHEADER, the receiver knows the PLFRAME duration and structure, the modulation and coding scheme of the XFECFRAME, the presence or absence of pilot symbols.

The PLHEADER (one SLOT of 90 symbols) shall be composed of the following fields:

- **SOF** (26 symbols), identifying the Start of Frame.
- **PLS** code (64 symbol): PLS (Physical Layer Signalling) code shall be a non-systematic binary code of length 64 and dimension 7 with minimum distance d_{min} = 32. It is equivalent to the first order Reed-Muller under permutation. It transmits 7 bits for physical layer signalling purpose. These 7 bits consists of two fields: MODCOD and TYPE defined as follows:
 - MODCOD (5 symbols), identifying the XFECFRAME modulation and FEC rate;
 - TYPE (2 symbols), identifying the FECFRAME length (64 800 bits or 16 200 bits) and the presence/absence of pilots.

The PLHEADER, represented by the binary sequence $(y_1, y_2,...y_{90})$ shall be modulated into 90 π /2BPSK symbols according to the rule:

$$I_{2i-1} = Q_{2i-1} = (1/\sqrt{2}) (1-2y_{2i-1}), I_{2i} = -Q_{2i} = -(1/\sqrt{2}) (1-2y_{2i})$$
 for $i = 1, 2, ..., 45$

5.5.2.1 SOF field

SOF shall correspond to the sequence 18D2E82_{HEX} (01-1000-....-0010 in binary notation, the left-side bit being the MSB of the PLHEADER).

5.5.2.2 MODCOD field

MODCOD shall correspond to 5 bits, identifying code rates in the set $\eta_C = [1/4, 1/3, 2/5, 1/2, 3/5, 2/3, 3/4, 4/5, 5/6, 8/9, 9/10]$ and modulations in the set of spectrum efficiencies $\eta_{MOD} = [2, 3, 4, 5]$ according to table 12.

Mode	MOD COD	Mode	MOD COD	Mode	MOD COD	Mode	MOD COD
QPSK 1/4	1 _D	QPSK 5/6	9 _D	8PSK 9/10	17 _D	32APSK 4/5	25 _D
QPSK 1/3	2 _D	QPSK 8/9	10 _D	16APSK 2/3	18 _D	32APSK 5/6	26 _D
QPSK 2/5	3 _D	QPSK 9/10	11 _D	16APSK 3/4	19 _D	32APSK 8/9	27 _D
QPSK 1/2	4 _D	8PSK 3/5	12 _D	16APSK 4/5	20 _D	32APSK 9/10	28 _D
QPSK 3/5	5 _D	8PSK 2/3	13 _D	16APSK 5/6	21 _D	Reserved	29 _D
QPSK 2/3	6 _D	8PSK 3/4	14 _D	16APSK 8/9	22 _D	Reserved	30 _D
QPSK 3/4	7 _D	8PSK 5/6	15 _D	16APSK 9/10	23 _D	Reserved	31 _D
QPSK 4/5	8 _D	8PSK 8/9	16 _D	32APSK 3/4	24 _D	DUMMY PLFRAME	0 _D

Table 12: MODCOD coding

5.5.2.3 TYPE field

The MSB of the TYPE field shall identify 2 FECFRAME sizes (0 = normal: 64 800 bits; 1 = short: 16 200 bits). The LSB of the TYPE field shall identify the pilot configurations (see clause 5.5.3) (0 = no pilots, 1 = pilots).

5.5.2.4 PLS code

The MODCODE and TYPE fields are bi-orthogonally coded with a (64,7) code. Such code is constructed starting from a bi-orthogonal (32,6) code according to the construction in figure 13a.

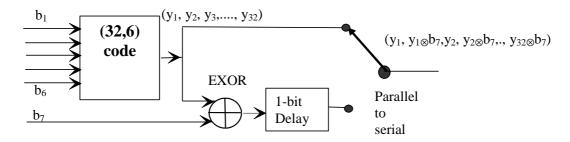


Figure 13a: (the symbol ⊗ stands for binary EXOR)

The particular construction guarantees that each odd bit in the (64,7) code is either always equal to the previous one or is always the opposite. Which of the two hypotheses is true depends on the bit b_7 . This fact can be exploited in case differentially coherent detection is adopted in the receiver.

The MODCOD and the MSB of the TYPE field shall be encoded by a linear block code of length 32 with the following generator matrix.

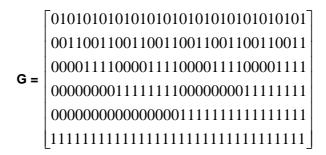


Figure 13b

The most significant bit of the MODCOD is multiplied with the first row of the matrix, the following bit with the second row and so on. The 32 coded bits is denoted as $(y_1y_2\cdots y_{32})$. When the least significant bit of the TYPE field is 0, the final PLS code will generate $(y_1y_1y_2y_2\cdots y_{32}y_{32})$ as the output, i.e. each symbol shall be repeated. When the least significant bit of the TYPE field is 1, the final PLS code will generate $(y_1\overline{y_1}y_2\overline{y_2}\cdots y_{32}\overline{y_{32}})$ as output, i.e. the repeated symbol is further binary complemented. The 64 bits output of the PLS code is further scrambled by the binary sequence:

5.5.3 Pilots insertion

Two PLFRAME configurations shall be possible:

- Without pilots.
- With pilots.

In this latter case a PILOT BLOCK shall be composed of P=36 pilot symbols. Each pilot shall be an un-modulated symbol, identified by $I=(1/\sqrt{2})$, $Q=(1/\sqrt{2})$. The first PILOT BLOCK shall be inserted 16 SLOTs after the PLHEADER, the second after 32 SLOTs and so on, as represented in figure 13. If the PILOT BLOCK position coincides with the beginning of the next SOF, then the PILOT BLOCK is not transmitted.

The pilot presence/absence in VCM and ACM can be changed on a frame-by-frame basis.

5.5.4 Physical layer scrambling

Prior to modulation, each PLFRAME, excluding the PLHEADER, shall be randomized for energy dispersal by multiplying the (I+jQ) samples by a complex randomization sequence (C_1+jC_0) :

$$I_{SCRAMBLED} = [I C_I - Q C_O];$$
 $Q_{SCRAMBLED} = (I C_O + Q C_I)$

NOTE 1: The randomization sequence rate corresponds to the I-Q PLFRAME symbol rate, thus it has no impact on the occupied signal bandwidth. The randomization sequence has a period greater than the maximum required duration of about 70 000 symbols).

The randomization sequence shall be reinitialized at the end of each PLHEADER (see figure 14). The PLFRAME duration depends on the modulation selected, thus the randomization sequence length shall be truncated to the current PLFRAME length.

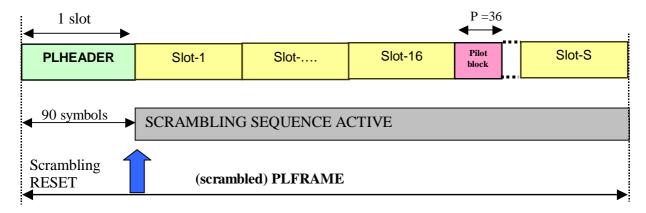


Figure 14: PL SCRAMBLING

The scrambling code sequences shall be constructed by combining two real m-sequences (generated by means of two generator polynomials of degree 18) into a complex sequence. The resulting sequences thus constitute segments of a set of Gold sequences.

Let x and y be the two sequences respectively. The x sequence is constructed using the primitive (over GF(2)) polynomial $1+x^7+x^{18}$. The y sequence is constructed using the polynomial $1+y^5+y^7+y^{10}+y^{18}$.

The sequence depending on the chosen scrambling code number n is denoted z_n in the sequel. Furthermore, let x(i), y(i) and $z_n(i)$ denote the ith symbol of the sequence x, y, and z_n respectively. The m-sequences x and y are constructed as:

- Initial conditions:
 - x is constructed with x(0) = 1, x(1) = x(2) = ... = x(16) = x(17) = 0.
 - y(0) = y(1) = ... = y(16) = y(17) = 1.
- Recursive definition of subsequent symbols:
 - $x(i+18) = x(i+7) + x(i) \text{ modulo } 2, i = 0, ..., 2^{18} 20.$
 - y(i+18) = y(i+10) + y(i+7) + y(i+5) + y(i) modulo 2, $i = 0, ..., 2^{18} 20$.

The nth Gold code sequence z_n $n = 0, 1, 2, ..., 2^{18}$ -2, is then defined as:

-
$$z_n(i) = [x((i+n) \text{ modulo } (2^{18}-1)) + y(i)] \text{ modulo } 2, i = 0,..., 2^{18} - 2.$$

These binary sequences are converted to integer valued sequences R_n (R_n assuming values 0, 1, 2, 3) by the following transformation:

$$R_n(i) = 2 z_n((i + 131\ 072)\ \text{modulo}\ (2^{18}-1)) + z_n(i)$$
 $i = 0, 1, ..., 66\ 419.$

Finally, the nth complex scrambling code sequence $C_I(i) + jC_O(i)$ is defined as:

$$C_{I}(i) + jC_{O}(i) = \exp(j R_n(i) \pi/2)$$

R _n	exp(j R _n π/2)	scrambled	Q _{scrambled}
0	1	1	Q
1	j	-Q	I
2	-1	-1	-Q
3	-i	Q	-1

Figure 15 gives a possible block diagram for PL scrambling sequences generation for n=0.

for $|f| < f_N(1-\alpha)$

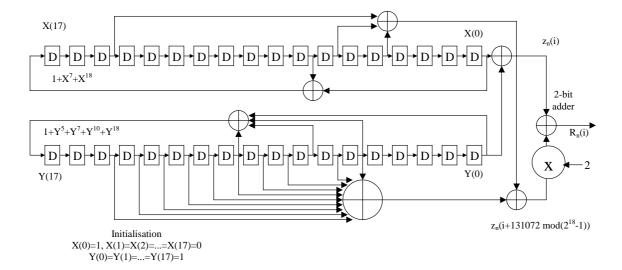


Figure 15: Configuration of PL scrambling code generator for n = 0

In case of broadcasting services, n = 0 shall be used as default sequence, to avoid manual receiver setting or synchronization delays.

NOTE 2: n, assuming values in the range 0 to 262 141, indicates the spreading sequence number. The use of different PL Scrambling sequences allows a reduction of interference correlation between different services. For the same purpose, it is possible to reuse a shifted version of the same sequence in different satellite beams. Furthermore n can be unequivocally associated to each satellite operator or satellite or transponder, thus permitting identification of an interfering signal via the PL Scrambling "signature" detection. There is no explicit signalling method to convey *n* to the receiver.

5.6 Baseband shaping and quadrature modulation

After randomization, the signals shall be square root raised cosine filtered. The roll-off factor shall be $\alpha = 0.35, 0.25$ and 0.20, depending on the service requirements.

The baseband square root raised cosine filter shall have a theoretical function defined by the following expression:

H(f) = 1

$$H(f) = \left\{ \frac{1}{2} + \frac{1}{2} \sin \frac{\pi}{2f_N} \left[\frac{f_N - |f|}{\alpha} \right] \right\}^{\frac{1}{2}}$$
 for $f_N(1 - \alpha)$

$$H(f) = 0$$
 for $|f| > f_N(1+\alpha)$,

where: $f_N = \frac{1}{2T_s} = \frac{R_s}{2}$ is the Nyquist frequency and α is the roll-off factor.

A template for the signal spectrum at the modulator output is given in annex A.

Quadrature modulation shall be performed by multiplying the in-phase and quadrature samples (after baseband filtering) by $\sin(2\pi f_0 t)$ and $\cos(2\pi f_0 t)$, respectively (where f_0 is the carrier frequency). The two resulting signals shall be added to obtain the modulator output signal.

6 Error performance

Table 13 summarizes performance requirements at QEF over AWGN (E_s = average energy per transmitted symbol). Ideal E_s /No (dB) is the figure achieved by computer simulation, 50 LDPC fixed point decoding iterations (see annex G), perfect carrier and synchronization recovery, no phase noise, AWGN channel. For short FECFRAMEs an additional degradation of 0,2 dB to 0,3 dB has to be taken into account.

For calculating link budgets, specific satellite channel impairments should be taken into account.

PER is the ratio between the useful transport stream packets (188 bytes) correctly received and affected by errors, after forward error correction.

Table 13: E_s/No performance at Quasi Error Free PER = 10⁻⁷ (AWGN channel)

Mode	Spectral efficiency	Ideal E _s /No (dB)
		for FECFRAME length = 64 800
QPSK 1/4	0,490243	-2,35
QPSK 1/3	0,656448	-1,24
QPSK 2/5	0,789412	-0,30
QPSK 1/2	0,988858	1,00
QPSK 3/5	1,188304	2,23
QPSK 2/3	1,322253	3,10
QPSK 3/4	1,487473	4,03
QPSK 4/5	1,587196	4,68
QPSK 5/6	1,654663	5,18
QPSK 8/9	1,766451	6,20
QPSK 9/10	1,788612	6,42
8PSK 3/5	1,779991	5,50
8PSK 2/3	1,980636	6,62
8PSK 3/4	2,228124	7,91
8PSK 5/6	2,478562	9,35
8PSK 8/9	2,646012	10,69
8PSK 9/10	2,679207	10,98
16APSK 2/3	2,637201	8,97
16APSK 3/4	2,966728	10,21
16APSK 4/5	3,165623	11,03
16APSK 5/6	3,300184	11,61
16APSK 8/9	3,523143	12,89
16APSK 9/10	3,567342	13,13
32APSK 3/4	3,703295	12,73
32APSK 4/5	3,951571	13,64
32APSK 5/6	4,119540	14,28
32APSK 8/9	4,397854	15,69
32APSK 9/10	4,453027	16,05

NOTE: Given the system spectral efficiency η_{tot} the ratio between the energy per information bit and single sided noise power spectral density $E_b/N_0 = E_s/N_0 - 10log_{10}(\eta_{tot})$.

Spectral efficiencies (per unit symbol rate) are computed for normal FECFRAME length and no pilots.

Examples of possible use of the System are given in annex H. Table H.1 gives examples of possible degradation introduced by the satellite channel model described in clause H.7.

Annex A (normative): Signal spectrum at the modulator output

For roll-off factor $\alpha = 0.35$, the signal spectrum at the modulator output shall be in accordance with EN 300 421 [2].

