

DIGITAL TELEVISION



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The Analogue TV World





TV Digital: What is it Really ?



All the information – video, audio, data - arrives to our houses as a discrete sequence of (pre-defined) symbols which together allow to resynthesize the original information with a minimum acceptable quality !



Why Digital TV ?

- More efficient usage of the spectrum
- More channels and services
- Interactivity
- Personalization
- Error robustness
- Audio and video quality control
- Easier processing
- Better relation with the computer world
- Easier multiplexing and encryption
- Possibility of information regeneration





TV of the Future: How will it Look like ?

- Set-top box + TV analogue
- Digital TV
- PC Card
- Mobile device
- Any type of digital receiver







The Digital Domestic Scenario







Which Arguments Convince the Users ?

- Satisfaction of important needs / added value / functionalities
- Interoperability at the application level users don't care much about the specific technical solution
- Quality and reliability
- Facility of usage
- Low cost of usage and equipment
- Variety and quality of content
- Interactivity





Interactivity

The digital representation of information facilitates the explosion of interactive capabilities – *user capability to select or change something, thus personalizing the television experience* associated to television and thus the capability of the users to:

- Access to thematic information
- Access to complementary information
- Control of the visualization sequence
- Select the visualization angle
- Express opinions, voting
- Use various services, e.g. tele-shopping, tele-banking





Winky Dink and You (1953-57, CBS, USA)...





Types of Interactivity



- Low Interactivity Zapping, audio control
- Medium Interactivity Defines program but does not change program, e.g.VOD, teletext
- **High Interactivity –** Changes the program, e.g. program personalization, definition of end, mix with Internet

Moreover, interactivity does not always require to use a feedback channel ...



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Main Digital TV Systems

After the satellite and the cable, the possibility to release bandwidth has brought digital TV also to the terrestrial systems ... and more ...

The main digital TV systems are:

- Digital Video Broadcasting (DVB) Driven by Europe
- Advanced Television Systems Committee (ATSC) Driven by USA
- Integrated Services Digital Broadcasting (ISDB) Driven by Japan (large similarities with DVB)
- Audio Video coding Standard (AVS) Driven by China
- Sistema Brasileiro de TV Digital Terrestre (SBTVD) Driven by Brazil (large similarities with ISDB)







What is DVB ?



- Consortium with 220 members from 30 countries (at the beginning mainly European), formed in September 1993:
 - Content producers
 - Equipment manufacturers
 - Telecom operators
 - Regulation organizations
- with the objective to define standards for digital television broadcasting over several transmission channels.
- Joint Technical Committee of ETSI / CENELEC / EBU



DVB: Initial Objectives



- High quality digital video delivery (up to HDTV)
- Delivery with good quality of TV programs using narrow bandwidth channels and increase the number of programs in current channels
- Reception in pocket terminals equipped with small reception antennas (portable reception)
- Mobile reception with good quality of TV programs
- Possibility of easy transmission over various telecom networks and integration with the PC world



From SDTV to HDTV ..













The New DVB Vision: Combining Worlds ...

DVB's vision is to build a content environment that combines the stability and interoperability of the world of broadcast with the vigor, innovation, and multiplicity of services of the world of the Internet."

DVB, 2000



The DVB Scenarios and Standards

- Satellite: DVB-S, DVB-S2
- Cable: DVB-C
- Terrestrial: DVB-T, DVB-T2
- DVB-MHP (Multimedia Home Platform) – middleware tools allowing to use a single set-top box for all services and applications (hardware abstraction)
- Portable: DVB-H







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DVB-S: Adoption ...





DVB-C: Adoption ...





DVB-T: Adoption ...





DVB Technologies





The DVB Specifications

The DVB specifications – also ETSI standards – define all the modules in the television delivery chain which need a normative specification; this is made using available standards defined by other standardization bodies or developing new (DVB) specifications.

The main modules specified are:

- Audio and Video Source Coding MPEG-2 Audio and MPEG-2 Video are adopted; later also H.264/AVC has been adopted
- Synchronization and Multiplexing MPEG-2 Systems is adopted
- Channel Coding
- Modulation
- Conditional Access



Note: No encryption is specified in MPEG-2 standards.



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MPEG-2 Standard





MPEG-2: Objectivos



Generic Coding of Moving Pictures and Associated Audio

Audio and video coding for high quality transmission and storage, e.g. high and medium definition television.

- The ISO/IEC MPEG-2 Video standard is a joint development with ITU-T where it is designated as Recommendation H.262.
- The MPEG-2 standard should have covered audiovisual coding up to 10 Mbit/s, leaving to MPEG-3 the higher rates and higher definition. However, since the MPEG-2 standard addressed well the HDTV space, MPEG-3 was never defined and MPEG-2 lost its upper bitrate limit.



MPEG-2: The Service Model





MPEG-2: Applications

- More channels due to the more efficient usage of the available bandwidth (mainly determined by coding and modulation)
- Cable, satellite, terrestrial digital TV
- HDTV, Stereoscopic TV
- Pay per view, Video on demand, Tele-shopping
- Games
- Storage, p.e. DVD
- High quality personal communications





MPEG-2: Which Advantages ?

