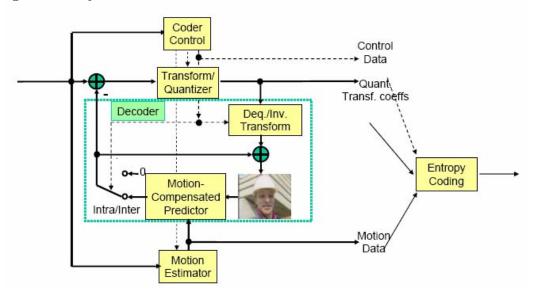
1. Video Standards and Systems¹

Framework: Developments in broadcasting, personal computers, communication technologies and services, such as better data compression algorithms, fiber optic networks, distributed & network computing, DSP devices, and digital recording offer a variety of IT products in the near future. Driving the R&D in this e field are the consumer and commercial applications:

Digital television broadcasting	2 6 Mbps (1020 Mbps for HD)	MPEG-2
DVD video	6 8 Mbps	MPEG-2
Internet video streaming	20 200 kbps	Proprietary, similar to H.263, MPEG-4, or H.264/JVT
Videoconferencing, videotelephony	20 320 kbps	H.261, H.263
Video over 3G wireless	20 100 kbps	H.263, MPEG-4

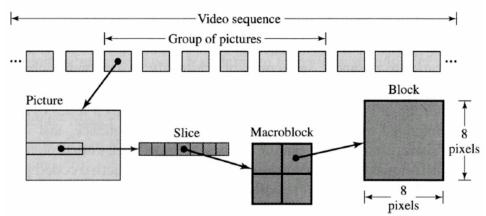
- Digital TV, including Advanced TV and HDTV @20 Mbps over 6.0 MHz taboo channels.
- Multimedia and desktop video @1.5 Mbps on CD-ROM or DVT (DAT) or HD storage and Video over IP (VOIP). (Do not mix with the term VoIP, which stands for "Voice over IP."
- Videoconferencing, H.261-H.264, MPEG4-MPEG21, and the new JVT Standard based on MPEG4.12, @ 384 Kbps and higher using *px64* ISDN channels and IP.
- Videophone & mobile image communication @ 10-33.4 Kbps on cellular & voice channels.
- Common tread to all of these systems is the principle of "Motion-Compensated Hybrid Coding" and the systems associated with it.



¹ The material in this chapter has been provided by Professor A. Murat Tekalp at Koc University, Istanbul, Turkey and VCDemo has been provided by Professor Inald Lagendijk of Delft University of Technology, The Netherlands.

Highlights of these systems:

1. Hierarchical Syntax



2. Syntax Hierarchy

SYNTAX HIERARCHY AS USED IN DIFFERENT VIDEO CODING STANDARDS*

Syntax layer	Functionality	Standard	
Sequence (SC) VOL (SC)	Definition of entire video sequence Definition of entire video object	H.261/3, MPEG-1/2 MPEG-4	
GOP (SC) GVOP (SC)	Enables random access in video stream Enables random access in video stream	MPEG-1/2 MPEG-4	
Picture (SC) VOP (SC)	Primary coding unit Primary coding unit	H.261/3, MPEG-1/2 MPEG-4	
GOB (SC)	Resynchronization, refresh, and error recovery in a picture	H.261/3	
Slice (SC)	Resynchronization, refresh, and error recovery in a picture	MPEG-1/2	
Video Packet (SC)	Resynchronization and error recovery in a picture	MPEG-4	
MB	Motion compensation and shape coding unit	H.261/3, MPEG-1/2/4	
Block	Transform and compensation unit	H.261/3, MPEG-1/2/4	

*Each layer starts with a header. An SC in a syntax layer indicates that the header of that layer starts with a start code.

All-Digital TV

TV being the most commonly used image communication system used in the world is most appealing to developers and manufacturers of IT to be associated with. Digital TV has been in the agenda for long time, but it been recently taken into an active playground due to breakthrough in compression algorithms and the DSP technology with the efforts in the area of *High Definition TV* (HDTV).