As an option, the signal spectrum can correspond to a narrower roll-off factor $\alpha = 0.25$ or $\alpha = 0.20$.

Figure A.1 gives a template for the signal spectrum at the modulator output.

Figure A.1 also represents a possible mask for a hardware implementation of the Nyquist modulator filter. The points A to S shown on figures A.1 and A.2 are defined in table A.1. The mask for the filter frequency response is based on the assumption of ideal Dirac delta input signals, spaced by the symbol period $T_S = 1/R_S = 1/2f_N$ while in the case of rectangular input signals a suitable x/sin x correction shall be applied on the filter response.

Relative power (dB) 10 C Ε G 0 D -10 Μ -20 -30 -40 S -50 0,5 1 1,5 2 2,5 3 f/f_N

Figure A.1: Template for the signal spectrum mask at the modulator output represented in the baseband frequency domain, the frequency axis is calibrated for roll-off factor $\alpha = 0.35$

Figure A.2 gives a mask for the group delay for the hardware implementation of the Nyquist modulator filter.

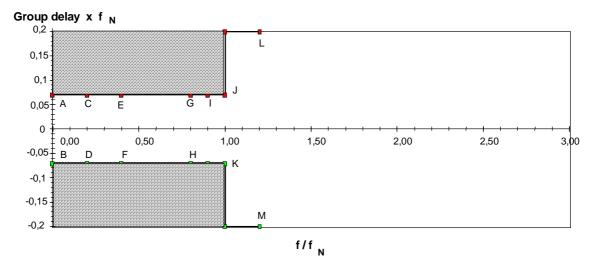


Figure A.2: Template of the modulator filter group delay

Table A.1: Definition of points given in figures A.1 and A.2

Point	Frequency for α = 0,35	Frequency for $\alpha = 0.25$	Frequency for α = 0,20	Relative power (dB)	Group delay
Α	0,0 f _N	0,0 f _N	0,0 f _N	+0,25	+0,07 / f _N
В	0,0 f _N	0,0 f _N	0,0 f _N	-0,25	-0,07 / f _N
С	0,2 f _N	0,2 f _N	0,2 f _N	+0,25	+0,07 / f _N
D	0,2 f _N	0,2 f _N	0,2 f _N	-0,40	-0,07 / f _N
Е	0,4 f _N	0,4 f _N	0,4 f _N	+0,25	+0,07 / f _N
F	0,4 f _N	0,4 f _N	0,4 f _N	-0,40	-0,07 / f _N
G	0,8 f _N	0,86f _N	0,89 f _N	+0,15	+0,07 / f _N
Н	0,8 f _N	0,86 f _N	0,89 f _N	-1,10	-0,07 / f _N
I	0,9 f _N	0,93 f _N	0,94 f _N	-0,50	+0,07 / f _N
J	1,0 f _N	1,0 f _N	1,0 f _N	-2,00	+0,07 / f _N
K	1,0 f _N	1,0 f _N	1,0 f _N	-4,00	-0,07 / f _N
L	1,2 f _N	1,13 f _N	1,11 f _N	-8,00	-
M	1,2 f _N	1,13 f _N	1,11 f _N	-11,00	-
N	1,8 f _N	1,60 f _N	1,5 f _N	-35,00	-
Р	1,4 f _N	1,30 f _N	1,23 f _N	-16,00	-
Q	1,6 f _N	1,45 f _N	1,4 f _N	-24,00	-
S	2,12 f _N	1,83 f _N	1,7 f _N	-40,00	-

Annex B (normative): Addresses of parity bit accumulators for $n_{ldpc} = 64800$

Example of interpretation of the table B.4.

```
p_{54} = p_{54} \oplus i_0 \quad p_{9318} = p_{9318} \oplus i_0 \quad p_{14392} = p_{14392} \oplus i_0 \quad p_{27561} = p_{27561} \oplus i_0 \quad p_{26909} = p_{26909} \oplus i_0 \quad p_{10219} = p_{10219} \oplus i_0
p_{2534} = p_{2534} \oplus i_0 \quad p_{8597} = p_{8597} \oplus i_0
p_{144} = p_{144} \oplus i_1 \quad p_{9408} = p_{9408} \oplus i_1 \quad p_{14482} = p_{14482} \oplus i_1 \quad p_{27651} = p_{27651} \oplus i_1 \quad p_{26999} = p_{26999} \oplus i_1 \quad p_{10309} = p_{10309} \oplus i_1
p_{2624} = p_{2624} \oplus i_1 \quad p_{8687} = p_{8687} \oplus i_1
\vdots \quad \vdots \quad \vdots \quad \vdots \quad \vdots \quad \vdots
p_{32364} = p_{32364} \oplus i_{359} \quad p_{9228} = p_{9228} \oplus i_{359} \quad p_{14302} = p_{14302} \oplus i_{359} \quad p_{27471} = p_{27471} \oplus i_{359} \quad p_{26819} = p_{26819} \oplus i_{359}
p_{10129} = p_{10129} \oplus i_{359} \quad p_{2444} = p_{2444} \oplus i_{359} \quad p_{8507} = p_{8507} \oplus i_{359}
p_{55} = p_{55} \oplus i_{360} \quad p_{7263} = p_{7263} \oplus i_{360} \quad p_{4635} = p_{4635} \oplus i_{360} \quad p_{2530} = p_{2530} \oplus i_{360} \quad p_{28130} = p_{28130} \oplus i_{360}
p_{3033} = p_{3033} \oplus i_{360} \quad p_{23830} = p_{23830} \oplus i_{360} \quad p_{3651} = p_{3651} \oplus i_{360}
\vdots \quad \vdots
```

Table B.1: Rate 1/4 (n_{Idoc} = 64 800)

23606 36098 1140 28859 18148 18510 6226 540 42014 20879 23802 47088 16419 24928 16609 17248 7693 24997 42587 16858 34921 21042 37024 20692 1874 40094 18704 14474 14004 11519 13106 28826 38669 22363 30255 31105 22254 40564 22645 22532 6134 9176 39998 23892 8937 15608 16854 31009 8037 40401 13550 19526 41902 28782 13304 32796 24679 27140 45980 10021 40540 44498 13911 22435 32701 18405 39929 25521 12497 9851 39223 34823 15233 45333 5041 44979 45710 42150 19416 1892 23121 15860 8832 10308 10468 44296 3611 1480 37581 32254 13817 6883 32892 40258 46538 11940 6705 21634 28150 43757 895 6547 20970 28914 30117 25736 41734 11392 22002 5739 27210 27828 34192 37992 10915 6998 3824 42130 4494 35739 8515 1191 13642 30950 25943 12673 16726 34261 31828 3340 8747 39225 18979 17058 43130 4246 4793 44030 19454 29511 47929 15174 24333 19354 16694 8381 29642 46516 32224 26344 9405 18292 12437 27316 35466 41992 15642 5871 46489 26723 23396 7257 8974 3156 37420 44823 35423 13541 42858 32008 41282 38773 26570 2702 27260 46974 1469 20887 27426 38553 22152 24261 8297 19347 9978 27802 34991 6354 33561 29782 30875 29523 9278 48512 14349 38061 4165 43878 8548 33172 34410 22535 28811 23950

20439 4027 24186 38618 8187 30947 35538 43880 21459 7091 45616 15063 5505 9315 21908

14615 15593 1642 29111 37061 39860

Table B.2: Rate 1/3 (n_{Idpc} = 64 800)

Table B.3: Rate 2/5 (n_{ldpc} = 64 800)

31413 18834 28884 947 23050 14484 14809 4968 455 33659 16666 19008 13172 19939 13354 13719 6132 20086 34040 13442 27958 16813 29619 16553 1499 32075 14962 11578 11204 9217 10485 23062 30936 17892 24204 24885 $32490\ 18086\ 18007\ 4957\ 7285\ 32073\ 19038\ 7152\ 12486\ 13483\ 24808\ 21759$ 32321 10839 15620 33521 23030 10646 26236 19744 21713 36784 8016 12869 35597 11129 17948 26160 14729 31943 20416 10000 7882 31380 27858 33356 14125 12131 36199 4058 35992 36594 33698 15475 1566 18498 12725 7067 17406 8372 35437 2888 1184 30068 25802 11056 5507 26313 32205 37232 15254 5365 17308 22519 35009 718 5240 16778 23131 24092 20587 33385 $27455\ 17602\ 4590\ 21767\ 22266\ 27357\ 30400\ 8732\ 5596\ 3060\ 33703\ 3596$ 6882 873 10997 24738 20770 10067 13379 27409 25463 2673 6998 31378 15181 13645 34501 3393 3840 35227 15562 23615 38342 12139 19471 15483 13350 6707 23709 37204 25778 21082 7511 14588 10010 21854 28375 33591 12514 4695 37190 21379 18723 5802 7182 2529 29936 35860 28338 10835 34283 25610 33026 31017 21259 2165 21807 37578 1175 16710 21939 30841 27292 33730 6836 26476 27539 35784 18245 16394 17939 23094 19216 17432 11655 6183 38708 28408 35157 17089 13998 36029 15052 16617 5638 36464 15693 28923 26245 9432 11675 25720 26405 5838 31851 26898 8090 37037 24418 27583 7959 35562 37771 17784 11382 11156 37855 7073 21685 34515 10977 13633 30969 7516 11943 18199 5231 13825 19589 23661 11150 35602 19124 30774 6670 37344 16510 26317 23518 22957 6348 34069 8845 20175 34985 14441 25668 4116 3019 21049 37308 24551 24727 20104 24850 12114 38187 28527 13108 13985 1425 21477 30807 8613 26241 33368 35913 32477 5903 34390 24641 26556 23007 27305 38247 2621 9122 32806 21554 18685 17287 27292 19033

Table B.4: Rate 1/2 (n_{Idpc} = 64 800)

54 9318 14392 27561 26909 10219 2534 8597
55 7263 4635 2530 28130 3033 23830 3651
56 24731 23583 26036 17299 5750 792 9169
57 5811 26154 18653 11551 15447 13685 16264
58 12610 11347 28768 2792 3174 29371 12997
59 16789 16018 21449 6165 21202 15850 3186
60 31016 21449 17618 6213 12166 8334 18212
61 22836 14213 11327 5896 718 11727 9308
62 2091 24941 29966 23634 9013 15587 5444
63 22207 3983 16904 28534 21415 27524 25912
64 25687 4501 22193 14665 14798 16158 5491
65 4520 17094 23397 4264 22370 16941 21526
66 10490 6182 32370 9597 30841 25954 2762
67 22120 22865 29870 15147 13668 14955 19235
68 6689 18408 18346 9918 25746 5443 20645
69 29982 12529 13858 4746 30370 10023 24828
70 1262 28032 29888 13063 24033 21951 7863
71 6594 29642 31451 14831 9509 9335 31552
72 1358 6454 16633 20354 24598 624 5265
73 19529 295 18011 3080 13364 8032 15323
74 11981 1510 7960 21462 9129 11370 25741
75 9276 29656 4543 30699 20646 21921 28050
76 15975 25634 5520 31119 13715 21949 19605
77 18688 4608 31755 30165 13103 10706 29224
78 21514 23117 12245 26035 31656 25631 30699
79 9674 24966 31285 29908 17042 24588 31857
80 21856 27777 29919 27000 14897 11409 7122
81 29773 23310 263 4877 28622 20545 22092
82 15605 5651 21864 3967 14419 22757 15896
83 30145 1759 10139 29223 26086 10556 5098
84 18815 16575 2936 24457 26738 6030 505
85 30326 22298 27562 20131 26390 6247 24791
86 928 29246 21246 12400 15311 32309 18608
87 20314 6025 26689 16302 2296 3244 19613
88 6237 11943 22851 15642 23857 15112 20947
89 26403 25168 19038 18384 8882 12719 7093
0 14567 24965
1 3908 100
2 10279 240
3 24102 764
4 12383 4173
5 13861 15918
6 21327 1046
7 5288 14579
8 28158 8069
9 16583 11098
10 16681 28363
11 13980 24725
12 32169 17989
13 10907 2767
14 21557 3818
15 26676 12422
16 7676 8754
17 14905 20232
18 15719 24646
19 31942 8589
20 19978 27197
21 27060 15071
22 6071 26649
23 10393 11176
24 9597 13370
25 7081 17677
26 1433 19513
27 26925 9014
28 19202 8900
29 18152 30647
30 20803 1737
31 11804 25221
32 31683 17783
33 29694 9345
34 12280 26611
35 6526 26122
36 26165 11241
37 7666 26962
38 16290 8480
39 11774 10120

Table B.5: Rate 3/5 (n_{Idoc} = 64 800)

Table B.6: Rate $2/3 (n_{ldpc} = 64800)$

Table B.7: Rate 3/4 (n_{ldpc} = 64 800)

Table B.8: Rate 4/5 (n_{ldpc} = 64 800)