- Offers more channels, e.g. thematic channels, regional channels
- Offers various angles of visualization, e.g. in the transmission of music or sports
- Introduction of high definition television
- Introduction of stereoscopic television
- Offers a large variety of television related services, e.g. VOD
- Releases bandwidth allocated to terrestrial TV, notably for the expansion of mobile networks



MPEG-2 Standard: Organization

- **Part 1 SYSTEMS** Specified the multiplexing, synchronization and protection of coded elementary bitstreams (audio, video and data).
- **Part 2 VIDEO Specifies the coded representation of video signals.**
- Part 3 AUDIO Specifies the coded representation of audio signals.
- Part 4 CONFORMANCE TESTING Specifies compliance tests for decoders and streams.
- **Part 5 REFERENCE SOFTWARE –** Includes software implementing the technical specification parts.
- Part 6 DSM-CC (Digital Storage Media Command Control) -Specifies user management and control protocols; they constitute and extension of the Systems parts.



MPEG-2 Standard

Part 1: Systems



MPEG-2 Systems: Objective

MPEG-2 Systems has the basic objective to combine and synchronize one or more coded audio and video bitstreams in a single multiplexed bitstream.

The main objectives of this standards regard:

- Multiplexing of various streams, e.g. audio and video from one program or several programs together
- Synchronization between streams, e.g. audio and video from one program or several programs



DTS - Decoding Time Stamp SCR - System Clock Reference (SCR) PTS - Presentation Time Stamp STC – System Time Clock



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Packetized Elementary Streams (PESs) & Packet Syntax

The audio and video coded elementary streams are divided into variable length packets - *the packets* – creating the so-called *Packetized Elementary Streams* (PESs), as for MPEG-1 Systems.



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Program Stream and Transport Stream

- Program Stream:
 - Stream with a single time base for all multiplexed streams
 - Adequate for transmission and storage in channels virtually without errors (BER < 10⁻¹⁰), e.g. CD-ROM, DVD, hard disks
 - Variable length packets as for MPEG-1 Systems
- Transport Stream:
 - Stream may include several time bases to combine programs with different time bases; however, each PES has a single time base
 - Adequate for transmission in error prone channels (BER > 10⁻⁴), e.g.. broadcasting
 - Packets with <u>a fixed length of 188 bytes</u>



Decoding Program Streams ...





Program Stream Syntax



MPEG-2 Program Streams are similar to MPEG-1 Systems streams.



Decoding Transport Streams ...





Transport Stream Syntax





'Surviving in the Labyrinth' ...



In order a user may find the elementary streams he/she needs in a MPEG-2 Transport Stream, e.g. audio and video for RTP 2 or SIC, some auxiliary data is needed !



Program Specific Information (PSI)

Program Specific Information (PSI) is delivered in the *transport stream* 'showing the path in the labyrinth'.

- PSI is carried using 4 tables (corresponding to a small bitrate budget)
- Each table is repeated many times (in a *carroussel*), e.g. 10-50/s, and corresponds to a different PID
- Tables are only applicable to Transport Streams
- A common syntax is defined to segment and carry the tables in Transport Packets
- The syntax allows a clean and backward compatible strategy to possibly extend the current standard with new tables, both standardized or privately (e.g. DVB) defined



Transport Stream PSI Tables

- *Program Association Table* (PAT) Corresponds to PID 0x00 and it is mandatory; it contains the PIDs for the PMTs corresponding to each program in each transport stream; it also contains the PID for the NIT.
- *Program Map Table* (PMT) Each PMT indicates the PIDs corresponding to the elementary streams <u>for each program</u>; it is always *on the clear* even if the programs are encrypted.
- Conditional Access Table (CAT) Corresponds to PID 0x01 and it contains the PIDs for the packets with conditional access data, e.g. corresponding to the DVB tables with the access keys for the encrypted programs.
- *Network Information Table* (NIT) Information about the network, e.g. the frequency for each RF channel (only the syntax is defined in MPEG-2).



Program Association Table (PAT)





- Delivered in the packets with PID = 0
- Indicates for all programs present in this transport stream, the relation between the program number (0 - 65535) and the PID of the packets transporting the map of that program, this means the Program Map Table
- The PAT is always sent without protection even if all programs in the transport stream are protected



Program Map Table (PMT)

- Provides detailed information about a specific program
- Identifies the packets (PIDs) transporting the audio and video elementary streams associated to the program it refers
- Identifies the PID for the packets transporting the temporal references associated to the relevant program clock (SCRs)
- May be enhanced with a set of descriptors (standard or user specified), e.g.
 - Video coding parameters
 - Audio coding parameters
 - Language identification
 - Conditional access information



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Relation between PAT and PMT

PID 0x0500





Network Information Table (NIT)

• Optional table with private content, i.e. its content is defined by the user and is not standardized by MPEG



- Should provide information about the physical network, e.g.
 - Channel frequencies
 - Satellite details
 - Modulation characteristics
 - Service provider
 - Alternative available networks
- When present, the PID for the NIT is contained in the PAT program zero



Conditional Access Table (CAT)



- Mandatory whenever there is, at least, one elementary stream in the transport stream which is protected
- Provides information about the used protection system (scrambling)
- Identifies the PIDs for the packets transporting the conditional access management and authorization information
- Its format is not specified by the MPEG-2 standard since it depends on the used protection mechanism which is typically operator dependent



Relation between PSI Tables ...

