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Digital Video Broadcasting (DVB); Implementation guidelines for the use of Video and Audio Coding in Broadcasting Applications based on the MPEG-2 Transport Stream

European Broadcasting Union



Union Européenne de Radio-Télévision



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Foreword

This Technical Specification (TS) 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 original TR 101 154 was based on the DVB document A001 and it covered only the 25 Hz SDTV Baseline IRD. The first revision of TR 101 154 extended the scope to encompass both the 25 Hz SDTV Baseline IRD and the 25 Hz SDTV IRD with a digital interface intended for connection to a bitstream storage device such as a digital VCR. The second revision covered both the Baseline IRD and the IRD with digital interface for 25 Hz SDTV, 25 Hz HDTV, 30 Hz SDTV and 30 Hz HDTV. Subsequent revisions added optional support for the video Active Format Description (annex B), AC-3 audio (annex C) and Ancillary Data for MPEG audio (annex D) and the Coding of Data Fields in the Private Data Bytes of the Adaptation Field (annex E). This revision adds optional support for DTS audio (annex F) and receiver-mixed audio (annex G). The revisions to TR 101 154 have been developed in a largely backwards compatible manner, i.e. no changes to the mandatory functionality of a previously defined IRD have been made between one edition of TR 101 154 and the next.

The present document is complementary to TS 102 154 [19], which provides Implementation Guidelines for the use of Video and Audio Coding in Contribution and Primary Distribution Applications based on the MPEG-2 Transport Stream.

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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Founded in September 1993, the DVB Project is a market-led consortium of public and private sector organizations in the television industry. Its aim is to establish the framework for the introduction of MPEG-2 based digital television services. Now comprising over 200 organizations from more than 25 countries around the world, DVB fosters market-led systems, which meet the real needs, and economic circumstances, of the consumer electronics and the broadcast industry.

Introduction

The present document presents guidelines covering coding and decoding using the MPEG-2 system layer, video coding and audio coding as defined in ISO/IEC 13818-1 [1], ISO/IEC 13818-2 [2], ISO/IEC 11172-3 [10] and ISO/IEC 13818-3 [3] respectively.

The guidelines presented in the present document for the Integrated Receiver-Decoder (IRD) are intended to represent a minimum functionality that all IRDs of a particular class are required to either meet or exceed. It is necessary to specify the minimum IRD functionality for basic parameters, if broadcasters are not to be prevented from ever using certain features. For example, if a significant population of IRDs were produced that supported only the Simple Profile, broadcasters would never be able to transmit Main Profile bit-streams.

IRDs are classified in three dimensions as:

- "25 Hz" or "30 Hz", depending on whether the nominal video frame rates based on 25 Hz or 30 000/1 001 Hz (approximately 29,97 Hz) are supported. It is expected that 25 Hz IRDs will be used in those countries where the existing analogue TV transmissions use 25 Hz frame rate and 30 Hz IRDs will be used in countries where the analogue TV transmissions use 30 000/1 001 Hz frame rate. There are also likely to be "dual-standard" IRDs which have the capabilities of both 25 Hz and 30 Hz IRDs.
- "SDTV" or "HDTV", depending on whether or not they are limited to decoding pictures of conventional TV resolution. The capabilities of an SDTV IRD are a sub-set of those of an HDTV IRD.
- "with digital interface" or "Baseline", depending on whether or not they are intended for use with a digital bitstream storage device such as a digital VCR. The capabilities of a Baseline IRD are a sub-set of those of an IRD with digital interface.

To give a complete definition of an IRD, all three dimensions need to be specified, e.g. 25 Hz SDTV Baseline IRD.

It should be noted that in DVB systems the source picture format, encoded picture format and display picture format do not need to be identical. For example, HDTV source material may be broadcast as an SDTV bitstream after down-conversion to SDTV resolution and encoding within the constraints of Main Profile at Main Level. The IRD receiving the bitstream may then up-convert the decoded picture for display at HDTV resolution.

Another notable feature of the DVB system is that a single Transport Stream may contain programme material intended for more than one type of IRD. A typical example of this is likely to be the simulcasting of SDTV and HDTV video material. In this case an SDTV IRD will decode and display SDTV pictures whilst an HDTV IRD will decode and display HDTV pictures from the same Transport Stream.

Where a feature described in the present document is mandatory, the word "shall" is used and the text is in *italic*; all other features are optional. The functionality is specified in the form of constraints on MPEG-2 systems, video and audio which the IRDs are required to decode correctly.

The specification of these baseline features in no way prohibits IRD manufacturers from including additional features, and should not be interpreted as stipulating any form of upper limit to the performance. The guidelines do not cover features, such as the IRD's up-sampling filter, which affect the quality of the displayed picture rather than whether the IRD is able to decode pictures at all. Such issues are left to the marketplace.

The guidelines presented for IRDs observe the following principles:

- wherever practical, IRDs should be designed to allow for future compatible extensions to the bit-stream syntax;
- all "reserved" and "private" bits in MPEG-2 systems, video and audio should be ignored by IRDs not designed to make use of them.

The rules of operation for the encoders are features and constraints which the encoding system should adhere to in order to ensure that the transmissions can be correctly decoded. These constraints may be mandatory or optional. Where a feature or constraint is mandatory, the word "shall" is used and the text is *italic*; all other features are optional.

Clauses 4 to 6 and the annexes, provide the guidelines for the Digital Video Broadcasting (DVB) systems layer, video, and audio respectively. For information, some of the key features are summarized below, but clauses 4 to 6 and the annexes should be consulted for all definitions:

Systems:

- MPEG-2 Transport Stream (TS) is used;
- Service Information (SI) is based on MPEG-2 program-specific information;
- Scrambling is as defined in ETR 289 [6];
- Conditional access uses the MPEG-2 Conditional Access CA_descriptor;
- Partial Transport Streams are used for digital VCR applications.

Video:

- MPEG-2 Main Profile at Main Level is used for SDTV;
- MPEG-2 Main Profile at High Level is used for HDTV;
- The 25 Hz SDTV IRD supports 25 Hz frame rate;
- The 25 Hz HDTV IRD supports frame rates of 25 Hz or 50 Hz;
- The 30 Hz SDTV IRD supports frame rates of 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz and 30 Hz;
- The 30 Hz HDTV IRD supports frame rates of 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz, 30 Hz, 60 000/1 001 Hz and 60 Hz;
- SDTV pictures may have either 4:3, 16:9 or 2.21:1 aspect ratio; IRDs support 4:3 and 16:9 and optionally 2.21:1 aspect ratio;
- HDTV pictures have 16:9 or 2.21:1 aspect ratio; IRDs support 16:9 and optionally 2.21:1 aspect ratio;
- IRDs support the use of pan vectors to allow a 4:3 monitor to give a full-screen display of a 16:9 coded picture of SDTV resolution;
- IRDs may also optionally support the use of the Active Format Description (refer to annex B) as part of the logic to control the processing and positioning of the reconstructed image for display.

Audio:

- Sampling rates of 32 kHz, 44,1 kHz and 48 kHz are supported by IRDs;
- The encoded bit-stream does not use emphasis;
- MPEG-1 or MPEG-2 stereo Layer I and Layer II are supported by all IRDs;
- IRDs support single channel, dual channel, joint stereo, stereo and the extraction of at least a stereo pair from MPEG-2 backwards compatible multi-channel audio;
- IRDs may also optionally support full multi-channel decoding of MPEG-2 Layer II backwards compatible multi-channel audio;
- Audio content shall be encoded using MPEG-1 Layer I, MPEG-1 Layer II or MPEG-2 Layer II backwards compatible audio, except in systems where IRDs are required to comply with annex C;
- The use of Layer II encoding is recommended for MPEG-1 audio bit-streams;
- IRDs may also optionally support Dolby AC-3 audio decoding (refer to annex C) or DTS audio decoding (refer to annex F);

- Where annex C is specified, the audio content may be encoded in one or more of the following modes; MPEG-1 layer I, MPEG-1 Layer II, MPEG-2 Layer II backwards compatible audio or AC-3. For MPEG audio systems refer to clause 6. For AC-3 audio systems refer to annex C;
- IRDs may also optionally support the decoding of MPEG audio streams which include ancillary data (see annex D);
- IRDs may also optionally support receiver-mixed audio (see annex G).

1 Scope

The present document provides implementation guidelines for the use of MPEG-2 audio-visual coding in satellite, cable and terrestrial broadcasting distribution systems. Both Standard Definition Television (SDTV) and High Definition Television (HDTV) are covered. Guidelines for devices equipped with a digital interface intended for digital VCR applications are also given in the present document. It does not cover applications such as contribution services which are likely to be the subject of subsequent "Guidelines" documents.

The rules of operation for the encoders are features and constraints which the encoding system should adhere to in order to ensure that the transmissions can be correctly decoded. These constraints may be mandatory, recommended or optional.

2 References

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

- References are either specific (identified by date of publication and/or edition number or version number) or non-specific.
- For a specific reference, subsequent revisions do not apply.
- For a non-specific reference, the latest version applies.

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

- [1] ISO/IEC 13818-1 (2nd edition, 2000): "Information technology - Generic coding of moving pictures and associated audio information - Part 1: Systems".
- [2] ISO/IEC 13818-2 (2nd edition, 2000): "Information technology - Generic coding of moving pictures and associated audio information - Part 2: Video".
- [3] ISO/IEC 13818-3 (2nd edition, 1998): "Information technology - Generic coding of moving pictures and associated audio information - Part 3: Audio".
- [4] ISO/IEC 13818-7 (1997): "Information technology - Generic coding of moving pictures and associated audio information - Part 7: Advanced Audio Coding (AAC)".
- [5] ISO/IEC 13818-9 (1996): "Information technology - Generic coding of moving pictures and associated audio information - Part 9: Extension for real time interface for systems decoders".
- [6] ETSI ETR 289: "Digital Video Broadcasting (DVB); Support for use of scrambling and Conditional Access (CA) within digital broadcasting systems".
- [7] ETSI EN 300 468: "Digital Video Broadcasting (DVB); Specification for Service Information (SI) in DVB systems".
- [8] ETSI ETR 211: "Digital Video Broadcasting (DVB); Guidelines on implementation and usage of Service Information (SI)".
- [9] ISO/IEC 11172-1 (1993): "Information technology - Coding of moving pictures and associated audio for digital storage media at up to about 1,5 Mbit/s - Part 1: Systems".
- [10] ISO/IEC 11172-3 (1993): "Information technology - Coding of moving pictures and associated audio for digital storage media at up to about 1,5 Mbit/s - Part 3: Audio".
- [11] ITU-T Recommendation J.17 (1988): "Pre-emphasis used on sound-programme circuits".
- [12] EBU Technical Recommendation R.68: "Alignment level in digital audio production equipment and in digital audio recorders".

- [13] ITU-R Recommendation BS.1196-1 (2001): "Audio coding for digital terrestrial television broadcasting".
- [14] ITU-R Recommendation BT.709: "Parameter values for the HDTV standards for production and international programme exchange".
- [15] ETSI EN 300 294: "Television systems; 625-line television Wide Screen Signalling (WSS)".
- [16] ETSI TS 102 114: "DTS Coherent Acoustics; Core and Extensions".
- [17] ISO/IEC 14496-1: "Information technology - Coding of audio-visual objects - Part 1: Systems".
- [18] SMPTE 170M: "Television - Composite Analog Video Signal - NTSC for Studio Applications".
- [19] ETSI TS 102 154: " Digital Video Broadcasting (DVB); Implementation guidelines for the use of Video and Audio Coding in Contribution and Primary Distribution Applications based on the MPEG-2 Transport Stream".
- [20] ISO 639: "Codes for the representation of names of languages".

3 Definitions and abbreviations

3.1 Definitions

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

25 Hz SDTV IRD: IRD which is capable of decoding and displaying pictures based on a nominal video frame rate of 25 Hz from MPEG-2 Main Profile, Main Level bitstreams as specified in TS 101 154

25 Hz SDTV Bitstream: bitstream which contains only Main Profile, Main Level video at 25 Hz frame rate as specified in TS 101 154

25 Hz HDTV IRD: IRD that is capable of decoding and displaying pictures based on a nominal video frame rate of 25 Hz or 50 Hz from MPEG-2 Main Profile, High Level bitstreams as specified in TS 101 154, in addition to providing the functionality of a 25 Hz SDTV IRD

25 Hz HDTV Bitstream: bitstream which contains only Main Profile, High Level (or simpler) video at 25 Hz or 50 Hz frame rates as specified in TS 101 154

30 Hz SDTV IRD: IRD which is capable of decoding and displaying pictures based on a nominal video frame rate of 24 000/1 001 Hz (approximately 23,98 Hz), 24 Hz, 30 000/1 001 Hz (approximately 29,97 Hz) or 30 Hz from MPEG-2 Main Profile at Main Level bitstreams as specified in TS 101 154

30 Hz SDTV Bitstream: bitstream which contains only Main Profile, Main Level video at 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz or 30 Hz frame rate as specified in TS 101 154

30 Hz HDTV IRD: IRD that is capable of decoding and displaying pictures based on nominal video frame rates of 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz, 30 Hz, 60/1 001 Hz or 60 Hz from MPEG-2 Main Profile, High Level bitstreams as specified in TS 101 154, in addition to providing the functionality of a 30 Hz SDTV IRD

30 Hz HDTV Bitstream: bitstream which contains only Main Profile, High Level (or simpler) video at 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz, 30 Hz, 60/1 001 Hz or 60 Hz frame rates as specified in TS 101 154

Baseline IRD: IRD which provides the minimum functionality to decode transmitted bitstreams as recommended in TS 101 154

NOTE: It is not required to have the ability to decode Partial Transport Streams as may be received from a digital interface connected to digital bitstream storage device such as a digital VCR.

IRD with Digital Interface: IRD which has the ability to decode Partial Transport Streams received from a digital interface connected to digital bitstream storage device such as a digital VCR as specified in TS 101 154, in addition to providing the functionality of a Baseline IRD

Pan Vector: horizontal offset in video frame centre position specified by non zero value in the frame_centre_horizontal_offset field in the MPEG video stream

Partial Transport Stream: bitstream derived from an MPEG-2 Transport Stream by removing those Transport Stream Packets that are not relevant to one particular selected programme, or a number of selected programmes

3.2 Abbreviations

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

AAC	Advanced Audio Coding according to ISO/IEC 13818-7 [4]
AC-3	dolby AC-3

NOTE: Audio coding system according to TS 102 114 [16].

AD	Audio Description
AFD	Active Format Description
AU	Access Unit
CA	Conditional Access
CRC	Cyclic Redundancy Check
DAB	Digital Audio Broadcasting
DAR	Display Aspect Ratio
DTH	Direct-To-Home
DTS	Digital Theatre Systems

NOTE: Audio coding system according to ITU-R Recommendation BS.1196-1 [13].

DVB	Digital Video Broadcasting
DVD	Digital Versatile Disc
ES	Elementary Stream
ESCR	Elementary Stream Clock Reference
HDTV	High Definition TeleVision
I-Frame	Intra-coded Frame
IRD	Integrated Receiver-Decoder
LFE	Low Frequency Effects
LSB	Least Significant Bit
MPEG	Moving Pictures Experts Group
MSB	Most Significant Bit
NIT	Network Information Table
PAT	Program Association Table
PCR	Program Clock Reference
PES	Packetized Elementary Stream
PID	Packet IDentifier
PMT	Program Map Table
PTS	Partial Transport Stream
PSI	Program Specific Information
ScF-CRC	Scale Factor - Cyclic Redundancy Check
SDTV	Standard Definition TeleVision
SI	Service Information
STD	System Target Decoder
TS	Transport Stream
TSDT	Transport Stream Description Table
T-STD	Transport stream - System Target Decoder
VCR	Video Cassette Recorder
WSS	Wide Screen Signalling

4 Systems layer

This clause describes the guidelines for encoding the systems layer of MPEG-2 in DVB broadcast bit-streams, and for decoding this layer in the IRD. The source bitstream may be transmitted via a satellite, cable or terrestrial channel, or via a digital interface. Clause 4.1 applies to the encoding of all source bitstreams and their decoding by a Baseline IRD. Clause 4.2 gives specific information relating to bitstreams transmitted via a digital interface intended for VCR applications and decoding by IRDs equipped with such an interface.

4.1 Broadcast bitstreams and baseline IRDs

The multiplexing of baseband signals and associated data conforms to ISO/IEC 13818-1 [1]. Some of the parameters and fields are not used in the DVB System and these restrictions are described below.

To allow full compliance to ISO/IEC 13818-1 [1] and upward compatibility with future enhanced versions, a DVB IRD shall be able to skip over data structures which are currently "reserved", or which correspond to functions not implemented by the IRD. As an example of this capability, a descriptor tag not yet defined within the DVB System shall be interpreted as a no-action tag, its length field correctly decoded and subsequent data skipped.

For the same reason, IRD design should be made under the assumption that any legal structure as permitted by ISO/IEC 13818-1 [1] may occur in the broadcast stream even if presently reserved or unused. Therefore the following is assumed:

- private data shall only be acted upon by decoders which are so enabled;
- filling out the bit-stream shall be carried out using the normal stuffing mechanism. Reserved fields shall not be used for this purpose. Data of reserved fields shall be set to 0xFF.

The headings in this clause are based on ISO/IEC 13818-1 [1]. The numbers in brackets after the headings are the relevant chapter and clause headings of ISO/IEC 13818-1 [1].

4.1.1 Introduction (ISO/IEC 13818-1 - Introduction)

MPEG-2 systems specify two types of multiplexed data stream: the transport stream and the program stream.

Encoding: The transmitted multiplex shall use the transport stream.

Decoding: All Baseline IRDs shall be able to demultiplex the MPEG-2 transport stream. Demultiplexing of program streams (as described in clauses Intro.2 and Intro.3 of ISO/IEC 13818-1 [1]) is optional.

4.1.2 Packetized Elementary Stream (PES) (ISO/IEC 13818-1 - clause Intro.4)

Encoding: The creation of a physical Packetized Elementary Stream (PES) by an encoder is not required. ESCR fields and ES rate fields need not be coded.