Table B.o. Rate	+/3 (II _{Idpc} = 64 600)
0 149 11212 5575 6360 12559 8108 8505 408 10026 12828	3 6970 5447
1 5237 490 10677 4998 3869 3734 3092 3509 7703 10305	4 3217 5638
2 8742 5553 2820 7085 12116 10485 564 7795 2972 2157	5 8972 669
3 2699 4304 8350 712 2841 3250 4731 10105 517 7516	6 5618 12472
4 12067 1351 11992 12191 11267 5161 537 6166 4246 2363	7 1457 1280
5 6828 7107 2127 3724 5743 11040 10756 4073 1011 3422	8 8868 3883
6 11259 1216 9526 1466 10816 940 3744 2815 11506 11573	9 8866 1224
7 4549 11507 1118 1274 11751 5207 7854 12803 4047 6484	10 8371 5972
8 8430 4115 9440 413 4455 2262 7915 12402 8579 7052	11 266 4405
9 3885 9126 5665 4505 2343 253 4707 3742 4166 1556 10 1704 8936 6775 8639 8179 7954 8234 7850 8883 8713	12 3706 3244 13 6039 5844
11 11716 4344 9087 11264 2274 8832 9147 11930 6054 5455	14 7200 3283
12 7323 3970 10329 2170 8262 3854 2087 12899 9497 11700	15 1502 11282
13 4418 1467 2490 5841 817 11453 533 11217 11962 5251	16 12318 2202
14 1541 4525 7976 3457 9536 7725 3788 2982 6307 5997	17 4523 965
15 11484 2739 4023 12107 6516 551 2572 6628 8150 9852	18 9587 7011
16 6070 1761 4627 6534 7913 3730 11866 1813 12306 8249	19 2552 2051
17 12441 5489 8748 7837 7660 2102 11341 2936 6712 11977 18 10155 4210	20 12045 10306 21 11070 5104
19 1010 10483	22 6627 6906
20 8900 10250	23 9889 2121
21 10243 12278	24 829 9701
22 7070 4397	25 2201 1819
23 12271 3887	26 6689 12925
24 11980 6836	27 2139 8757
25 9514 4356	28 12004 5948
26 7137 10281 27 11881 2526	29 8704 3191 30 8171 10933
28 1969 11477	31 6297 7116
29 3044 10921	32 616 7146
30 2236 8724	33 5142 9761
31 9104 6340	34 10377 8138
32 7342 8582	35 7616 5811
33 11675 10405	0 7285 9863
34 6467 12775 35 3186 12198	1 7764 10867 2 12343 9019
0 9621 11445	3 4414 8331
1 7486 5611	4 3464 642
2 4319 4879	5 6960 2039
3 2196 344	6 786 3021
4 7527 6650	7 710 2086
5 10693 2440	8 7423 5601
6 6755 2706 7 5144 5998	9 8120 4885 10 12385 11990
8 11043 8033	11 9739 10034
9 4846 4435	12 424 10162
10 4157 9228	13 1347 7597
11 12270 6562	14 1450 112
12 11954 7592	15 7965 8478
13 7420 2592	16 8945 7397 17 6590 8316
14 8810 9636	18 6838 9011
15 689 5430	19 6174 9410
16 920 1304	20 255 113
17 1253 11934	21 6197 5835
18 9559 6016	22 12902 3844
19 312 7589	23 4377 3505
20 4439 4197 21 4002 9555	24 5478 8672 25 4453 2132
21 4002 9333 22 12232 7779	26 9724 1380
23 1494 8782	27 12131 11526
24 10749 3969	28 12323 9511
25 4368 3479	29 8231 1752
26 6316 5342	30 497 9022
27 2455 3493	31 9288 3080
28 12157 7405 29 6598 11495	32 2481 7515 33 2696 268
30 11805 4455	34 4023 12341
31 9625 2090	35 7108 5553
32 4731 2321	
33 3578 2608	
34 8504 1849	
35 4027 1151 0 5647 4035	
0 5647 4935 1 4219 1870	
2 10968 8054	

Table B.9: Rate 5/6 (n_{ldpc} = 64 800)

Table B.9: Rate 5/6 (n _{ldpc} = 64 800)	
0 4362 416 8909 4156 3216 3112 2560 2912 6405 8593 4969 6723	15 9027 3415
1 2479 1786 8978 3011 4339 9313 6397 2957 7288 5484 6031 10217	16 1690 3866
2 10175 9009 9889 3091 4985 7267 4092 8874 5671 2777 2189 8716	17 2854 8469
3 9052 4795 3924 3370 10058 1128 9996 10165 9360 4297 434 5138	18 6206 630
4 2379 7834 4835 2327 9843 804 329 8353 7167 3070 1528 7311	19 363 5453
5 3435 7871 348 3693 1876 6585 10340 7144 5870 2084 4052 2780	20 4125 7008
6 3917 3111 3476 1304 10331 5939 5199 1611 1991 699 8316 9960 7 6883 3237 1717 10752 7891 9764 4745 3888 10009 4176 4614 1567	21 1612 6702 22 9069 9226
8 10587 2195 1689 2968 5420 2580 2883 6496 111 6023 1024 4449	23 5767 4060
9 3786 8593 2074 3321 5057 1450 3840 5444 6572 3094 9892 1512	24 3743 9237
10 8548 1848 10372 4585 7313 6536 6379 1766 9462 2456 5606 9975	25 7018 5572
11 8204 10593 7935 3636 3882 394 5968 8561 2395 7289 9267 9978	26 8892 4536
12 7795 74 1633 9542 6867 7352 6417 7568 10623 725 2531 9115	27 853 6064
13 7151 2482 4260 5003 10105 7419 9203 6691 8798 2092 8263 3755	28 8069 5893
14 3600 570 4527 200 9718 6771 1995 8902 5446 768 1103 6520	29 2051 2885
15 6304 7621 16 6498 9209	0 10691 3153 1 3602 4055
17 7293 6786	2 328 1717
18 5950 1708	3 2219 9299
19 8521 1793	4 1939 7898
20 6174 7854	5 617 206
21 9773 1190	6 8544 1374
22 9517 10268	7 10676 3240
23 2181 9349 24 1949 5560	8 6672 9489 9 3170 7457
25 1556 555	10 7868 5731
26 8600 3827	11 6121 10732
27 5072 1057	12 4843 9132
28 7928 3542	13 580 9591
29 3226 3762	14 6267 9290
0 7045 2420	15 3009 2268
1 9645 2641	16 195 2419 17 8016 1557
2 2774 2452 3 5331 2031	18 1516 9195
4 9400 7503	19 8062 9064
5 1850 2338	20 2095 8968
6 10456 9774	21 753 7326
7 1692 9276	22 6291 3833
8 10037 4038	23 2614 7844
9 3964 338	24 2303 646
10 2640 5087 11 858 3473	25 2075 611 26 4687 362
12 5582 5683	27 8684 9940
13 9523 916	28 4830 2065
14 4107 1559	29 7038 1363
15 4506 3491	0 1769 7837
16 8191 4182	1 3801 1689
17 10192 6157 18 5668 2205	2 10070 2359
18 5668 3305 19 3449 1540	3 3667 9918 4 1914 6920
20 4766 2697	5 4244 5669
21 4069 6675	6 10245 7821
22 1117 1016	7 7648 3944
23 5619 3085	8 3310 5488
24 8483 8400	9 6346 9666
25 8255 394 26 6338 5042	10 7088 6122 11 1291 7827
20 0338 3042 27 6174 5119	12 10592 8945
28 7203 1989	13 3609 7120
29 1781 5174	14 9168 9112
0 1464 3559	15 6203 8052
1 3376 4214	16 3330 2895
2 7238 67	17 4264 10563
3 10595 8831 4 1221 6513	18 10556 6496 19 8807 7645
5 5300 4652	20 1999 4530
6 1429 9749	21 9202 6818
7 7878 5131	22 3403 1734
8 4435 10284	23 2106 9023
9 6331 5507	24 6881 3883
10 6662 4941	25 3895 2171
11 9614 10238	26 4062 6424
12 8400 8025 13 9156 5630	27 3755 9536 28 4683 2131
14 7067 8878	29 7347 8027
	-> 13T1 0041

Table B.10: Rate 8/9 (n_{ldpc} = 64 800)

0.6225.2049.2222	13 1969 3869	6 5821 4932	19 5736 1399	12 2644 5073
0 6235 2848 3222				13 4212 5088
1 5800 3492 5348	14 3571 2420	7 6356 4756 8 3930 418	0 970 2572 1 2062 6599	14 3463 3889
2 2757 927 90	15 4632 981			
3 6961 4516 4739	16 3215 4163	9 211 3094	2 4597 4870	15 5306 478
4 1172 3237 6264	17 973 3117	10 1007 4928	3 1228 6913	16 4320 6121
5 1927 2425 3683	18 3802 6198	11 3584 1235	4 4159 1037	17 3961 1125
6 3714 6309 2495	19 3794 3948	12 6982 2869	5 2916 2362	18 5699 1195
7 3070 6342 7154	0 3196 6126	13 1612 1013	6 395 1226	19 6511 792
8 2428 613 3761	1 573 1909	14 953 4964	7 6911 4548	0 3934 2778
9 2906 264 5927	2 850 4034	15 4555 4410	8 4618 2241	1 3238 6587
10 1716 1950 4273	3 5622 1601	16 4925 4842	9 4120 4280	2 1111 6596
11 4613 6179 3491	4 6005 524	17 5778 600	10 5825 474	3 1457 6226
12 4865 3286 6005	5 5251 5783	18 6509 2417	11 2154 5558	4 1446 3885
13 1343 5923 3529	6 172 2032	19 1260 4903	12 3793 5471	5 3907 4043
14 4589 4035 2132	7 1875 2475	0 3369 3031	13 5707 1595	6 6839 2873
15 1579 3920 6737	8 497 1291	1 3557 3224	14 1403 325	7 1733 5615
16 1644 1191 5998	9 2566 3430	2 3028 583	15 6601 5183	8 5202 4269
17 1482 2381 4620	10 1249 740	3 3258 440	16 6369 4569	9 3024 4722
18 6791 6014 6596	11 2944 1948	4 6226 6655	17 4846 896	10 5445 6372
19 2738 5918 3786	12 6528 2899	5 4895 1094	18 7092 6184	11 370 1828
0 5156 6166	13 2243 3616	6 1481 6847	19 6764 7127	12 4695 1600
1 1504 4356	14 867 3733	7 4433 1932	0 6358 1951	13 680 2074
2 130 1904	15 1374 4702	8 2107 1649	1 3117 6960	14 1801 6690
3 6027 3187	16 4698 2285	9 2119 2065	2 2710 7062	15 2669 1377
4 6718 759	17 4760 3917	10 4003 6388	3 1133 3604	16 2463 1681
5 6240 2870	18 1859 4058	11 6720 3622	4 3694 657	17 5972 5171
6 2343 1311	19 6141 3527	12 3694 4521	5 1355 110	18 5728 4284
7 1039 5465	0 2148 5066	13 1164 7050	6 3329 6736	19 1696 1459
8 6617 2513	1 1306 145	14 1965 3613	7 2505 3407	
9 1588 5222	2 2319 871	15 4331 66	8 2462 4806	
10 6561 535	3 3463 1061	16 2970 1796	9 4216 214	
11 4765 2054	4 5554 6647	17 4652 3218	10 5348 5619	
12 5966 6892	5 5837 339	18 1762 4777	11 6627 6243	

Table B.11: Rate 9/10 (n_{ldpc} = 64 800)

0 5611 2563 2900	0 4165 884	0 855 4475	0 3171 4690	0 5779 1596
1 5220 3143 4813	1 2896 3744	1 4097 3970	1 5204 2114	1 2403 1237
2 2481 834 81	2 874 2801	2 4433 4361	2 6384 5565	2 2217 1514
3 6265 4064 4265	3 3423 5579	3 5198 541	3 5722 1757	3 5609 716
4 1055 2914 5638	4 3404 3552	4 1146 4426	4 2805 6264	4 5155 3858
5 1734 2182 3315	5 2876 5515	5 3202 2902	5 1202 2616	5 1517 1312
6 3342 5678 2246	6 516 1719	6 2724 525	6 1018 3244	6 2554 3158
7 2185 552 3385	7 765 3631	7 1083 4124	7 4018 5289	7 5280 2643
8 2615 236 5334	8 5059 1441	8 2326 6003	8 2257 3067	8 4990 1353
9 1546 1755 3846	9 5629 598	9 5605 5990	9 2483 3073	9 5648 1170
10 4154 5561 3142	10 5405 473	10 4376 1579	10 1196 5329	10 1152 4366
11 4382 2957 5400	11 4724 5210	11 4407 984	11 649 3918	11 3561 5368
12 1209 5329 3179	12 155 1832	12 1332 6163	12 3791 4581	12 3581 1411
13 1421 3528 6063	13 1689 2229	13 5359 3975	13 5028 3803	13 5647 4661
14 1480 1072 5398	14 449 1164	14 1907 1854	14 3119 3506	14 1542 5401
15 3843 1777 4369	15 2308 3088	15 3601 5748	15 4779 431	15 5078 2687
16 1334 2145 4163	16 1122 669	16 6056 3266	16 3888 5510	16 316 1755
17 2368 5055 260	17 2268 5758	17 3322 4085	17 4387 4084	17 3392 1991
0 6118 5405	0 5878 2609	0 1768 3244	0 5836 1692	
1 2994 4370	1 782 3359	1 2149 144	1 5126 1078	
2 3405 1669	2 1231 4231	2 1589 4291	2 5721 6165	
3 4640 5550	3 4225 2052	3 5154 1252	3 3540 2499	
4 1354 3921	4 4286 3517	4 1855 5939	4 2225 6348	
5 117 1713	5 5531 3184	5 4820 2706	5 1044 1484	
6 5425 2866	6 1935 4560	6 1475 3360	6 6323 4042	
7 6047 683	7 1174 131	7 4266 693	7 1313 5603	
8 5616 2582	8 3115 956	8 4156 2018	8 1303 3496	
9 2108 1179	9 3129 1088	9 2103 752	9 3516 3639	
10 933 4921	10 5238 4440	10 3710 3853	10 5161 2293	
11 5953 2261	11 5722 4280	11 5123 931	11 4682 3845	
12 1430 4699	12 3540 375	12 6146 3323	12 3045 643	
13 5905 480	13 191 2782	13 1939 5002	13 2818 2616	
14 4289 1846	14 906 4432	14 5140 1437	14 3267 649	
15 5374 6208	15 3225 1111	15 1263 293	15 6236 593	
16 1775 3476	16 6296 2583	16 5949 4665	16 646 2948	
17 3216 2178	17 1457 903	17 4548 6380	17 4213 1442	

Annex C (normative): Addresses of parity bit accumulators for $n_{ldpc} = 16200$

Table C.1: Rate 1/4 (n_{Idoc} = 16 200)

6295 9626 304 7695 4839 4936 1660 144 11203 5567 6347 12557 10691 4988 3859 3734 3071 3494 7687 10313 5964 8069 8296 11090 10774 3613 5208 11177 7676 3549 8746 6583 7239 12265 2674 4292 11869 3708 5981 8718 4908 10650 6805 3334 2627 10461 9285 11120 7844 3079 10773 3385 10854 5747 1360 12010 12202 6189 4241 2343 9840 12726 4977

Table C.2: Rate 1/3 ($n_{ldpc} = 16200$)

416 8909 4156 3216 3112 2560 2912 6405 8593 4969 6723 6912 8978 3011 4339 9312 6396 2957 7288 5485 6031 10218 2226 3575 3383 10059 1114 10008 10147 9384 4290 434 5139 3536 1965 2291 2797 3693 7615 7077 743 1941 8716 6215 3840 5140 4582 5420 6110 8551 1515 7404 4879 4946 5383 1831 3441 9569 10472 4306 1505 5682 7778 7172 6830 6623 7281 3941 3505 10270 8669 914 3622 7563 9388 9930 5058 4554 4844 9609 2707 6883 3237 1714 4768 3878 10017 10127 3334 8267