PAT (PID 0) CAT (PID 1) Program 016 Program 1/22 NIT Conditional Program 3 33 Access Data **Private** Program 95 Network Data PMT (PIØ 22) PMT (PID 33) Video 1 547 /ideo Q 48 81 Audio 1 udio Audio 2 49 Audio 82 66 ECM 88 FCM Prog 1 EMM Prog 1 PAT Prog 3 Prog 3 Prog 3 Prog 3 Prog 1 Prog 3 PMT PMT Audio Video Audio 1 Audio Video Video 22 0 1 33 49 19 81 19 54 82

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DVB Service Information (SI) Tables

DVB specifies additional tables which, among other things, allow the receiver to automatically configure itself and the user to navigate using an electronic program guide (EPG).

- Service Description Table (SDT) Includes the names and parameters for the services in the multiplexed stream.
- **Event Information Table (EIT)** Includes information related to events (current and future) in the same stream or in other multiplexed streams.
- **Time and Date Table (TDT)** Allows to update the internal clock of the settop box.
- **Bouquet Association Table (BAT)** Allows to group services in bouquets; one program may be part of one or more bouquets.
- **Running Status Table (RST)** Serves to update the situation of some events.
- **Stuffing Table (ST)** Serves to substitute tables that became invalid.



EPG: Program Timelining



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Zappping or Filtering ?





Description	
undefined content	
Movie/Drama:	
movie/drama (general)	
detective/thriller	
adventure/western/war	
science fiction/fantasy/horror	
comedy	
soap/melodrama/folkloric	
romance	
serious/classical/religious/historical movie/drama	
adult movie/drama	
reserved for future use	
user defined	
News/Current affairs:	
news/current affairs (general)	
news/weather report	
news magazine	
documentary	
discussion/interview/debate	
reserved for future use	
user defined	
Show/Game show:	
show/game show (general)	
game show/quiz/contest	
variety show	
talk show	
reserved for future use	
user defined	

Description		
Sports:		
sports (general)		
special events (Olympic Games, World Cup etc.)		
sports magazines		
football/soccer		
tennis/squash		
team sports (excluding football)		
athletics		
motor sport		
water sport		
winter sports		
equestrian		
martial sports		
reserved for future use		
user defined		
Children's/Youth programmes:		
children's/youth programmes (general)		
pre-school children's programmes		
entertainment programmes for 6 to14		
entertainment programmes for 10 to 16		
informational/educational/school programmes		
cartoons/puppets		
reserved for future use		
user defined		

Music/Ballet/Dance:

music/ballet/dance (general)	
rock/pop	
serious music/classical music	
folk/traditional music	
jazz	
musical/opera	
ballet	
reserved for future use	
user defined	



MPEG-2 Standard Part 2: Video



MPEG-2 Video (also H.262): Quality Objectives

The following quality objectives have been initially defined:

- Secondary distribution For broadcasting to the users, the signal quality at 3-5 Mbit/s must be better, or at least similar, to the quality of available analogue systems, i.e. PAL, SECAM and NTSC.
- **Primary distribution** For contribution, e.g. transmission between studios, the signal quality at 8-10 Mbit/s must be similar to the quality of Recommendation ITU-R 601 (using PCM).







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MPEG-2 Video: the Quality



The quality requirements depend on the application (thus type of content) and are strongly related to

- Resolution (in space and time) of the video signal
- Bitrate available (and thus compression factor)

Other important requirements related to quality:

- Quality robustness of the coding scheme to sudden changes of the signal statistics, e.g. scene changes
- Quality robustness to cascading this means successive coding and decoding processes



MPEG-2 Video: Requirements

- Large range of spatial and temporal resolutions, both in progressive and interlaced formats
- Several chrominance subsampling formats, e.g. 4:4:4, 4:2:2 and 4:2:0
- Flexibility in terms of bitrates, constant or variable
- Special modes, e.g. random access for edition and channel hoping, fast modes, conditional access, and easy transcoding to MPEG-1 Video, H.261 and JPEG
- Flexibility in adapting to different transmission and storage channels, e.g. in terms of synchronization and error resilience



MPEG-2 Video: the Compatibility

The compatibility among standards allows to offer some continuity regarding the already available standards – JPEG, H.261, MPEG-1 Video – providing some interoperability between the various applications.

Two main types of compatibility are relevant:

- **Backward compatibility** A MPEG-2 Video decoder is able to decode a coded bitstream compliant with a previously available standard.
- Forward compatibility A decoder compliant with a previously available standard, e.g. MPEG-1 Video, is able to, totally or partially, decode in a useful way a bitstream compliant with MPEG-2 Video.

MPEG-2 Video foresees some compatibility mechanisms with MPEG-1 Video (intrinsic to the MPEG-2 Video syntax) and H.261 (using spatial scalability).



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MPEG-2 Video: the Complexity



The complexity assessment of the encoders and decoders is essential for the adaptation to the technological constraints and adoption by the market.

- Assymmetric Applications For the 'one encoder, many decoders' type of applications, it is possible to develop high quality encoders even if at the cost of additional complexity since the overall system cost is mainly related to the decoders which should have a reduced complexity (and cost).
 - **Symmetric Applications** For the 'one to one' type of applications, both the encoders and decoder should have a reasonable (low) complexity.