Major Development Efforts for HDTV Research, Delivery and Products			
Japan	NHK		
Europe	EUREKA-95		
USA	Grand Alliance	ATT&, General Instruments, MIT, Philips N.A., David Sarnoff Research Center, Thompson Consumer Electronics, Zenith	

NHK system being a hybrid system has had very limited success and eventual abandonment of the whole thing! All-digital TV is the big thing today. Although, the present TV standards are analog, digital TV signals find routine use in digital TV applications and it might even be a must for all-digital TV to be backward compatible with the good old TV sets! Here are the existing TV standards:

- NTSC
 - 1. 2:1 interlace with 4:3 aspect ratio.
 - 2. 525 lines/frame, 29.97 frames/s, (262.5 lines/field, 59.94 files/s.)
 - 3. Perceptually 340 lines/frame, 420 resolvable pels/line (Kell factor).
 - 4. Analog transmission over taboo 6.0 MHz channels.
 - 5. 68 channels in US: (54-88 MHz; Ch.2-6) + (174-216; Ch.7-13 MHz) + (470-806 MHz; Ch.14-69).
- PAL
 - 6. 2:1 interlace with 4:3 aspect ratio.
 - 7. 625 lines/frame, 50 fields/s.
 - 8. Analog transmission over 8.0 MHz channel.
- SECAM (Similar features as in PAL but different setup.)

US Grand Alliance:

After the initial efforts on ATV in Japan and IDTV and EQTV in Europe, FCC in US initiated a HDTV broadcast standard in 1987. After 6 years of competitive efforts of various groups to come up with a winner system, 4 candidates with all-digital architectures were taken into the test phase. All had flaws and nobody was declared winner.

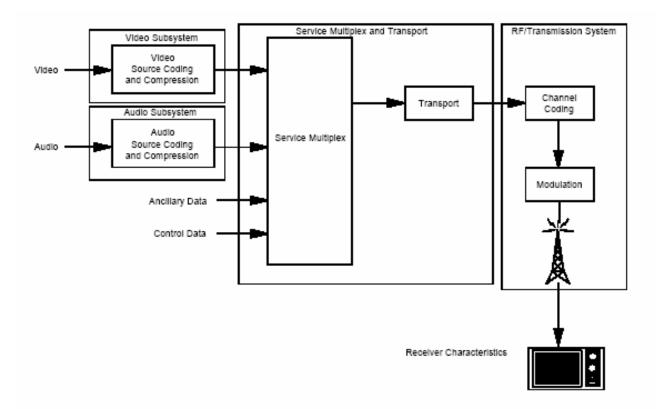
In May 1993, 4 groups consisting of (a) AT&T and Zenith Electronics Corporation; (b) General Instrument Corporation & MIT; (3) Philips Consumer Electronics & Thomson Consumer Electronics, and (4) the David Sarnoff Research Center came together to form the "*Digital HDTV Grand Alliance*." Their combined proposal became the HDTV Standard. Key points:

- 1. Video compression algorithm will be MPEG-2, main profile, high-level.
- 2. MPEG-2 transport mechanism will be used.
- 3. Dolby AC-3 audio system will be used, and
- 4. Three modulation techniques: 4-level VSB, 6-level VSB, and 32 QAM will be tested to complete specification.

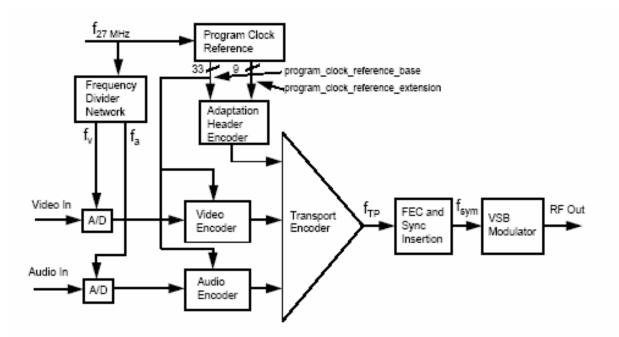
System Block Diagram:

A basic block diagram representation of the ATSC standard is shown below, which is based on a model adopted by the International Telecommunication Union, Radio-communication Sector (ITU-R), Task Group 11/3 (Digital Terrestrial Television Broadcasting). According to this model, the digital television system can be seen to consist of three subsystems.3

- Source coding and compression
- Service multiplex and
- RF/transmission



ITU-R digital terrestrial television broadcasting model.

