

III.3 Scalable Video Coding

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Outline

- Introduction
- Pyramidal coding
- Scalability in the standard codecs
- Layered coding with wavelets
- MPEG-4 FGS
- MPEG-21 SVC
- Summary

Introduction

- Layered video coding (scalable coding): a concept that enables video layers to interwork
- The codec generates two bit-streams
 - Base layer: most vital video information
 - Enhancement layer: residual information to enhance the quality of the base layer image
- three general layered coding techniques:
 - Pyramidal coding
 - Scalability in the standard video codecs (MPEG-2, H.263+, MPEG-4)
 - Wavelet-based coding (MPEG-4 I-frame, JPEG 2000)

Pyramidal Coding

- Pyramid: a data structure that provides successively condensed information of an image
- Coding schemes based on the pyramid structure are called *pyramidal coding*
 - the *apex* picture: the top of the pyramid, which gives the minimum acceptable picture resolution
 - other levels reconstruct images of higher quality by including additional information
 - lower levels toward to the bottom of the pyramid are of less significant importance

Pyramidal Coding (Cont.)

- can be used to reconstruct images of varying quality, depending on the network resources
- Two methods of pyramidal image coding:
 - Laplacian pyramid (Burt and Adelson 1983)
 - DCT pyramid

Laplacian Pyramid

- LPC (Laplacian pyramidal coding) includes two distinct types of pyramid:

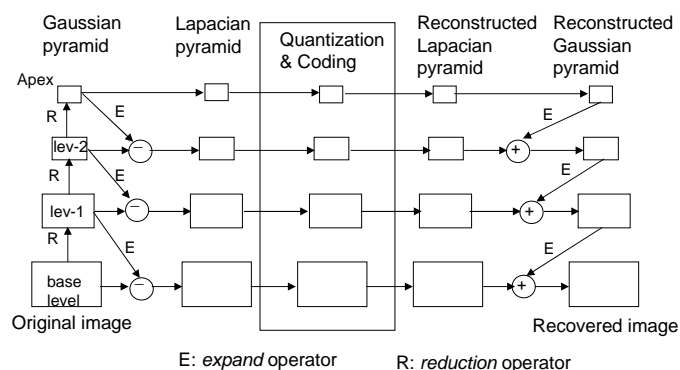
- The Gaussian pyramid
- The Laplacian pyramid

- Coding efficiency is limited

$$G_{k+1} = \text{reduce}(G_k)$$

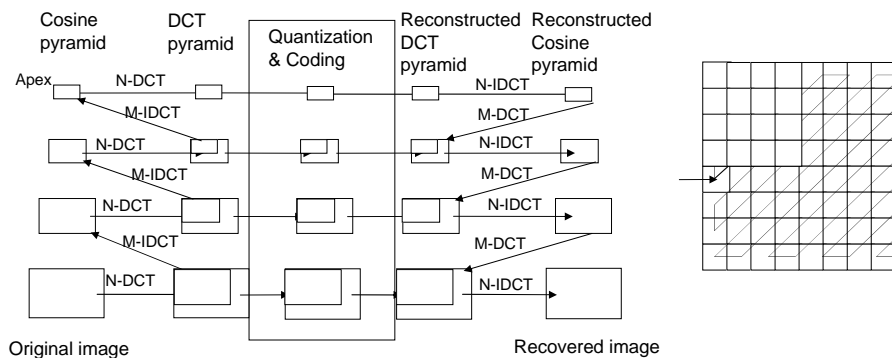
$$G_k = L_k + \text{expand}(G_{k+1})$$

$$L_k = G_k - \text{expand}\{\text{reduce}(G_k)\}$$



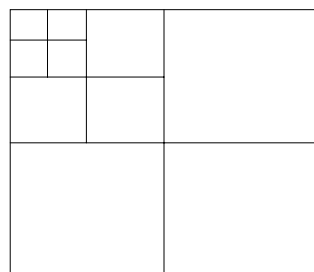
DCT Pyramid

- N->M decimation:
 - retains only the $M \times M$ low-freq. components out of $N \times N$ DCT coef.
 - then performs $M \times M$ IDCT to obtain the decimated image
 - forms the cosine pyramid
- The remaining high-freq. coef. are quantized and coded (DCT pyramid)