Decoding: ESCR fields and ES rate fields need not be decoded.

4.1.3 Transport stream system target decoder (ISO/IEC 13818-1 - clause 2.4.2)

Encoding: The system clock frequency shall conform to the tolerance specified in clause 2.4.2.1 of ISO/IEC 13818-1 [1]. It is recommended that the tolerance is within 5 parts per million.

Decoding: The IRD shall operate over the full tolerance range of the system clock frequency specified in clause 2.4.2.1 of ISO/IEC 13818-1 [1].

4.1.4 Transport packet layer (ISO/IEC 13818-1 - clause 2.4.3.2)

4.1.4.1 Null packets

Encoding: The encoding of null packets (those with PID value 0x1FFF) shall be as specified in ISO/IEC 13818-1 [1].

4.1.4.2 Transport packet header

4.1.4.2.1 transport_error_indicator

Encoding: It is recommended that any error detecting devices in a transmission path should set the **transport_error_indicator** bit when uncorrectable errors are detected.

Decoding: Whenever the **transport_error_indicator** flag is set in the transmitted stream it is recommended that the IRD should then invoke a suitable concealment or error recovery mechanism.

4.1.4.2.2 transport_priority

Decoding: The **transport_priority** bit has no meaning to the IRD, and may be ignored.

4.1.4.2.3 transport_scrambling_control

Encoding: The transport_scrambling_control bits shall be set according to table 1, in accordance with ETR 289 [6].

Table 1: Coding of transport_scrambling_control bits

Value	Description
00	no scrambling of TS packet payload
01	reserved for future DVB use
10	TS packet scrambled with Even key
11	TS packet scrambled with Odd key

Decoding: These bits shall be read by the IRD, and the IRD shall respond in accordance with table 1.

4.1.4.2.4 Packet Identifier (PID) values for Service Information (SI) tables

Encoding: The assignment of PID values for SI data is given in EN 300 468 [7].

4.1.5 Adaptation field (ISO/IEC 13818-1 - clause 2.4.3.4)

4.1.5.1 Random_access_indicator

Encoding: It is recommended that the **random_access_indicator** bit is set whenever a random access point occurs in video streams (i.e. video sequence header immediately followed by an I-frame).

4.1.5.2 elementary_stream_priority_indicator

Decoding: The **elementary_stream_priority_indicator** bit may be ignored by the IRD.

4.1.5.3 Program Clock Reference (PCR)

Encoding: The time interval between two consecutive PCR values of the same program shall not exceed 100 ms as specified in clause 2.7.2 of ISO/IEC 13818-1 [1]. It is recommended that this interval should be no greater than 40 ms.

Decoding: The IRD shall operate correctly with PCRs for a program arriving at intervals not exceeding 100 ms.

4.1.5.4 Other fields

This clause covers the following fields:

- original_program_clock_reference_base;
- original_program_clock_reference_extension;
- splice_countdown;
- private_data_byte;
- adaptation_field_extension (including fields within).

Encoding: These fields are optional in a DVB bit-stream. The flags that indicate the presence or absence of each of these fields shall be set appropriately.

Decoding: IRDs shall be able to accept bit-streams which contain these fields. IRDs may ignore the data within the fields.

4.1.6 Packetized Elementary Stream (PES) packet (ISO/IEC 13818-1 - clause 2.4.3.6)

4.1.6.1 stream_id and stream_type

Encoding: Elementary streams shall be identified by stream_id and stream_type in accordance with ISO/IEC 13818-1 [1], tables 2-18 and 2-29.

4.1.6.2 PES_scrambling_control

Encoding: The **PES_scrambling_control** bits shall be set according to table 2, in accordance with ETR 289 [6].

Table 2: Coding of PES_scrambling_control bits

Value	Description
00	no scrambling of PES packet payload
01	reserved for future DVB use
10	PES packet scrambled with Even key
11	PES packet scrambled with Odd key

Decoding: The **PES_scrambling_control** bits shall be read by the IRD, and the IRD shall respond in accordance with table 2.

4.1.6.3 PES_priority

Decoding: The **PES_priority** bit may be ignored by the IRD.

4.1.6.4 copyright and original_or_copy

Encoding: The copyright and **original_or_copy** bits may be set as appropriate.

Decoding: The IRD need not interpret these bits. The setting of these bits shall not be altered in any digital output from the IRD.

4.1.6.5 Trick mode fields

This clause covers the following fields:

- `trick_mode_control`;
- `field_id`;
- `intra_slice_refresh`;
- `frequency_truncation`;
- `field_rep_cntrl`.

Encoding: *These trick mode fields shall not be transmitted in a broadcast bit-stream.* Bit-streams for other applications (e.g. for non-broadcast interactive services, storage applications, etc.) may use these fields.

Decoding: The IRD may skip over any data which is flagged as being in a trick mode, if it does not support decoding of trick modes. If the IRD has a digital interface intended for digital VCR applications, it is recommended that it supports decoding of trick modes as indicated in clause 2.2.

4.1.6.6 `additional_copy_info`

Encoding: This field may be used as appropriate.

Decoding: The IRD need not interpret this field. *The coding of the field shall not be altered in any digital output from the IRD.*

4.1.6.7 Optional fields

This clause covers the following fields:

- `ESCR`;
- `ESCR_extension`;
- `ES_rate`;
- `previous_PES_packet_CRC`;
- `PES_private_data`;
- `pack_header()`;
- `program_packet_sequence_counter`;
- `MPEG1_MPEG2_identifier`;
- `original_stuff_length`;
- `P-STD_buffer_scale`;
- `P-STD_buffer_size`.

Encoding: These fields are optional in a DVB bit-stream. The flags that indicate the presence or absence of each of these fields shall be set appropriately.

Decoding: The IRD shall be able to accept bit-streams which contain these fields. The IRD may ignore the data within the fields.

4.1.6.8 PES_extension_field

The **PES_extension_field** data field is currently "reserved".

Encoding: This extension field shall not be coded unless specified in the future by MPEG.

Decoding: The IRD shall be able to accept bit-streams which contain this field. The IRD may ignore the data within the field.

4.1.7 Program Specific Information (PSI) (ISO/IEC 13818-1 - clause 2.4.4)

The data formats for the Transport Stream Description Table (TSDT) and Network Information Table (NIT) in DVB bit-streams are given in EN 300 468 [7]. The present document also defines additional tables for service information which use Program Specific Information (PSI) private_section structure defined in ISO/IEC 13818-1 [1].

It is recommended that the Program Association Table (PAT) and Program Map Table (PMT) are repeated with a maximum time interval of 100 ms between repetitions. It is recommended that the Transport Stream Description Table (TSDT) is repeated with a maximum time interval of 10 s between repetitions.

4.1.8 Program and elementary stream descriptors (ISO/IEC 13818-1 - clause 2.6)

4.1.8.1 video_stream_descriptor and audio_stream_descriptor

Encoding: The **video_stream_descriptor** shall be used to indicate video streams containing still picture data, otherwise these descriptors may be used when appropriate. If **profile_and_level_indication** is not present, then the video bit-stream shall comply with the constraints of Main Profile at Main Level. The appropriate **profile_and_level_indication field** shall always be transmitted for Profiles and Levels other than Main Profile at Main Level.

If the **audio_stream_descriptor** is not present, then the audio bit-stream shall not use sampling frequencies of 16 kHz, 22,05 kHz or 24 kHz, and all audio frames in the stream shall have the same bit rate.

Decoding: The IRD may use these descriptors when present to determine if it is able to decode the streams.

4.1.8.2 hierarchy_descriptor

Encoding: The **hierarchy_descriptor** shall be used if, and only if, audio is coded as more than one hierarchical layer.

4.1.8.3 registration_descriptor

Encoding: The **registration_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.4 data_stream_alignment_descriptor

Encoding: The **data_stream_alignment_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.5 target_background_grid_descriptor

Encoding: The **target_background_grid_descriptor** shall be used when the horizontal or vertical resolution is other than 720×576 pixels for a 25 Hz bitstream or is other than 720×480 pixels for a 30 Hz bitstream, otherwise its use is optional.

Decoding: If this descriptor is absent, a default grid of 720×576 pixels shall be assumed by a 25 Hz IRD, a default grid of 720×480 pixels shall be assumed by a 30 Hz IRD. The display of correctly windowed video on background grids other than 720×576 pixels is optional for a 25 Hz SDTV IRD, the display of correctly windowed video on background grids other than 720×480 pixels is optional for a 30 Hz SDTV IRD. The HDTV IRD shall read this descriptor, when present, to override the default values.

4.1.8.6 video_window_descriptor

Encoding: The **video_window_descriptor** may be used when appropriate, to indicate the required position of the video window on the screen.

Decoding: *The IRD shall read this descriptor, when present, and position the video window accordingly.*

4.1.8.7 Conditional Access CA_descriptor

Encoding: The **CA_descriptor** shall be encoded as defined in ETR 289 [6].

Decoding: The IRD shall interpret this descriptor as defined in ETR 289 [6].

4.1.8.8 ISO_639_Language_descriptor

Encoding: The **ISO_639_Language_descriptor** shall be present if more than one audio (or video) stream with different languages is present within a program. It is optional otherwise. The use of the **ISO_639_Language_descriptor** is recommended for all audio, video and data streams.

Decoding: The IRD shall use the data from this descriptor to assist the selection of appropriate audio (or video) stream of program, if more than one stream is available.

4.1.8.9 system_clock_descriptor

Encoding: It is recommended that the **system_clock_descriptor** is included in the program_info part of the Program Map Table for each program.

Decoding: The IRD need not make use of this descriptor.

4.1.8.10 multiplex_buffer_utilization_descriptor

Encoding: The **multiplex_buffer_utilization_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.11 copyright_descriptor

Encoding: The **copyright_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.12 maximum_bitrate_descriptor

Encoding: The **maximum_bitrate_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.13 private_data_indicator_descriptor

Encoding: The **private_data_indicator_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.14 smoothing_buffer_descriptor

Encoding: It is recommended that the **smoothing_buffer_descriptor** is included in the program_info part of the Program Map Table for each program.

Decoding: The IRD need not make use of this descriptor, but the information may be of assistance to digital VCRs.

4.1.8.15 STD_descriptor

Encoding: The **STD_descriptor** shall be used as specified in ISO/IEC 13818-1 [1].

Decoding: The IRD need not make use of this descriptor.

4.1.8.16 IBP_descriptor

Encoding: The **IBP_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.17 MPEG-4_video_descriptor

Encoding: The **MPEG-4_video_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.18 MPEG-4_audio_descriptor

Encoding: The **MPEG-4_audio_descriptor** may be used when appropriate.

Decoding: The IRD need not make use of this descriptor.

4.1.8.19 Descriptors related to ISO/IEC 14496-1

This clause covers the following descriptors:

- IOD_descriptor;
- SL_descriptor;
- FMC_descriptor;
- External_ES_ID_descriptor;
- MuxCode_descriptor;
- FmxBufferSize_descriptor;
- MultiplexBuffer_descriptor.

Encoding: These descriptors may be used when appropriate.

Decoding: The IRD need not make use of these descriptors.

Additional descriptors to those defined in ISO/IEC 13818-1 [1] are defined in EN 300 468 [7], and guidelines for their use are provided in ETR 211 [8].

4.1.9 Compatibility with ISO/IEC 11172-1 (ISO/IEC 13818-1 - clause 2.8)

Decoding: Compatibility with ISO/IEC 11172-1 [9] (MPEG-1 Systems) is optional.

4.1.10 Storage media interoperability

It is recommended that the total bitrate of the set of components, associated PMT and PCR packets for an SDTV service anticipated to be recorded by a consumer, should not exceed 9 000 000 bit/s. It is recommended that the total bitrate of the set of components, associated PMT and PCR packets for an HDTV service anticipated to be recorded by a consumer, should not exceed 28 000 000 bit/s.

It is recommended that the parameters `sb_size` and `sb_leak_rate` in the `smoothing_buffer_descriptor` remain constant for the duration of an event. The value of the `sb_leak_rate` should be the peak attained during the event. The `short_smoothing_buffer_descriptor` is defined in EN 300 468 [7] and guidelines for its use are provided in ETR 211 [8].

4.2 Bitstreams from storage applications and IRDs with digital interfaces

This Clause covers both the treatment of Partial Transport Streams which result from external program selection and Trick Play information received from a storage device. MPEG-2 PSI and DVB SI tables for use specifically in storage applications are defined in EN 300 468 [7].

4.2.1 Partial transport streams

Partial transport streams for transfer on a digital interface, e.g. for digital VCR applications, have been defined in IEC CD - 100C/1883. A Partial Transport Stream may be created by selection of Transport Stream Packets from one or more program(s), including PSI Packets.

Encoding: The partial transport stream shall be fully MPEG compliant with reference to MPEG-2 "Extension for Real-Time-Interface for systems decoders" (ISO/IEC 13818-9 [5]).

Decoding: Devices equipped with a digital interface intended for digital VCR applications shall accept the bursty character of a Partial Transport Stream with gaps of variable length between the Transport Stream Packets.

4.2.2 Decoding of trick play data (ISO/IEC 13818-1 - clause 2.4.3.7)

Encoding: Trick mode operation shall be signalled by use of the `DSM_trick_mode` flag in the header of the video Packetized Elementary Stream (PES) packets. During trick mode playback the storage device shall construct a bitstream which is syntactically and semantically correct, except as outlined in the note below.

Decoding: It is recommended that devices decode the `DSM_trick_mode_flag` and the eight bit trick mode field. Devices which decode the trick mode data shall follow the normative requirements detailed in ISO/IEC 13818-1 [1], for all values of the `trick_mode_control` field.

NOTE: Trick mode semantic constraints.

The bitstream delivered to the decoder during trick mode shall comply with the syntax defined in the MPEG-2 standard. However, for the following video syntax elements, semantic exceptions apply in the presence of the `DSM_trick_mode` field:

- `bit_rate`;
- `vbv_delay`;
- `repeat_first_field`;
- `v_axis_positive`;
- `field_sequence`;
- `subcarrier`;
- `burst_amplitude`;

- subcarrier_phase.

A decoder cannot rely on the values encoded in these fields when in trick mode.

Similarly, for the systems layer, the following semantic exceptions apply in the presence of the DSM_trick_mode field:

- maximum spacing of PSI information may exceed 400 ms;
- maximum spacing of PTS or DTS occurrences may exceed 700 ms;
- PES packets may be void of video data to indicate a change in trick mode byte;
- a PES packet void of video data may contain a PTS to indicate effective presentation time of new trick mode control;
- when trick_mode status is true, the elementary stream buffers in the T-STD may underflow.

5 Video

This clause describes the guidelines for encoding MPEG-2 video in DVB broadcast bit-streams, and for decoding this bit-stream in the IRD.

- Clause 5.1 applies to 25 Hz SDTV IRDs and broadcasts intended for reception by such IRDs.
- Clause 5.2 applies to 25 Hz HDTV IRDs and broadcasts intended for reception by such IRDs.
- Clause 5.3 applies to 30 Hz SDTV IRDs and broadcasts intended for reception by such IRDs.
- Clause 5.4 applies to 30 Hz HDTV IRDs and broadcasts intended for reception by such IRDs.

The video encoding shall conform to ISO/IEC 13818-2 [2]. Some of the parameters and fields are not used in the DVB System and these restrictions are described below. The IRD design should be made under the assumption that any legal structure as permitted by ISO/IEC 13818-2 [2] may occur in the broadcast stream even if presently reserved or unused.

To allow full compliance to the MPEG-2 standard and upward compatibility with future enhanced versions, a DVB IRD shall be able to skip over data structures which are currently "reserved", or which correspond to functions not implemented by the IRD.

This clause is based on ISO/IEC 13818-2 [2].

5.1 25 Hz SDTV IRDs and bitstreams

5.1.1 Profile and level

Encoding: Encoded bit-streams shall comply with the Main Profile Main Level restrictions, as described ISO/IEC 13818-2 [2], clause 8.2. The **profile_and_level_indication** is "01001000" or, if appropriate, "Onnnnnnn", where "Onnnnnnn">"01001000", indicating a "simpler" profile or level than Main Profile, Main Level.

Decoding: The 25 Hz SDTV IRD shall support the decoding of Main Profile Main Level bitstreams. Support for profiles and levels beyond Main Profile, Main Level is optional. If the IRD encounters an extension which it cannot decode, such as one whose identification code is Reserved, Picture Sequence Scaleable, Picture Spatial Scaleable or Picture Temporal Scaleable, it shall discard the following data until the next start code (to allow backward compatible extensions to be added in the future).

5.1.2 Frame rate

Encoding: The frame rate shall be 25 Hz, i.e. **frame_rate_code** is "0011".

Still pictures may be encoded by use of a video sequence consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

Decoding: All 25 Hz SDTV IRDs shall support the decoding and display of video material with a frame rate of 25 Hz interlaced (i.e. **frame_rate_code** of "0011"). Support of other frame and field rates is optional.

25 Hz SDTV IRDs shall be capable of decoding and displaying still pictures, i.e. video sequences consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

5.1.3 Aspect ratio

Encoding: The source aspect ratio in 25 Hz SDTV bit-streams shall be either 4:3, 16:9 or 2.21:1. Note that decoding of 2.21:1 aspect ratio is optional for the 25 Hz SDTV IRD.

*The **aspect_ratio_information** in the sequence header shall have one of the following three values:*

- 4:3 aspect ratio source: "0010";
- 16:9 aspect ratio source: "0011";
- 2.21:1 aspect ratio source: "0100".

It is recommended that pan vectors for a 4:3 window are included in the transmitted bit-stream when the source aspect ratio is 16:9 or 2.21:1. *The vertical component of the transmitted pan vector shall be zero.*

If pan vectors are transmitted then the **sequence_display_extension** shall be present in the bit-stream and the **aspect_ratio_information** shall be set to "0010" (4:3 display). The **display_vertical_size** shall be equal to the **vertical_size**. The **display_horizontal_size** shall contain the resolution of the target 4:3 display. The value of the **display_horizontal_size** field may be calculated by the following equation:

$$\text{display_horizontal_size} = \frac{4}{3} \times \frac{\text{horizontal_size}}{\text{source aspect ratio}}$$

Table 3 gives some typical examples.