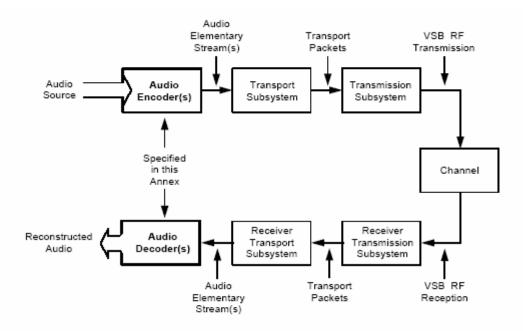


High level view of encoding equipment.

Possible Video Input Signals

While not required by this standard, there are certain television production standards, shown in the table below that define video formats that relate to compression formats specified by the HDTV standard.

Video Standard	Active Lines	Active Samples/ Line	
SMPTE 274M	1080	1920	
SMPTE 296M	720	1280	
ITU-R BT.601-4	483	720	



Audio subsystem in the digital television system.

Broadcast Applications:





Further details on ATSC Standard can be found in: ATSC Standard: Digital Television Standard (A/53), Revision C at he URL: <u>http://www.atsc.org</u> **Videoconferencing:** It refers to interactive distance conferencing with multi-media features over limited bandwidth channels (56 kb through DSL) with clear sound and sharp full-motion video. Two major product levels: Personal & Group

- Personal
 - Desktop, primarily for one-to-one communication
 - PC-based software, with simple video camera
 - PC handles audio and video compression
 - · Requires Pentium-III or better PC
 - Either video capture card or direct via-USB
 - NetMeeting or similar software

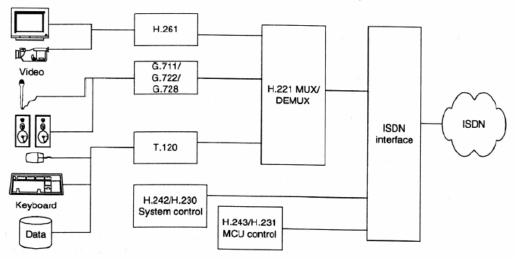


Videoconferencing Standards:

	H.320	H.321	H.322	H.323	H.324
Approved Date	1990	1995	1995	1996	1996
Network	Narrowband switched digital ISDN	Broadband ISDN, ATM	Guaranteed bandwidth packet- switched networks	Packet- switched networks (LAN/ WAN, ATM)	PSTN or POTS - the analog phone system
Video	H.261	H.261	H.261	H.261	H.261
	H.263	H.263	H.263	H.263	H.263
Audio	G.711	G.711	G.711	G.711	G.723
	G.722	G.722	G.722	G.722	
	G.728	G.728	G.728	G.723	
				G.728	
				G.729	
Multiplexing	H.221	H.221	H.221	H.225.0	H.223
Control	H.230	H.242	H.242	H.245	H.245
	H.242		H.230		
Multipoint	H.231	H.231	H.231		
	H.243	H.243	H.243		
Data	T.120	T.120	T.120	T.120	T.120
Comm.	1.400	AAL	1.400 &	TCP/IP	V.34 modem
Interface		1.363	TCP/IP	AAL	
		AJM I.361 PHY I.400			



Videoconferencing Using H.320 Protocol over ISDN:



H.320 Protocol Highlights:

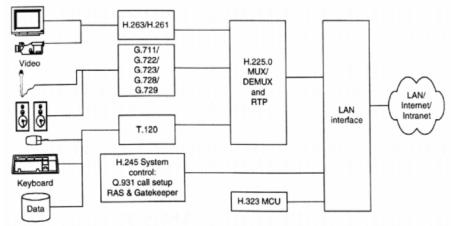
- 1. Supports QCIF and CIF resolutions
- 2. It uses H.261 Video Codec as a default codec.
- 3. Frame rates of 7.5, 10, 15 or 30 frames/second

Class 1: QCIF at 7.5 frames/second, no encoding, no motion compensation Class 2: QCIF/CIF at 17 frames/second max, limited motion compensation Class 3: CIF at 30 frames/second max, full encoding; motion compensation

- 4. Three possible Audio Codecs
 - G.711 for Class 1

G.722 and G.728 for Classes 2 &3.