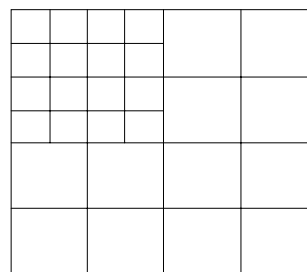


Wavelet Decomposition vs. DCT Pyramid

- The Laplacian coding increase the coding are
 - $1 + 1/4 + 1/16 + 1/64 + \dots = 4/3$
- The DCT pyramid and the wavelet transform do not increase the coding area



(a) wavelet



(b) DCT pyramid

Subband Decomposition

- The DCT pyramid implicitly embodies subband decomposition
- The effective bandwidth of these bands decreases from level to level
- Quantization and coding of each band of the pyramid can be adapted to reflect the sensitivity of the HVS
 - Coarser quantization for the higher frequency bands
 - Finer quantization for the lower frequency bands

Performance of DCT Pyramidal Coding with Data Loss from Levels



(a) 1



(b) 1+2



(c) 1+2+3



(d) All levels except apex

PSNR Performance of DCT Pyramidal Coding: Parrot

Layers received	Bit /picture [kbits]	Bit/pixel	Discard rate [%]	Quality [dB]
Apex = 5	8.1	0.02	92	21.75
4+5	28.4	0.07	72	26.48
3+4+5	56.8	0.14	44	31.06
2+3+4+5	77	0.19	24	34.78
All	101.4	0.25	0	39.2

Scalability in Standard Codecs

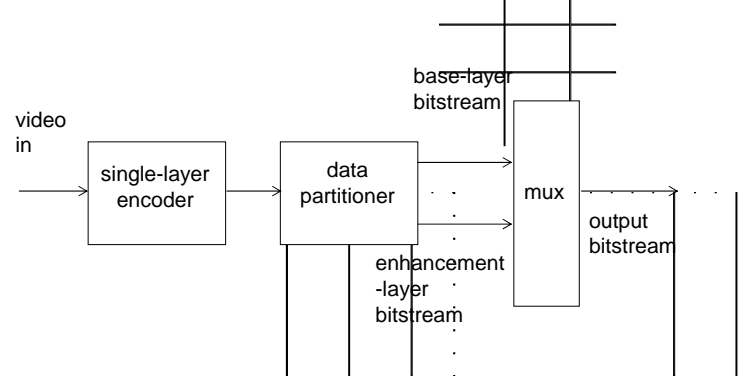
- The basic scalability tools offered are:
 - Data partitioning
 - SNR scalability
 - Spatial scalability
 - Temporal scalability
 - Hybrid scalability

Data Partitioning

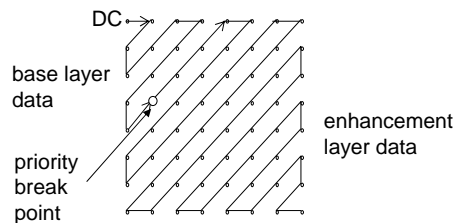
- not a true scalable coding
- a means of dividing the bitstream of a single-layer nonscalable DCT-based codec into two parts(layers):
 - The first layer
 - comprises the critical parts of the bitstream (e.g., headers, motion vectors, lower order DCT coefficients)
 - The second layer
 - is made of less critical data (e.g., higher DCT coefficients)
- Data from the second layer cannot be used unless the decoded base layer data are available

Data Partitioning (Cont.)

- At the encoder, during the quantization and zigzag scanning of each 8 x 8 DCT coefficient, the scanning is broken at the *priority break point (PBP)*



Data Partitioning (Cont.)