Table C.3: Rate 2/5 (n_{ldpc} = 16 200)

5650 4143 8750 583 6720 8071 635 1767 1344 6922 738 6658 5696 1685 3207 415 7019 5023 5608 2605 857 6915 1770 8016 3992 771 2190 7258 8970 7792 1802 1866 6137 8841 886 1931 4108 3781 7577 6810 9322 8226 5396 5867 4428 8827 7766 2254 4247 888 4367 8821 9660 324 5864 4774 227 7889 6405 8963 9693 500 2520 2227 1811 9330 1928 5140 4030 4824 806 3134 1652 8171 1435 3366 6543 3745 9286 8509 4645 7397 5790 8972 6597 4422 1799 9276 4041 3847 8683 7378 4946 5348 1993 9186 6724 9015 5646 4502 4439 8474 5107 7342 9442 1387 8910 2660

Table C.4: Rate 1/2 (n_{ldpc} = 16 200)

11 8935 4996 12 3028 764 13 5988 1057 14 7411 3450

Table C.5: Rate 3/5 (n_{ldpc} = 16 200)

Table C.6: Rate 2/3 (n_{ldpc} = 16 200)

0 2084 1613 1548 1286 1460 3196 4297 2481 3369 3451 4620 2622	1 2583 1180
1 122 1516 3448 2880 1407 1847 3799 3529 373 971 4358 3108	2 1542 509
2 259 3399 929 2650 864 3996 3833 107 5287 164 3125 2350	3 4418 1005
3 342 3529	4 5212 5117
4 4198 2147	5 2155 2922
5 1880 4836	6 347 2696
6 3864 4910	7 226 4296
7 243 1542	8 1560 487
8 3011 1436	9 3926 1640
9 2167 2512	10 149 2928
10 4606 1003	11 2364 563
11 2835 705	12 635 688
12 3426 2365	13 231 1684
13 3848 2474	14 1129 3894
14 1360 1743	
0 163 2536	

Table C.7: Rate 3/4 (n_{ldpc} = 16 200)

3 3198 478 4207 1481 1009 2616 1924 3437 554 683 1801 4 2681 2135 5 3107 4027 6 2637 3373 7 3830 3449 8 4129 2060 9 4184 2742 10 3946 1070 11 2239 984 0 1458 3031 1 3003 1328 2 1137 1716 3 132 3725 4 1817 638 5 1774 3447 6 3632 1257	8 1015 1945 9 1948 412 10 995 2238 11 4141 1907 0 2480 3079 1 3021 1088 2 713 1379 3 997 3903 4 2323 3361 5 1110 986 6 2532 142 7 1690 2405 8 1298 1881 9 615 174 10 1648 3112 11 1415 2808
7 342 3074	

Table C.8: Rate 4/5 (n_{Idpc} = 16 200)

5 896 1565	7 951 2068	9 2116 1855
6 2493 184	8 3108 3542	0 722 1584
7 212 3210	9 307 1421	1 2767 1881
8 727 1339	0 2272 1197	2 2701 1610
9 3428 612	1 1800 3280	3 3283 1732
0 2663 1947	2 331 2308	4 168 1099
1 230 2695	3 465 2552	5 3074 243
2 2025 2794	4 1038 2479	6 3460 945
3 3039 283	5 1383 343	7 2049 1746
4 862 2889	6 94 236	8 566 1427
5 376 2110	7 2619 121	9 3545 1168
6 2034 2286	8 1497 2774	

Table C.9: Rate 5/6 (n_{ldpc} = 16 200)

Table C.10: Rate 8/9 ($n_{ldpc} = 16\ 200$)

0 1558 712 805	4 1496 502	3 544 1190
1 1450 873 1337	0 1006 1701	4 1472 1246
2 1741 1129 1184	1 1155 97	0 508 630
3 294 806 1566	2 657 1403	1 421 1704
4 482 605 923	3 1453 624	2 284 898
0 926 1578	4 429 1495	3 392 577
1 777 1374	0 809 385	4 1155 556
2 608 151	1 367 151	0 631 1000
3 1195 210	2 1323 202	1 732 1368
4 1484 692	3 960 318	2 1328 329
0 427 488	4 1451 1039	3 1515 506
1 828 1124	0 1098 1722	4 1104 1172
2 874 1366	1 1015 1428	
3 1500 835	2 1261 1564	

Annex D (normative): Additional Mode Adaptation and ACM tools

D.1 "ACM Command" signalling interface

"ACM Command" signalling input (see figure D.1) shall allow setting, by an external "transmission mode control unit", of the transmission parameters to be adopted by the DVB-S2 modulator, for a specific portion of input data.

"ACM Command" shall carry the following information:

- MODCOD (5 bits, according to table 12).
- TYPE (2 bits, according to clause 5.5.2.3).
- CVALID (Command Valid).
- SEND (deliver Data).

The transmission format specified by MODCOD and TYPE shall be applied to user data received after CVALID = active and before SEND = active. When SEND = active, the modulator shall deliver user data immediately, even if a FECFRAME is not completed, by inserting the PADDING field (see clause 5.2.1). The user data included in the interval between CVALID = active and SEND = active shall not exceed the capacity of $(K_{bch}-80)$ bits, K_{bch} being the transmittable bits associated with a specific MODCOD and TYPE.

For input Transport Streams, ACM is implemented via null-packet deletion function, therefore input user data do not correspond directly to the transmitted data. In this case, the SEND function is not relevant, and CVALID, MODCOD and TYPE shall become active at least 10 times a second. The ACM modulator shall continuously apply the specified MODCOD and TYPE to user data after CVALID = active.

An example temporization of ACM Command is given in figure D.1, using a single serial interface to convey MODCOD, TYPE, CVALID(active = high-to-low transition) and SEND (active = low-to-high transition).

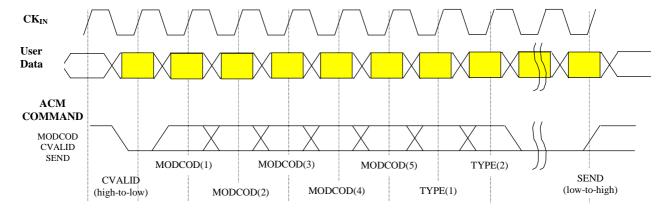


Figure D.1: Example temporization of ACM Command (serial format)

D.2 Input stream synchronizer

Delays and packet jitter introduced by DVB-S2 modems may depend from the transmitted bit-rate and may change in time during ACM rate switching. The "Input Stream Synchronizer" (see figures D.1 and D.2) shall provide a mechanism to regenerate, in the receiver, the clock of the Transport Stream (or Generic Packetized Stream) at the modulator Mode Adapter input, in order to guarantee end-to-end constant bit rates and delays (see also figure G.3, example receiver implementation). Table D.1 indicates the applications in which the Input Stream Synchronizer is normative or optional.

When ISSYI = 1 in MATYPE field (see table 3), a counter shall be activated (22 bits), clocked by the modulator symbol rate (frequency R_s). The Input Stream SYnchronization field (ISSY, 2 or 3 bytes) shall be appended after each input packet (in the case of Transport Streams, before null-packet deletion takes place), as shown in figure D.2. ISSY shall be coded according to table D.1, sending the following variables:

- ISCR (short: 15 bits; long: 22 bits) (ISCR = Input Stream Time Reference), loaded with the LSBs of the counter content at the instant the relevant input packet is processed (at constant rate R_{IN}), and specifically the instant the MSB of the relevant packet arrives at the modulator input stream interface.
- BUFS (2+10 bits) (BUFS = maximum size of the requested receiver buffer to compensate delay variations). It is assumed that a receiver FIFO buffer (see clause G.3, figure G.3) operates on a single stream input (i.e. corresponding to a specific MATYPE-2 configuration for SIS/MIS = 0 in MATYPE-1); the FIFO buffer input is the recovered packet stream after FEC error correction, at the channel arriving rate, and after null packet reinsertion, its output is the modulator output stream (to be sent to the TS demultiplexer in case of Transport Stream), read with the recovered (transport) stream clock. If ISSYI = 1 and optional BUFS is used, this variable shall be transmitted at least 5 times per second, replacing ISCR. The maximum buffer size required in the receiver shall be 20 Mbits.
- BUFSTAT (2+10 bits) (BUFSTAT = actual status to reset the receiver buffer = number of filled bits). If ISSYI = 1 and optional BUFSTAT is used, this variable shall be transmitted at least 5 times per second, replacing ISCR. This value can be used to set the receiver buffer status during reception start-up procedure, and to verify normal functioning in steady state.

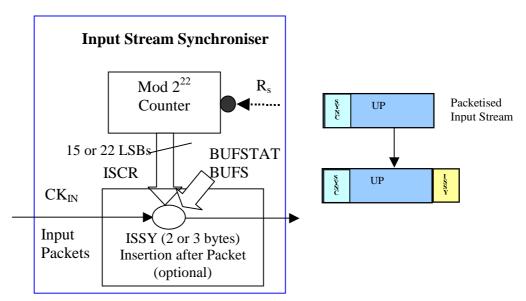


Figure D.2: Input stream synchronizer block diagram

Table D.1: ISSY field coding (2 or 3 bytes)

First Byte				Second Byte	Third Byte	
bit-7 (MSB)	bit-6	bit-5 and bit-4	bit-3 and bit-2	bit-1 and bit-0	bit-7 to bit-0	bit-7 bit-0
0 = ISCR _{short}	ISCR _{short}	next 6 bits of ISCR _{short}			next 8 bits of SCR _{short}	not present
1	0 = ISCR _{long}	6 MSBs of ISCR _{lon}	g		next 8 bits of ISCR _{long}	next 8 bits of ISCR _{long}
1	1	00 = BUFS	BUFS unit 00 = bits 01 = Kbits 10 = Mbits 11 = reserved	2 MSBs of BUFS	next 8 bits of BUFS	not present when ISCR _{short} is used; else reserved
1	1	10 = BUFSTAT	BUFSTAT unit 00 = bits 01 = Kbits 10 = Mbits 11 = reserved	2 MSBs of BUFSTAT	next 8 bits of BUFSTAT	not present when ISCR _{short} is used; else reserved
1	1	others = reserved	reserved	reserved	reserved	not present when ISCR _{short} is used; else reserved

An example receiver scheme to regenerate the output packet stream and the relevant clock R'_{IN} is given in figure G.3.

D.3 Null-packet Deletion (normative for input transport streams and ACM)

Transport Stream rules require that the bit rates at the output of the MUX and the input of the DEMUX are constant in time, and the end-to-end delay is also constant. In order to fulfil such requirements in an ACM environment, the null-packet deletion function shall be activated (see annex H.4 for application examples).

As shown in figure D.3, Useful Packets (i.e. packets with $PID \neq 8191_D$) (including the optional ISSY appended field) shall be transmitted while null-packets ($PID = 8191_D$) (including the optional ISSY appended field) shall be removed.

After transmission of a UP, a counter called DNP (Deleted Null-Packets, 1 byte) shall be first reset and then incremented at each deleted null-packet. The counter content shall be appended after the Least Significant Byte of the next transmitted useful packet, then DNP shall be reset. When DNP reaches the maximum allowed value $DNP = 255_D$, then if the following packet is again a null-packet this null-packet is kept as a useful packet and transmitted.

The resulting stream has $UPL = (188 + 1) \times 8$ bits (for ISSYI = 0) or $UPL = (188 + 2 + 1) \times 8$ bits (for ISSYI = 1 and $ISCR_{short}$), or $UPL = (188 + 3 + 1) \times 8$ bits (for ISSYI = 1 and $ISCR_{long}$), since the Transport Stream packets are extended by the DNP and ISSY (optional) fields.

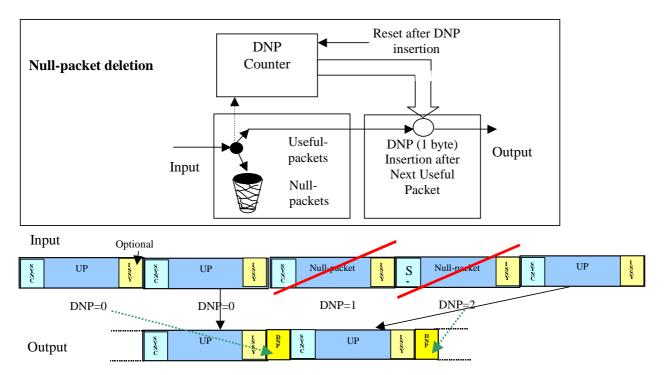


Figure D.3: Null-packet deletion and DNP field (1 byte) insertion

D.4 BBHEADER and Merging/slicing Policy for various application areas

According to the application area, BBHeader coding and Merging/slicing policy shall be according to table D.2.

Table D.2: BBHeader coding for various application areas and Merging/Slicing policy

Application area/configuration	MATYPE-1	MATYPE-2	UPL	DFL	SYNC	SYNCD	CRC-8	Merging/ slicing policy
Broadcasting/CCM, single TS	111100Y	Х	188 _D x8	K _{bch} - 80 _D	47 _{HEX}	Y	Υ	Break No timeout No Padding No Dummy
Broadcasting, differentiated protection level per stream/VCM, constant protection level per TS, Multiple TS	1100Y0Y	Y	188 _D x8 (+16 or 24 if ISSYI = 1)	K _{bch} - 80 _D	47 _{HEX}	Υ	Υ	Break Read (1) No timeout No Padding Yes Dummy
DSNG with time variable protection level/ACM, single TS input, NP- deletion, ACM Command active	111011Y	X	189 _D x8+ (16 or 24)	K _{bch} - 80 _D	47 _{HEX}	Υ	Υ	Break Read (0) No timeout No Padding Yes Dummy
Interactive services with ACM over TS, differentiated protection per stream/ ACM, constant protection level per TS, Multiple TS, NP- deletion	1100Y1Y	Y	189 _D x8 (+16 or 24 if ISSYI = 1)	Y ≤K _{bch} - 80 _D	47 _{HEX}	Y	Υ	Read(1) or (2) Yes Padding Yes Dummy YES shortframe (see note)
Interactive services (IP) with ACM over GS, differentiated protection per stream/ ACM, constant protection level per input stream, Multiple Generic Stream	010000Y	Y	0	Y ≤K _{bch} - 80 _D	X	X	Υ	Read(1) or (2) Yes Padding Yes Dummy YES shortframe (see note)
Interactive services (IP) with ACM over GS, time variable protection/ ACM, time variable protection level, Single Generic Stream, ACM Command active	011000Y	X	0	Y ≤K _{bch} - 80 _D	X	X	Υ	According to ACM Command Yes Padding Yes Dummy YES shortframe
BC Broadcasting services	111100Y	Х	188 _D x8	K _{bch} - 80 _D	47 _{HEX}	Y	Y	Break No timeout No Padding No Dummy

X = not defined; Y = according to configuration/computation Break = break packets in subsequent DATAFIELDs; Timeout: maximum delay in merger/slicer buffer.