The complexity of a codec is assessed based on parameters such as memory size to contain the reference images, required access to memory speed, number of operations per second, size of coding tables and number of coding table accesses per second.



Video Structure

- The video data is organized in a structure with 5 hierarchical layers (as for MPEG-1 Video):
 - Sequence
 - Group of Pictures (GOP)
 - Picture
 - Slice
 - Macroblock (MB)
 - Block





MPEG-2 Video: the Coding Tools

Temporal Redundancy



Predictive coding: temporal differences and motion compensation (uni and bidirectional; ¹/₂ pixel accuracy)

Spatial Redundancy

Transform coding (DCT)

Statistical Redundancy

Huffman entropy coding

• Irrelevancy



DCT coefficients quantization



Starting Again with the same Architecture ...





Temporal Prediction Structure



- The "conflict" between coding efficiency and random access led to the definition of 3 frame types depending on the coding tools used:
- Intra frames (I) Don't use temporal predictions
- Predicted frames (P) May only use forward prediction from previous I/P frame
- Bidirectionally predicted frames (B)

 May use both forward and backward prediction from previous and future I/P frame



MPEG-2 Video versus MPEG-1 Video

- The main differences between the MPEG-1 Video and MPEG-2 Video standards are related to:
- INTERLACING Coding of interlaced video content with MPEG-2 Video (which is not possible with MPEG-1 Video)
- SCALABILITY Availability of scalable coding in MPEG-2 Video (only temporal scalabilility with the I/P/B structure is possible with MPEG-1 Video)





MPEG-2 Video

Interlaced Coding



TV World: Progressive and Interlaced









Interlaced Content Coding

To more efficiently code interlaced content, MPEG-2 Video classifies each coded picture as:

- **Frame-Picture** The MBs to code are defined in the frame resulting from the combination of the 2 fields (top and bottom)
- **Field-Pictures-** The MBs to code are defined within each of the fields (top or bottom) which are independently processed





Main Prediction Modes

- Frame Mode for Frame-Pictures Similar to MPEG-1 Video, frames are coded as I, P or B frames with current and prediction MBs defined in the frames; gives good results for content with low or moderate motion or pannings over detailed backgrounds.
- Field Mode for Field-Pictures Conceptually similar to the previous mode but now with the MBs defined within each field and the predictions also coming from a single field, top or bottom (not necessarily with the same parity).
- Field Mode for Frame-Pictures Each MB in the frame-picture is divided in the pixels corresponding to the top and bottom fields; than, predictions are made for 16×8 matrices from one of the fields of the reference pictures.
- 16×8 Blocks for Field-Pictures A motion vector is allocated to each half of each MB for each field.



Frame-Pictures: Frame Mode and Field Mode Predictions



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Alternate Scanning Order ...



For frame-pictures, the correlation between lines is reduced for the pictures with more motion. Thus, it is possible to use another scanning order – ALTERNATE order – where the DCT coefficients corresponding to the vertical transitions (meaning horizontal edges) are privileged in terms of scanning order.



MPEG-2 Video

Scalable Coding



Scalability is a functionality regarding the <u>useful decoding of parts</u> of a coded bitstream, ideally

- i) while achieving an RD performance at any supported spatial, temporal, or SNR resolution that is comparable to single-layer (nonscalable) coding at that particular resolution, and
- ii) without significantly increasing the decoding complexity.





Scalability Types

Temporal: change of frame rate



Spatial: change of frame size





Fidelity: change of quality (a.k.a. SNR)

HDTV



High rate



TV



Scalable Coding Types: Spatial Scalability

• **SPATIAL SCALABILITY** – The original video signal is scalable coded with several spatial resolution layers.



Spatial scalability in MPEG-2



Scalable Coding Types: Quality Scalability

• **QUALITY (SNR) SCALABILITY** – Special case of spatial scalability where the spatial resolution is kept the same between layers (base and enhancement); the enhancement layers contain the data produced after the requantization of the residual signal between the original signal and the previous layer decoded signal.





Temporal and Frequency Scalability

- **TEMPORAL SCALABILITY** The original signal is coded with 2 or more layers with increasing temporal resolution; an example is also the coding of the interlaced signal in two layers where one layer corresponds to the top field and the other layer to the bottom field. <u>Temporal scalability is</u> <u>already provided by the temporal I/P/B prediction structure.</u>
- **FREQUENCY SCALABILITY** (designated *data partitioning* in MPEG-2 Video) – The coded information is structured in layers corresponding to subsets of DCT coefficients with increasing frequency; in the specific case of MPEG-2 Video, the partition is made in two layers.

Hybrid scalability combines two types of scalability in three or more scalable layers.



Combining the Coding Tools ...



The MPEG-2 Video Symbolic Model



A video sequence (interlaced or progressive) is represented, in a scalable way or not, as a succession of GOPs including pictures coded as frames or fields and classified as I, P or B, structured in macroblocks, each of them represented using motion vectors and/or DCT coefficients, following the constraints imposed by the picture coding type.



MPEG-2 Video: Encoder



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MPEG-2 Video: Decoder



To display

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MPEG-2 Video Syntax





MPEG-2 Video

Profiles and Levels



MPEG-2 Video: Very Big or Just Enough ?