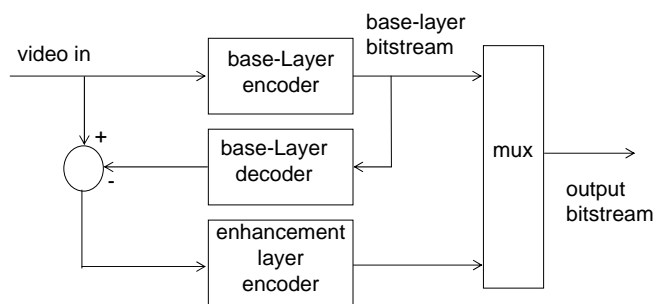


- May cause *picture drift*
- I-pictures can clean up the drift, but cause higher bit rates
- One of the limitations in data partitioning is the need for a high allocated bit rate to the base layer to avoid “blockiness”
- The simplest kind of scalability, has no extra complexity over the nonscalable encoder

SNR Scalability

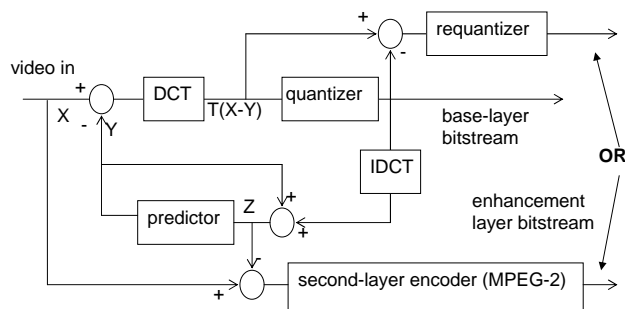
- Generates two video layers
 - Base layer and enhancement layer
 - With same spatio-temporal resolution, but different video qualities
 - The enhancement layer enhances the *signal-to-noise ratio* (SNR) of the base layer, so called SNR scalability

Two-Layer SNR Scalable Coder



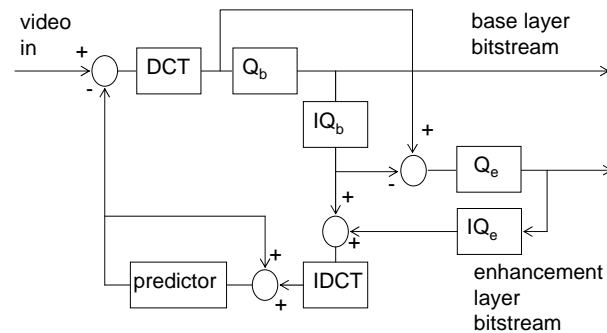
- First, the input video is coded at a low bit rate (lower image quality), to generate the base layer bitstream
- With a higher precision, to generate the enhancement layer
- May use an identical or different encoder at two layers
- The encoder is much more complex than the data-partitioning, it requires at least two nonscalable encoders

DCT-Based Base-Layer Encoder



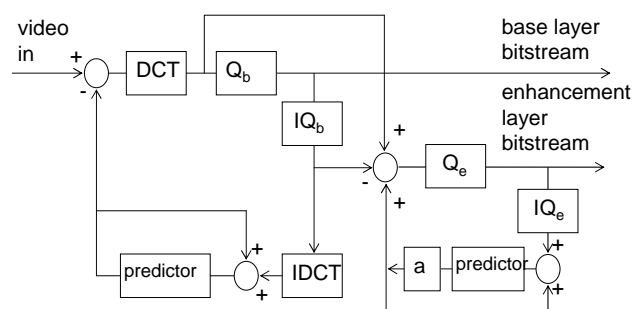
- Input pixels block X and motion-compensated predictions Y
- After transform coding: $T(X-Y)$
- After quantization (a quantization distortion Q): $T(X-Y)-Q$
- After the inverse DCT: $T^{-1}[T(X-Y)-Q] \rightarrow T^{-1}T(X-Y)-T^{-1}(Q)$
 $\rightarrow X-Y-T^{-1}(Q)$
- When this error is added to the Y : $Z = Y + X - Y - T^{-1}(Q) = X - T^{-1}(Q)$
- Thus the signal coded by the second-layer encoder is
 $X - Z = X - X + T^{-1}(Q) = T^{-1}(Q)$