D.5 Signalling of reception quality via return channel (Normative for ACM)

In ACM modes, the receiver shall signal the reception quality via an available return channel, according to the various DVB interactive systems, such as for example DVB-RCS (EN 301 790 [6]), DVB-RCP (ETS 300 801 [7]), DVB-RCG (EN 301 195 [8]), DVB-RCC (ES 200 800 [9]).

DVB "Network Independent Protocols for DVB Interactive Services" (ETS 300 802 [11]) may be adopted to achieve maximum network interoperability. Other simpler or optimized solutions (e.g. to guarantee minimum signalling delay) may be adopted to directly interface with the aforementioned DVB interactive systems.

Read (0) = Read [K_{bch} (Normal FECFRAME) - 80] bits when available, otherwise dummy.

Read (1) = Round-robin polling. Read [K_{bch} (Normal FECFRAME) - 80] bits from port i when available, otherwise poll the next port.

Read (2) = On timeout, read DFL bits from port i and select the shortest FECFRAME containing DFL.

NOTE: Additional merging policy modes may be optionally implemented by manufacturers.

The receiver shall evaluate quality-of-reception parameters, in particular carrier to noise plus interference ratio in dB available at the receiver, indicated as **CNI**. CNI format shall be:

```
CNI = 20 + 10 \{10 \text{ Log}_{10}[C / (N + I)]\} (positive integer, 8 bits, in the range 0 to 255).
```

In fact for DVB-S2 10 $Log_{10}[C / (N + I)]$ may be in the range -2 dB to +23,5 dB.

 $10 \, \text{Log}_{10}[\text{C} \, / \, (\text{N} + \text{I})]$ shall be evaluated with a quantized accuracy better than 1 dB (accuracy = mean error + 3 σ , where σ is the standard deviation). Since modulation and coding modes for DVB-S2 are typically spaced 1 dB to 1,5 dB apart, a quantized precision better than 0,3 dB is recommended in order to fully exploit system capabilities. The measurement process is assumed to be continuous. A possible method to evaluate CNI is by using symbols known a-priori at the receiver, such as those in the SOF field of the PLFRAME Header and, when available, pilot symbols.

CNI and other optional reception quality parameters (such as for example the BER on the channel evaluated by counting the errors corrected by the LDPC decoder, the packet error rate detected by CRC-8, the CNI distance from the QEF threshold) may optionally be used by the receiver to identify the maximum throughput DVB-S2 transmission mode that it may decode at QEF, indicated by MODCOD_RQ (7 bits, b_6 , ..., b_0) where:

- (b₄, ..., b₀) are coded according to MODCOD in table 12;
- b_5 indicates the presence/absence of pilots: ($b_5 = 0$ no pilots, $b_5 = 1$ pilots);
- $b_6 = 1$ indicates $(b_5, ..., b_0)$ are valid; $b_6 = 0$ indicates $(b_5, ..., b_0)$ information is not available by the terminal.

As a minimum, the CNI and MODCOD_RQ parameters shall be sent to the satellite network operator Gateway every time the protection on the DVB-S2 channel has to be changed. When no modification of the protection level is requested, the optional message from the terminal to the Gateway shall indicate MODCOD_RQ = actual MODCOD and pilot configuration of the frames received by the terminal. In specific applications, CNI and MODCOD_RQ fields may be extended to an integer number of byte(s), by padding zeroes in MSB positions.

The maximum delay required for CNI and MODCOD evaluation and delivery to the Gateway via the interaction channel shall be no more than 300 ms, but this delay should be minimized if services interruptions are to be avoided under fast fading conditions (C/N+I variations as fast as 0,5 dB/s to 1 dB/s may occur in Ka band). Optionally the gateway may acknowledge the reception of the message and the execution of the command by a message containing the new adopted MODCOD, coded according to table 12. The allocated protection shall be equal or more robust than that requested by the terminal.

Example Transmission Protocol using [11]

DVBS2_Change_Modcod message shall be sent from the receiving terminal to the satellite network operator gateway, every time the protection on the DVB-S2 channel has to be changed.

```
DVBS2_Change_Modcod() length in bits (big-endian notation)
{
    CNI;    8
    MODCOD_RQ;  8
}
```

DVBS2_Ack_Modcod message shall optionally be sent from the Gateway to the receiving terminal to acknowledge the DVB-S2 protection level modification. MODCOD_ACK shall be coded according to the MODCOD_RQ conventions.

```
DVBS2_Ack_Modcod() length in bits (big-endian notation)
{
     MODCOD_ACK; 8
}
```

Annex E (normative): SI and signal identification for DSNG and contribution applications

In DSNG transmissions, editing of the SI tables in the field may be impossible due to operational problems. Therefore, only the following MPEG.2-defined SI tables PAT, PMT and Transport Stream Descriptor Table (TSDT) are mandatory. DSNG transmission using DVB-S2 shall implement SI according to annex D of EN 301 210 [3].

Satellite transmissions may be affected by interference problems, which may be generated by SNG stations not strictly adhering to standard operating regulations. Although solutions to this problem are mostly based on operational rules, DVB-S2 provides technical means to allow interfering station identification. DVB-S2 up-link stations (except stations for broadcast services) shall make their signal identifiable by applying the Physical Layer Scrambling initialization sequence n (n in the range 0 to 262 141; see clause 5.5.4) assigned to each station owner.

Annex F (normative): Backwards Compatible modes (optional)

Optional Backwards Compatible (BC) modes are intended to send, on a single satellite channel, two Transport Streams, the first (High Priority, HP) being compatible with DVB-S receivers (according to EN 300 421 [2]) as well as with DVB-S2 receivers, the second (Low Priority, LP) being compatible with DVB-S2 receivers only. Backwards compatibility can be optionally implemented according to two approaches:

- layered modulations, where a DVB-S2 and a DVB-S signals are asynchronously combined on the radio-frequency channel (therefore this operational mode does not require any specific tool in the DVB-S2 specification);
- hierarchical modulation, where the two HP and LP Transport Streams are synchronously combined at modulation symbol level on an asymmetric 8PSK constellation.
- NOTE 1: Hierarchical modes are also used in EN 300 744 [i.2].
- NOTE 2: Also other non-DVB legacy broadcast services based on QPSK may transmit a DVB-S2 LP stream, using the aforementioned hierarchical and layered modulation schemes.

Figure F.1 schematically represents the hierarchical transmission system structure: it consists of two branches, the first compliant with DVB-S standard for the high priority layer, the second increasing the constellation dimensionality to a non-uniform 8PSK for the low priority layer.

Figure F.2 shows the non-uniform 8PSK transmitted constellation.

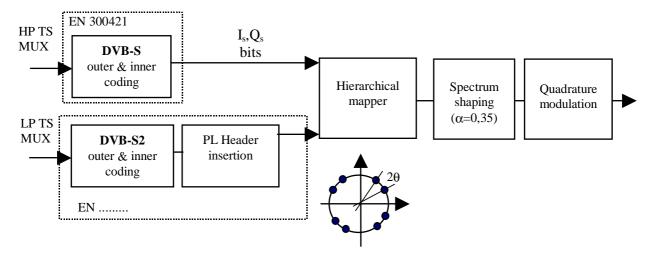


Figure F.1: Functional block diagram of hierarchical backward compatible DVB-S2 system

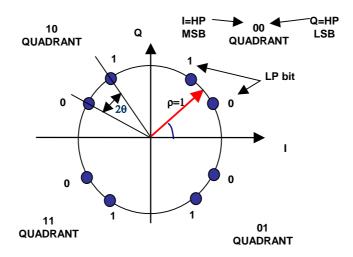


Figure F.2: Non-uniform 8PSK constellation

For hierarchical modulation, the LP DVB-S2 compliant signal shall be BCH and LDPC encoded, with LDPC code rates 1/4, 1/3, 1/2 or 3/5. The LP stream shall be generated by the DVB-S2 system according to figure 1. The hierarchical mapper shall map three bits per symbol as in the figure F.2:

 one bit from the DVB-S2 LDPC encoded signal, following PLHEADER insertion; two bits from the HP DVB-S encoder.

The deviation angle θ may vary according to user requirements.

NOTE 3: θ may be constrained by legacy receiver performance, such as carrier recovery cycle slips: large θ figures improve LP ruggedness against noise and interference, but penalize HP. clause H.6 gives examples of the HP and LP performance versus θ .

The PLHEADER sequence shall comprise 90 bits to indicate the LP code rate being transmitted, as follow, shown octal encoded, MSB (leftmost bit) transmitted first:

- 441647550603300214057262124712 indicates 1/4 rate;
- 753012530643221406326217103300 indicates 1/3 rate;
- 247124232750142006603055713411 indicates 1/2 rate;
- 006604742326301422613065240657 indicates 3/5 rate.

Dummy Frames and short frames shall not be used for BC modes.

Table F.1 gives the number of SLOTs per FECFRAME in the LP stream. Table F.2 gives the bit-rate ratio (%) of the LP stream versus the HP stream.

Table F.1: S = number of SLOTs (M = 90 symbols) per FECFRAME

n _{ldpc} = 64 800 (normal frame)			
η _{MOD} (bit/s/Hz) for LP stream	S		
1	720		

Table F.2: (LP bit-rate/HP bit-rate)x100 [%]

	DVB-S2 Coding (LP)			
DVB-S Coding (HP)	1/4	1/3	1/2	3/5
1/2	26,6	35,7	53,7	64,6
2/3	20,0	26,7	40,3	48,4
3/4	17,8	23,8	35,8	43,0
5/6	16,0	21,4	32,2	38,7
7/8	15,2	20,4	30,7	36,9

Annex G (informative): Supplementary information on receiver implementation

Receiver specification is not under the scope of the present document. Nevertheless the DVB-S2 specification has been developed devoting a large effort to technical evaluations on the receiver design, in order to guarantee that the end-to-end performance target may be met. Typical impairments that may significantly impact the performance of the receiver are:

- phase noise of the LNB and tuner;
- quality of the transmitter and/or receiver oscillators;
- adjacent channel interference;
- satellite non-linearity.

This clause includes some tutorial material on receiver implementation, although other techniques may be used offering the target functionalities and receiver performance.

G.1 Carrier recovery

A significant portion of DVB-S2 deployment will replace/enhance the current DVB-S users. In this situation, it is highly desirable that the outdoor equipment including the LNB will be retained. Extensive simulation studies have been carried out, using the critical phase noise model of consumer LNB from annex H.8, and implementing variants of three baseline carrier recovery techniques:

- decision directed digital phase locked loops with "blind" phase error detection (DD-PLL), operating on the modulated symbols;
- pilot-symbol aided DD-PLL (phase estimations extracted from the pilot symbols and from the PLHEADER are used to reset the DD-PLL circuit, which otherwise operates on the modulated symbols);
- pilot-symbol interpolation techniques (phase estimations extracted from the pilot symbols are interpolated over the modulated symbols).

With the receiver algorithms taken into consideration, the simulation results demonstrate that carrier synchronization can be achieved in presence of phase noise and worst-case thermal noise for any DVB-S2 mode. Some of the transmission modes, such as for example 8PSK rate 3/5 and rate 2/3, 16APSK rate 3/4, 32APSK 4/5 required the use of pilot symbols to avoid cycle slips. For ACM operation, the use of pilot symbols can guarantee continuous receiver synchronization.

G.2 FEC decoding

LDPC codes are linear block codes with sparse parity check matrices $H_{(n-k)xn}$.

As an example, an LDPC code of length N = 8 and rate $\frac{1}{2}$ can be specified by the following parity check matrix.

$$H = \begin{bmatrix} n_1 & n_2 & n_3 & n_4 & n_5 & n_6 & n_7 & n_8 \\ 1 & 0 & 0 & 1 & 1 & 0 & 0 & 1 \\ 0 & 1 & 1 & 0 & 1 & 0 & 1 & 0 \\ 1 & 0 & 1 & 0 & 0 & 1 & 0 & 1 \\ 0 & 1 & 0 & 1 & 0 & 1 & 1 & 0 \end{bmatrix} \begin{matrix} m_1 \\ m_2 \\ m_3 \\ m_4 \\ \end{matrix}$$

The same code can be equivalently represented by the bipartite graph in figure G.1(a) which connects each check equation (check node) to its participating bits (bit nodes).

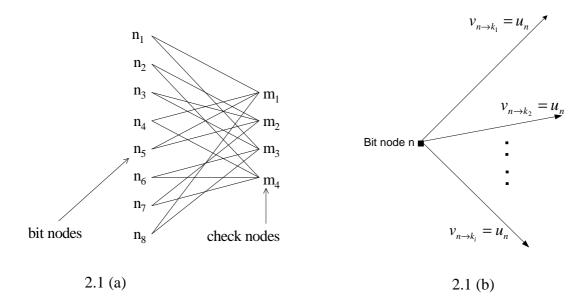


Figure G.1: (a) Bipartite graph of an LDPC code; (b) Initialization of outgoing messages from bit nodes

The purpose of the decoder is to determine the transmitted values of the bits. Bit nodes and check nodes communicate with each other to accomplish that. The decoding starts by assigning the channel values to the outgoing edges from bit nodes to check nodes. Upon receiving that, the check nodes make use of the parity check equations to update the bit node information and send it back. Each bit node then performs a soft majority vote among the information reaching him. At this point, if the hard decisions on the bits satisfy all of the parity check equations, it means a valid codeword has been found and the process stops. Otherwise bit nodes go on sending the result of their soft majority votes to the check nodes. In the following clauses, we describe the decoding algorithm in detail. The number of edges adjacent to a node is called the degree of that node.