Table 3: Values for display_horizontal_size

horizontal_size × vertical_size	Source aspect ratio	display_horizontal_size
720 × 576	16:9	540
544 × 576	16:9	408
480 × 576	16:9	360
352 × 576	16:9	264
352 × 288	16:9	264

Decoding: The 25 Hz SDTV IRD shall be able to decode bit-streams with values of **aspect_ratio_information** of "0010" and "0011", corresponding to 4:3 and 16:9 aspect ratio respectively. If the IRD has a digital interface, this should be capable of outputting bit-streams with aspect ratios which are not directly supported by the IRD to allow their decoding and display via an external unit.

All 25 Hz SDTV IRDs shall support the use of pan vectors and up sampling to allow a 4:3 monitor to give a full-screen display of a selected portion of a 16:9 coded picture with the correct aspect ratio. IRDs implementing the 2.21:1 aspect ratio should support the use of pan vectors and up sampling to allow a 4:3 monitor to give a full screen display of a selected portion of the 2.21:1 picture with the correct aspect ratio. Support for pan vectors with non-zero vertical components is optional. When no pan vectors are present in the transmitted bit-stream, the central portion of the wide-screen picture shall be displayed. The support of vertical resampling to obtain the correct aspect ratio for a letterbox display of a 16:9 or 2.21:1 coded picture on a 4:3 monitor is optional.

5.1.4 Luminance resolution

Encoding: *The encoded picture shall have a full-screen luminance resolution (horizontal \times vertical) of one of the following values:*

- 720 \times 576;
- 544 \times 576;
- 480 \times 576;
- 352 \times 576;
- 352 \times 288.

In addition, non full-screen pictures may be encoded for display at less than full-size (when using one of the standard up-conversion ratios at the IRD).

Decoding: The 25 Hz SDTV IRD shall be capable of decoding pictures with luminance resolutions as shown in table 4 and applying up sampling to allow the decoded pictures to be displayed at full-screen size. In addition, IRDs shall be capable of decoding lower picture resolutions and displaying them at less than full-size after using one of the standard up-conversions, e.g. a horizontal resolution of 704 pixels within the 720 pixels full-screen display.

Table 4: Resolutions for full-screen display from IRD

Coded picture		Displayed picture horizontal up sampling	
Luminance resolution (horizontal × vertical)	Aspect ratio	4:3 monitors	16:9 monitors
720 × 576	4:3 16:9 2.21:1	× 1 × 4/3 (see note 2) × 5/3 (see note 3)	× 3/4 (see note 1) × 1 × 5/4 (see note 4)
544 × 576	4:3 16:9 2.21:1	× 4/3 × 16/9 (see note 2) × 20/9 (see note 3)	× 1 (see note 1) × 4/3 × 5/3 (see note 4)
480 × 576	4:3 16:9 2.21:1	× 3/2 × 2 (see note 2) × 5/2 (see note 3)	× 9/8 (see note 1) × 3/2 × 15/8 (see note 4)
352 × 576	4:3 16:9 2.21:1	× 2 × 8/3 (see note 2) × 10/3 (see note 3)	× 3/2 (see note 1) × 2 × 5/2 (see note 4)
352 × 288	4:3 16:9 2.21:1	× 2 × 8/3 (see note 2) × 10/3 (see note 3) (and vertical up sampling × 2)	× 3/2 (see note 1) × 2 × 5/2 (see note 4) (and vertical up sampling × 2)

NOTE 1: Up sampling of 4:3 pictures for display on a 16:9 monitor is optional in the IRD, as 16:9 monitors can be switched to operate in 4:3 mode.

NOTE 2: The up sampling with this value is applied to the pixels of the 16:9 picture to be displayed on a 4:3 monitor.

NOTE 3: The up sampling with this value is applied to the pixels of the 2.21:1 picture to be displayed on a 4:3 monitor. Up sampling from 2.21:1 pictures for display on a 4:3 monitor is optional in the IRD.

NOTE 4: The up sampling with this value is applied to the pixels of the 2.21:1 picture to be displayed on a 16:9 monitor. Up sampling from 2.21:1 pictures for display on a 16:9 monitor is optional in the IRD.

5.1.5 Chromaticity parameters

Encoding: It is recommended that the chromaticity co-ordinates of the ideal display, opto-electronic transfer characteristic of the ideal display and matrix coefficients used in deriving luminance and chrominance signals from the red, green and blue primaries be explicitly signalled in the encoded bit stream by setting the appropriate values for each of the following 3 parameters in the **sequence_display_extension()**:

- **colour_primaries, transfer_characteristics; and**
- **matrix_coefficients.**

Within 25 Hz SDTV bit streams, if the **sequence_display_extension()** is not present in the bit stream or **colour_description** is zero, the chromaticity shall be implicitly defined to be that corresponding to **colour_primaries** having the value 5, the transfer characteristics shall be implicitly defined to be those corresponding to **transfer_characteristics** having the value 5 and the matrix coefficients shall be implicitly defined to be those corresponding **matrix_coefficients** having the value 5. This set of parameter values corresponds signals compliance with ITU-R Recommendation BT.470-3 System B,G,I (see bibliography).

5.1.6 Chrominance

Encoding: The operation used to down sample the chrominance information from 4:2:2 to 4:2:0 shall be indicated by the parameter **chroma_420_type** in the picture coding extension. A value of zero indicates that the fields have been down sampled independently. A value of one indicates that the two fields have been combined into a single frame before down sampling. It is desirable that the fields are down sampled independently (i.e. **chroma_420_type** = 0) to allow the IRD to use less memory for picture reconstruction.

Decoding: It is desirable that the operation used to up sample the chrominance information from 4:2:0 to 4:2:2 should be dependent on the parameter **chroma_420_type** in the picture coding extension.

5.1.7 Video sequence header

Encoding: It is recommended that a video sequence header, immediately followed by an I-frame, be encoded at least once every 500 ms. If quantizer matrices other than the default are used, the appropriate **intra_quantizer_matrix** and/or **non_intra_quantizer_matrix** are recommended to be included in every sequence header.

NOTE 1: Increasing the frequency of video sequence headers and I-frames will reduce channel hopping time but will reduce the efficiency of the video compression.

NOTE 2: Having a regular interval between I-frames may improve trick mode performance, but may reduce the efficiency of the video compression.

5.2 25 Hz HDTV IRDs and bit streams

5.2.1 Profile and level

Encoding: Encoded 25 Hz HDTV bit-streams shall comply with the Main Profile High Level restrictions, as described ISO/IEC 13818-2 [2], clause 8.2. The **profile_and_level_indication** is "01000100" or, if appropriate, "0nnnnnnn", where "0nnnnnnn">"01000100", indicating a "simpler" profile or level than Main Profile, High Level.

Decoding: *The 25 Hz HDTV IRD shall support the decoding of Main Profile High Level bit streams. This requirement includes support for "simpler" profiles and levels, including Main Profile at Main Level, as defined in table 8-15 of ISO/IEC 13818-2 [2]. Support for profiles and levels beyond Main Profile, High Level is optional. If the IRD encounters an extension which it cannot decode, such as one whose identification code is Reserved, Picture Sequence Scaleable, Picture Spatial Scaleable or Picture Temporal Scaleable, it shall discard the following data until the next start code (to allow backward compatible extensions to be added in the future).*

5.2.2 Frame rate

Encoding: The frame rate shall be 25 Hz or 50 Hz, i.e. **frame_rate_code** is "0011" or "0110".

The source video format for 50 Hz frame rate material shall be progressive. The source video format for 25 Hz frame rate material may be interlaced or progressive.

Still pictures may be encoded by use of a video sequence consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

Decoding: All 25 Hz HDTV IRDs shall support the decoding and display of video material with a frame rate of 25 Hz progressive, 25 Hz interlaced or 50 Hz progressive (i.e. **frame_rate_code** of "0011" or "0110") within the constraints of Main Profile at High Level. Support of other frame and field rates is optional.

25 Hz HDTV IRDs shall be capable of decoding and displaying still pictures, i.e. video sequences consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

5.2.3 Aspect ratio

Encoding: The source aspect ratio in 25 Hz HDTV bit-streams shall be 16:9 or 2.21:1. Note that decoding of 2.21:1 aspect ratio is optional for the 25 Hz HDTV IRD.

*The **aspect_ratio_information** in the sequence header shall have the value "0011" or "0100".*

Decoding: The 25 Hz HDTV IRD shall be able to decode bit-streams with **aspect_ratio_information** of value "0011", corresponding to 16:9 aspect ratio. The support of the aspect ratio 2.21:1 is optional. If the IRD has a digital interface, this should be capable of outputting bit-streams with aspect ratios which are not directly supported by the IRD to allow their decoding and display via an external unit.

5.2.4 Luminance resolution

Encoding: The encoded picture shall have a full-screen luminance resolution within the constraints set by Main Profile at High Level, i.e. it shall not have more than:

- 1 152 lines per frame;
- 1 920 luminance samples per line;
- 62 668 800 luminance samples per second.

It is recommended that the source video for 25 Hz HDTV Bit streams has a luminance resolution of:

- 1 080 lines per frame;
- 1 920 luminance samples per line;
- with an associated frame rate of 25 Hz, with two interlaced fields per frame.

The source video may or may not be down-sampled prior to encoding.

The use of other encoded video resolutions within the constraints of Main Profile at High Level is also permitted. Annex A provides examples of supported full screen luminance resolutions. In addition, non full-screen pictures may be encoded for display at less than full-size.

NOTE 1: The limit of 62 668 800 luminance samples per second of Main Profile at High Level excludes the use of the maximum allowed picture resolution at 50 Hz frame rate.

NOTE 2: If the recommended source video format is encoded without down-sampling it gives 51 840 000 luminance samples per second and therefore falls within the allowed range for Main Profile at High Level.

Decoding: The 25 Hz HDTV IRD shall be capable of decoding and displaying pictures with luminance resolutions within the constraints set by Main Profile at High Level.

5.2.5 Chromaticity parameters

Encoding: The chromaticity co-ordinates of the ideal display, opto-electronic transfer characteristic of the source picture and matrix coefficients used in deriving luminance and chrominance signals from the red, green and blue primaries shall be explicitly signalled in the encoded HDTV bit stream by setting the appropriate values for each of the following 3 parameters in the **sequence_display_extension()**:

- **colour_primaries, transfer_characteristics; and**
- **matrix_coefficients.**

It is recommended that ITU-R Recommendation BT.709 [14] colorimTRy is used in the 25 Hz HDTV bit stream, which is signalled by setting **colour_primaries** to the value 1, **transfer_characteristics** to the value 1 and **matrix_coefficients** to the value 1.

Decoding: The 25 Hz HDTV IRD shall be capable of decoding bit streams with any allowed values of **colour_primaries, transfer_characteristics** and **matrix_coefficients**. It is recommended that appropriate processing be included for the accurate representation of pictures using ITU-R Recommendation BT.709 [14] colorimTRy.

5.2.6 Chrominance

Encoding: The operation used to down sample the chrominance information from 4:2:2 to 4:2:0 shall be indicated by the parameter **chroma_420_type** in the picture coding extension. A value of zero indicates that the fields have been down sampled independently. A value of one indicates that the two fields have been combined into a single frame before down sampling. It is desirable that the fields are down sampled independently (i.e. **chroma_420_type** = 0) to allow the IRD to use less memory for picture reconstruction.

Decoding: It is desirable that the operation used to up sample the chrominance information from 4:2:0 to 4:2:2 should be dependent on the parameter **chroma_420_type** in the picture coding extension.

5.2.7 Video sequence header

Encoding: It is recommended that a video sequence header, immediately followed by an I-frame, be encoded at least once every 500 ms. If quantizer matrices other than the default are used, the appropriate **intra_quantizer_matrix** and/or **non_intra_quantizer_matrix** are recommended to be included in every sequence header.

NOTE 1: Increasing the frequency of video sequence headers and I-frames will reduce channel hopping time but will reduce the efficiency of the video compression.

NOTE 2: Having a regular interval between I-frames may improve trick mode performance, but may reduce the efficiency of the video compression.

5.2.8 Backwards compatibility

Decoding: In addition to the above, a 25 Hz HDTV IRD shall be capable of decoding any bit stream that a 25 Hz SDTV IRD is required to decode, as described in clause 5.1.

5.3 30 Hz SDTV IRDs and bit streams

5.3.1 Profile and level

Encoding: Encoded bit-streams shall comply with the Main Profile Main Level restrictions, as described ISO/IEC 13818-2 [2], clause 8.2. The **profile_and_level_indication** is "01001000" or, if appropriate, "Onnnnnnn", where "Onnnnnnn">"01001000", indicating a "simpler" profile or level than Main Profile, Main Level.

Decoding: The IRD shall support the syntax of Main Profile. Support for profiles and levels beyond Main Profile, Main Level is optional. If the IRD encounters an extension which it cannot decode, such as one whose identification code is Reserved, Picture Sequence Scaleable, Picture Spatial Scaleable or Picture Temporal Scaleable, it shall discard the following data until the next start code (to allow backward compatible extensions to be added in the future).

5.3.2 Frame rate

Encoding: The frame rate shall be either 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz or 30 Hz, i.e. the **frame_rate_code** field shall be encoded with one of the following values:

- "0001";
- "0010";
- "0100"; or
- "0101".

Still pictures may be encoded by use of a video sequence consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

Decoding: All 30 Hz SDTV IRDs shall support the decoding and display of Main Profile @ Main Level video with a frame rate of 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz or 30 Hz. Support of other frame rates is optional.

IRDs shall be capable of decoding and displaying still pictures, i.e. video sequences consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

5.3.3 Aspect ratio

Encoding: The source aspect ratio in 30 Hz SDTV bit-streams shall be either 4:3, 16:9 or 2.21:1. Note that decoding of 2.21:1 aspect ratio is optional for the 30 Hz SDTV IRD.

*The **aspect_ratio_information** in the sequence header shall have one of the following three values:*

- 4:3 aspect ratio source: "0010";
- 16:9 aspect ratio source: "0011";
- 2.21:1 aspect ratio source: "0100".

It is recommended that pan vectors for a 4:3 window are included in the transmitted bit-stream when the source aspect ratio is 16:9 or 2.21:1. *The vertical component of the transmitted pan vector shall be zero.*

If pan vectors are transmitted then the **sequence_display_extension** shall be present in the bit-stream and the **aspect_ratio_information** shall be set to "0010" (4:3 display). The **display_vertical_size** shall be equal to the **vertical_size**. The **display_horizontal_size** shall contain the resolution of the target 4:3 display. The value of the **display_horizontal_size** field may be calculated by the following equation:

$$\text{display_horizontal_size} = \frac{4}{3} \times \frac{\text{horizontal_size}}{\text{source aspect ratio}}$$

Table 5 gives some typical examples.

Table 5: Values for display_horizontal_size

horizontal_size × vertical_size	Source aspect ratio	display_horizontal_size
720 × 480	16:9	540
640 × 480	16:9	480
544 × 480	16:9	408
480 × 480	16:9	360
352 × 480	16:9	264
352 × 240	16:9	264

Decoding: 30 Hz SDTV IRD shall be able to decode bit-streams with values of **aspect_ratio_information** of "0010" and "0011", corresponding to 4:3 and 16:9 aspect ratio respectively. If the IRD has a digital interface, this should be capable of outputting bit-streams with aspect ratios which are not directly supported by the IRD to allow their decoding and display via an external unit.

All 30 Hz SDTV IRDs shall support the use of pan vectors and up sampling to allow a 4:3 monitor to give a full-screen display of a selected portion of a 16:9 coded picture with the correct aspect ratio. IRDs implementing the 2.21:1 aspect ratio should support the use of pan vectors and up sampling to allow a 4:3 monitor to give a full screen display of a selected portion of the 2.21:1 picture with the correct aspect ratio. Support for pan vectors with non-zero vertical components is optional. When no pan vectors are present in the transmitted bit-stream, the central portion of the wide-screen picture shall be displayed. The support of vertical resampling to obtain the correct aspect ratio for a letterbox display of a 16:9 or 2.21:1 coded picture on a 4:3 monitor is optional.

5.3.4 Luminance resolution

Encoding: The encoded picture shall have a full-screen luminance resolution (horizontal \times vertical) of one of the following values:

- 720 \times 480;
- 640 \times 480;
- 544 \times 480;
- 480 \times 480;
- 352 \times 480;
- 352 \times 240.

In addition, non full-screen pictures may be encoded for display at less than full-size (when using one of the standard up-conversion ratios at the IRD).

Decoding: The 30 Hz SDTV IRD shall be capable of decoding pictures with luminance resolutions as shown in table 6 and applying up sampling to allow the decoded pictures to be displayed at full-screen size. In addition, IRDs shall be capable of decoding lower picture resolutions and displaying them at less than full-size after using one of the standard up-conversions, e.g. a horizontal resolution of 704 pixels within the 720 pixels full-screen display.

Table 6: Resolutions for full-screen display from IRD

Coded picture		Displayed picture horizontal up sampling	
Luminance resolution (horizontal \times vertical)	Aspect ratio	4:3 monitors	16:9 monitors
720 \times 480	4:3 16:9 2:21:1	$\times 1$ $\times 4/3$ (see note 2) $\times 5/3$ (see note 3)	$\times 3/4$ (see note 1) $\times 1$ $\times 5/4$ (see note 4)
640 \times 480	4:3	$\times 9/8$	$\times 27/32$ (see note 1)
544 \times 480	4:3 16:9 2:21:1	$\times 4/3$ $\times 16/9$ (see note 2) $\times 20/9$ (see note 3)	$\times 1$ (see note 1) $\times 4/3$ $\times 5/3$ (see note 4)
480 \times 480	4:3 16:9 2:21:1	$\times 3/2$ $\times 2$ (see note 2) $\times 5/2$ (see note 3)	$\times 9/8$ (see note 1) $\times 3/2$ $\times 15/8$ (see note 4)
352 \times 480	4:3 16:9 2:21:1	$\times 2$ $\times 8/3$ (see note 2) $\times 10/3$ (see note 3)	$\times 3/2$ (see note 1) $\times 2$ $\times 5/2$ (see note 4)
352 \times 240	4:3 16:9 2:21:1	$\times 2$ $\times 8/3$ (see note 2) $\times 10/3$ (see note 3) (and vertical up sampling $\times 2$)	$\times 3/2$ (see note 1) $\times 2$ $\times 5/2$ (see note 4) (and vertical up sampling $\times 2$)

NOTE 1: Up sampling of 4:3 pictures for display on a 16:9 monitor is optional in the IRD, as 16:9 monitors can be switched to operate in 4:3 mode.