H.323 Protocol for LAN/ATM



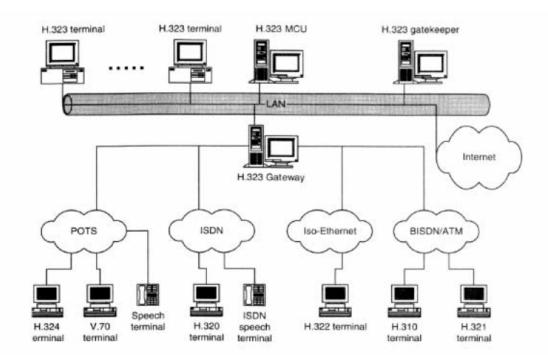
H.323 Protocol Highlights

- Extends H.320 to LANs (Intranet and Internet) with no guaranteed QoS support Also specifies video conferencing over ATM including ATM QoS. Includes standards for system control of clients, multipoint servers, gateways and gatekeepers
- Defines the Real-time Transport Protocol (RTP)
- H.261 and H.263 Video Codecs and G.729 Audio Codec

Audio Codecs for Videoconferencing Standards and Systems:

- 1. G.711: Pulse Coded Modulation PCM encoding of 3.1KHz audio to 64kbps digital using either A-law (Europe) or μlaw mu-law encoding (Japan & North America).
- 2. G.721: Adaptive DPCM (ADPCM) of 3.1KHz audio to 32Kbps digital.
- 3. G.722: Defines how 7.5KHz audio is encoded to a 64Kbps using ADPCM.
- 4. G.723: Dual rate speech coder for transmitting 5.3 and 6.4Kbps.
- 5. G.728: Compression using Low Delay Code Excited Linear Prediction (LDCELP) to achieve 3.4KHz audio signal to a digitized signal of 16Kbps.

Videoconferencing Network Layout and the System Block Diagram:



The ISO MPEG-4 and Beyond

The MPEG Ad-Hoc Group of 1993 was commissioned to develop a fundamentally generic video coding standard at rates below 64 Kbps. Later, this mission has been modified to "provide an audio-visual coding standard allowing for

- 1. Interactivity,
- 2. Compression based on Discrete Wavelet Transform instead of DCT
- 3. Universal accessibility with high degree of flexibility, and
- 4. Scalability and Extensibility.
- 5. Very similar to XML.

MPEG: Moving Picture Experts Group Coding of Moving Video and Audio

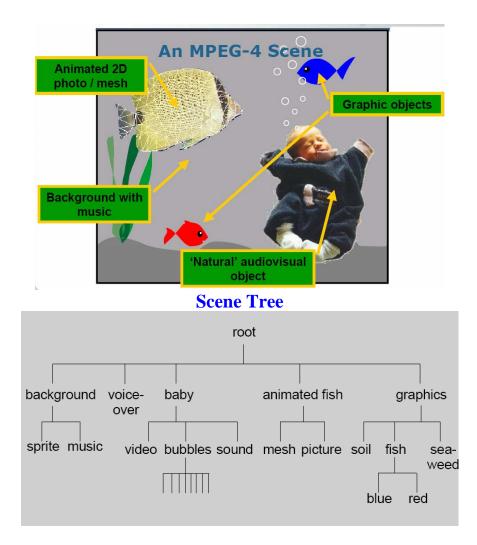
- MPEG-1:CD-i, (VoD trials), ... 1992
- MPEG-2:... + TV, HDTV 1994
- MPEG-3: HDTV → merged into MPEG-2
- MPEG-4: Coding of Audiovisual Objects –1998, 1999 Extensions ongoing
- MPEG-7: MM Description Interface Fall 2001 `Describing' audiovisual material
- MPEG-21: Digital Multimedia Framework 1st parts ready 'The Big Picture and The Glue'
- As of June 2003 MPEG-4 part 10 --(JVT Codec, Audio extensions and some Systems refinements—is the latest standard (AVC H.264).