• Initialization:

$$v_{n \to k_i} = u_n$$
, $n = 0,1,...,N-1$, $i = 1,2,...,\deg(bit \, node \, n)$

Here $v_{n \to k_i}$ denotes the message that goes from bit node n to its adjacent check node k_i , u_n denotes the channel value for the bit n and N is the codeword size. The initialization process is also shown in figure G.1(b).

Check node update:

Let us denote the incoming messages to the check node k from its d_c adjacent bit nodes by $v_{n_1 \to k}, v_{n_2 \to k}, ..., v_{n_{dc} \to k}$ (see figure G.2(a)). Our aim is to compute the outgoing messages from the check node k back to d_c adjacent bit nodes. Let us denote these messages by $w_{k \to n_1}, w_{k \to n_2}, ..., w_{k \to n_{dc}}$.

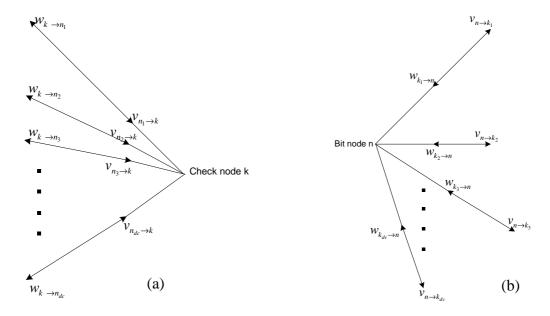


Figure G.2: Message update at check nodes (a), and at bit nodes (b)

$$w_{k \to n_i} = g(v_{n_1 \to k}, v_{n_2 \to k}, ..., v_{n_{i-1} \to k}, v_{n_{i+1} \to k}, ..., v_{n_{dc} \to k})$$

where $g(a,b) = sign(a) \times sign(b) \times \{\min(|a|,|b|)\} + LUT_g(a,b);$

and
$$LUT_{\sigma}(a,b) = \log(1 + e^{-|a+b|}) - \log(1 + e^{-|a-b|})$$
.

• Bit Node Update:

Let us denote the incoming messages to the bit node n from its d_v adjacent check nodes by $w_{k_1 \to n}, w_{k_2 \to n}, \dots, w_{k_{dv} \to n}$ (see figure G.2(b)). Our aim is to compute the outgoing messages from the bit node n back to d_v adjacent check nodes. Let us denote these messages by $v_{n \to k_1}, v_{n \to k_2}, \dots, v_{n \to k_{dv}}$.

They are computed as follows:
$$v_{n \to k_i} = u_n + \sum_{i \neq i} w_{k_i \to n}$$
.

• Hard Decision Making:

After the bit node updates, hard decision can be made for each bit n by looking at the sign of $v_{n \to k_i} + w_{k_i \to n}$ for any k_i . If the hard decisions satisfy all the parity check equations, it means a valid codeword has been found, therefore the process stops. Otherwise another check node/bit node update is performed. If no convergence is achieved after a pre-determined number of iterations, the current output is given out. As SNR increases, the decoder converges with fewer iterations.

G.3 ACM: Transport Stream regeneration and clock recovery using ISCR

When the modulator operates in ACM mode (null-packet deletion active), the receiver may regenerate the Transport Stream by inserting, before each useful packet, DNP in the reception FIFO buffer. As shown in figure G.3, the Transport Stream clock R'_{IN} may be recovered by means of a Phase Locked Loop (PLL). The recovered symbol-rate R_s may be used to clock a local counter (which by definition runs synchronously with the input stream synchronization counter of figure D.2, apart from the Satellite Doppler frequency shift). The PLL compares the local counter content with the transmitted ISCR of each TS packet, and the phase difference may be used to adjust the R'_{IN} clock. In this way R'_{IN} remains constant, and the reception FIFO buffer automatically compensates the chain delay variations. Since the reception FIFO buffer is not self-balancing, the BUFSTAT and the BUFS information may be used to set its initial state.

As an alternative, when dynamic variations of the end-to-end delay and bit-rate may be acceptable by the source decoders, the receiver buffer filling condition may be used to drive the PLL. In this case the reception buffer is self-balancing (in steady state half of cells are filled), and the ISSY field may be omitted at the transmitting side.

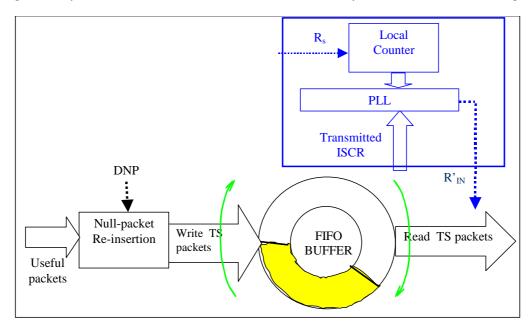


Figure G.3: Example receiver block diagram for Null-packet re-insertion and R_{TS} clock recovery

G.4 Non linearity pre-compensation and Intersymbol Interference suppression techniques

A significant feature of DVB-S2 standard is to support high-order modulation such as 16 APSK and 32 APSK. These modulation schemes, although specifically designed for non-linear channels, are particularly sensitive to the characteristics of the satellite transponders. Computer simulation studies, based on the use of the satellite transponder model of clause H.7, demonstrated that there are significant opportunities to further enhance the performance by pre-distortion of the transmitted signal and/or intersymbol interference suppression technique in the receiver.

G.5 Interactive services using DVB-RCS return link: user terminal synchronization

Interactive services can be operated with a DVB-RCS (EN 301 790 [6]) return path, provided that an absolute time reference (NCR, Network Clock Reference) can be generated in the user terminal for transmissions alignment. In DVB-RCS the hub broadcasts the NCR in the form of special transport packets over the forward link. In case of DVB-S2 forward link, NCR is associated to the emission time, at the transmitting side, of the first symbol of the SOF field.

In order to facilitate RCS synchronization at user terminal, a "SOF flag" output should be included in the DVB-S2 receiver chipset. Furthermore, in order to allow alignment of the SOF flag with the relevant NCR, the receiver chipset should implement an internal counter of the received physical layer frames (e.g. modulo M=32), with arbitrary start-up. The counter content should label both the "SOF flag" and the decoded data at the chip output. In practical implementations the SOF flag label could be signalled serially on the SOF flag signal and the frame label on another signal.

Annex H (informative): Examples of possible use of the System

H.1 CCM digital TV broadcasting: bit rate capacity and C/N requirements

The DVB-S2 system may be used in "single carrier per transponder" or in "multi-carriers per transponder" (FDM) configurations. In single carrier per transponder configurations, the transmission symbol rate R_S can be matched to given transponder bandwidth BW (at -3 dB), to achieve the maximum transmission capacity compatible with the acceptable signal degradation due to transponder bandwidth limitations. To take into account possible thermal and ageing instabilities, reference can be made to the frequency response mask of the transponder. Group delay equalization at the transmitter may be used to increase the transmission capacity or to reduce degradation.

In the multi-carrier FDM configuration, R_S can be matched to the frequency slot BS allocated to the service by the frequency plan, to optimize the transmission capacity while keeping the mutual interference between adjacent carriers at an acceptable level.

Figure H.1 gives examples of the useful bit rate capacity R_u achievable by the System versus the LDPC code rate, assuming unit symbol rate R_S . The symbol rate R_S corresponds to the -3 dB bandwidth of the modulated signal. R_S (1+ α) corresponds to the theoretical total signal bandwidth after the modulator. The figures refer to Constant Coding and Modulation, normal FECFRAME length (64 800 bit), no PADDING field, no pilots (the pilots would reduce the efficiency by about 2,4 %). Typical BW/ R_S or BS/ R_S ratios are 1 + α = 1,35 and 1,20, where α is the roll-off factor of the modulation. This choice allows to obtain a negligible E_S/N_o degradation due to transponder bandwidth limitations, and also to adjacent channel interference on a linear channel. BW/ R_S factors < 1 + α may also be adopted, but careful studies should be carried-out on a case-by-case basis to avoid unacceptable interference and distortion levels.

Figure H.2 shows the required C/N (Carrier-to-Noise power ratio measured in a bandwidth equal to the symbol rate) versus the spectrum efficiency (useful bit-rate for unit symbol rate R_S), obtained by computer simulations on the AWGN channel (ideal demodulator, no phase noise). The performance of DVB-S and DVB-DSNG is also shown for comparison, for the same quality target PER = 10^{-7} . Before Nyquist filtering in the modulator, the peak-to-average power ratio is 0 dB for QPSK and 8PSK, while it is in the range $[1,05 \div 1,11]$ dB for 16APSK (the exact value can be calculated using expression 4 $\gamma^2/(3\gamma^2+1)$) and $[1,97 \div 2,12]$ dB for 32APSK (the exact expression is $8\gamma_2^2/(4\gamma_2^2+3\gamma_1^2+1)$). When DVB-S2 is transmitted by satellite, quasi-constant envelope modulations, such as QPSK and 8PSK, are power efficient in single carrier per transponder configuration, since they can operate on transponders driven near saturation. 16APSK and 32APSK, which are inherently more sensitive to non-linear distortions and would require quasi-linear transponders (i.e. with larger Output-Back-Off, OBO) may be improved in terms of power efficiency by using non-linear compensation techniques in the up-link station.

The use of the narrower roll-off $\alpha = 0.25$ and $\alpha = 0.20$ may allow a transmission capacity increase but may also produce larger non-linear degradations by satellite for single carrier operation.

In the FDM configuration, the satellite transponder should be quasi-linear (i.e. with large Output-Back-Off, OBO) to avoid excessive intermodulation interference between signals.

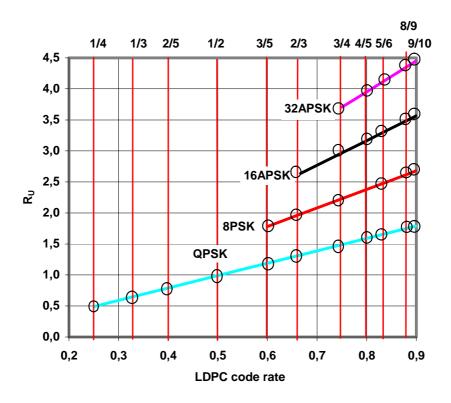


Figure H.1: Examples of useful bit rates R_u versus LDPC code rate per unit symbol rate R_s

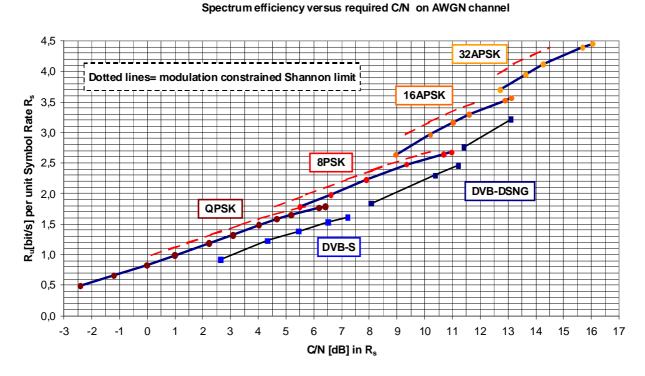


Figure H.2: Required C/N versus spectrum efficiency, obtained by computer simulations on the AWGN channel (ideal demodulator) (C/N refers to average power)

ETSI

Table H.1 shows the C/N degradation measured by computer simulations using the satellite channel models given in annexes H.7 (non linearized TWTA) and H.8. The following parameters have been simulated: $R_S = 27.5$ Mbaud, roll-off = 30 % (not available in DVB-S2, but giving performance between roll-off 0,35 and 0,25), dynamic pre-distortion memory M = 5 (for QPSK, 8PSK and 16APSK), M = 3 (for 32APSK). C_{SAT} is the un-modulated carrier power at HPA saturation, OBO is the measured power ratio (dB) between the un-modulated carrier at saturation and the modulated carrier (after OMUX). The considered phase noise mask is relevant to a low-cost consumer LNB, while for professional applications lower degradations may be expected.

Transmission Mode	C _{SAT} /N loss [dB] no predistortion without Phase Noise	C _{SAT} /N loss [dB] with dynamic predistortion without Phase Noise	C _{SAT} /N loss [dB] with dynamic predistortion with Phase Noise
QPSK 1/2	0,62 (IBO = 0; OBO = 0,33)	0.5 (IBO = 0 dB; OBO = 0.38)	0,63
8PSK 2/3	0,95 (IBO = 0,5; OBO = 0,35)	0,6 (IBO = 0;OBO = 0,42)	0,85
16APSK 3/4	3,2 (IBO = 5; OBO = 1,7)	1,5 (IBO = 1; OBO = 1,1)	1,8
32APSK 4/5	6,2 (IBO = 9; OBO = 3,7)	2,8 (IBO = 3,6; OBO = 2,0)	3,5

Table H.1: C/N degradation [dB] on the satellite channel (simulation results)

H.2 Distribution of multiple TS multiplexes to DTT Transmitters (Multiple TS, CCM)

The DVB-S2 system is suitable for the distribution of N MPEG multiplexes to digital terrestrial transmitters, using a single carrier per transponder configuration, thus optimizing the power efficiency by saturating the satellite HPA (with the DVB-S system N carriers per transponder should be transmitted, requiring a large HPA OBO). For example, assuming the availability of a BW = 36 MHz transponder, a symbol rate of 30 Mbaud may be transmitted using $\alpha = 0.20$. Thus to transmit two DTT MUXes at 24 Mbit/s each, a spectrum efficiency of 1,6 [bit/s/Hz] is required, corresponding to QPSK rate 5/6 (required C/N = 5,2+0,7 dB + implementation margin). Figure H.3 shows an example of possible configuration at the transmitting side.