NOTE 2: The up sampling with this value is applied to the pixels of the 16:9 picture to be displayed on a 4:3 monitor.

NOTE 3: The up sampling with this value is applied to the pixels of the 2.21:1 picture to be displayed on a 4:3 monitor. Up sampling from 2.21:1 pictures for display on a 4:3 monitor is optional in the IRD.

NOTE 4: The up sampling with this value is applied to the pixels of the 2.21:1 picture to be displayed on a 16:9 monitor. Up sampling from 2.21:1 pictures for display on a 16:9 monitor is optional in the IRD.

5.3.5 Chromaticity parameters

Encoding: It is recommended that the chromaticity co-ordinates of the ideal display, opto-electronic transfer characteristic of the ideal display and matrix coefficients used in deriving luminance and chrominance signals from the red, green and blue primaries be explicitly signalled in the encoded bit stream by setting the appropriate values for each of the following 3 parameters in the **sequence_display_extension()**:

- **colour_primaries, transfer_characteristics; and**
- **matrix_coefficients.**

*Within 30 Hz SDTV bit streams, if the **sequence_display_extension()** is not present in the bit stream or **colour_description** is zero, the chromaticity shall be implicitly defined to be that corresponding to **colour_primaries** having the value 6, the transfer characteristics shall be implicitly defined to be those corresponding to **transfer_characteristics** having the value 6 and the matrix coefficients shall be implicitly defined to be those corresponding **matrix_coefficients** having the value 6. This set of parameter values signals compliance with SMPTE 170M [18].*

5.3.6 Chrominance

Encoding: The operation used to down sample the chrominance information from 4:2:2 to 4:2:0 shall be indicated by the parameter **chroma_420_type** in the picture coding extension. A value of zero indicates that the fields have been down sampled independently. A value of one indicates that the two fields have been combined into a single frame before down sampling. It is desirable that the fields are down sampled independently (i.e. **chroma_420_type** = 0) to allow the IRD to use less memory for picture reconstruction.

Decoding: It is desirable that the operation used to up sample the chrominance information from 4:2:0 to 4:2:2 should be dependent on the parameter **chroma_420_type** in the picture coding extension.

5.3.7 Video sequence header

Encoding: It is recommended that a video sequence header, immediately followed by an I-frame, be encoded at least once every 500 ms. If quantizer matrices other than the default are used, the appropriate **intra_quantizer_matrix** and/or **non_intra_quantizer_matrix** are recommended to be included in every sequence header.

NOTE 1: Increasing the frequency of video sequence headers and I-frames will reduce channel hopping time but will reduce the efficiency of the video compression.

NOTE 2: Having a regular interval between I-frames may improve trick mode performance, but may reduce the efficiency of the video compression.

5.4 30 Hz HDTV IRDs and bit streams

5.4.1 Profile and level

Encoding: Encoded 30 Hz HDTV bit-streams shall comply with the Main Profile High Level restrictions, as described ISO/IEC 13818-2 [2], clause 8.2.

The **profile_and_level_indication** is "01000100" or, if appropriate, "Onnnnnnn", where "Onnnnnnn">"01000100", indicating a "simpler" profile or level than Main Profile, High Level.

Decoding: *The 30 Hz HDTV IRD shall support the decoding of Main Profile High Level bit streams. This requirement includes support for "simpler" profiles and levels, including Main Profile at Main Level, as defined in table 8-15 of ISO/IEC 13818-2 [2]. Support for profiles and levels beyond Main Profile, High Level is optional. If the IRD encounters an extension which it cannot decode, such as one whose identification code is Reserved, Picture Sequence Scaleable, Picture Spatial Scaleable or Picture Temporal Scaleable, it shall discard the following data until the next start code (to allow backward compatible extensions to be added in the future).*

5.4.2 Frame rate

Encoding: The frame rate shall be 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz, 30 Hz, 60 000/1 001 Hz or 60 Hz, i.e. **frame_rate_code** is "0001", "0010", "0100", "0101", "0111" or "1000".

The source video format for 24 000/1 001 Hz, 24 Hz, 60 000/1 001 Hz and 60 Hz frame rate material shall be progressive. The source video format for 30 000/1 001 Hz and 30 Hz frame rate material may be interlaced or progressive.

Still pictures may be encoded by use of a video sequence consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

Decoding: All 30 Hz HDTV IRDs shall support the decoding of video material with a frame rate of 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz, 30 Hz, 60 000/1 001 Hz or 60 Hz (i.e. **frame_rate_code** of "0001", "0010", "0100", "0101", "0111" or "1000") within the constraints of Main Profile at High Level. Support of other frame rates is optional.

30 Hz HDTV IRDs shall support the display of video whose source frame rate is 24 000/1 001 Hz, 24 Hz, 30 000/1 001 Hz, 30 Hz, 60 000/1001 Hz or 60 Hz progressive. 30 Hz HDTV IRDs shall support the display of video whose source frame rate is 30 000/1 001 Hz or 30 Hz interlaced.

30 Hz HDTV IRDs shall be capable of decoding and displaying still pictures, i.e. video sequences consisting of a single intra-coded picture (see definition of still pictures in ISO/IEC 13818-1 [1], clause 2.1.48).

5.4.3 Aspect ratio

Encoding: The source aspect ratio in 30 Hz HDTV bit-streams shall be 16:9 or 2.21:1. Note that decoding of 2.21:1 aspect ratio is optional for the 30 Hz HDTV IRD.

The **aspect_ratio_information** field in the sequence header shall have the value "0011" or "0100".

Decoding: The 30 Hz HDTV IRD shall be able to decode bit-streams with **aspect_ratio_information** of value "0011", corresponding to 16:9 aspect ratio. If the IRD has a digital interface, this should be capable of outputting bit-streams with aspect ratios which are not directly supported by the IRD to allow their decoding and display via an external unit.

5.4.4 Luminance resolution

Encoding: The encoded picture shall have a full-screen luminance resolution within the constraints set by Main Profile at High Level, i.e. it shall not have more than:

- 1 152 lines per frame;
- 1 920 luminance samples per line;
- 62 668 800 luminance samples per second.

It is recommended that the source video for 30 Hz HDTV Bit streams has a luminance resolution of:

- 1 080 lines per frame and 1 920 luminance samples per line, with an associated frame rate of 30 000/1 001 Hz (approximately 29,97) with two interlaced fields per frame.
- The source video may or may not be down-sampled prior to encoding.
- The use of other encoded video resolutions within the constraints of Main Profile at High Level is also permitted. Annex A provides examples of supported full screen luminance resolutions. In addition, non full-screen pictures may be encoded for display at less than full-size.
- The limit of 62 668 800 luminance samples per second of Main Profile at High Level excludes the use of the maximum allowed picture resolution at 60 Hz and 60 000/1 001 Hz frame rates.

NOTE: If the recommended source video format is encoded without down-sampling it gives 62 145 854 luminance sample per second and therefore falls within the allowed range for Main Profile at High Level.

Decoding: The 30 Hz HDTV IRD shall be capable of decoding and displaying pictures with luminance resolutions within the constraints set by Main Profile at High Level.

5.4.5 Chromaticity parameters

Encoding: The chromaticity co-ordinates of the ideal display, opto-electronic transfer characteristic of the source picture and matrix coefficients used in deriving luminance and chrominance signals from the red, green and blue primaries shall be explicitly signalled in the encoded HDTV bit stream by setting the appropriate values for each of the following 3 parameters in the **sequence_display_extension()**:

- **colour_primaries, transfer_characteristics; and**
- **matrix_coefficients.**

It is recommended that ITU-R Recommendation BT.709 [14] colorimTRy is used in the 30 Hz HDTV bit stream, which is signalled by setting **colour_primaries** to the value 1, **transfer_characteristics** to the value 1 and **matrix_coefficients** to the value 1.

Decoding: The 30 Hz HDTV IRD shall be capable of decoding bit streams with any allowed values of **colour_primaries, transfer_characteristics** and **matrix_coefficients**. It is recommended that appropriate processing be included for the accurate representation of pictures using ITU-R Recommendation BT.709 [14] colorimTRy.

5.4.6 Chrominance

Encoding: The operation used to down sample the chrominance information from 4:2:2 to 4:2:0 shall be indicated by the parameter **chroma_420_type** in the picture coding extension. A value of zero indicates that the fields have been down sampled independently. A value of one indicates that the two fields have been combined into a single frame before down sampling. It is desirable that the fields are down sampled independently (i.e. **chroma_420_type** = 0) to allow the IRD to use less memory for picture reconstruction.

Decoding: It is desirable that the operation used to up sample the chrominance information from 4:2:0 to 4:2:2 should be dependent on the parameter **chroma_420_type** in the picture coding extension.

5.4.7 Video sequence header

Encoding: It is recommended that a video sequence header, immediately followed by an I-frame, be encoded at least once every 500 ms. If quantizer matrices other than the default are used, the appropriate **intra_quantizer_matrix** and/or **non_intra_quantizer_matrix** are recommended to be included in every sequence header.

NOTE 1: Increasing the frequency of video sequence headers and I-frames will reduce channel hopping time but will reduce the efficiency of the video compression.

NOTE 2: Having a regular interval between I-frames may improve trick mode performance, but may reduce the efficiency of the video compression.

5.4.8 Backwards compatibility

Decoding: In addition to the above, a 30 Hz HDTV IRD shall be capable of decoding any bit stream that a 30 Hz SDTV IRD is required to decode, as described in clause 5.3.

6 Audio

This clause describes the guidelines for encoding MPEG backward compatible audio in DVB broadcast bit-streams, and for decoding this bit-stream in the IRD. Additional optional audio coding systems and ancillary data are described in annexes C and D.

The recommended level for reference tones for transmission is 18 dB below clipping level, in accordance with EBU Recommendation R.68 [12].

The audio encoding shall conform to either ISO/IEC 11172-3 [10] or ISO/IEC 13818-3 [3], except in systems where IRDs are required to comply with annex C. Some of the parameters and fields in ISO/IEC 11172-3 [10] and ISO/IEC 13818-3 [3] are not used in the DVB System and these restrictions are described below.

The IRD design should be made under the assumption that any legal structure as permitted by ISO/IEC 11172-3 [10] or ISO/IEC 13818-3 [3] may occur in the broadcast stream even if presently reserved or unused. *To allow full compliance to ISO/IEC 11172-3 [10] and ISO/IEC 13818-3 [3] and upward compatibility with future enhanced versions, a DVB IRD shall be able to skip over data structures which are currently "reserved", or which correspond to functions not implemented by the IRD. For example, an IRD which is not designed to make use of the ancillary data field shall skip over that portion of the bit-stream.*

This clause is based on ISO/IEC 11172-3 [10] (MPEG-1 audio) and ISO/IEC 13818-3 [3] (MPEG-2 backwards compatible audio coding).

6.1 Audio mode

Encoding: The audio shall be encoded in one of the following modes:

- ISO/IEC 11172-3 [10] single channel;
- ISO/IEC 11172-3 [10] joint stereo;
- ISO/IEC 11172-3 [10] stereo;
- ISO/IEC 13818-3 [3] multi-channel audio, backwards compatible to ISO/IEC 11172-3 [10] (dematrix procedure = 0, 1 or 2).

In addition, audio may be encoded in ISO/IEC 11172-3 [10] dual channel mode, as specified by TS 102 154 [19], in a transmission intended both as a contribution feed and for Direct-To-Home (DTH) reception. However, this is not recommended. Care needs to be taken to ensure that the optional dual channel decoding mode is supported in the DTH IRD. Furthermore, there may be problems due to the left/right channel selection being performed by different equipment from the decoding unit (e.g. decoding may be by a set-top-box but left/right channel selection and audio balance may be performed by the TV set).

Decoding: The IRD shall be capable of decoding the following audio modes:

- ISO/IEC 11172-3 [10] single channel;
- ISO/IEC 11172-3 [10] joint stereo;
- ISO/IEC 11172-3 [10] stereo.

The IRD shall be capable of decoding at least the ISO/IEC 11172-3 [10] compatible basic stereo information from an ISO/IEC 13818-3 [3] multi-channel audio bit-stream. Full decoding of an ISO/IEC 13818-3 [3] multi-channel audio bit-stream is optional.

Support for decoding of ISO/IEC 11172-3 [10] dual channel is optional.

6.2 Layer

Encoding: An ISO/IEC 11172-3 [10] encoded bit-stream shall use either Layer I or Layer II coding (**layer** = "11" or "10" respectively). Use of Layer II is recommended.

An ISO/IEC 13818-3 [3] multi-channel encoded bit-stream shall use Layer II coding (**layer** = "10").

Decoding: IRDs shall be capable of decoding Layer I and Layer II.

6.3 Bit rate

Encoding: The value of **bitrate_index** in the encoded bit-stream shall be one of the 14 values from "0001" to "1110" (inclusive).

For Layer I, these correspond to bit rates of: 32 kbits/s, 64 kbits/s, 96 kbits/s, 128 kbits/s, 160 kbits/s, 192 kbits/s, 224 kbits/s, 256 kbits/s, 288 kbits/s, 320 kbits/s, 352 kbits/s, 384 kbits/s, 416 kbits/s or 448 kbits/s.

For Layer II, these correspond to bit rates of: 32 kbits/s, 48 kbits/s, 56 kbits/s, 64 kbits/s, 80 kbits/s, 96 kbits/s, 112 kbits/s, 128 kbits/s, 160 kbits/s, 192 kbits/s, 224 kbits/s, 256 kbits/s, 320 kbits/s, 384 kbits/s.

For ISO/IEC 13818-3 [3] encoded bit-streams with total bit rates greater than 384 kbit/s, an extension bit-stream shall be used. The bit rate of that extension may be in the range of 0 kbit/s to 682 kbit/s.

Decoding: IRDs shall be capable of decoding bit-streams with a value of **bitrate_index** from "0001" to "1110" (inclusive). Support for the free format bit rate (**bitrate_index** = "0000") is optional.

6.4 Sampling frequency

Encoding: *The audio sampling rate of primary sound services shall be 32 kHz, 44,1 kHz or 48 kHz. Sampling rates of 16 kHz, 22,05 kHz, 24 kHz, 32 kHz, 44,1 kHz or 48 kHz may be used for secondary sound services.*

Decoding: *The IRD shall be capable of decoding audio with sampling rates of 32 kHz, 44,1 kHz and 48 kHz. Support for sampling rates of 16 kHz, 22,05 kHz and 24 kHz is optional.*

6.5 Emphasis

Encoding: The encoded bit-stream shall have no emphasis (**emphasis** = "00").

Decoding: The IRD shall be capable of decoding audio with no emphasis. Support for 50/15 microseconds de-emphasis and ITU-T Recommendation J.17 [11] de-emphasis (**emphasis** = "01" or "11") is optional.

6.6 Cyclic redundancy code

Encoding: The parity check word (**crc_check**) shall be included in the encoded bit-stream.

Decoding: It is recommended that the IRD use **crc_check** to detect errors and subsequently invoke suiTable concealment or muting mechanisms.

6.7 Prediction

Encoding: ISO/IEC 13818-3 [3] multichannel encoded bit-streams shall not use `mc_prediction` (`mc_prediction_on` equals "0").

Decoding: The IRD shall be capable of decoding ISO/IEC 13818-3 [3] multichannel encoded bit-streams which do not use `mc_prediction`.

6.8 Multilingual

Encoding: ISO/IEC 13818-3 [3] multichannel encoded bit-streams shall not contain multilingual channels (`no_of_multilingual_channels` equals "0").

Decoding: The IRD shall be capable of decoding ISO/IEC 13818-3 [3] multichannel encoded bit-streams which do not contain multilingual channels.

6.9 Extension stream

Encoding: When an ISO/IEC 13818-3 [3] encoded bit-stream uses an extension stream, it is recommended that a continuous stream of extension frames is maintained for the duration of a programme, even if a total bit rate of less than 384 kbits/s would be sufficient to encode individual frames. This prevents undesired resets of the audio decoder.

6.10 Ancillary data

Encoding: ISO/IEC 13818-3 [3] stereo or multichannel encoded bitstreams may contain ancillary data as described in annex D. It is recommended to include the data in the bitstream.

Decoding: The IRD may interpret the ancillary data field in an ISO/IEC 13818-3 [3] stereo or multichannel bitstream as described in annex D and it is recommended that the contribution IRD make use of this data.

Annex A (informative): Examples of full screen luminance resolutions for SDTV and HDTV

Table A.1

vertical_size_ value	horizontal_size_ value	aspect_ratio_ information	frame_rate_ code (see note)	progressive_ sequence	Decodeable by SDTV IRD
1 152	1 440	16:9	25	0	
1 080	1 920	16:9	25	1	
			23,976, 24, 29,97, 30	1	
			25	0	
			29,97, 30	0	
1 035	1 920	16:9	25	0	
			29,97, 30	0	
720	1 280	16:9	25, 50	1	
			23,976, 24, 29,97, 30, 59,94, 60	1	
			50	1	
576	720	4:3, 16:9	25	1	✓
			25	0	✓
	544	4:3, 16:9	25	1	✓
			25	0	✓
	480	4:3, 16:9	25	1	✓
			25	0	✓
	352	4:3, 16:9	25	1	✓
			25	0	✓
480	720	4:3, 16:9	59,94, 60	1	
			23,976, 24, 29,97, 30	1	✓
			29,97, 30	0	✓
	640	4:3	59,94, 60	1	
			23,976, 24, 29,97, 30	1	✓
			29,97, 30	0	✓
	544	4:3, 16:9	23,976, 29,97	1	✓
			29,97	0	✓
	480	4:3, 16:9	23,976, 29,97	1	✓
			29,97	0	✓
	352	4:3, 16:9	23,976, 29,97	1	✓
			29,97	0	✓
288	352	4:3, 16:9	25	1	✓
240	352	4:3, 16:9	23,976, 29,97	1	✓

NOTE: Shaded "frame_rate_code" values indicate 30 Hz bit streams, clear values 25 Hz bit streams.