- MPEG-2 Video is already a big standard !
- The MPEG-2 Video tools address many requirements from several application domains.
- Some tools are very likely useless in certain application domains.

It is essential to define adequate subsets of tools in terms of functionalities and complexity !



Profiles and Levels: Why ?

The profile and level concepts were first adopted by the MPEG-2 Video standard and they provide a trade-off between:

- Implementation complexity for a certain class of applications
- Interoperability between applications

while guaranteeing the necessary compression efficiency capability required by the class of applications in question and limiting the codec complexity and associated costs.

- **PROFILE –** Subset of coding tools corresponding to the requirements of a certain class of applications.
- LEVEL Establishes for each profile constraints on relevant coding parameters, e.g. bitrate and memory



Some MPEG-2 Video Profiles and Levels

1920×1152 pixels high (960×576) level 80 Mbit/s 100(80.25) Mbit/s 1440×1152 pixels 1440×1152 pixels 1440×1152 pixels high-1440 (720×576) (720×576) level 60 Mbit/s 60(40.15) Mbit/s 80(60.20) Mbit/s 720×576 pixels 720×576 pixels 720×576 pixels main 720×576 pixels (352×288) level 15 Mbit/s 15 Mbit/s 15(10) Mbit/s 20(15.4) Mbit/s low 352×288 pixels 352×288 pixels level 4 Mbit/s 4(3) Mbit/s levels. SNR scalable simple spatial scalable main high profile profile profile profile profile profiles (main profile, (main profile, (4:2:0, (SNR profile, (spatial profile, + SNR scalability) + spat. scalability) without B-pictures) no scalabilty) +4:2:2 (coding)

1920×1152 pixels



MPEG-2 Video: the Profile and Level Hierarchies



Some profiles are syntactically hierarchical this means one profile is syntactically a superset of another and so on.

For a profile, the syntactic elements do not vary with the level, just the parametric constraints.

Also the levels may be hierarchical meaning that the constraints become less strict for higher levels, e.g. bitrate increases.

Compliance points for decoder and bitstreams correspond to a profile@level combination.



- If an encoder produces a bitstream which is over, even if only slightly, the predefined limits for a certain profile and/or level, than it is classified with the profile or/and level immediately above (to guarantee decoding).
- If the decoding capabilities of a decoder are below, even if only slightly, from those predefined for a certain profile and/or level, than it is classified with the profile and/or level immediately below (to guarantee decoding).

This type of classification is important for the deployment and compliance of MPEG-2 Video content and decoders !



MPEG-2 Video in DVB

- Standard Definition TV (SDTV) uses MP@ML
 - Frame rate 25 or 30 Hz
 - Aspect ratio 4:3, 16:9 or 2.21:1
 - Spatial resolution (720, 576, 480) × 576 or 352 × (576, 288) or (720, 640, 544, 480, 352) × 480 or 352 × 540
 - Chrominance subsampling 4:2:2 or 4:2:0
- HDTV uses MP@HL
 - Frame rate 25, 50 or 30 e 60 Hz
 - Aspect ratio 16:9 or 2.21:1
 - Spatial resolution 1152 rows per frame at most and 1920 luminance samples per row at most
 - Complexity: 62 688 800 luminance samples per second at most



MPEG-2 Standard Part 3: Audio



Audio in MPEG-2: Objective

Efficient high quality audio coding targeting the broadcasting and storage of TV or TV like signals.

There are two parts in the MPEG-2 standard specifying audio codecs:

- Audio (Part 3) Codes up to 5 channels + 1 low frequency channel with high quality, at 384 kbit/s or less per channel, using the following additional sampling rates: 16, 22.05 and 24 kHz; offers backward and forward compatibilities with MPEG-1 Audio, thus the name of *MPEG-2 Audio Backward Compatible* (BC).
- Advanced Audio Coding (Part 7) Gives up on any compatibility with MPEG-1 Audio, increasing its rate-distortion performance, reaching higher quality for the same rate; codes 1 to 48 canais, with sampling rates from 8 to 96 kHz); it was initially designated as *MPEG-2 Audio Non-Backward Compatible* (NBC), now *Advanced Audio Coding* (AAC).



MPEG-2 Audio (Part 2): What's New ?

There are two main technical innovations in MPEG-2 Audio (BC or Part 2) regarding MPEG-1 Audio:

- Lower sampling frequencies (MPEG-2 Audio LSF): adding 16, 22.05 and 24 kHz to 32, 44.1 and 48 kHz
 - Motivated by the increase of low data rate applications over the Internet, it has the main goal to achieve MPEG-1 Audio or better audio quality at lower data rates using a lower bandwidth
- Multichannel coding
 - Motivated by the need to increase the user experience, notably with HDTV.



The three MPEG-1 Audio layers with different complexityperformance tradeoffs are again defined in MPEG-2 Audio Part 2.



The 5.1 multichannel configuration includes 5 full bandwidth channels and a low frequency enhancement (LFE) channel covering frequencies below 200 Hz (less than 10% of the full bandwidth).



MPEG-2 Audio: the Secret !



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MPEG-2 and MPEG-1 Audio Compatibility



Compatibility is provide through a MPEG-1 Audio compliant stereo pair and additional MPEG-2 Audio compliant data for the other channels.