Two-Layer SNR Scalable Encoder with a Drift-Free Enhancement Layer



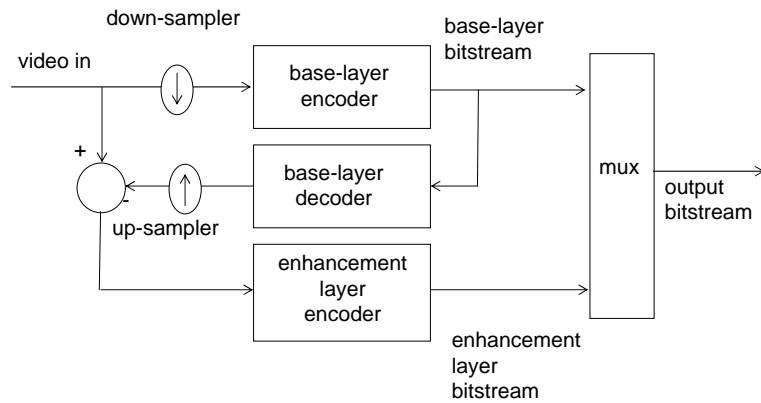
- Q_b and Q_e are the base and enhancement layer quantization step sizes, which $Q_e < Q_b$
- Compared to data partitioning, this requires only a second quantizer, an inverse quantizer, and two adders, and the complexity is not extremely great
- May cause picture drift

A Two-Layer SNR Scalable Encoder without Drift

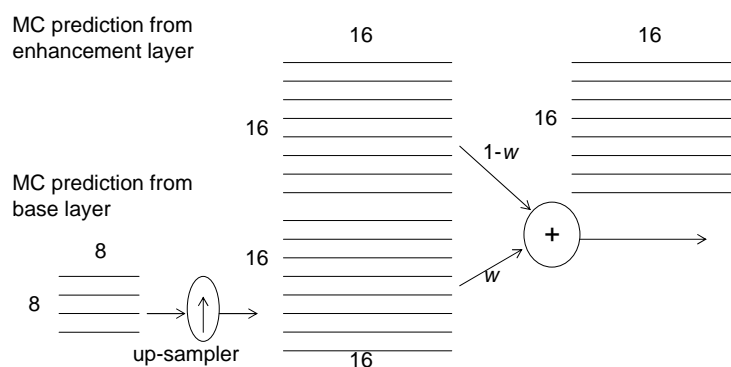


- One way to prevent picture drift is not to feed back the enhancement data into the base-layer prediction
- Intra-coding is used in the enhancement layer, which results in a very high bit rate
- To reduce bit rate, needs another encoder, then becomes much more complex
- To reduce the complexity, a leaky prediction is used
- Optimal tradeoff between bit-rate and picture drift can be achieved for $a = 0.9 \sim 0.95$