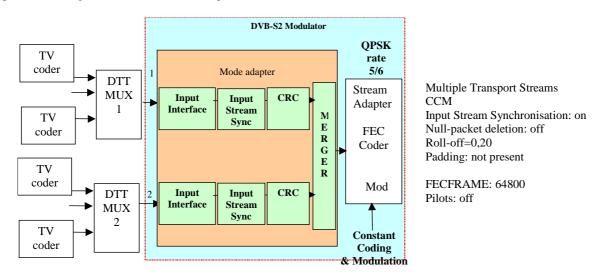


Figure H.3: Example of DVB-S2 configuration for multiple DTT multiplexes distribution by satellite

H.3 SDTV and HDTV broadcasting with differentiated protection (VCM, Multiple TS)

The DVB-S2 system may deliver broadcasting services over multiple Transport Streams, providing differentiated error protection per Mux (VCM mode). A typical application is broadcasting of a highly protected Mux for SDTV, and of a less protected Mux for HDTV. It should be noted that the DVB-S2 system is unable to differentiate error protection within the same TS Mux. Figure H.4 shows an example configuration at the transmitting side. Assuming to transmit 27,5 Mbaud and to use 8PSK 3/4 and QPSK 2/3, 40 Mbit/s would be available for two HDTV programmes and 12 Mbit/s for two SDTV programmes. The difference in C/N requirements would be around 5 dB to 5,5 dB.

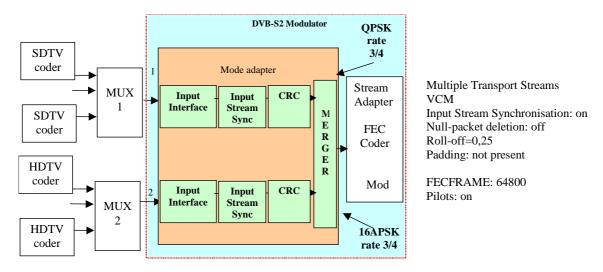


Figure H.4: Example DVB-S2 configuration for TV and HDTV broadcasting using VCM

H.4 DSNG Services using ACM (Single transport Stream, information rate varying in time)

In point-to-point ACM links, where a single TS is sent to a unique receiving station (e.g. DSNG), the TS packets protection should follow the C/N+I variations on the satellite channel in a given receiving location. When propagation conditions change (see figure H.5, arrow), the PLFRAMEs F_i switch from protection mode M_j to protection mode M_K to guarantee the service continuity.

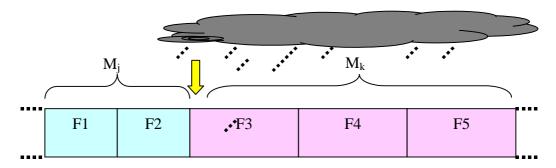


Figure H.5: PLFRAMEs changing protection during a rain fading

The DVB-S2 system may operate as follows (see figure H.6, showing also example evaluation of the chain loop delays D.1 to D.8):

1) The bit-rate control unit keeps the VBR source bit-rate (e.g. video encoder) at the maximum level compatible with the actual channel conditions C/N+I. In parallel, it sets the DVB-S2 modulator transmission mode via the "ACM Command" input port.

- 2) The TS bit-rate is set at the maximum level receivable at QeF in clear sky. The TS multiplexer adds null-packets to generate the constant bit-rate R_{TS}.
- 3) Null Packets (NP) are deleted in the Mode Adapter, so that the actual bit-rate on the channel corresponds to the source bit-rate. The deleted NPs are signalled in the DNP byte.
- 4) The receiver re-inserts Null Packets exactly in the original position (by decoding DNP), and the Transport Stream clock is regenerated using the Input Stream Clock Reference (ISCR).

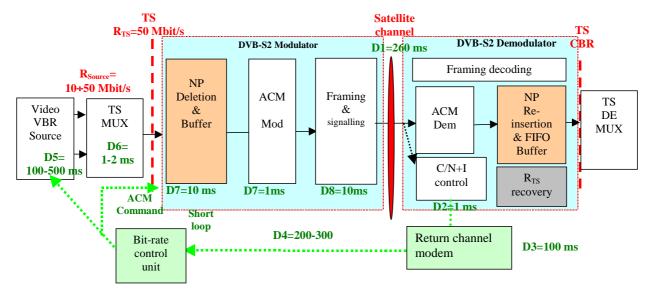


Figure H.6: Single TS - uniform protection for long periods: transmission and receiving schemes

With reference to figure H.6, during a fast fading the bit rate control unit may impose a rate reduction first on the source encoder, and only after the command has been executed (e.g. after 100 ms to 500 ms), to the DVB-S2 modulator (via ACM Command). A drawback of this configuration is that the video encoder and MUX delays (D5 and D6 in figure H.6) are included in the control loop, with the risk of service outage under fast fading conditions. To overcome this additional delay the ACM Command can be instantly delivered also to the modulator, but to avoid packet losses large buffers have to be inserted in the DVB-S2 modulator and demodulator.

H.5 IP Unicast Services (Non-uniform protection on a user-by-user basis)

Figure H.7 shows a possible exchange of information (info request and info response) between the user, the Satellite Gateway and the information provider during an Internet navigation session by satellite (forward high capacity link).

These interactive data services may take advantage of:

- non-uniform error protection (ACM);
- differentiated service levels (priority in the delivery queues).

According to the negotiation between the user terminal and the "ACM routing manager", an "ACM router" may in principle separate IP packets per user, per required error protection and per service level. The aggregate input traffic on the various protection levels should not overload the available channel capacity; this applies to the average input traffic, while the peak traffic may temporarily exceed it, compatibly with the input buffering capacity and the service requirements on maximum delays.

To fulfil this rule, when the total offered traffic becomes larger than the channel capacity, lower priority IP packets may be delayed (or even dropped) in favour of high priority packets. If the control-loop delays (including routing manager and ACM router) are too large to allow error free reception under fast-fading conditions, real time services (e.g. video/audio streaming) may be permanently allocated to a high protection branch, while lower priority services (e.g. best effort) may exploit the higher efficiency (i.e. lower cost) provided by ACM. It should be noted that the polling strategy of the input buffers may be statically or dynamically profiled according to the traffic statistics, the propagation characteristics, and the traffic prioritization policy of the service operator.

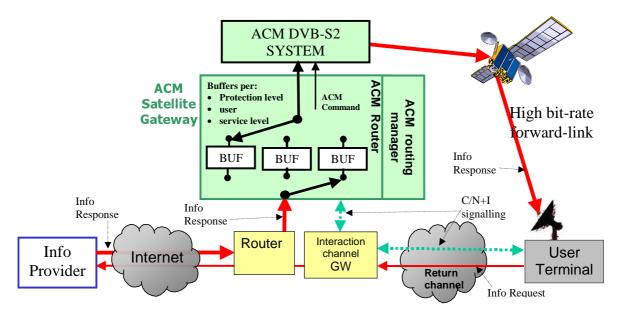


Figure H.7: Example of IP services using a DVB-S2 ACM link

The ACM router may interface with the DVB-S2 modulator:

- via a Single Generic Stream input and the ACM Command input. In this case the ACM router is independent from the DVB-S2 modulator, and may implement any routing policy. The DVB-S2 modulator immediately transmits the user data according to the ACM Command, therefore the loop delays may be minimized;
- via Multiple (Transport or Generic) Stream inputs, one per each active protection level (the ACM Command interface needs not be active). In this case the DVB-S2 Merger/Slicer partially covers the functionality of the ACM router.

This latter case is represented in more detail in figure H.8. The ACM router splits the users" packets per service level (priority) and per required protection level, and sends them to the multiple DVB-S2 input interfaces, each stream being permanently associated to a given protection level. Therefore each input stream merges the traffic of all the users needing a specific protection level, and its useful bit-rate may (slowly) change in time according to the traffic characteristics. According to table D.2, the Merger/Slicer in figure H.8 cyclically polls the input TS buffers, and conveys to the ACM modulator a block of "users" data ready to fill (or partially fill) a PLFRAME. A timeout may be defined in order to avoid long delays in each merger/slicer buffer.

Dotted boxes in figure H.8 address the specific case of IP services encapsulated in Transport Streams (Multi-Protocol Encapsulation - MPE), according to EN 301 192 [4]. In this case, K MPE gateways (GTW_i) are associated to K TS Multiplexers, to feed K DVB-S2 input streams (one per active protection level). Null-packet deletion, applied to each branch, reduces the transmitted bit-rate. The decoded TS, after null packets re-insertion, is a valid TS (the input stream synchronizer may optionally be activated). To fully exploit the potential ACM advantages, the additional control-loop delays introduced by the TS-specific equipment (Gateways, TS Muxes) should be minimized.

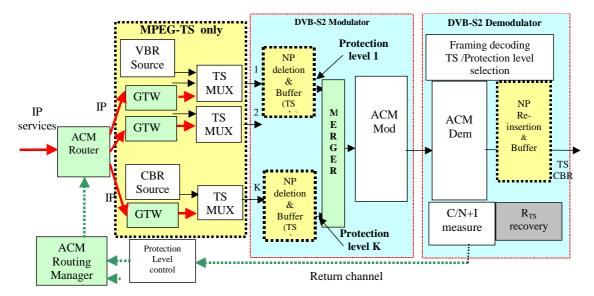


Figure H.8: IP Unicasting and ACM: Multiple input streams - uniform protection per stream (for Generic input Streams, GTWs, TS Muxes and null-packet deletion are not required)

H.6 Example performance of BC modes

Figure H.9 shows the C/N performance (linear AWGN channel) of the HP stream (DVB-S) and of LP stream (DVB-S2) versus the deviation angle θ , achieved under the simplifying assumption that the LP stream degrades the HP stream as a Gaussian noise of equal power (the C/N figures are according to EN 300 421 [2] for HP and according to table 13 - scaled to BPSK format - for LP, including 0,8 dB implementation margin for both streams). The points of intersection of the curves correspond to balanced HP and LP performance. For example, assuming an available C/N = 10 dB, adopting QPSK 7/8 for the HP DVB-S stream, $\theta = 12^{\circ}$ and rate 1/3 LP DVB-S2 stream, the total bit rate per unit bandwidth is 1,61 (HP) + 0,33 (LP) = 1,94 bit/s/Hz. In comparison, a DVB-S-only transmission would not be able to fully exploit the available C/N to maximize the total bit rate. The most significant advantages (additional bit-rate at a given C/N) of the BC DVB-S2 system over the legacy DVB-S system can be obtained for large available C/N (i.e. > 9 dB) and for spectral efficiencies above the maximum level offered by DVB-S (QPSK 7/8).

In practical cases, transmission parameters (code rates, θ) should be optimized on a case-by-case basis, taking into account the real channel characteristics.

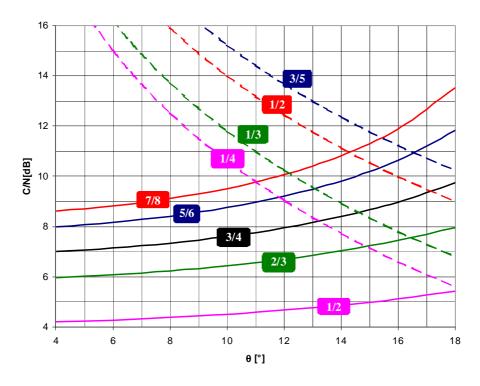


Figure H.9: Example performance of HP and LP backwards compatible streams versus θ

H.7 Satellite transponder models for simulations

For simulations, the "transparent" (i.e. non regenerative) satellite transponder model may be composed of an input filter (IMUX), a power amplifier (TWT or SSA) and an output filter (OMUX). Two amplifier models are here defined, the linearized TWTA (LTWTA) and the non-linearized TWTA. SSPAs have not been considered since they are less critical than TWTAs in terms of degradations.

The reference symbol rate with the specified IMUX/OMUX filter bandwidth is Rs = 27,5 Mbaud.

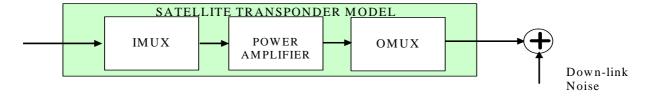
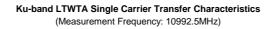


Figure H.10: Satellite transponder model

Figures H.11 and H.12 give the AM/AM and AM/PM TWTA characteristics.



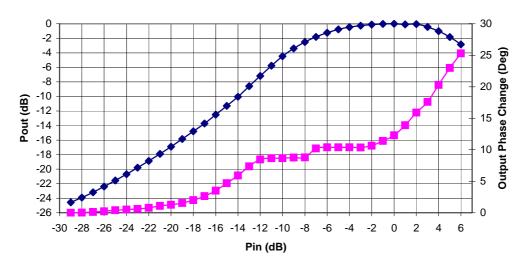


Figure H.11: Linearized TWTA characteristic

Ka-band TWTA - Single Carrier Transfer Characteristics

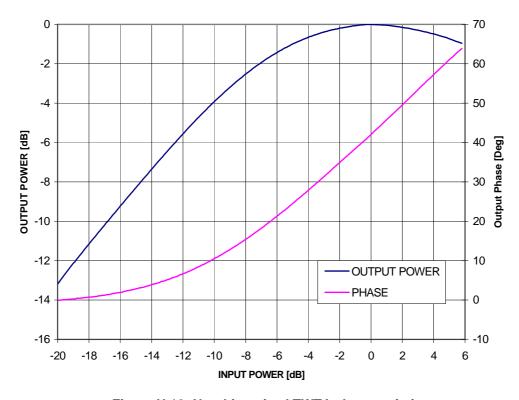
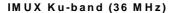
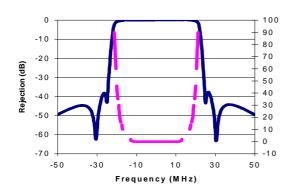


Figure H.12: Non-Linearized TWTA characteristic



OMUX Ku-band (36 MHz)



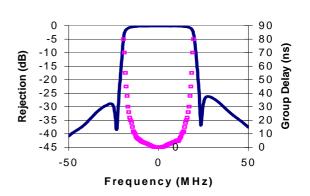


Figure H.13: IMUX and OMUX characteristics

Other transponder bandwidths BW [MHz] may be obtained by scaling the IMUX and OMUX characteristics:

- $R(f) = Rejection [f \times (BW/36)].$
- $G(f) = [(36/BW)] \times Group-delay [f \times (BW/36)].$

The band-centre insertion loss is not indicated, but should be included in C_{SAT} for link budget computation.

H.8 Phase noise masks for simulations

The following phase noise masks for consumer reception systems may be used to evaluate the carrier recovery algorithms. The mask represents single side-band power spectral densities. The "aggregate" masks combine the phase noise contributions of the LNB and of the relevant Tuner. Other sources of phase noise within the chain (e.g. satellite transponder, up-link station, etc.) are usually negligible, and therefore the proposed masks may be considered as representative of the full chain.