Annex B (informative): Active Format Description (AFD)

B.1 Overview

The Active Format Description (AFD) describes the portion of the coded video frame that is "of interest". It is intended for use in networks that deliver mixed formats to a heterogeneous receiver population. The format descriptions are informative in nature and are provided to assist receiver systems to optimize their presentation of video.

Transmission of this description, and use of this description by a receiver, are both optional.

The AFD is intended for use where there are compatibility problems between the source format of a programme, the format used for the transmission of that programme, and the format of the target receiver population. For example, a wide-screen production may be transmitted as a 14:9 letter-box within a 4:3 coded frame, thus optimized for the viewer of a 4:3 TV, but causing problems to the viewer of a wide screen TV. The appropriate AFD may be transmitted with the video to indicate to the receiver the "area of interest" of the image, thereby enabling a receiver to present the image in an optimum fashion (which will depend on the format and functionality of the receiving equipment combined with the viewer's preferences). In this example, the functionality provided by the AFD is analogous to that provided by Wide Screen Signalling (WSS) described in EN 300 294 [15].

However, the AFD extends WSS by allowing the "area of interest" of a full-frame 16:9 (anamorphic) image to be described, for example to indicate that the centre 4:3 portion of the image has been protected such that a set-top box connected to a 4:3 set may perform a centre cut-out without removing any essential picture information.

The AFD itself does not describe the aspect ratio of the coded frame (as this is described elsewhere in the MPEG-2 video syntax).

B.2 Coding

The AFD is carried in the user data of the video elementary stream. After each sequence start (and repeat sequence start) the default aspect ratio of the area of interest is that signalled by the sequence header and sequence display extension parameters. After introduction, an AFD persists until the next sequence start or until another AFD is introduced.

Encoding: Support for the encoding of AFD is optional.

The AFD may be inserted wherever user data may be inserted in the video elementary stream (after the sequence extension, and/or GOP header, and/or picture coding extension, as specified in ISO/IEC 13818-2 [2]). For example, it could be inserted once per sequence after each sequence extension, once per GOP after each GOP header, or once per picture after each picture coding extension. It may be changed for each picture.

Decoding: Support for the decoding of AFD is optional.

A decoder that supports the decoding of AFD shall be capable of decoding it from wherever user data may be inserted in the video stream (i.e. after the sequence extension, and GOP header, and picture coding extension).

B.3 Syntax and semantics

The AFD is carried in the user data of the video elementary stream as defined in ISO/IEC 13818-2 [2]. The syntax is illustrated in table B.1.

Table B.1: Active Format Description

Syntax	Number of bits	Identifier
user_data_start_code	32	bslbf
afd_identifier	32	bslbf
"0"	1	bslbf
active_format_flag	1	bslbf
reserved (set to '00 0001')	6	bslbf
if (active_format_flag == 1) {		
reserved (set to '1111')	4	bslbf
active_format	4	bslbf
}		

afd_identifier: A 32 bit field that identifies that the syntax of the user data is as specified here. Its value is 0x44544731.

active_format_flag: A 1 bit flag. A value of "1" indicates that an active format is described in this data structure.

active_format: A 4 bit field describing the "area of interest" in terms of its aspect ratio within the coded frame as defined in ISO/IEC 13818-2 [2].

The active_format is used by the decoder in conjunction with the "source aspect ratio". The source aspect ratio is derived from the "Display Aspect Ratio"(DAR) signalled in the aspect_ratio_information, the horizontal_size, vertical_size, and display_horizontal_size and display_vertical_size if present (see ISO/IEC 13818-2 [2]):

- If sequence_display_extension() is not present:

$$\text{source aspect ratio} = \text{DAR}$$

- If sequence_display_extension() is present:

$$\text{source aspect ratio} = \text{DAR} \times \frac{\text{display_horizontal_size}}{\text{display_vertical_size}} \times \frac{\text{vertical_size}}{\text{horizontal_size}}$$

The combination of source aspect ratio and active_format allows the decoder to identify whether the "area of interest" is the whole of the frame (e.g. source aspect ratio 16:9, active_format 16:9 centre), a letterbox within the frame (e.g. source aspect ratio 4:3, active_format 16:9 centre), or a "pillar-box" (see note) within the frame (e.g. source aspect ratio 16:9, active_format 4:3 centre).

NOTE: "Pillar-box" describes a frame that the image fails to fill horizontally, in the same way that a "Letterbox" describes a frame that the image fails to fill vertically.

Table B.2: active_format

Active_format	Aspect ratio of the "area of interest"
0000 - 0001	reserved
0010	box 16:9 (top)
0011	box 14:9 (top)
0100	box > 16:9 (centre)
0101 - 0111	reserved
1000	Active format is the same as the coded frame
1001	4:3 (centre)
1010	16:9 (centre)
1011	14:9 (centre)
1100	reserved
1101	4:3 (with shoot and protect 14:9 centre)
1110	16:9 (with shoot and protect 14:9 centre)
1111	16:9 (with shoot and protect 4:3 centre)

The complete set of Active Formats described in the present document is illustrated in table B.3. Note that for each format two example illustrations have been given, corresponding to the source aspect ratio of the coded frame being 4:3 and 16:9. The AFD may also be used with coded frames of other aspect ratios. For example a coded frame of 2.21:1 with active_format 10 would represent a 16:9 image centred (pillar-box) within a 2.21:1 frame.

The Active Formats are illustrated using figure B.1 diagrammatic representation.

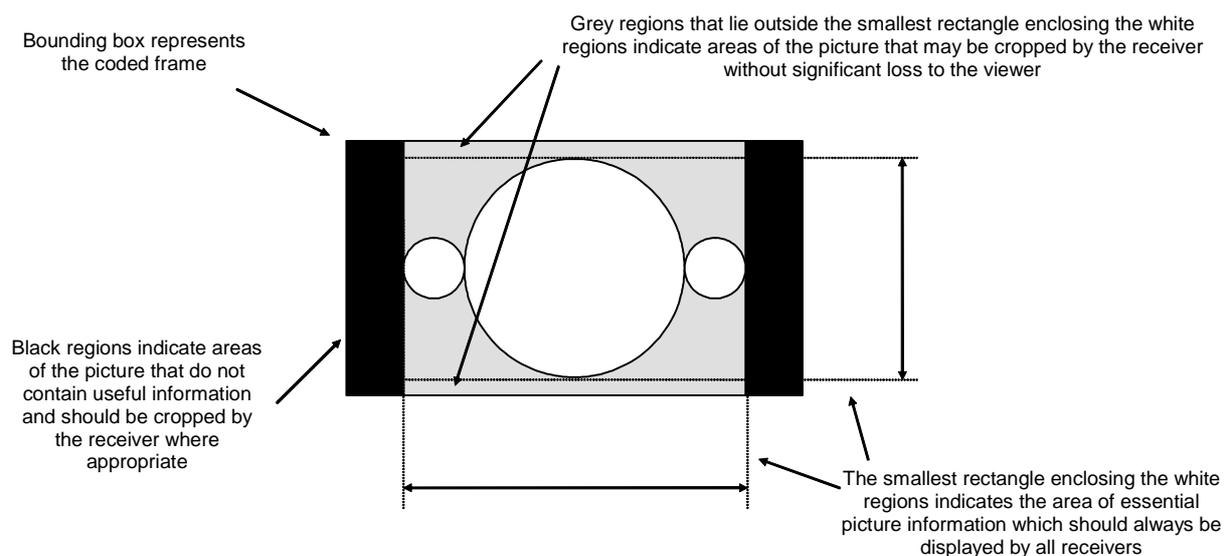
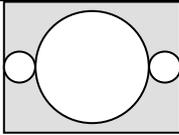
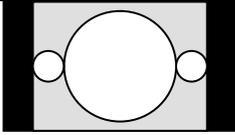
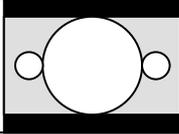
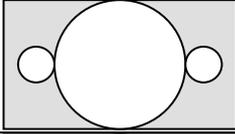
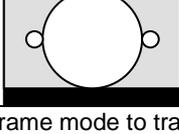
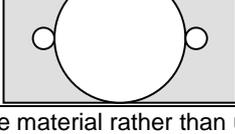


Figure B.1

Table B.3: Active formats illustrated

Active_format		Illustration of described format	
value	description	in 4:3 coded frame	in 16:9 coded frame
0000 - 0001	reserved		
0010	box 16:9 (top)		
0011	box 14:9 (top)		
0100	box > 16:9 (centre)		
0101 - 0111	reserved		
1000	As the coded frame		
1001	4:3 (centre)		
1010	16:9 (centre)		
1011	14:9 (centre)		

Active_format		Illustration of described format	
value	description	in 4:3 coded frame	in 16:9 coded frame
1100	reserved		
1101	4:3 (with shoot and protect 14:9 centre)		
1110	16:9 (with shoot and protect 14:9 centre)		
1111	16:9 (with shoot and protect 4:3 centre)		

NOTE: It is recommended to use the 4:3 coded frame mode to transmit 4:3 source material rather than using a pillar box to transmit it in a 16:9 coded frame. This allows for higher horizontal resolution on both 4:3 and 16:9 sets.

B.4 Relationship with Wide Screen Signalling (WSS)

The AFD provides a super-set of the aspect ratio signalling specified in EN 300 294 [15]. The mapping of source aspect ratio and active_format to WSS Aspect Ratio is given in table B.4.

Table B.4: Support for WSS

Sequence header	Active Format Description	WSS	
source aspect ratio	value	code (bits 0-3)	description
4:3	1001	0001	full format 4:3
	1011	1000	box 14:9 centre
	0011	0100	box 14:9 top
	1010	1101	box 16:9 centre
	0010	0010	box 16:9 top
	0100	1011	box > 16:9 centre
	1101	0111	full format 4:3 (shoot and protect 14:9 centre)
16:9	1010	1110	full format 16:9 (anamorphic)

B.5 Aspect ratio ranges

The labels 4:3, 14:9, 16:9 and > 16:9 used in the AFD shall correspond to the aspect ratio ranges specified in EN 300 294 [15]. (Note that the corresponding active lines specified in EN 300 294 [15] do not, in general, apply).

B.6 Relationship with pan vectors

Encoding: Encoded bit-streams may optionally include pan vectors and AFDs.

Decoding: The decoder may use the AFD as part of the logic that decides how the IRD processes and positions the reconstructed image for display on a monitor, where the monitor aspect ratio does not match the source aspect ratio (e.g. whether to use pan vectors, or generate a letterbox display).

Annex C (informative): Guidelines for the implementation of AC-3 Audio in DVB compliant transport streams

C.1 Scope

The inclusion of AC-3 audio streams in a DVB multiplex is optional, and IRDs may optionally decode these streams. This annex contains the guidelines to include one or more AC-3 elementary streams in a DVB Transport Stream in compliance with ISO/IEC 13818-1 [1]. The coding and decoding of an AC-3 elementary stream is based upon ITU-R Recommendation BS.1196-1 [13] - annex 2. However, annex 1 to annex 2 of ITU-R Recommendation BS.1196-1 [13] should be disregarded, as it is not applicable to the DVB system.

It is recommended that implementations of DVB systems that include AC-3 audio streams should comply with this annex.

The AC-3 packetized elementary stream shall conform to the requirements of a user private stream type 1, as described in ISO/IEC 13818-1 [1].

The IRD design should be made under the assumption that any legal structure as permitted by ISO/IEC 13818-1 [1], including private data streams, may occur in the Transport Stream, even if presently reserved or unused. To allow full compliance to the MPEG-2 standard and upward compatibility with future enhanced versions, a DVB IRD shall be able to skip over data structures which are currently "reserved", or which correspond to functions not implemented by the IRD.

This Clause is based on ISO/IEC 13818-1 [1] and ITU-R Recommendation BS.1196-1 [13] - annex 2.

C.2 Introduction

An AC-3 elementary bit stream may be multiplexed into an MPEG-2 transport stream in much the same way an MPEG-1 audio stream would be included. The AC-3 elementary stream is packetized into PES packets with a structure similar to an MPEG audio PES. An MPEG-2 transport stream containing AC-3 elementary stream(s) must meet the constraints described in the STD model in clause C.4.5.

It is necessary to unambiguously indicate that an MPEG private stream is, in fact, an AC-3 stream. A public DVB descriptor, the AC-3_descriptor has been specified for this purpose. The syntactical elements that need to be specified in order to include AC-3 within an MPEG-2 transport stream are: the MPEG stream_type, stream_id and the DVB AC - 3_descriptor.

The ISO 639 [20] language descriptor may be used to indicate the language of the content of the AC-3 stream.

IRDs shall decode all bit rates and sample rates listed in ITU-R Recommendation BS.1196-1 [13] - annex 2.

Some constraints are placed on the PES layer for the case of multiple audio streams intended to be reproduced in exact sample synchronism as described in clause C.5.

C.3 DVB compliant streams

The AC-3 PES shall be carried as an MPEG private data stream type, conforming to the structure of a private_stream_1 as described in ISO/IEC 13818-1 [1] tables 2-18 (stream_id) and 2-29 (stream_type).

When an AC-3 stream is included in a DVB transport stream, the AC-3_descriptor shall also be included. The AC-3_descriptor is defined in EN 300 468 [7], but for information a description is included here in clause C.4.4. The AC-3_descriptor is located in the PMT and the Selection Information Table of the DVB SI tables defined in EN 300 468 [7].

Certain other of the DVB Service Information descriptors defined in EN 300 468 [7] can provide additional means of identifying the existence of an AC-3 stream without accessing the PMT. The component_descriptor (see clause C.4.2) may have values assigned to its syntactical elements, which indicate both the presence and type of AC-3 stream(s) in the DVB-SI.

C.4 Detailed specification

C.4.1 MPEG transport stream compliance

C.4.1.1 Stream_id

- Semantics:** The semantics of the stream_id field are described in ISO/IEC 13818-1 [1] table 2-18. Multiple AC-3 streams may share the same value of stream_id since each stream is carried with a unique PID value. The mapping of values of PID to stream_type is indicated in the transport stream Program Map Table (PMT).
- Encoding:** The value of the stream_id field for an AC-3 elementary stream shall be 0xBD (indicating private_stream_1).
- Decoding:** This field shall be read by the IRD, and the IRD shall interpret this field in accordance with MPEG systems syntax.

C.4.1.2 Stream_type

- Semantics:** The semantics of the stream_type field are described in ISO/IEC 13818-1 [1] table 2-29.
- Encoding:** The recommended value of stream_type for an AC-3 elementary stream shall be 0x06 (indicating PES packets containing private data).
- Decoding:** This field shall be read by the IRD, and the IRD shall interpret this field in accordance with MPEG systems syntax.

C.4.2 Use of the DVB-SI component_descriptor and multilingual_component_descriptor

- Semantics:** The semantics of the component_descriptor and multilingual_component_descriptor are defined in EN 300 468 [7]. The stream_content and component_type assigned values for DVB AC-3 audio streams are listed in EN 300 468 [7].
- Encoding:** The values for the elements of the component_descriptor and multilingual_component_descriptor shall be set in accordance with EN 300 468 [7].
- Decoding:** These fields shall be read by the IRD, and the IRD shall interpret these fields to indicate the type of audio service present.

C.4.3 AC-3_descriptor

The syntax of the AC-3_descriptor is described in table C.1.

NOTE: Horizontal lines in the table indicate allowable termination points for the descriptor.

The AC-3_descriptor syntax provides information about individual AC-3 elementary streams to be identified in the PSI PMT sections. The descriptor is located in the PSI PMT, and used once in a program map section following the relevant ES_info_length field for any stream containing AC-3 audio coded in accordance with ITU-R Recommendation BS.1196-1 [13] - annex 2.

Table C.1: AC-3 descriptor syntax

Syntax	Number of bits	Identifier
AC-3_descriptor() {		
descriptor_tag	8	uimsbf
descriptor_length	8	uimsbf
AC-3_type_flag	1	bslbf
bsid_flag	1	bslbf
mainid_flag	1	bslbf
asvc_flag	1	bslbf
reserved	1	bslbf
If (AC-3_type_flag)==1{		
AC-3_type	8	uimsbf
}		
If (bsid_flag)==1{		
bsid	8	uimsbf
}		
If (mainid_flag)==1{		
mainid	8	uimsbf
}		
If (asvc_flag)==1{		
asvc	8	bslbf
}		
For (I=0;I<N;I++){		
additional_info [I]	N x 8	uimsbf
}		
}		

C.4.3.1 descriptor_tag

Encoding: The descriptor tag is an 8-bit field, which identifies each descriptor. The value assigned to the AC-3 descriptor_tag is 0x6A (see EN 300 468 [7], table E.1).

Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

C.4.3.2 descriptor_length

Semantics: This 8-bit field specifies the total number of bytes of the data portion of the descriptor following the byte defining the value of this field. The AC-3 descriptor has a minimum length of one byte but may be longer depending on the use of the optional flags and the additional_info_loop.

Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

C.4.3.3 AC-3_type_flag

Semantics: This 1-bit field is mandatory for AC-3 streams. If set to "1" the optional AC-3_type field is included in the descriptor.

Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended that IRDs decode this field.

C.4.3.4 bsid_flag

- Semantics: This 1-bit field is mandatory for AC-3 streams. If set to "1" the optional bsid field is included in the descriptor.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended that IRDs decode this field.

C.4.3.5 mainid_flag

- Semantics: This 1-bit field is mandatory for AC-3 streams. If set to "1" the optional mainid field is included in the descriptor.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended that IRDs decode this field.

C.4.3.6 asvc_flag

- Semantics: This 1-bit field is mandatory for AC-3 streams. If set to "1" the optional asvc field is included in the descriptor.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended that IRDs decode this field.