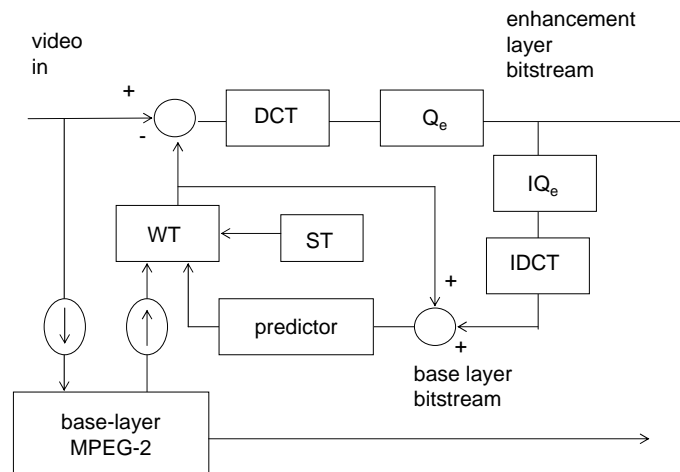
Spatial Scalability



Principle of Spatio-Temporal Prediction in The Spatial Scalable Encoder



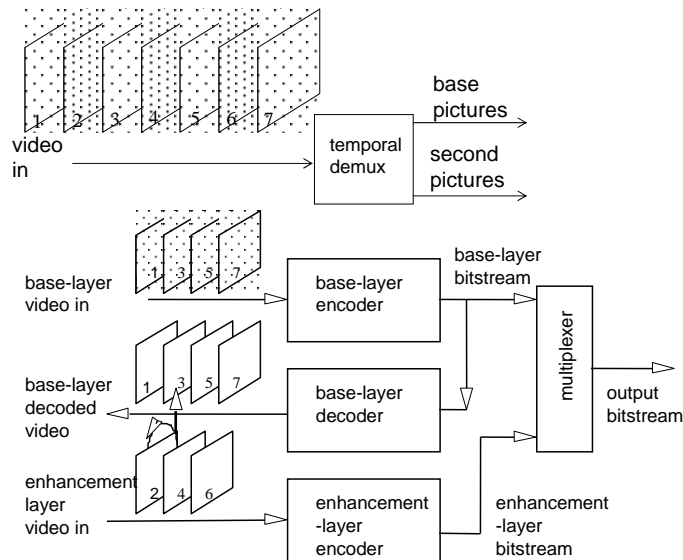
Details of Spatial Scalable Encoder



Comments on Spatial Scalable Encoder

- Comparing to data-partitioning and SNR scalable coders
 - Base-layer picture is almost free from blockiness
 - Some of very high frequency information is still missing
 - Base-layer picture can be used alone without picture drift
 - Higher price and more complexity

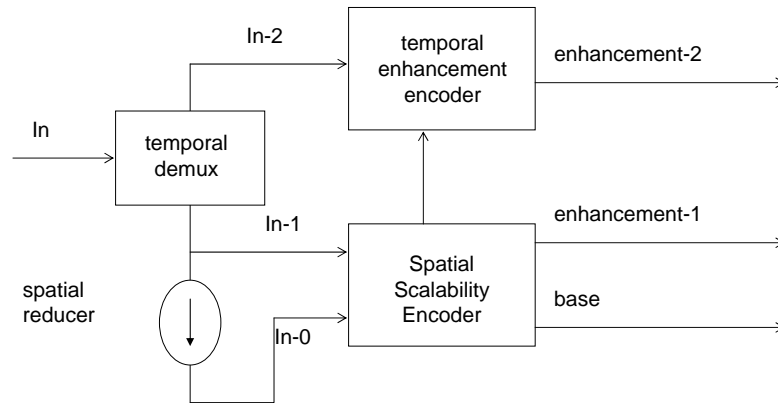
Block Diagram of a Two-Layer Temporal Scalable Encoder



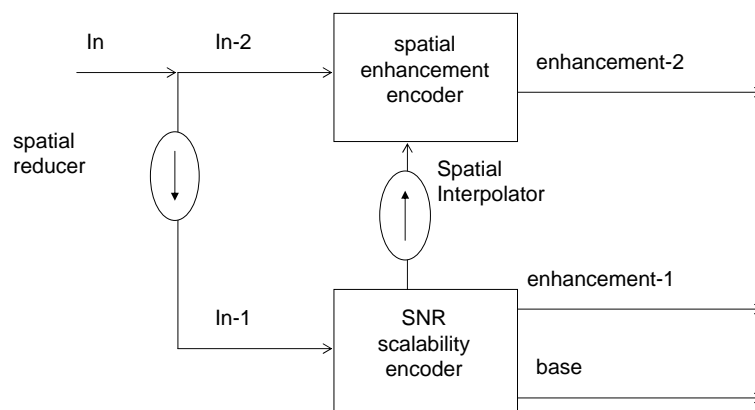
Temporal Scalability

- In fact the B pictures in MPEG-1 and MPEG-2 provide a very simple temporal scalability
 - Base layer: I and P pictures
 - Enhancement layer: B picture
- The encoder needs not be more complex than a single-layer encoder
- Free from picture drift