Table H.2: Aggregate Phase Noise masks for Simulation (in dBc/Hz)

frequency ⇒	100 Hz	1 kHz	10 kHz	100 kHz	1 MHz	> 10 MHz
Aggregate1 (typical)	-25	-50	-73	-93	-103	-114
Aggregate2 (critical)	-25	-50	-73	-85	-103	-114

Annex I (normative): Mode Adaptation input interfaces (optional)

I.1 Mode Adaptation input interface with separate signalling circuit (optional)

Mode Adaptation optional input interface (see figure 1) shall allow implementing the merging of multiple input streams by an external "Mode Adaptation Unit", respecting all the rules of the DVB-S2 specification. To allow to vary the transmission parameters to be adopted by the DVB-S2 modulator, it shall also transport the ACM command associated to each specific Data Field.

According to figure 3 Mode Adaptation shall be a sequence of Data Fields (according to clause 5.1.5), where each individual Data Field is preceded by a BBHEADER, according to clause 5.1.6 and to figure 3, and a Stream Adaptation command (SA command), transporting the transmission parameters to be adopted by the DVB-S2 modulator for each specific Data Field and corresponding BBHEADER.

"SA Command" (similar to the ACM command format, see clause D.1) shall carry the following information:

- MODCOD (5 bits, according to table 12).
- TYPE (2 bits, according to clause 5.5.2.3).
- CVALID (Command Valid).
- SEND (end of MA Packet).

The CVALID=active indicates the start of a MA Packet (MSB of the BB Header).

The transmission format specified by MODCOD and TYPE shall be applied to MA Packet received after CVALID=active and before SEND=active. When SEND=active, the modulator shall deliver user data immediately, even if a FECFRAME is not completed, by inserting the PADDING field (see clause 5.2.1). The user data included in the interval between CVALID=active and SEND=active shall not exceed the capacity of $(K_{bch}-80)$ bits, K_{bch} being the transmittable bits associated with a specific MODCOD and TYPE.

An example temporisation of SA Command is given in figure I.1, using a single serial interface to convey MODCOD, TYPE, CVALID(active= high-to-low transition) and SEND (active= low-to-high transition).

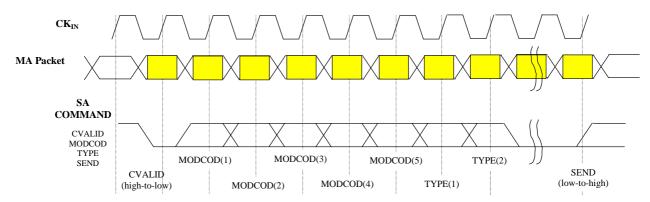


Figure I.1: Example temporisation of SA Command (serial format)

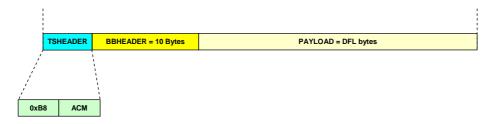
I.2 Mode Adaptation input interface with in-band signalling (optional)

Alternatively to clause I.1, the SA command can be mapped into a Transport Header to be prepended to the data generated by the external Mode Adaptation Unit. According to figure I.1, Mode Adaptation shall be a sequence of Data Fields (according to clause 5.1.5), where each individual Data Field is preceded by a BBHEADER, according to clause 5.1.6 and to figure 3, and a Transport Header.

The Transport Header shall consist of 2 bytes as illustrated in figure I. 2 and defined in table I.1. The first byte identifies the start of the Mode Adaptation packet and shall correspond to the sequence 0xB8. The second byte shall indicate the ACM command, defining the dynamic transmission parameters (MODCOD, TYPE) for the BBFRAME, according to table I. 2.

The BBFRAME shall consist of a valid BBHEADER, followed by the payload with length DFL, without padding bytes.

Stream Adaptation shall synchronize to the baseband frames (using the 0xB8 syncmarker and the DFL field inside the BBHEADER.



Transport Header: 2 Bytes

Figure I.2: Mode Adaptation format at the Mode Adaptation input interface

Table I.1: Transport Header format

Byte	Contents	Purpose
Byte 0	0xB8 syncmarker	For BBF synchronization
Byte 1	ACM command byte	Defines modcod, frametype and pilot insertion

Table I.2: ACM command byte definition (acm[0] is the least significant bit)

Bit fields	Description					
Acm[4:0]	MODCOD (as defined in table 12)					
Acm[5]	pilots configuration (0 = no pilots, 1 = pilots)					
Acm[6]	FECFRAME sizes (0 = normal: 64 800 bits; 1 = short: 16 200 bits)					
Acm[7]	reserved bit (set to 0)					

Annex J (informative): Bibliography

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ETSI ETS 300 327: "Satellite Earth Stations and Systems (SES); Satellite News Gathering (SNG) Transportable Earth Stations (TES) (13-14/11-12 GHz)".

ETSI EN 300 673: "Electromagnetic compatibility and Radio spectrum Matters (ERM); ElectroMagnetic Compatibility (EMC) standard for Very Small Aperture Terminal (VSAT), Satellite News Gathering (SNG), Satellite Interactive Terminals (SIT) and Satellite User Terminals (SUT) Earth Stations operated in the frequency ranges between 4 GHz and 30 GHz in the Fixed Satellite Service (FSS)".

Annex K: For future use

Annex L:

For future use

Annex M (normative):

Transmission format for wideband satellite transponders using time-slicing (optional)

This section specifies the optional transmission format for high symbol-rate satellite carriers for broadcasting, professional and interactive services. This format may optionally be adopted for wideband satellite transponders (e.g. 200 MHz to 500 MHz), where the transmission of a single or few wide-band carriers is preferable to the transmission of a multiplicity of narrow-band carriers, for power and efficiency optimisation or other needs. This format is intended to permit the operation of time-slicing receivers, which are characterised by realtime high-speed coherent-demodulation and PL-Header processing capabilities, but FEC decoding speed significantly lower than that of the wideband carrier. In order to allow such receivers to select and decode a specific stream carrying one or more service(s) within its performance capabilities, while discarding the other streams and services of the wide-band carrier, the transmitter shall map the input services into streams (identified by a specific Time Slice Number, TSN). Such streams shall be transmitted in time-slices (i.e. bursts) suitably spaced in time. A time-slicing burst (identified by a specific TSN) shall correspond to one PL-Frame.

The Time Slice Number TSN -8 bits- may optionally correspond to MATYPE2 ISI field in the BB-Header (clause 5.1.6).

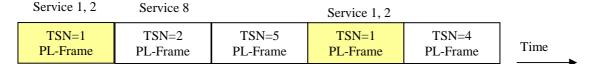


Figure M.1: Example of time-sliced transmission

The receiver can select TSN=1 and decode Service 1 or Service 2, and discard other TSNs and associated services.

Depending on the applications, the time-sliced transmission may correspond to a periodic sequence of slices (e.g. TSN=1, TSN=2,...,TSN=20, TSN=1, TSN=20,...) or to a non-ordered sequence of slices (e.g. TSN=1, TSN=22, TSN=4,...) which may be decided "on-the-fly" at the transmitting side, according to service/traffic needs.

This annex specifies physical layer signalling that shall be introduced in the transmitted waveform to allow receiver configuration in time slicing modes. Algorithms to define the slicing sequence at the transmitter site are left open to optimisation according to use cases. Such algorithms shall satisfy the receiver capabilities as defined in clause M.1. As an example, in broadcasting applications the total wideband symbol rate can be constantly assigned (in static mode) to "virtual carriers" of equal or different capacity, using CCM per virtual carrier. In unicasting ACM applications, where the slice structure should follow the traffic requirements, "on the fly" allocation of resources (in dynamic mode) may offer the best efficiency and flexibility.

Upper layer signalling shall be according to EN 300 468 [12].

M.1 Definition of Time-slicing receiver

Time-slicing receivers are characterized by:

- (i) real-time high-speed coherent demodulation and PL-Header processing capabilities, including continuous PL-frame synchronisation;
- (ii) maximum average decoding speed at FECFRAME level R_{FEC} (e.g. $R_{FEC} = 100 \text{ Mbit/s}$);

NOTE: R_{FEC} may be significantly lower than the wide-band carrier bit-rate).

(iii) a minimum Guard Time T_G in μ sec which must separate two adjacent slices received by the decoder (which may be time variable, and can be better defined for different receiver classes).

M.2 TIME SLICE MODE CODING

This mode shall comply with all the clauses in the main body of this specification, with the exception of the PL-Header structure of clauses 5.5.2 to 5.5.4 which shall be coded according to the following clauses.

M.2.1 PL signalling

The PLHEADER is intended for receiver synchronization and physical layer signalling.

NOTE 1: After decoding the PLHEADER, the receiver knows the PLFRAME duration and structure, the modulation and coding scheme of the XFECFRAME, the presence or absence of pilot symbols.

The PLHEADER shall be extended to two SLOTs of 90 symbols, and shall be composed of the following fields:

- **SOF** (26 symbols), identifying the Start of Frame;
- **PLS** code (154 symbols): PLS (Physical Layer Signalling) code shall be a constraint length k=5, rate 1/5 convolutional code (77,16), whose output bits $(c_0,c_1,c_2,...,c_{76})$ are repeated twice to produce the (154,16) codeword $(c_0,c_0,c_1,c_1,c_2,c_2,...,c_{76},c_{76})$, described by the following generator polynomials:

 $gi=(g_{i,0},g_{i,1},g_{i,2},g_{i,3},g_{i,4})$

g0=(10101); g1=(10111); g2=(11011); g3=(11111); g4=(11001);

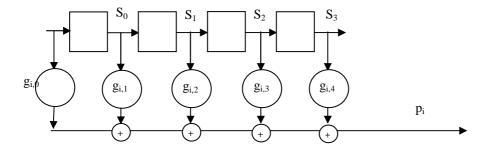


Figure M.2: Convolutional encoding scheme

To output only 77 coded bits instead of 80, the following bits shall be punctured:

If $(u_0, u_1, u_2, u_3, ..., u_{15})$ are the information bits, then each information bit shall generate 5 parity bits $(p_0, p_1, p_2, ..., p_4)$. Then, for information bits u_3, u_8 , and u_{13} , the parity bit p_4 shall be punctured.

"Tail biting" shall be used to complete the encoding process: depending on the input bits, the initial state shall be chosen so that the initial and final states are the same. The encoder initial state shall thus be set as:

$$S_0=u_{15}$$
; $S_1=u_{14}$; $S_2=u_{13}$; $S_3=u_{12}$

Tail bits shall not be transmitted.

NOTE 2: The repetitive structure of the PLS code may be exploited in the receivers for differential detection synchronisation, in presence of frequency and phase errors.

The resulting 154 coded bits shall be scrambled with the following sequence:

The PLHEADER, represented by the binary sequence $(y_1, y_2,...y_{180})$, shall be modulated into 180 π /2BPSK symbols according to the rule:

$$I._{2i-1} = Q._{2i-1} = (1/\sqrt{2}) \ (1-2y_{2i-1}), \ I._{2i} = -Q._{2i} = - \ (1/\sqrt{2}) \ (1-2y_{2i}) \ for \ i=1, \, 2, \, ..., \, 90$$

M.2.2 SOF field

SOF shall be coded according to clause 5.5.2.1.

M.2.3 MODCOD field

• $(u_0, u_1, u_2, u_3, ..., u_5) = MODCOD$

The MODCOD field shall be extended with respect to clause 5.5.2.2, in order to allow additional modulation and coding configurations. The two MSB u_0 and u_1 shall be coded as follows:

 $\begin{array}{lll} u_0\!=\!0 & \text{modes according to table } 12. \\ u_0\!=\!1 \; , & (u_1, \ldots, u_5)\!=\!0 \text{D to } 5 \text{D} & \text{additional modes using QPSK constellations} \\ u_0\!=\!1 \; , & (u_1, \ldots, u_5)\!=\![6 \text{D to } 11 \text{D}] & \text{additional modes using } 8 \text{PSK constellations} \\ u_0\!=\!1 \; , & (u_1, \ldots, u_5)\!=\![12 \text{D to } 17 \text{D}] & \text{additional modes using } 16 \text{APSK constellations} \\ u_0\!=\!1 \; , & (u_1, \ldots, u_5)\!=\![18 \text{D to } 23 \text{D}] & \text{additional modes using } 32 \text{APSK constellations} \\ u_0\!=\!1 \; , & (u_1, \ldots, u_5)\!=\![24 \text{D to } 31 \text{D}] & \text{additional modes using } 64 \text{APSK constellations} \end{array}$

A table to define the extended MODCODE configurations will be included in a later version of the present document.

M.2.4 TYPE field

• $(u_6, u_7) = TYPE$ (according to clause 5.5.2.3).

M.2.5 TSN code

• $(u_8, ..., u_{15})$ = TSN (may correspond to MATYPE-2 field, ISI).

M.3 Phase noise masks

The following typical phase noise masks shall be taken into account for receiver implementation.

Table M.1: PROFILE "Ku- DTH" and "Ka-DTH", SSB (dBc/Hz)

frequency ⇒	100 Hz	1 kHz	10 kHz	100 kHz	1 MHz	10 MHz	> 50 MHz
Aggregate1 (typical)	-25	-50	-73	-93	-103	-114	-117
Aggregate2 (critical)	-25	-50	-73	-85	-103	-114	-117

Table M.2: PROFILE "Ku- Non DTH", SSB (dBc/Hz)

frequency ⇒	10 Hz	100 Hz	1 kHz	10 kHz	100 kHz	1 MHz	10 MHz	> 50 MHz
Aggregate	-33	-62	-79	-89	-99	-109	-119	-120

Table M.3: PROFILE "Ka- Non DTH", SSB (dBc/Hz)

frequency ⇒	10 Hz	100 Hz	1 kHz	10 kHz	100 kHz	1 MHz	10 MHz	> 50 MHz
Aggregate1 (typical)	-33	-62	-79	-89	-95	-106	-116	-118

History

Document history							
V1.1.1	March 2005	Publication					
V1.1.2	June 2006	Publication					
V1.2.1	August 2009	Publication					
V1.3.1	November 2012	EN Approval Procedure	AP 20130301: 2012-11-01 to 2013-03-01				
V1.3.1	March 2013	Publication					