MPEG-1/2 Audio in DVB



- All DVB audio decoders use MPEG-1 Audio, Layers 1 and 2, or MPEG-2 Audio Part 3 (BC), Layers 1 and 2.
- For MPEG-1 Audio, it is recommended to use Layer 2.
- Due to backward compatibility, it is possible to recover, with a MPEG-1 Audio decoder, a stereo pair from a multichannel MPEG-2 Audio BC coded bitstream (through downmixing).
- Sampling frequencies: 32, 44.1 and 48 kHz.



New Systems and ... Business Models ...



iPod Vídeos, Músicas, Fotos. 30GB, 80GB





iPod is able to play the following audio formats: MP3, WAV, AAC, <u>Protected AAC</u>, AIFF and Apple Lossless.



Technologies Developed by DVB



Channel Coding



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Channel Coding



- At sender, additional redundancy is included in the compressed signal in order to allow the channel decoder the detection and correction of channel errors.
- The introduction of added redundancy results in a bitrate increase. The channel coding selection must consider the channel characteristics and the modulation.
- The compressed signal needs a channel with a small amount of (RESIDUAL) errors, e.g. BER of 10⁻¹⁰- 10⁻¹² which means 0.1-1 erred bits per hour for a rate of 30 Mbit/s.





DVC Channel Coding Tools





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Interleaving

The interleaver does not provide error correction capabilities by itself; it rather reorganizes the symbols to have burst and bit errors more efficiently corrected when also using a channel code, e.g. a RS code.





Reed-Solomon Code

- The Reed-Solomon (RS) code is a block code:
 - Allowing the detection of corrupted symbols (up to a certain limit)
 - Allowing the correction of corrupted symbols (up to a certain limit)
- Good performance for burst errors ... of course, in combination with the interleaver.
- The RS code used in DVB is RS(204,188), this means 188 source bytes in each full block of 204 bytes; this implies a 16/188 = 8 % overhead.
- The RS(204,188) code has the capacity to correct 8 bytes in each block; if there are more than 8 bytes corrupted in a block, the channel decoder signals the lack of capability to correct the errors in the block.


Convolutional Coding and Puncturing



- Convolutional channel coding is introduced as a complement to Reed Solomon coding.
- For every *m* input bits, there are
 n output bits, typically with a m/n
 = ½ coding rate which means that
 the source rate is half the total
 rate.
- The coding rate is the ratio of the source rate to the total rate (1 when there is no channel coding)
 - To improve the coding rate (to make it higher), puncturing is used which means that certain bits at the convolutional encoder output are not transmitted, reducing the overall rate.



Puncturing Example

- Source coded data:
- 1 0 1 1 0 0 0
- Channel coded data, ¹/₂ coding rate:
- **11 10 00 01 01 11 00**
- *Puncturing with rate* $\frac{3}{4}$ (regarding the input data to the channel encoder: $\frac{3}{4} = \frac{1}{2} \times \frac{3}{2}$); when *puncturing*, 4 bits in each 6 are transmitted with a YYNYYN pattern:
- $11 \quad (1)0 \quad 0(0) \quad 01 \quad (0)1 \quad 1(1) \quad 00$
- Transmitted data:
- 11 00 01 11 00
- Reconstruction for decoding:
- 11 X0 0X 01 X1 1X 00



DVB-S2: Channel Coding



- DVB-S2 (second generation of DVB specifications for satellite) uses a more complex and more powerful channel coding solution.
- The Reed- Solomon *outer code* in DVB-S is substituted by a BCH (*Bose, Ray-Chaudhuri, Hocquenghem*) code with the capacity to correct 8 to 12 bytes in the block.
- The convolutional *inner code* in DVB-S is substituted by a LDPC (*low density parity check*) code.
- The overall BCH&LDPC block length is 64800 bits for applications without critical delay requirements, and 16200 bits otherwise.
- Depending on the needs, the following coding rates may be used: 1/4, 1/3, 2/5, 1/2, 3/5, 2/3, 3/4, 4/5, 5/6, 8/9 and 9/10.



Modulation



About Modulation ...

- Factors to consider when selecting a modulation:
 - Channel characteristics
 - Spectral efficiency, i.e. how many bits are transmitted per Hertz
 - Robustness to channel distortion
 - Tolerance to transmitter and receiver imperfections
 - Minimization of requirements for interference protection
- Main basic digital modulation techniques:
 - Amplitude modulation (ASK)
 - Frequency modulation (FSK)
 - Phase modulation (PSK)
 - Combined amplitude and phase modulation (QAM)



Amplitude Modulation: ASK

The information is transmitted in the signal amplitude !





Phase Modulation: PSK

The information is transmitted in the signal phase !





QAM Modulation

The digital signal is decomposed into 2 multilevel components corresponding to two carriers I and Q; the information is transmitted in the signal amplitude and phase, simultaneously.



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64-QAM Modulation Constellation ...





DVB Modulations

- **DVB-S QPSK** (low SNR and rather high available bandwidth); amplitude modulation is difficult due to the high attenuation.
- **DVB-S2** QPSK, 8PSK, 16APSK, 32APSK (Asymmetric Phase Shift Keying, also called Amplitude and Phase Shift Keying).
 - APSK has advantages over QAM due to the lower number of possible amplitude levels, resulting in less problems with non-linear amplifiers.
- **DVB-C** Essentially 64-QAM.
- **DVB-T and DVB-H** Orthogonal Frequency Division Multiplex (OFDM) based on QPSK and QAM modulations (very robust to multipath effects).