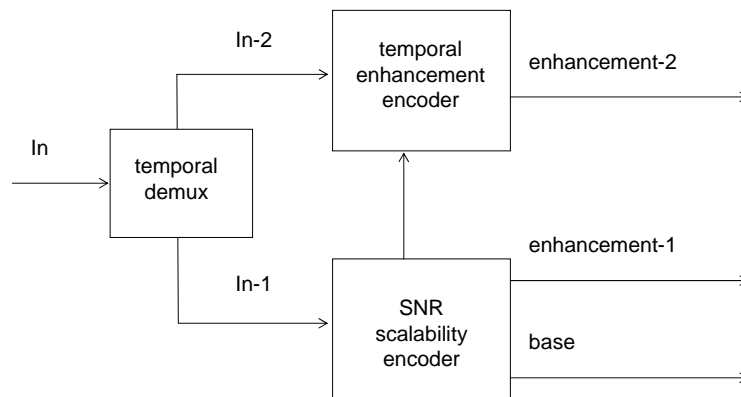
Hybrid Scalability



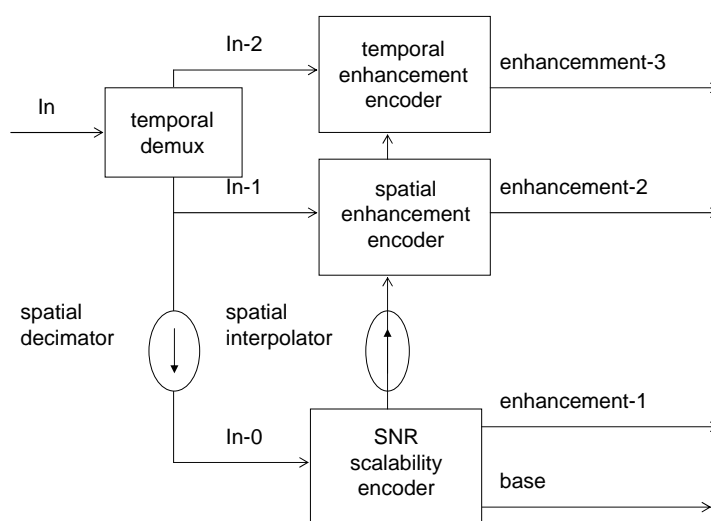
Hybrid Scalability: SNR and Spatial



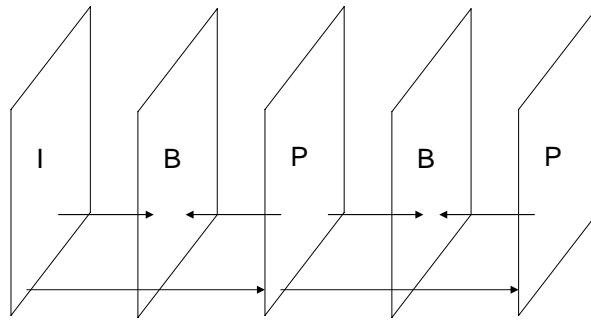
Hybrid Scalability: SNR and Temporal



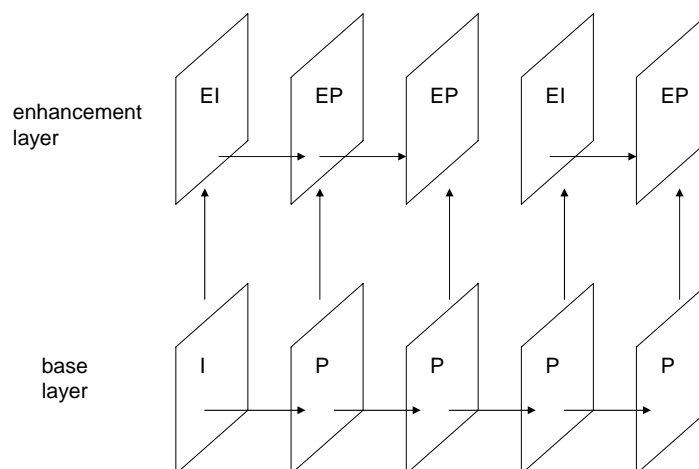
Hybrid Scalability: Spatial and Temporal