C.4.3.7 reserved flags

- Semantics: These 1-bit fields are reserved for future use. They should always be set to "0".
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. IRDs may ignore the data within this field.

C.4.3.8 AC-3_type

- Semantics: This optional 8-bit field indicates the type of audio carried in the AC-3 elementary stream.
- Encoding: This field is set to the same value as the component_type field of the component descriptor (see EN 300 468 [7], table E.3).
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. IRDs may ignore the data within this field.

C.4.3.9 bsid

- Semantics: This optional 8-bit field indicates the AC-3 coding version.
- Encoding: The three MSBs should always be set to "0". The five LSBs are set to the same value as the bsid field in the AC-3 elementary stream.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. IRDs may ignore the data within this field.

C.4.3.10 mainid

- Semantics: This 8-bit field is optional. It contains a number in the range 0 to 7 which identifies a main audio service. Each main service should be tagged with a unique number. This value is used as an identifier to link associated services with particular main services.
- Encoding: Each main service should be tagged with a unique number in the range 0 to 7.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. IRDs may ignore the data within this field.

C.4.3.11 asvc

- Semantics: This 8-bit field is optional.
- Encoding: Each bit (0 to 7) indicates to which main service(s) this associated service belongs. The left most bit, bit 7, indicates whether this associated service may be reproduced along with main service number 7. If the bit has a value of 1, the service is associated with main service number 7. If the bit has a value of 0, the service is not associated with main service number 7.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. IRDs may ignore the data within this field.

C.4.3.12 additional_info

- Semantics: These optional bytes are reserved for future use.
- Decoding: IRDs shall be able to accept bit-streams, which contain these bytes. IRDs may ignore the data within these bytes.

C.4.4 STD audio buffer size

It is recommended that for AC-3 audio in a DVB system, the main audio buffer size (BS_n) has a fixed value of 5 696 bytes. Refer to ISO/IEC 13818-1 [1] for the derivation of (BS_n) for audio elementary streams.

C.5 PES constraints

C.5.1 Encoding

In some applications, the audio decoder may be capable of simultaneously decoding two elementary streams containing different programme elements, and then combining the programme elements into a complete programme.

Most of the programme elements are found in the main audio service. Another programme element (such as a narration of the picture content intended for the visually impaired listener) may be found in the associated audio service.

In order to have the audio from the two elementary streams reproduced in exact sample synchronism, it is necessary for the original audio elementary stream encoders to have encoded the two audio programme elements frame synchronously; i.e. if audio stream 1 has sample 0 of frame n taken at time t_0 , then audio stream 2 should also have frame n beginning with its sample 0 taken the identical time t_0 . If the encoding of multiple audio services is done frame and sample synchronous, and decoding is intended to be frame and sample synchronous, then the PES packets of these audio services shall contain identical values of PTS, which refer to the audio access units intended for synchronous decoding.

Audio services intended to be combined together for reproduction shall be encoded at an identical sample rate.

C.5.2 Decoding

If audio access units from two audio services which are to be simultaneously decoded have identical values of Partial Transport Stream (PTS) indicated in their corresponding PES headers, then the corresponding audio access units shall be presented to the audio decoder for simultaneous synchronous decoding. Synchronous decoding means that for corresponding audio frames (access units), corresponding audio samples are presented at the identical time.

If the PTS values do not match (indicating that the audio encoding was not frame synchronous) then the audio frames (access units) of the main audio service may be presented to the audio decoder for decoding and presentation at the time indicated by the PTS. An associated service, which is being simultaneously decoded, may have its audio frames (access units), which are in closest time alignment (as indicated by the PTS) to those of the main service being decoded, presented to the audio decoder for simultaneous decoding. In this case the associated service may be reproduced out of sync by as much as 1/2 of a frame time. (This is typically satisfactory; a visually impaired narration does not require highly precise timing).

C.5.3 Byte-alignment

The AC-3 elementary stream shall be byte-aligned within the MPEG-2 data stream. This means that the initial 8 bits of an AC-3 frame shall reside in a single byte, which is carried by the MPEG-2 data stream.

Annex D (informative): Implementation of ancillary data for MPEG audio

D.1 Scope

This annex contains the guidelines required to include ancillary data in the MPEG Audio elementary stream.

The IRD design should be made under the assumption that any structure as permitted by this annex may occur in the broadcast stream. The IRD is not required to make use of this data but its use is recommended.

D.2 Introduction

An MPEG audio elementary stream provides for the inclusion of ancillary data. This data can be used to convey specific information about the audio content to the decoder, allowing the broadcaster to control rendering of the content to a greater extent. The data includes dynamic range control information and dialogue normalization information.

D.3 DVB compliance

The ancillary data format described in this annex does not introduce any additional elements to the DVB transport stream. It is compliant with the present document and compatible with all MPEG audio decoders.

D.4 Detailed specification

D.4.1 DVD-Video ancillary data

The transmission of "dynamic_range_control" in MPEG audio is optional. If applied, 16 bits of ancillary data [b15.b0] (situated at the end of each MPEG audio base frame) shall be used.

In case of MPEG1 streams or MPEG2 streams without an extension stream (MPEG audio format1), ancillary data described in this Annex is placed at the end of each base frame.

In case of MPEG2 streams with extension stream (MPEG audio format 2), the ancillary data described in this Annex is placed at the end of each base frame.

Table D.1: DVD-Video ancillary data syntax

Syntax	Number of bits	Mnemonic
dvd_ancillary_data() {		
dynamic_range_control	8	bslbf
dynamic_range_control_on	1	bslbf
reserved (set to '000 0000b')	7	bslbf
}		

Semantics: The 8-bit dynamic_range_control field leads to the following gain control value by considering the upper 3 bits as unsigned integer X and the binary value of the lower 5 bits as unsigned integer Y:

- linear: $G = 2^{4-(X + Y/30)}$
- $(0 \leq X \leq 7, 0 \leq Y \leq 29)$

- in dB: $G = 24,082 - 6,0206 X - 0,2007 Y$
- $(0 \leq X \leq 7, 0 \leq Y \leq 29)$

If the `dynamic_range_control_on` field is set to "0b", the `dynamic_range_control` field does not convey useful information.

- Encoding: When dynamic range control is temporarily not applied, that value of `dynamic_range_control` shall be set to "1000 0000b" or `dynamic_range_control_on` shall be set to "0b".
- Decoding: The decoder shall read this field, and the decoder shall interpret the value G as a gain value applied to all sub band samples, before the reconstruction filter. This value may be scaled in the decoder to allow user control of the amount of dynamic range compression that is applied.

D.4.2 Extended ancillary data syntax

The syntax of the extended ancillary data field is described in table D.2.

The extended ancillary data is inserted beginning from the end of the base frame. It is recommended that it be parsed from the end. The description in table D.2 is in the reverse order of the transmission. The bit order in each byte is, however, such that the MSB comes first in the transmission.

Table D.2: Extended ancillary data syntax

Syntax	Number of bits	Mnemonic
<code>extended_ancillary_data() {</code>		
<code>dvd_ancillary_data</code>	16	bslfb
<code>extended_ancillary_data_sync</code> (set to 0xBC)	8	bslfb
<code>bs_info</code>	8	bslbf
<code>ancillary_data_status</code>	8	bslbf
<code>if(advanced_dynamic_range_control_status == 1)</code>		
<code>advanced_dynamic_range_control</code>	24	bslbf
<code>if(dialog_normalization_status == 1)</code>		
<code>dialog_normalization</code>	8	bslbf
<code>if(reproduction_level_status == 1)</code>		
<code>reproduction_level</code>	8	bslbf
<code>if(downmixing_levels_MPEG2_status == 1)</code>		
<code>downmixing_levels_MPEG2</code>	8	bslbf
<code>if(audio_coding_mode_and_compression_status == 1) {</code>		
<code>audio_coding_mode</code>	8	bslbf
<code>compression</code>	8	bslbf
<code>}</code>		
<code>if(coarse_grain_timecode_status == 1)</code>		
<code>coarse_grain_timecode</code>	16	bslbf
<code>if(fine_grain_timecode_status == 1)</code>		
<code>fine_grain_timecode</code>	16	bslbf
<code>if(scale_factor_CRC_status == 1)</code>		
<code>scale_factor_CRC</code>	16 to 32	bslbf
<code>}</code>		

The elements of the ancillary data structure are described in the following paragraphs. The order of the bits is in transmission order, MSB first.

D.4.2.1 ancillary_data_sync

- Encoding: This field shall be set to 0xBC.
- Decoding: The decoder may use this field to verify the availability of the extended ancillary data. If the IRD indicates that this information is present, this takes precedence.

D.4.2.2 bs_info

The detailed syntax is described in table D.3.

Table D.3: bs_info syntax

Syntax	Number of bits	Mnemonic
<code>bs_info() {</code>		
<code>mpeg_audio_type</code>	2	bslbf
<code>dolby_surround_mode</code>	2	bslbf
<code>ancillary_data_bytes</code>	4	uimsbf
<code>}</code>		

D.4.2.3 mpeg_audio_type

Table D.4: MPEG audio type table

mpeg_audio_type	Description
"00"	Reserved
"01"	Only MPEG1 audio data
"10"	MPEG2 audio data
"11"	Reserved

Decoding: The decoder may ignore this field.

D.4.2.4 dolby_surround_mode

Table D.5: Dolby surround mode table

mpeg_audio_type	Description
"00"	Reserved
"01"	MPEG1 part is not Dolby surround encoded
"10"	MPEG1 part is Dolby surround encoded
"11"	Reserved

Decoding: It is recommended that the decoder parse this field and provides this information to the reproduction set-up.

D.4.2.5 ancillary_data_bytes

This field indicates the amount of ancillary data bytes that precede this byte in the transmission. This field may be used by the decoder as an indication of how many bytes it needs to buffer.

D.4.2.6 ancillary_data_status

The detailed syntax is described on table D.6.

Table D.6: ancillary_data_status syntax

Syntax	Number of bits	Mnemonic
<code>ancillary_data_status() {</code>		
<code>advanced_dynamic_range_control_status</code>	1	bslbf
<code>dialog_normalization_status</code>	1	bslbf
<code>reproduction_level_status</code>	1	bslbf
<code>downmix_levels_MPEG2_status</code>	1	bslbf
<code>scale_factor_CRC_status</code>	1	bslbf
<code>audio_coding_mode_and_compression_status</code>	1	bslbf
<code>coarse_grain_timecode_status</code>	1	bslbf
<code>fine_grain_timecode_status</code>	1	bslbf
<code>}</code>		

Semantics: The bits in this field indicate the presence of the associated fields in the ancillary data.

Encoding: A bit in this field shall be set to "1" if the associated field is present in the bit stream.

Decoding: It is recommended that the decoder parse this field to allow parsing of the following fields in the ancillary data section.

D.4.2.7 advanced_dynamic_range_control

The detailed syntax is described on table D.7.

Table D.7: advanced_dynamic_range_control syntax

Syntax	Number of bits	Mnemonic
<code>advanced_dynamic_range_control() {</code>		
<code>advanced_drc_part_0</code>	8	bslbf
<code>advanced_drc_part_1</code>	8	bslbf
<code>advanced_drc_part_2</code>	8	bslbf
<code>}</code>		

Semantics: Each field consists of an unsigned integer value X in the three MSBs and an unsigned integer value Y in the five LSBs. The actual value is $24,082 + 6,0206 X - 0,2007 Y$ dB. The 1 152 samples of an MPEG2 frame are divided in 3 parts of 384 samples. The advanced_drc values are applicable for the corresponding part of the audio frame.

Decoding: If this field is present and the decoder supports this type of dynamic range control, these values shall be used rather than the DVD-Video ancillary data. The decoder shall apply these values to the sub band samples, before the reconstruction filter. These values may be scaled in the decoder to allow user control of the amount of dynamic range compression that is applied.

D.4.2.8 dialog_normalization

The detailed syntax is described on table D.8.

Table D.8: dialog_normalization syntax

Syntax	Number of bits	Mnemonic
<code>dialog_normalization() {</code>		
<code>dialog_normalization_on</code>	2	bslbf
<code>dialog_normalization_value</code>	6	uimsbf
<code>}</code>		

D.4.2.9 dialog_normalization_on

Table D.9: Dialog normalization table

dialog_normalization_on	Description
"00"	dialog_normalization_value is not valid
"01"	reserved
"10"	dialog_normalization_value is valid
"11"	Reserved

D.4.2.10 dialog_normalization_value

Semantics: This field represents the headroom in dB of the dialogue component in the MPEG1 compatible part, relative to full-scale sine wave. Values 41 through 63 are reserved. When dialogue normalization is temporarily not applied, "Dialogue_Normalization_on" shall be set to "00" and "Dialogue_Normalization_value" shall be set to "000000".

Decoding: It is recommended that the decoder parse this field. The decoder should apply these values to the sub band samples, before the reconstruction filter, in order to allow reproduction of different programmes with the same dialogue level.

D.4.2.11 reproduction_level

The detailed syntax is described on table D.10.

Table D.10: reproduction_level syntax

Syntax	Number of bits	Mnemonic
<code>reproduction_level () {</code>		
<code>surround_reproduction_level</code>	1	bslbf
<code>production_roomtype</code>	2	bslbf
<code>reproduction_level_value</code>	5	uimsbf
<code>}</code>		

D.4.2.12 surround_reproduction_level

Table D.11: Surround reproduction level table

surround_reproduction_level	Description
"0"	The surround channels have the correct level for reproduction
"1"	The surround channels should be attenuated by 3 dB during reproduction

Decoding: It is recommended that the decoder parse this field and pass the value to the reproduction unit to allow correct adjustment of the surround levels.

D.4.2.13 production_roomtype

Table D.12: Production room type table

production_roomtype	Description
"00"	not indicated
"01"	large room
"10"	small room
"11"	reserved

Decoding: It is recommended that the decoder parse this field and pass the value to the reproduction unit to allow correct adjustment of the monitoring equipment.

D.4.2.14 reproduction_level_value

Semantics: This field represents the absolute acoustic sound pressure level in dB SPL during the final audio mixing session.

Decoding: The decoder may ignore this field.

D.4.2.15 downmixing_levels_MPEG2

The detailed syntax is described on table D.13. The down mixing levels describe the down mix in the decoder for stereo reproduction.

Table D.13: downmixing_levels_MPEG2 syntax

Syntax	Number of bits	Mnemonic
<code>downmixing_levels_MPEG2 () {</code>		
<code>center_mix_level_on</code>	1	bslbf
<code>center_mix_level_value</code>	3	bslbf
<code>surround_mix_level_on</code>	1	bslbf
<code>surround_mix_level_value</code>	3	bslbf
<code>}</code>		

D.4.2.16 center_mix_level_on

Semantics: If this field is set to "1" the center_mix_value field indicates nominal down mix level of the centre channel with respect to the left and right front channels. If this field is set to "0" the center_mix_value field shall be set to "000".

Decoding: It is recommended that the decoder parse this field.

D.4.2.17 surround_mix_level_on

Semantics: If this field is set to "1" the surround_mix_value field indicates nominal down mix level of the surround channels with respect to the left and right front channels. If this field is set to "0" the surround_mix_value field shall be set to "000".

Decoding: It is recommended that the decoder parse this field.

D.4.2.18 mix_level_value

Table D.14: Mix level value Table

mix_level_value	Multiplication factor
"000"	1,000 (0,0 dB)
'001'	0,841 (-1,5 dB)
'010'	0,707 (-3,0 dB)
'011'	0,596 (-4,5 dB)
'100'	0,500 (-6,0 dB)
'101'	0,422 (-7,5dB)
'110'	0,355 (-9,0 dB)
'111'	0,000 (-∞ dB)

Decoding: The multi-channel decoder may apply these values as gain factors to the individual channels when a down mix for stereo listening has to be created. The values need to be scaled to avoid overload after the mixing process.

D.4.2.19 audio_coding_mode

The detailed syntax is described in table D.15.

Table D.15: audio coding mode syntax

Syntax	Number of bits	Mnemonic
<code>audio_coding_mode () {</code>		
MPEG2_extension_stream_present	1	bslbf
MPEG2_center	2	bslbf
MPEG2_surround	2	bslbf
MPEG2_lfeon	1	bslbf
MPEG2_copyright_ident_present	1	bslbf
compression_on	1	bslbf
<code>}</code>		

Semantics: The semantics of the fields MPEG2_extension_stream_present, MPEG2_center, MPEG2_surround and MPEG2_lfeon is as defined in the mc_header field in ISO/IEC 13818-3 [3].

If MPEG2_copyright_ident_present is set to "0" the copyright identification in the MPEG 2mc_header is not filled in. If MPEG2_copyright_ident_present is set to "1" the copyright identification in the MPEG 2mc_header is used.

Decoding: The decoder may ignore this field. It may be parsed by multiplexers and bit stream monitors to simplify extraction of these parameters from a bit stream.

D.4.2.20 compression_on

Semantics: If this field is set to "1" the compression_value field indicates the heavy compression factor used for monophonic down mix reproduction. If this field is set to "0" the compression_value field shall be "0000 0000".

Decoding: It is recommended that the decoder parse this field.

D.4.2.21 compression_value

Semantics: This field consists of a value X in the four MSBs and a value Y in the four LSBs. The actual value is $48,164 + 6,0206 X - 0,4014 Y$ dB.

Decoding: These values shall be applied to the sub band samples, before the reconstruction filter when the decoder has to create a mix for monophonic listening where overloading of a subsequent analog transmission is highly undesirable.

D.4.2.22 coarse_grain_timecode

The detailed syntax is described on table D.16.

Table D.16: coarse grain time code syntax

Syntax	Number of bits	Mnemonic
<code>coarse_grain_timecode () {</code>		
coarse_grain_timecode_on	2	bslbf
coarse_grain_timecode_value	14	bslbf
<code>}</code>		

Semantics: If coarse_grain_timecode_on is set to "10" the five MSBs of this value represents the time in hours, the next six bits represent time in minutes, and the final three bits represent the time in eight second increments. If coarse_grain_timecode_on is not set to "10" all the bits of coarse_grain_timecode_value shall be set to "0".