DVB-S2 versus DVB-S



- The spectral efficiency depends on the selected modulation constellation and coding rate; it may vary between 0.5 and 4-5 bit/symbol.
- The 16APSK and 32APSK performances are comparable to the 16-QAM and 32-QAM performances.
- **QPSK and 8PSK are typically used for television due to their constant amplitude (and higher reliability).**
- DVB-S2 increases the DVB-S transmission capacity in about 30%.



DVB Systems Architecture



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DVB-T: Terrestrial Broadasting





Digital Terrestrial TV: Requirements

- Fixed, portable and mobile reception
- Immunity to multipath effects
- Single frequency networks



- Configuration flexibility, e.g. coverage/bitrate trade-offs, configuration hierarchies
- Robustness to analogue services interferences without interfering with those services
- Easy transcoding to and from other transmission channels, e.g. satellite, cable, optical fiber
- Low cost receivers



Main DVB-T Technical Characteristics



- Many characteristics common to the DVB-S and DVB-C systems
- Inclusion of the convolutional channel coding from DVB-S
- OFDM modulation based on QPSK and QAM (very robust to multipath effects) with 2k and 8k options
- Two hierarchical layers of channel coding and modulation
- MPEG-2 Video (Main profile) and later H.264/AVC source coding
- Definition of national and regional broadcasting networks (Single Frequency Networks (SFN) and Multiple Frequency Networks (MFN))



Single Frequency Networks



While in analogue reception, the user tunes the best behaving frequency for a certain channel (from different senders), in digital SFN reception all received signals for a certain channel are in the same frequency; thus, it is important to filter the signals from the other transmitters using an antenna with an adequate radiation diagram.



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One way to reduce the number of symbols which mutually interfere is to increase their duration; this can be achieved by transmitting symbols in parallel and not only sequentially; instead of a single carrier modulated at a high rate, many carriers are used, each modulated at a lower rate.

Each sub-symbol s_k may be modulated in amplitude or phase.



Orthogonal Sub-Carriers



The sub-carriers are said orthogonal if they are uniformly spaced in frequency in a way that all other sub-carriers are zero at the central position of any specific sub-carrier which means

 $w_k = 2 \pi k f_0$ with k=0, 1, ..., n-1

where f_0 is the base frequency.

The orthogonality of the subcarriers eliminates the intercarrier interference and provides a high spectral efficiency by allowing spectral overlapping.



Orthogonal Frequency Division Multiplex

For orthogonal sub-carriers, multi-carrier modulation corresponds to applying the *Inverse Discrete Fourier Transform* (IDFT) to the subcarriers in parallel, creating the so-called *Orthogonal Frequency Division Multiplex* (OFDM) modulation.





adjacent interfering symbols !

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OFDM Symbol: Union is Strength ...





- The adoption of a guard interval allows to create a time zone free of interferences between different modulated symbols received through multiple paths.
- The length of the guard interval must be longer than the biggest delay corresponding to the interfering signals (and this depends on the diffusion cells).



Guard Interval: an Example



The attenuation and delay of the signal received from another emission depends on the distance between transmitters.



The COFDM (Coded OFDM or OFDM) Variants

DVB-T defines two variants for data transmission (in 8 MHz channels):

- 2k Variant (1512 signal sub-carriers and 193 synchronization sub-carriers)

 Solution adequate for small areas coverage; less robust to interferences, less complex; 224 μs/symbol; 4464 Hz between sub-carriers.
- 8k Variant (6048 signal sub-carriers and 769 synchronization sub-carries Solution adequate for large areas coverage; more robust to interferences, more complex; 896 μs/symbol; 1116 Hz between sub-carriers.

The modulation of each sub-carrier may be made with QPSK (2 bit/symbol), 16-QAM (4 bit/symbol) or 64-QAM (6 bit/symbol), with guard intervals of $T_S/4$, $T_S/8$ or $T_S/32$, and 7.6 MHz between the extreme sub-carriers (for a 8 MHz channel).



Bitrate (Mbit/s) versus Modulation for each 8 MHz Channel ...

Modulation	Coding rate ·	Relative length of the guard interval				
WIOdulation		1/4	1/8	1/16	1/32	
QPSK	1/2	4.98	5.53	5.85	6.03	
	2/3	6.64	7.37	7.81	8.04	
	3/4	7.46	8.29	8.78	9.05	
	5/6	8.29	9.22	9.76	10.05	
	7/8	8.71	9.68	10.25	10.56	
16-QAM	1/2	9.95	11.06	11.71	12.06	
	2/3	13.27	14.75	15.61	16.09	
	3/4	14.93	16.59	17.56	18.10	
	5/6	16.59	18.43	19.52	20.11	
	7/8	17.42	19.35	20.49	21.11	
64-QAM	1/2	14.93	16.59	17.56	18.10	
	2/3	19.91	22.12	23.42	24.13	
	3/4	22.39	24.88	26.35	27.14	
	5/6	24.88	27.65	29.27	30.16	
	7/8	26.13	29.03	30.74	31.67	



Hierarchical Modulation

64-QAM (4+2 bit/symbol)