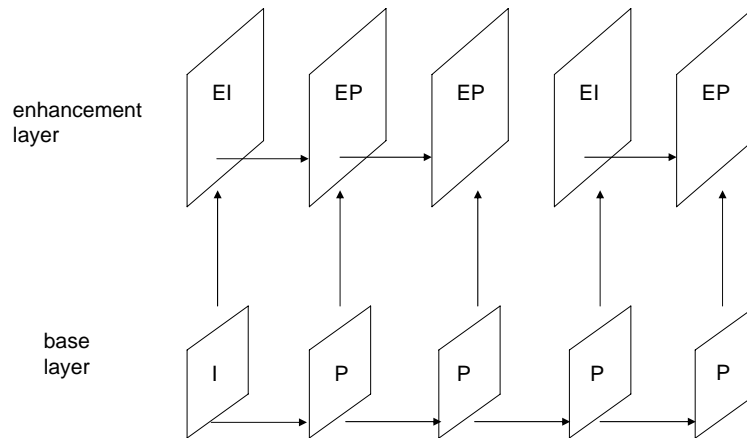
H.263+ Temporal Scalability



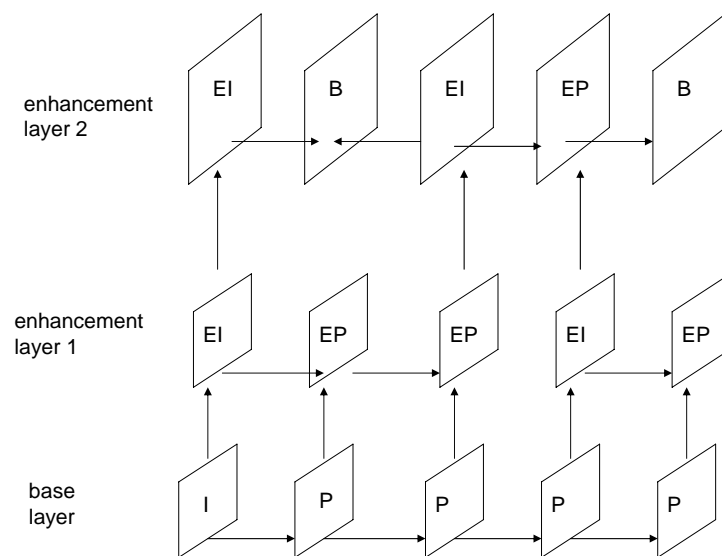
H.263+ SNR Scalability



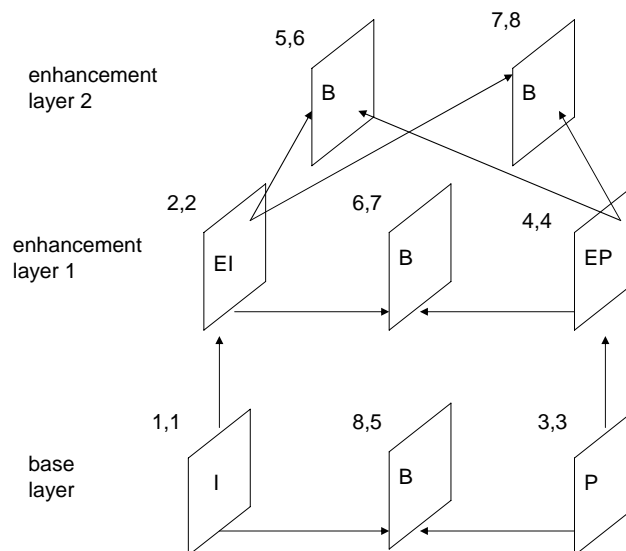
H.263+ Spatial Scalability



H.263+ Multilayer Scalability



Transmission Order of Pictures



Applications of Scalability

- Data partitioning (simplest):
 - Video over low-loss networks (e.g., ATM with congestion control)
- SNR scalability:
 - Transmission of video at different qualities
 - multiquality video, video on demand, broadcasting of TV and enhanced TV
 - Video over networks with high error or packet loss rates
 - the Internet,
 - heavily congested ATM networks

Applications of Spatial Scalability

- Spatial scalability (most complex):
 - Interworking between two different standard video codecs or heterogeneous data networks
 - Simulcasting of drift-free, good-quality video at two spatial resolutions, such as standard TV and HDTV
 - Distribution of video over computer networks
 - Video browsing
 - Reception of good quality low spatial resolution pictures over mobile networks
 - Similar to other scalable coders, transmission of error resilient video over packet networks.

Applications of Temporal Scalability