Decoding: The decoder may ignore this field.

D.4.2.23 fine_grain_timecode

The detailed syntax is described in table D.17.

Table D.17: fine grain time code syntax

Syntax	Number of bits	Mnemonic
<code>fine_grain_timecode () {</code>		
<code>fine_grain_timecode_on</code>	2	bslbf
<code>fine_grain_timecode_value</code>	14	bslbf
<code>}</code>		

Semantics: If `fine_grain_timecode_on` is set to "10" the three msb's of this value represents the time in seconds, the next five bits represent time in video frames, and the final six bits represent the time in fractions of 1/64 of a video frame. If `fine_grain_timecode_on` is not set to "10" all the bits of `fine_grain_timecode_value` shall be set to "0".

Decoding: The decoder may ignore this field.

D.4.2.24 scale_factor_CRC

Semantics: The `scale_factor_CRC` permits to verify the integrity of the MPEG Audio scale factors. The coding is according to TS 102 114 [16].

Encoding: It recommended that `scale_factor_CRC` be included for mobile applications.

Decoding: It is recommended to parse the data from the end. The length of the field depends on the bit rate index of the MPEG 1 header of the following frame. It is recommended to always parse the full 32 possible bits.

D.4.3 Announcement switching data

The transmission of announcement switching data in the ancillary data field of MPEG audio frames is optional. The syntax of the announcement switching data field is described in table D.18. Note that the description in table D.18 is in the reverse order of the transmission. The bit order in each byte is, however, such that the MSB comes first in the transmission. The data field length gives the number of bytes following this byte within this data field.

Table D.18: Announcement switching data field

Syntax	Number of bits	Mnemonic
<code>announcement_switching_data() {</code>		
<code>announcement_switching_data_sync</code>	8	bslbf
<code>data_field_length</code>	8	bslbf
<code>announcement_switching_flag_field_1</code>	16	bslbf
<code>announcement_switching_flag_field_2</code>	16	bslbf
<code>}</code>		

Semantics: The `announcement_switching_data_sync` should be set to 0xAD.

The `announcement_switching_flag_fields` are 16-bit flag fields specifying which type of announcements are actually running. The association between the bits of the flag field and the announcement types shall be according to the `announcement_support_indicator` (EN 300 468 [7]). A bit shall be set to "1" if the announcement is running and it shall be set to "0" if the announcement is not running.

The `announcement_switching_flag_field_1` shall be used for announcements within the audio elementary stream that is actually decoded.

The `announcement_switching_flag_field_2` shall be used for announcements within other audio elementary streams. Corresponding links shall be provided by means of the `announcement_support_descriptor` (EN 300 468 [7]).

Encoding: The `announcement_switching_data_field` is allowed to be embedded at the end of a MPEG audio packet, between the end of the audio data and another data field that is part of the ancillary data field or between two other data fields that are part of the ancillary data field.

If data fields according to DVD-video, extended ancillary data or ancillary data according to the DAB specification TS 102 114 [16] are used, then the `announcement_switching_data_field` is not allowed to be inserted at the end of an audio packet.

Decoding: It is recommended to parse the data from the end.

D.4.4 Scale factor error check

The transmission of a scale factor error check in the ancillary data field of MPEG audio frames is optional. The syntax of the corresponding data field is described in table D.19. Note that the description in table D.19 is in the reverse order of the transmission. The bit order in each byte is, however, such that the MSB comes first in the transmission. The `data_field_length` gives the number of bytes following this byte within this data field.

Table D.19: Scale factor error check data field

Syntax	Number of bits	Mnemonic
<code>scale_factor_error_check_data() {</code>		
<code>scale_factor_error_check data_sync</code>	8	bslbf
<code>data_field_length</code>	8	bslbf
<code>scale factor CRC</code>	32	bslbf
<code>}</code>		

Semantics: The `scale_factor_error_check data_sync` should be set to 0xFE.

The `scale_factor CRC` permits to verify the integrity of the MPEG Audio scale factors.

Encoding: The `scale_factor_error_check` is allowed to be embedded at the end of a MPEG audio packet, between the end of the audio packet and another data field that is part of the ancillary data field or between two other data fields that are part of the ancillary data field.

If data fields according to DVD-video, extended ancillary data (as described in annex D) or ancillary data according to the DAB specification TS 102 114 [16] are used, then the `scale_factor_error_check_data_field` is not allowed to be inserted at the end of an audio packet.

Decoding: It is recommended to parse the data from the end.

Annex E (informative): Coding of data fields in the private data bytes of the adaptation field

E.1 Introduction

This annex contains the guidelines required to include and to decode data fields in the private data bytes of the adaptation field (ISO/IEC 13818-1 [1]).

E.2 Detailed specification

Transport Stream (TS) packets coded according to ISO/IEC 13818-1 [1] may include an adaptation field. The presence of an adaptation field is indicated by means of the `adaptation_field_control`, i.e. a 2-bit field in the header of the TS packet. The adaptation field itself may contain `private_data_bytes`. The presence of private data bytes is signalled by means of the `transport_private_data_flag` coded at the beginning of the adaptation field. If private data bytes exist the total number of private data bytes is specified by means of the `transport_private_data_length`, an 8-bit field that is directly followed by the private data bytes. The private data bytes may be composed of one or more data fields as shown in figure E.1. Gaps are not allowed between two data fields.

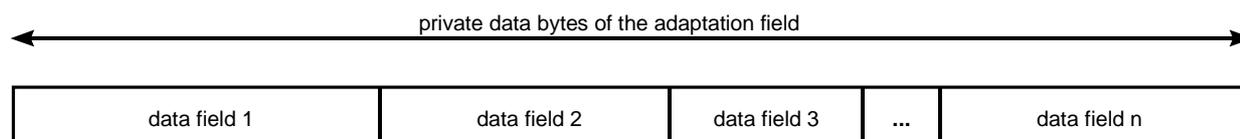


Figure E.1: Coding scheme for private data bytes within the adaptation field

Encoding: The support of data fields that are specified in this annex shall be indicated by means of the `adaptation_field_data_descriptor` EN 300 468 [7]. This descriptor shall be inserted in the corresponding `ES_info` loop of the.

Moreover, the following semantics apply to all data fields specified in this annex.

data_field_tag: The data field tag is an 8-bit field which identifies the type of each data field. The values of `data_field_tag` are defined in table E.1.

data_field_length: The data field length is an 8-bit field specifying the total number of bytes of the data portion of the data field following the byte defining the value of this field.

Table E.1: Allocation of data_field_tags

data_field_tag	Description
0x00	Reserved
0x01	Announcement switching data field
0x02 to 0x9F	Reserved for future use
0xA0 to 0xFF	User defined

Decoding: The IRD design should be made under the assumption that any structure as permitted by this Annex may occur in the broadcast stream. The IRD is not required to make use of this data.

E.2.1 Announcement switching data

The announcement switching data field is used to indicate whether spoken announcements are actually running or not. In comparison with that, the general support of announcements is indicated by means of the announcement_support_descriptor EN 300 468 [7].

The transmission of the announcement switching data field is optional but it shall be continuously provided in those audio streams that may carry announcements at some point in time. The announcement switching data field shall be present at least every 100 ms. The syntax of the announcement switching data field is described in table E.2.

Table E.2: Announcement switching data field

Syntax	Number of bits	Mnemonic
<code>announcement_switching_data() {</code>		
<code>data_field_tag</code>	8	uimsbf
<code>data_field_length</code>	8	uimsbf
<code>announcement_switching_flag_field</code>	16	bslbf
<code>}</code>		

Semantics: announcement_switching_flag_field: This 16-bit flag field specifies which type of announcements are actually running. The association between the bits of the flag field and the announcement types shall be according to the announcement_support_indicator that is specified for the announcement_support_descriptor EN 300 468 [7]. A bit shall be set to "1" if the announcement is running and it shall be set to "0" if the announcement is not running.

Annex F (informative): Guidelines for the implementation of DTS coded audio in DVB compliant transport streams

F.1 Scope

The inclusion of DTS coded audio streams in a DVB multiplex is optional, and IRDs may optionally decode these streams. This annex contains the guidelines to include one or more DTS coded elementary streams in a DVB Transport Stream in compliance with ISO/IEC 13818-1 [1]. The coding and decoding of a DTS coded elementary stream is based upon TS 102 114 [16].

It is recommended that implementations of DVB systems that include DTS coded audio streams should comply with this annex.

The DTS packetized elementary stream shall conform to the requirements of a user private stream type 1, as described in ISO/IEC 13818-1 [1].

The IRD design should be made under the assumption that any legal structure as permitted by ISO/IEC 13818-1 [1], including private data streams, may occur in the Transport Stream, even if presently reserved or unused. To allow full compliance to the MPEG-2 standard and upward compatibility with future enhanced versions, a DVB IRD shall be able to skip over data structures which are currently "reserved", or which correspond to functions not implemented by the IRD.

This clause is based on ISO/IEC 13818-1 [1].

F.2 Introduction

A DTS coded elementary bit stream may be multiplexed into an MPEG-2 transport stream in much the same way an MPEG-1 or AC-3 audio stream would be included. The DTS coded elementary stream is packetized into PES packets with a structure similar to an MPEG audio PES.

It is necessary to unambiguously indicate that an MPEG private stream is, in fact, a DTS coded stream. A public DVB descriptor, the `DTS_audio_descriptor` will be specified for this purpose and is defined as 0x73. The DTS registration_descriptor outlined in table F.1 must also be specified. The syntactical elements that need to be specified in order to include DTS within an MPEG-2 transport stream are: the MPEG stream_type, stream_id and the DVB `DTS_audio_descriptor`.

IRDs shall decode all bit rates and sample rates listed herein.

Some constraints are placed on the PES layer for the case of multiple audio streams intended to be reproduced in exact sample synchronism as described in clause 5.

F.3 DVB compliant streams

The DTS PES shall be carried as an MPEG private data stream type, conforming to the structure of a `private_stream_1` as described in ISO/IEC 13818-1 [1] tables 2-18 (`stream_id`) and 2-29 (`stream_type`).

When a DTS stream is included in a DVB transport stream, the DTS Audio descriptor (`DTS_audio_descriptor`) shall also be included. The DTS Audio descriptor is defined in annex F of EN 300 468 [7], but for information a description is included here in clause 4.3. Either the DTS Audio Descriptor or the DTS registration descriptor must be located in the PMT to identify the DTS stream as such; similarly one of DTS Audio Descriptor or DTS registration descriptor must be located in the SIT. The DTS Audio descriptor is located in the PMT and the Selection Information Table of the DVB SI tables in annex F of EN 300 468 [7].

DTS streams may also be signalled by the presence of a component_descriptor where the stream_content value is 0x05 (see EN 300 468 [7], clause 6.2.7) in the relevant service information tables.

Certain other of the DVB Service Information in EN 300 468 [7] can provide additional means of identifying the existence of a DTS stream without accessing the PMT.

F.4 Detailed specification

F.4.1 MPEG Transport Stream compliance

F.4.1.1 stream_id

- Semantics:** The semantics of the stream_id field are described in ISO/IEC 13818-1 [1] table 2-18. Multiple DTS streams may share the same value of stream_id since each stream is carried with a unique PID value. The mapping of values of PID to stream_type is indicated in the transport stream Program Map Table (PMT).
- Encoding:** The value of the stream_id field for a DTS elementary stream shall be 0xBD (indicating private_stream_1). If multiple DTS elementary streams are carried in a program stream the stream_id shall use values 110x xxxx where x xxxx indicates a stream number. Confusion may be avoided by use of a Program Stream Map, which associates values of a stream_id with a stream_type.
- Decoding:** This field shall be read by the IRD, and the IRD shall interpret this field in accordance with MPEG systems syntax.

F.4.1.2 stream_type

- Semantics:** The semantics of the stream_type field are described in ISO/IEC 13818-1 [1].
- Encoding:** The recommended value of stream_type for a DTS elementary stream shall be 0x06 (indicating PES packets containing private data) or any value which the MPEG-2 specification has assigned as "user private".
- Decoding:** This field shall be read by the IRD, and the IRD shall interpret this field in accordance with MPEG systems syntax.

F.4.2 DTS registration descriptor

The DTS registration descriptor is shown in table F.1. It is mandatory that the IRD decodes of the registration descriptor so that the stream is clearly identified as carrying DTS data.

Table F.1: DTS registration descriptor

Syntax	Number of bits	Mnemonic
Registration_descriptor(){		
descriptor_tag	8	uimsbf
descriptor_length	8	uimsbf
format_identifier	32	uimsbf
}		

F.4.2.1 descriptor_tag

- Encoding: The registration descriptor tag is an 8-bit field, which identifies each descriptor. The value assigned to the DTS descriptor_tag is 0x05.
- Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

F.4.2.2 descriptor_length

- Semantics: This 8-bit field specifies the total number of bytes of the data portion of the registration descriptor following the byte defining the value of this field. The value assigned to the DTS registration descriptor_length is 0x04.
- Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

F.4.2.3 format_identifier

- Encoding: The SMPTE registered format identifier sets the frame size for the DTS coded stream and is set according the values as follows;
- DTS format identifier is 0x44545331 ("DTS1") for frame size 512.
 - DTS format identifier is 0x44545332 ("DTS2") for frame size 1 024.
 - DTS format identifier is 0x44545333 ("DTS3") for frame size 2 048.
- Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

F.4.3 DTS audio descriptor

The DTS audio descriptor is shown in table F.2. It is optional that the IRD decodes the DTS audio descriptor.

Table F.2: DTS audio descriptor

Syntax	Number of bits	Mnemonic
DTS_audio_stream_descriptor() {		
descriptor_tag	8	uimsbf
descriptor_length	8	uimsbf
sample_rate_code	4	bslbf
bit_rate_code	6	bslbf
nblks	7	bslbf
fsize	14	uimsbf
surround_mode	6	bslbf
lfe_flag	1	uimsbf
extended_surround_flag	2	uimsbf
for(i=0;i<N;i++)	8*N	
{		
additional_info[N]		bslbf
}		
}		

F.4.3.1 descriptor_tag

- Encoding: The audio descriptor tag is an 8-bit field, which identifies each descriptor. The proposed value assigned to the audio descriptor_tag is defined as 0x73.
- Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

F.4.3.2 descriptor_length

- Semantics: This 8-bit field specifies the total number of bytes of the data portion of the audio descriptor following the byte defining the value of this field.
- Decoding: This field shall be read by the IRD, and the IRD shall interpret this field in accordance with ISO/IEC 13818-1 [1].

F.4.3.3 sample_rate_code

- Semantics: This 4-bit field is equivalent to SFREQ in DTS Coherent Acoustics. Specification and details are listed in table F.3. While broadcasters may use only a subset of these the complete table is given for consistency with the DTS Coherent Acoustics specification as defined in TS 102 114 [16].
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

Table F.3: Sample rate code

sample_rate_code	Sample rate
0b0000	Invalid
0b0001	8 kHz
0b0010	16 kHz
0b0011	32 kHz
0b0100	64 kHz
0b0101	128 kHz
0b0110	11,025 kHz
0b0111	22,05 kHz
0b1000	44,1 kHz
0b1001	88,02 kHz
0b1010	176,4 kHz
0b1011	12 kHz
0b1100	24 kHz
0b1101	48 kHz
0b1110	96 kHz
0b1111	192 kHz

F.4.3.4 bit_rate_code

- Semantics: The specification and details of typical broadcast bit_rate_code are listed in table F.4. While broadcasters may use only a subset of these, the complete table of fixed transmission bit rate values is given for consistency with the DTS Coherent Acoustics specification as defined in TS 102 114 [16].
- NOTE: It is recommended that DTS 5.1 compressed audio streams be transmitted at data rate of 384 kbps or above.
- Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

Table F.4: Bit rate table

bit_rate_code	Transmission bit rate
0bx00101	128 kbps
0bx00110	192 kbps
0bx00111	224 kbps
0bx01000	256 kbps
0bx01001	320 kbps
0bx01010	384 kbps
0bx01011	448 kbps
0bx01100	512 kbps
0bx01101	576 kbps
0bx01110	640 kbps
0bx01111	768 kbps
0bx10000	960 kbps
0bx10001	1 024 kbps
0bx10010	1 152 kbps
0bx10011	1 280 kbps
0bx10100	1 344 kbps
0bx10101	1 408 kbps
0bx10110	1 411,2 kbps
0bx10111	1 472 kbps
0bx11000	1 536 kbps
0bx11001	1 920 kbps
0bx11010	2 048 kbps
0bx11011	3 072 kbps
0bx11100	3 840 kbps
0bx11101	open
0bx11110	variable
0bx11111	lossless

NOTE: "x" indicated the bit is reserved and should be ignored.

F.4.3.5 nblks

Semantics: This 7-bit word is equivalent to NBLKS in listed in TS 102 114 [16]. This equals the number of PCM Sample Blocks. It indicates that there are (NBLKS+1) blocks (a block = 32 PCM samples per channel, corresponding to the number of PCM samples that are fed to the filterbank to generate one subband sample for each subband) in the current frame. The actual encoding window size is $32 \times (\text{NBLKS} + 1)$ PCM samples per channel. Valid range: 5 to 127. Invalid range: 0 to 4. For normal frames, this indicates a window size of either 2 048, 1 024 or 512 samples per channel. For termination frames, NBLKS can take any value in its valid range.

Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

F.4.3.6 Fsize

Semantics: This 14-bit word is equivalent to FSIZE listed in TS 102 114 [16]. (FSIZE+1) is the byte size of the current primary audio frame. The valid range for fsize is 95 to 8 192. The invalid range for fsize is 0 to 94, 8 193 to 16 384.

Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

F.4.3.7 surround_mode

Semantics: This 6-bit word is equivalent to AMODE in DTS Coherent Acoustics Specification. The values for surround_mode are given in table F.5. While broadcasters may use only a subset of these the complete table is given for consistency in TS 102 114 [16], table 5.4.

Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

Table F.5: Surround mode

Surround_mode	Number of channels/Channel layout
0b000000	1 / mono
0b000010	2 / L + R (stereo)
0b000011	2 / (L+R) + (L-R) (sum-difference)
0b000100	2 / LT +RT (left and right total)
0b000101	3 / C + L + R
0b000110	3 / L + R+ S
0b000111	4 / C + L + R+ S
0b001000	4 / L + R+ SL+SR
0b001001	5 / C + L + R+ SL+SR
0b001010	User defined
0b001011	User defined
0b001100	User defined
0b001101	User defined
0b001110	User defined
0b001111	User defined
0b010000 - 0b111111	User defined

Legends:
L = Left;
R = Right;
C = Centre;
SL = Surround Left;
SR = Surround Right;
T = Total.

F.4.3.8 lfe_flag

Semantics: The lfe flag shall be set to 0 when the Low Frequency Effects (LFE) audio channel is OFF. The flag shall be set to 1 when the LFE audio channel is ON.

Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

F.4.3.9 extended_surround_flag

Semantics: The extended_surround_flag indicates the presence of DTS ES rear centre audio as defined in TS 102 114 [16]. Its values are given in table F.6.

Decoding: IRDs shall be able to accept bit-streams, which contain this field. It is recommended IRDs decode this field.

Table F.6: extended_surround_flag values

Value	Description
00	No Extended Surround
01	Matrixed Extended Surround
10	Discrete Extended Surround
11	undefined

F.4.4 Use of the DVB-SI component_descriptor

Semantics: The semantics of the component_descriptor is defined in EN 300 468 [7]. The stream_content and component_type assigned values for DVB DTS audio stream are listed in annex F of EN 300 468 [7].

Encoding: The values for the elements of the component_descriptor shall be set in accordance with annex F of EN 300 468 [7].

Decoding: This field shall be read by the IRD, and the IRD shall interpret this field to indicate the type of audio service present.

F.5 PES constraints

F.5.1 Encoding

In some applications, the audio decoder may be capable of simultaneously decoding two elementary streams containing different programme elements, and then combining the programme elements into a complete programme.

Most of the programme elements are found in the main audio service. Another programme element (such as a narration of the picture content intended for the visually impaired listener) may be found in the associated audio service.

In order to have the audio from the two elementary streams reproduced in exact sample synchronism, it is necessary for the original audio elementary stream encoders to have encoded the two audio programme elements frame synchronously; i.e. if audio stream 1 has sample 0 of frame n taken at time t_0 , then audio stream 2 should also have frame n beginning with its sample 0 taken the identical time t_0 . If the encoding of multiple audio services is done frame and sample synchronous, and decoding is intended to be frame and sample synchronous, then the PES packets of these audio services shall contain identical values of PTS, which refer to the audio access units intended for synchronous decoding.

Audio services intended to be combined together for reproduction shall be encoded at an identical sample rate.

F.5.2 Decoding

If audio access units from two audio services which are to be simultaneously decoded have identical values of PTS indicated in their corresponding PES headers, then the corresponding audio access units shall be presented to the audio decoder for simultaneous synchronous decoding. Synchronous decoding means that for corresponding audio frames (access units), corresponding audio samples are presented at the identical time.

If the PTS values do not match (indicating that the audio encoding was not frame synchronous) then the audio frames (access units) of the main audio service may be presented to the audio decoder for decoding and presentation at the time indicated by the PTS. An associated service, which is being simultaneously decoded, may have its audio frames (access units), which are in closest time alignment (as indicated by the PTS) to those of the main service being decoded, presented to the audio decoder for simultaneous decoding. In this case the associated service may be reproduced out of sync by as much as 1/2 of a frame time. (This is typically satisfactory; a visually impaired narration does not require highly precise timing).

F.5.3 DTS PES field constraints

The DTS Audio format PES packet is defined according to ISO/IEC 13818-1 [1] with the following exceptions.

F.5.3.1 stream_id

In Program Streams, the stream_id for DTS is "private_stream_1" = 1011 1101 = 0xBD.

F.5.3.2 data_alignment_indicator

This is a 1 bit flag. When set to a value of "1" it indicates that the PES packet header is immediately followed by the DTS audio syncword.

F.5.3.3 PTS_flags

This is a 2 bit field. If the PTS_flags field equals "10", the PTS fields shall be present in the PES packet header. If the PTS_flags field equals "00" no PTS fields shall be present in the PES packet header. The value "01" is forbidden and 11 is invalid for audio PES streams.

F.5.3.4 DSM_trick_mode_flag

A 1 bit flag, which when set to "1" it indicates the presence of an 8 bit trick mode field. This has no meaning for DTS audio and is hence 0.

F.5.3.5 PES_extension_flag

A 1 bit flag, which when set to "1" indicates that an extension field exists in this PES packet header. When set to a value of "0" it indicates that this field is not present. It is always set to zero for DTS audio packets.

F.5.3.6 stuffing_byte

fixed 8-bit value equal to "1111 1111" that can be inserted. It should not be sent to the decoder unless it is placed at the end of the DTS data prior to the next sync word. A maximum of 32 stuffing bytes may be inserted.

F.5.4 Byte-alignment

The DTS elementary stream shall be byte-aligned within the MPEG-2 data stream. This means that the initial 8 bits of a DTS frame shall reside in a single byte, which is carried by the MPEG-2 data stream.

Annex G (informative): Receiver-mixed Audio Description and other supplementary audio services

G.1 Overview

Audio Description (AD) delivers a description of the scene as an ancillary component associated with a TV service. It is intended to aid understanding and enjoyment particularly, but not exclusively, for viewers who have visual impairments.

Loud sound effects or music could make added description hard to discern so an important requirement is to adjust, on a passage-by-passage basis, the relative level of programme sound in the mix which the AD user hears. The programme maker is best able to determine the level under controlled conditions when authoring the AD - information to modulate the level of programme sound in the AD-capable receiver is thus transmitted within the AD stream.

Individual AD users will have different aural acuity, describers will have different styles of delivery (voice pitch and timbre), several voices may be used to describe one programme and there are, in practice, differences in audio signal level for different home receivers. An essential requirement is for the user to be able to adjust the volume of the description signal to suit his/her condition.

The ability to optionally mix one or more supplementary additional audio channels with the main programme sound can have other applications, including multi-language commentaries, use for interactivity, and educational purposes.

G.2 Coding

Description content is voice only and is conveyed as a mono signal coded in accordance with ISO/IEC 11172-3 [10].

The principles of processing in a basic AD decoder are shown diagrammatically below in figure G.1.

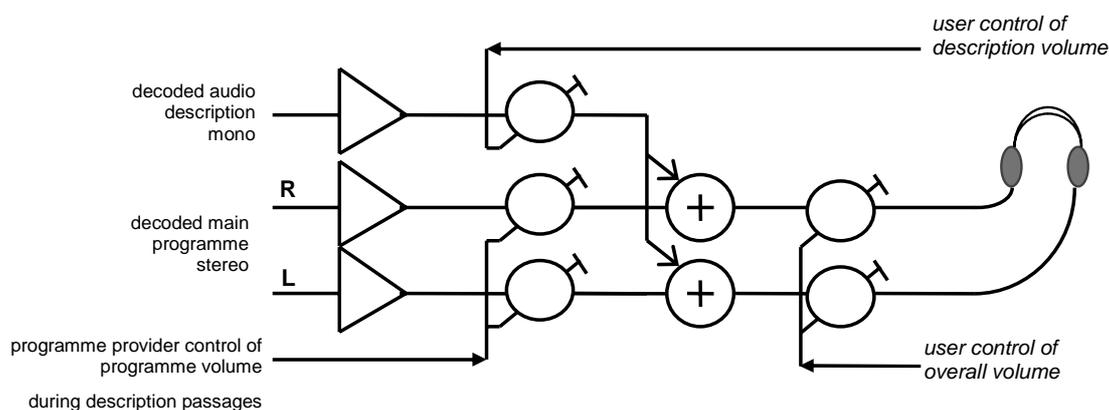


Figure G.1: Functionality of AD decoder processing

The level by which the programme sound should be attenuated during a description passage is signalled in PES_private_data within the PES encapsulation of the coded AD component (as specified in ISO/IEC 13818-1 [1]).

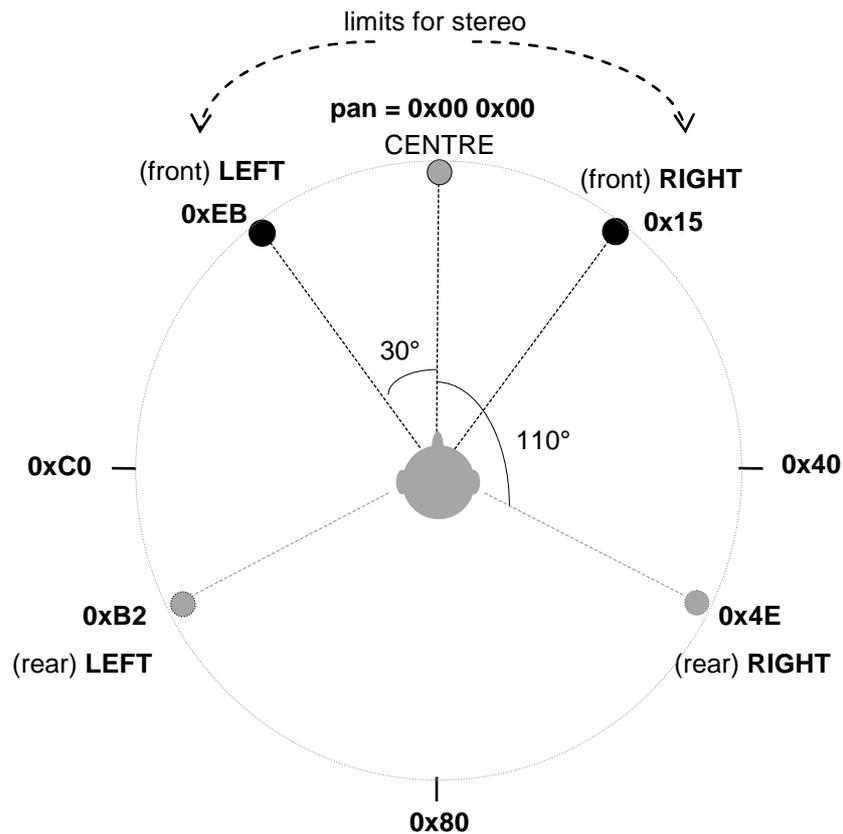
Coding: Support for the encoding of AD is optional.

Decoding: Support for the encoding of AD is optional.

The signalled fade value is an unsigned byte value, 0x00 representing 0 dB, each increment representing a nominal 0,3 dB, 0xFE representing approximately -77 dB whilst the fade value 0xFF represents completely mute programme sound.

A pan control value is also included within the transmitted data structure, enabling the decoded AD signal to be panned around the sound stage of the main programme sound and thus allowing the programme maker to place the "describer" at any preferred position within the sound field. As with fade, transmitted pan is a byte value, 0x00 representing centre front where each increment represents about 1,4° clockwise looking down on the listener (see figure G.2). For stereo the pan value will be restricted to $\pm 30^\circ$ of the centre front (i.e. to the range 0xEB..0xFF & 0x00..0x15) but the syntax of the signalling allows for any future use in which an AD component might be provided with a surround-sound main programme audio.

The values of fade and pan signalled in a PES packet apply to each access unit of AD sound contained within that same PES packet. This allows a fade (and a pan) to be relatively gradual or to be abrupt as the programme material allows.



NOTE: Seen from above the listener; includes mapping onto multi-channel sound presentation.

Figure G.2: Interpretation of audio description pan value

G.3 Syntax and semantics

AD fade and pan control information is coded in PES_private_data within the PES encapsulation of the coded AD component in accordance with ISO/IEC 13818-1 [1].

Table G.1: AD_descriptor

Syntax	value	Number of bits	Identifier
AD_descriptor {			
Reserved	1111	4	bslbf
AD_descriptor_length	1000	4	bslbf
AD_text_tag	0x4454474144	40	bslbf
revision_text_tag	0x31	8	bslbf
AD_fade_byte	0xXX	8	bslbf
AD_pan_byte	0xYY	8	bslbf
Reserved	0xFFFFFFFFFFFFFFF	56	bslbf
}			

AD_descriptor_length: The number of significant bytes following the length field (i.e. 8).

AD_text_tag: A string of 5 bytes forming a simple and unambiguous means of distinguishing this from any other PES_private_data. A receiver which fails to recognize this tag should not interpret this audio stream as audio description.

revision_text_tag: The AD_text_tag is extended by a single ASCII character version designator (here "1" indicates revision 1). Descriptors with the same AD_text_tag but a higher revision number shall be backwards compatible with the present document - the syntax and semantics of the fade and pan fields will be identical but some of the reserved bytes may be used for additional signalling.

AD_fade_byte: Takes values between 0x00 (representing no fade of the main programme sound) and 0xFF (representing a full fade). Over the range 0x00 to 0xFE one LSB represents a step in attenuation of the programme sound of approximately 0,3 dB giving a range of about 77 dB. The fade value of 0xFF represents no programme sound at all (i.e. mute). The rate of signalling and the expected behaviour of a decoder to changes in fade byte are described below.

AD_pan_byte: Takes values between 0x00 representing a central forward presentation of the audio description and 0xFF, each increment representing a 360°/256° step clockwise looking down on the listener (i.e. just over 1,4°, see figure G.2). The rate of signalling and the expected behaviour of a decoder are described below.

reserved: The remaining 7 bytes are set to 0xFF and reserved for future developments if and when required.

The maximum rate of signalling of fade and pan values is determined by the number of audio PES packets per second for that AD stream. For efficiency several Access Units (AUs) of audio are typically encapsulated within one PES packet and the fade and pan values in each AD_descriptor are deemed to apply to each AU encapsulated within, and which commences in, that PES packet. In typical efficient encapsulation fade and pan values are transmitted every 120 ms to 200 ms. This allows the control over the attack and decay of a fade where a particular gap in the narrative permits.

An AD decoder must maintain the relative timing between the decoded description signal and the decoded programme sound signal and between the appropriate fade and pan values and the decoded description signal.

During programmes for which there is no description there is little reason to transmit an AD stream of continual silence; in these cases the bit-rate accorded to AD may be reassigned for other purposes. Decoders should therefore be able to respond promptly to the restoration of the AD component at the start of a described programme.

The streams for programme sound and for AD are distinguished in the PSI by the use of the ISO_639_language descriptor. The audio_type field within the descriptor associated with programme sound is typically assigned the value 0x00 ("undefined") whilst the equivalent descriptor associated with AD has its audio_type field assigned the value 0x03 ("visual impaired commentary"). If a service has AD in several languages the PMT reference to each stream will have the appropriate ISO_639_language_code and the AD-capable decoder should discriminate between them on the basis of the preferred language chosen in the user settings.

G.4 Decoder behaviour

If there is a valid AD descriptor in the encoded description signal for the selected service, the AD decoder should present the appropriate mix of programme sound and description signal to the user, attenuating the programme sound by 0,3 dB per fade value increment. If the AD decoder cannot support such small steps then the implemented attenuation should match the intended attenuation as closely as possible.

EXAMPLE: If only 1 dB steps are possible then fade values of 0x00 and 0x01 should map to 0 db, 0x02, 0x03 and 0x04 should map to 1 dB, 0x05, 0x06, 0x07 and 0x08 to -2 db, etc.

When the fade value is 0x00 (or in the absence of an AD stream) the programme sound level should be unattenuated. Care should be taken to ensure that the default levels of programme sound and description are consistent when fed with streams coding standard level signals. It is also important that the mono description is matrixed to the stereo output so as to achieve a constant perceived description volume as the description is panned from stereo left through stereo centre to stereo right (see notes 1 and 2).

NOTE 1: E.g. using a model based on constant power as the description is panned across the stereo sound stage.

NOTE 2: The perceived loudness level of the main programme audio may well vary between different broadcast services. If the main programme audio is derived from a system using gain control metadata, for example AC-3, then the perceived loudness of the programme dialogue should be constant but it is likely to be different to that of a service for which the programme sound is delivered as MPEG-1 layer II. For any receiver which can decode main audio sources other than MPEG-1 layer II, the manufacturer may need to consider implementing different default gain levels for the audio description signal to provide a reasonable match of loudness to that of the programme dialogue. The ability of the user to adjust the relative level of description should nevertheless be retained.

In a stereo environment the AD decoder should interpret any pan values outside the ranges 0xEB..0xFF and 0x00..0x15 in the following manner. Pan values from 0x16 to 0x7F inclusive should be mapped to the value 0x15 (i.e. stereo hard right). Pan values from 0x80 to 0xEA should be mapped to the value 0xEB (i.e. stereo hard left).

When the user selects a new service or if the AD decoder detects an error in, or absence of, the AD descriptor in the encoded AD signal, the AD decoder should have a strategy which leads to muting the decoded description signal, restoring the programme sound to its default unfaded amplitude and setting the effective fade and pan values to 0x00. This restoration should not be abrupt - it is recommended that under such conditions the value of fade and of pan are ramped to the default values (0x00) over a period of at least 1 second. Equally, if the AD stream component is suddenly regained the implemented value of fade and of pan should be ramped to the signalled values from the default values (0x00) over a similar period.

G.5 Decoder user indicators

Description is typically confined to gaps in the programme narrative; these opportunities are therefore dependent on the programme. Some programmes are more suited to description than others; one may be effectively self-describing whilst another (e.g. news or a studio interview) might offer no opportunity for descriptive interpolation. Receiver implementations of AD should therefore allow the user to confirm that, in what may be extended gaps between description passages, description silence does not necessarily imply failure in delivery of the service or in the receiving equipment.

Many potential users of AD will be visually impaired. The user interface should not, therefore, rely solely on visual clues (lights or on-screen display logos) to indicate status (e.g. presence or absence of description). Audible indications are desirable and designers should consider how to distinguish different states using, for example, contrasting tones.

Annex H (informative): Bibliography

- ITU-R Recommendation BT.470-3: "Television systems".
- ETSI EN 300 401: "Radio Broadcasting Systems; Digital Audio Broadcasting (DAB) to mobile, portable and fixed receivers".
- ETSI TR 101 162: "Digital Video Broadcasting (DVB); Allocation of Service Information (SI) codes for DVB systems".
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History

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