100000	100010	101010	101000	001000	001010	000010	000000
100001	100011	101011	101001	001001	001011	000011	000001
							- -
100101	100111	101111	101101	001101	001111	000111	000101
100100	100110	101110	101100	001100	001110	000110	000100
				X	•		
110100	110110	111110	111100	011100	011110	010110	010100
110100	110110	111110	111100	011100	011110	010110	010100
110100 110101	110110 110111	111110 111111	111100 111101	011100 011101	011110 011111	010110 010111	010100
110100 110101	110110 110111	111110 111111	111100 1111101	011100 011101	011110 011111 011111	010110 010111	010100 010101
110100 110101 110001	110110 110111 110011	111110 111111 111011	111100 111101 111001	011100 011101 011001	011110 011111 011101	010110 010111 010011	010100 010101 010001
110100 110101 110001 110000	110110 110111 110011 110010	111110 111111 111011 111010	111100 111101 111001 111000	011100 011101 011001	011110 011111 011001 011010	010110 010111 010011 010010	010100 010101 010001
110100 110101 110001 110000	110110 110111 110011 110010	111110 111111 111011 111010	111100 111101 111001 111000	011100 011101 011001 011000	011110 011111 011001 011010	010110 010111 010011 010010	010100 010101 010001 010000

64-QAM hierarchical modulation allows the simultaneous diffusion of a priority stream (2 MSB bits) in QPSK and another stream (remaining 4 bits), e.g. for different programs or difference resolutions.

When the transmission conditions degrade, 16 points in the 64-QAM constellation may be taken as a single point in a QPSK constellation, allowing to receive, in good conditions, at least the 2 MSB bits.



DVB-T: Excellent Mobile Reception





Reception with spatial, temporal and frequency diversity ...







- TDT in Portugal will use 6 multiplexers (A, B, C, D, E e F) and Single Frequency Networks (SFN).
- Multiplexer A will transmit the free channels already with license (RTP 1, RTP 2, SIC e TVI); the fifth channel was intended for this multiplexer but plans for it were withdrawn.
- Multiplexers B to F should be for pay TV.
- Multiplexers B and C are national and Multiplexers D, E, F have partial coverage with a save zone of 80 km from the border with Spain (meaning that part of the population will not see these channels).
- By the 26th April 2012, the switch-off to digital should be finished ...







DVB Terminals



What Does a Set-top Box ?



DVB Integrated Receiver-Decoders (IRDs)

The DVB IRDs are classified according to 5 dimensions:

- **"25 Hz" or "30 Hz"** depending if they use 25 Hz or 30000/1001 Hz (approximately 29,97 Hz) picture rates; some IRDs may be *dual-standard* which means they may accept both 25 Hz and 30 Hz video content.
- **"SDTV" or "HDTV"** depending if they are limited or nor to decode conventional resolution images (ITU-R 601); a SDTV IRD has capabilities which are a sub-set of an HDTV IRD capabilities.
- **"With digital interface" or "Baseline"** depending if they can be used for storage as with a VCR (*Video Cassete Recorder*) or not; a *Baseline* IRD has capabilities which are a sub-set of the digital interface IRD capabilities.
- **"MPEG-2 Video" or "H.264/AVC"** depending if they use one or the other video coding format.
- Audio Coding Format, several, e.g. MPEG-1/2 Audio (Layers 1 e 2), Dolby AC-3, and recently MPEG-4 Audio HE AAC.



Video in DVB

- MPEG-2 Main Profile @ Main Level is used to code SDTV with MPEG-2 Video
- MPEG-2 Main Profile @ High Level is used to code HDTV with MPEG-2 Video
- H.264/AVC Main Profile @ Level 3 is used to code SDTV with H.264/AVC
- H.264/AVC High Profile @ Level 4 is used to code HDTV with H.264/AVC
- Both the 25 Hz MPEG-2 SDTV IRDs and 25 Hz H.264/AVC SDTV IRDs use 25 Hz
- The 25 Hz MPEG-2 HDTV IRDs and the 25 Hz H.264/AVC HDTV IRDs use both 25 and 50 Hz



Audio in DVB

- The DVB audio formats are MPEG-1 Audio Layer I, MPEG-1 Audio Layer II or MPEG-2 Audio Layer II backward compatible.
- The usage of Layer II is recommended when MPEG-1 Audio is used.
- Sampling rates are 32 kHz, 44,1 kHz and 48 kHz.
- IRDs may, optionally, decode multi-channel MPEG-2 Audio Layer II backwards compatible audio (Part 2).
- The usage of MPEG-4 Audio High Efficiency AAC (HE AAC) is optional, and thus the IRDs may, optionally, decode or not these streams.


Final Remarks



- The DVB solutions for digital TV are recognized as the best, notably for mobile and portable reception.
- There are many hundreds of millions of MPEG-2 set-top boxes sold, especially in USA and Europe.
- Both Europe (DVB) and US (ATSC) decided to use the MPEG-2 Systems and MPEG-2 Video standards (unfortunately with small differences). While DVB also uses MPEG-2 Audio, ATSC uses Dolby AC-3, another audio coding format.
- Digital Video Disc (DVD) has adopted MPEG-2 standards.

Deployed digital TV is currently mostly MPEG-2 based ... however, another more efficient video coding solution is quickly taking over: H.264/AVC (see next episode)!

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