MPEG-4, 7, 21 levels include eight functionalities that are not supported by the existing platforms:

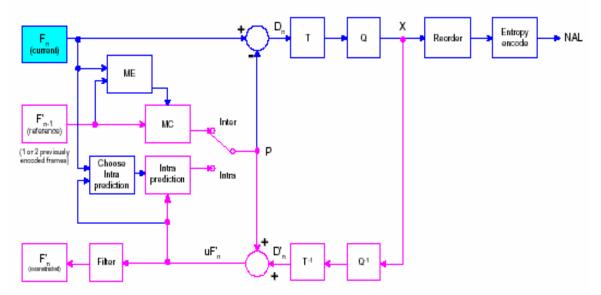
- 1. Content-based manipulation & bit stream editing.
- 2. Content-based multimedia data access tools.
- 3. Content-based scalability.
- 4. Coding of multiple concurrent data streams.
- 5. Hybrid natural and synthetic data coding.
- 6. Improved coding efficiency.
- 7. Improved temporal access at very low bit rates.
- 8. Robustness in error-prone environments.

Possible Application Areas:

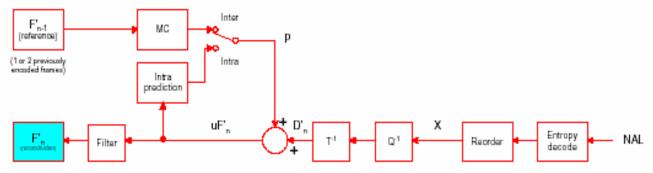
- Architecture, real estate, and interior design (searching for ideas)
- Broadcast media selection (radio channel, TV channel)
- Cultural services (history, museums, art galleries, etc.) and digital libraries
- E-Commerce (personalized advertising, on Commerce (personalized advertising, on-line catalogues, various directories.
- Education (repositories of multimedia courses, search for support material)
- Home entertainment
- Investigation (human/human characteristics recognition, forensics)
- Journalism
- Directory services (yellow pages, Tourism, Geographical information systems)
- Multimedia editing
- Remote sensing
- Social (even dating services)
- Surveillance



MPEG-4 Encoder Block Diagram:



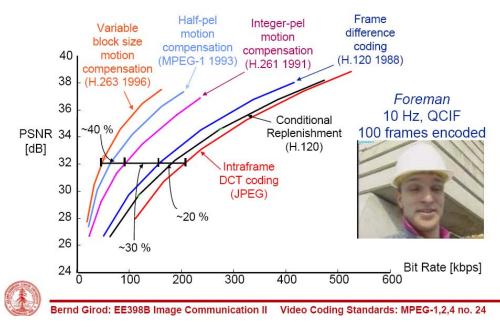
MPEG-4 Decoder Block Diagram:



Emerging H.26L Video Compression Standards²

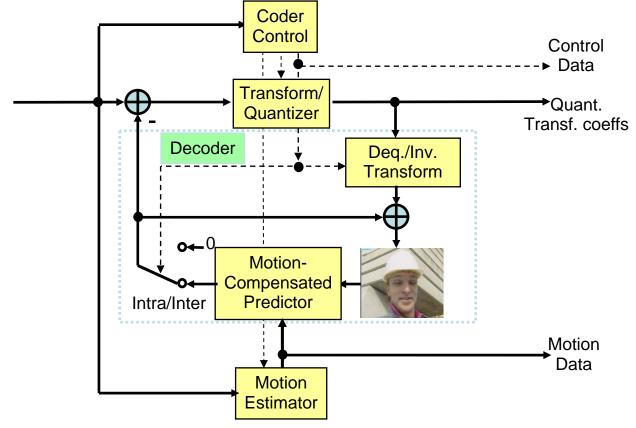
- H.26L family codecs (ITU-T Q.6/SG16 (VCEG) standardization activity for video coding especially aimed at 3G mobile networks and broadcast
- Possible formation of a joint video team with MPEG
- Goal for H.26L: 50% bit-rate reduction for same fidelity wrt earlier standards
- 1999: 3 proposals for definition of a first test model
 - HHI: Warping/OBMC motion model, wavelet- and context-based adaptive coding (CABAC)
 - Nokia: Affine motion model, multiple block transforms
 - Telenor: Block-matching with variable block-sizes, 4x4-DCT

Status: Finalized on June 2003 as the 8th test model (TML-8) based on Telenor proposal.
Video Compression Performance Progress: (From Berndt Girod's Video Coding Standards Course lecture notes at Stanford.)



² The material on H.26L are provided by Thomas Wiegand of Heinrich Hertz Institute Berlin, Germany and Gary Sullivan of Microsoft. This topic also appears as the AVC Standard prepared by the Joint Video Team (JVT) formed by ITU-T Video Coding Experts Group (VCEG) and ISO/IEC Moving Picture Experts Group (MPEG). formed a Joint Video Team (JVT)

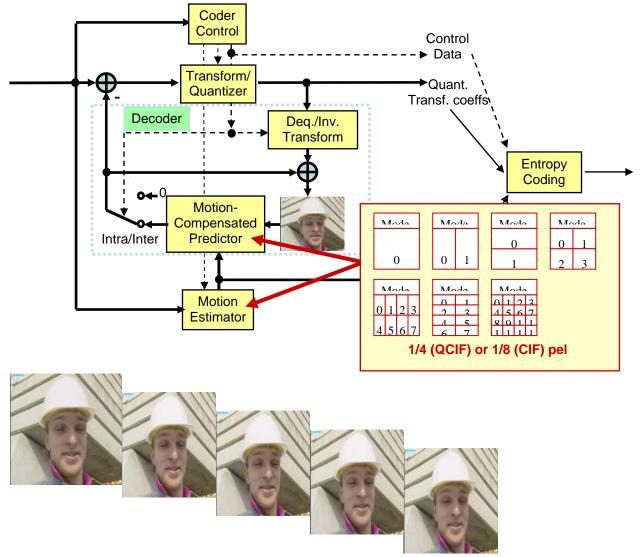
H.26L Structure:



H.26L TML-8 Design (Part 1):

- Still using a hybrid of DPCM and transform coding as in prior standards.
- Common elements with other standards include:
 - 16x16 macroblocks
 - Conventional sampling of chrominance and association of luminance and chrominance data
 - Block motion displacement
 - Motion vectors over picture boundaries
 - Variable block-size motion
 - Block transforms (not wavelets or fractals)
 - Scalar quantization

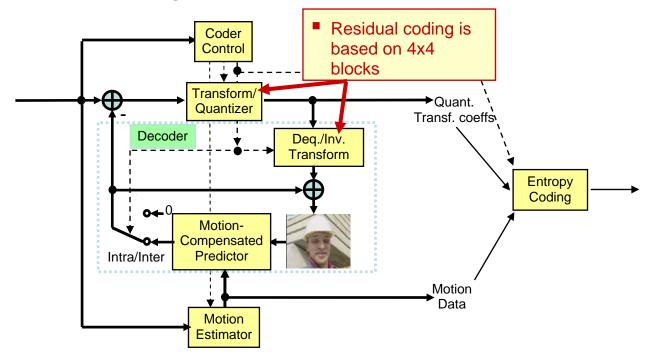




Multiple Reference Frames are used for Motion Compensation The H.26L TML-8 Design (Part 2): Motion Compensation

- Various block sizes and shapes for motion compensation (7 segmentations of the macroblock: 16x16, 16x8, 8x16, 8x8, 8x4, 4x8, 4x4)
- Multiple reference pictures (per H.263++ Annex U)
- Temporally-reversed motion
- B picture prediction weighting
- New "SP" transition pictures for sequence switching
- 1/4 sample (sort of per MPEG-4) and 1/8 sample accuracy motion:
- 1. 6x6 tap filtering to 1/2 sample accuracy, bilinear filtering to 1/4 sample accuracy, special position with heavier filtering
- 2. 8x8 tap filtering applied repeatedly for 1/8 pel motion

H.26L: Residual Coding:

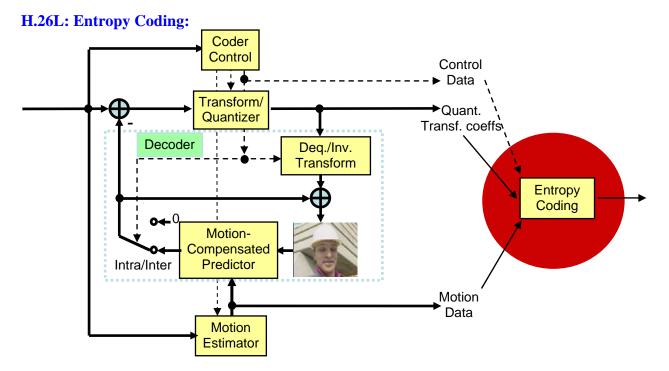


The H.26L TML-8 Design (Part 3):

- Transform
 - Integer transform approximating a DCT
 - Based primarily on 4x4 transform size (all prior standards used 8x8)
 - Expanded to 8x8 for chroma by 2x2 DC transform
- Intra Coding Structure
 - Directional spatial prediction (6 types luma, 1 chroma)
 - Expanded to 16x16 for luma intra by 4x4 DC Xfm

The H.26L TML-8 Design (Part 4):

- Quantization
 - Two inverse scan patterns
 - Logarithmic step size control
 - Smaller step size for chroma (per H.263 Annex T)
- Deblocking Filter (in loop)
- Distinct Network Adaptation Layer (NAL) design for network transport
 - Slice-structure coding
 - Data partitioning



1. Entropy Coding in H.26L Based on Universal Variable Length Code (UVLC): Relatively simple design with some disadvantages:

- Probability distribution is static
- Correlations between symbols are ignored, i.e. no conditional probabilities are used
- Code words must have integer number of bits (Low coding efficiency for highly peaked probability densities, pdfs)

2. Context-based adaptive binary arithmetic codes (CABAC):

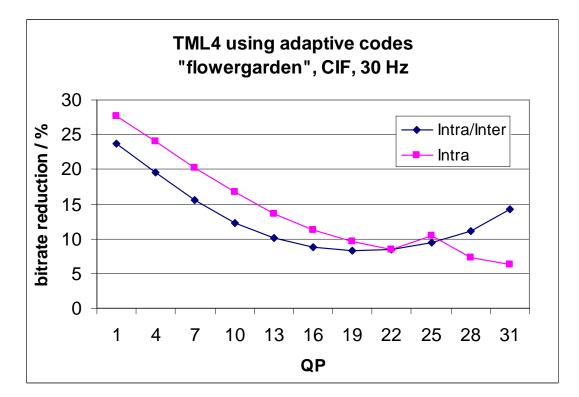
- Usage of adaptive probability models
- Exploiting symbol correlations by using contexts
- Non-integer number of bits per symbol by using arithmetic codes
- Restriction to binary arithmetic coding
 - o Simple and fast adaptation mechanism
 - o But: Binarization is needed for non-binary symbols
 - o Binarization enables partitioning of state space

Binary Arithmetic Coding:

- Standard implementations use integer arithmetic
- Fast, multiplication-free variants of binary arithmetic coder exists: e.g. MQ-coder used in JBIG-2, JPEG-LS, JPEG-2000
- *Estimation*: Increase in computational complexity lower than 10% (MQ) and 20% (Standard-AC) of the total decoder execution time at medium bitrate

Results: Bit-Rate Reduction:

Test Sequence: Flower Garden with CIF Format and 30 frames/second.

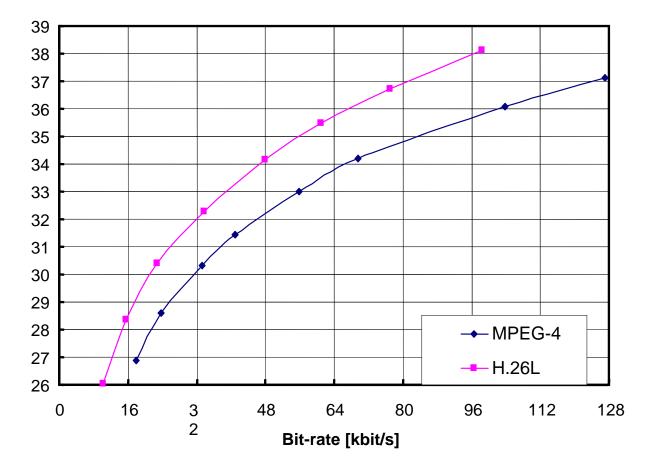


Comparison of H.26L to MPEG-4:

- MPEG-4: Advanced Simple Profile (ASP)
 - Motion Compensation: 1/4 pel
 - o Global Motion Compensation
- H.26L:
 - Motion Compensation: 1/4 pel (QCIF), 1/8 pel (CIF)
 - Using CABAC entropy coding
 - o 5 reference frames in 7 of 8 cases (News: 17 / 25)
- Both:
 - Sequence structure *IBBPBBP*...
 - QPB=QPP+2 (step size: +25%)
 - Search range: 32x32 around 16x16 predictor

RD Curves:

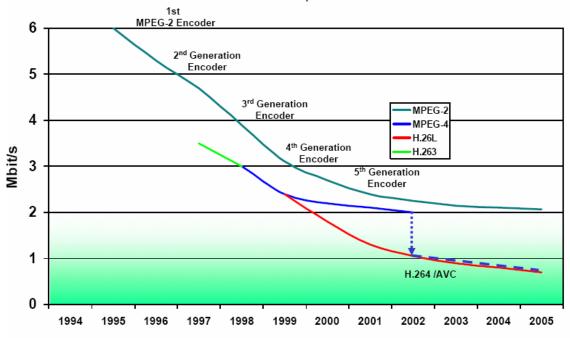
Test Sequence: Foreman with QCIF format at 10 frames/second Peak Signal-to-Noise Ratio (**PSNR**) in dB versus Bit-Rate in kbits/second



Average A,B,C = 2.0 dB			Average PSNR of Luminance		
Average A,B,C,D,E,F = 2.1 dB		H.26L TML-8	MPEG-4 ASP	Gain	
		Foreman	31.9 dB	30.1 dB	+1.8 dB
Case A	32 kbit/s	News	36.0 dB	33.3 dB	+2.8 dB
(Avg. 1.9)	QCIF, 10 fps	Container Ship	38.5 dB	36.8 dB	+1.8 dB
(Avg. 1.5)	•	Tempete	29.3 dB	27.8 dB	+1.5 dB
		Foreman	34.7 dB	32.8 dB	+1.9 dB
Case B	64 kbit/s	News	39.4 dB	35.8 dB	+3.7 dB
(Avg. 2.3)	QCIF, 15 fps	Container Ship	40.5 dB	38.6 dB	+1.9 dB
(Avg. 2.0)	•	Tempete	31.3 dB	29.4 dB	+1.9 dB
		Foreman	33.3 dB	31.3 dB	+2.0 dB
Case C	128 kbit/s	News	38.7 dB	35.9 dB	+2.9 dB
(Avg. 1.9)	CIF, 15 fps	Container Ship	36.7 dB	35.4 dB	+1.3 dB
(•	Tempete	28.8 dB	27.6 dB	+1.3 dB

PSNR Results: H.26L vs. MPEG-4 Average PSNR for Luminance at Three Different Bit Rates Test Sequences: Foreman, News Anchor, Container Ship and Tempete

MPEG "Adopts" H.26L



Conclusions

- H.26L Standard design is based on hybrid video coding
- Similar in spirit to other standards but with important differences
- Entropy coding can be conducted using
 - One VLC
 - Context-based adaptive arithmetic coding
- Context-based adaptive arithmetic coding provides improvements of 5-15 %
- H.26L delivers significant performance gain over existing standards including H.261-MPEG-4
- Bit-rate savings up to 50 % against MPEG-4.

1. More Information can be found in the H.26L standardization Documents and the associated websites:

http://www.itu.int/ITU-T/studygroups/com16/jvt/JVTToR.pdf

2. Draft Document can be downloaded from the ftp site: <u>ftp://ftp.imtc-files.org/jvt-experts/2002_12_Awaji/JVT-F100d2ncm.zip</u>

- 3. JVT FTP Site: <u>ftp://ftp.imtc-files.org/jvt-experts/</u>
- 4. Wiegand's website: <u>http://bs.hhi.de/</u>
- 5. VQEG website: <u>http://www-ext.crc.ca/vqeg/frames.html</u>