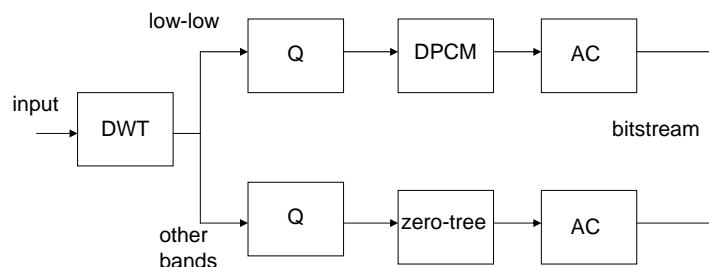
- Temporal scalability (moderately complex):
 - Migration to progressive HDTV from the current interlaced broadcast TV.
 - Internetworking between lower bit rate mobile and higher bit rate fixed networks.
 - Video over LANs, Internet and ATM for computer work stations.
 - Video over packet (Internet/ATM) networks for loss resilience.

Layered Coding with Wavelets

- One of the advantages of wavelet over DCT-based codecs is the absence of blocking artifacts
- With wavelet transforms, one can generate several layers having various spatial and quality resolutions
- The number of data layers can be much higher than what with the DCT-based codecs
- Better delivery of images over networks

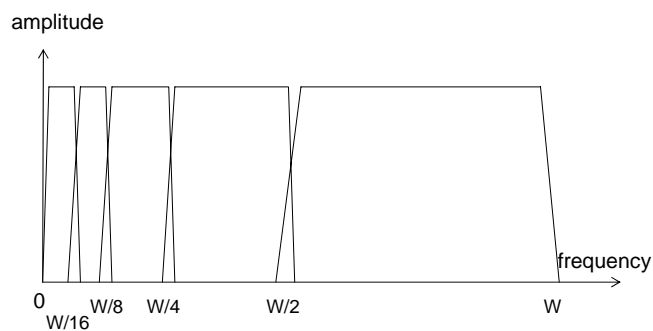
Wavelet-Based Still Image Coder

- The coding principle is based on the discrete wavelet transform, which is a subclass of subband coding
- The lowest subband is coded with a differential pulse code modulation (DPCM)
- Higher bands with the zero-tree coding technique

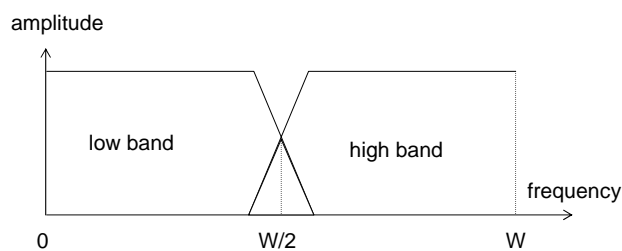


Discrete Wavelet Transform

- The basic principle is the partitioning of the signal spectrum into several frequency bands

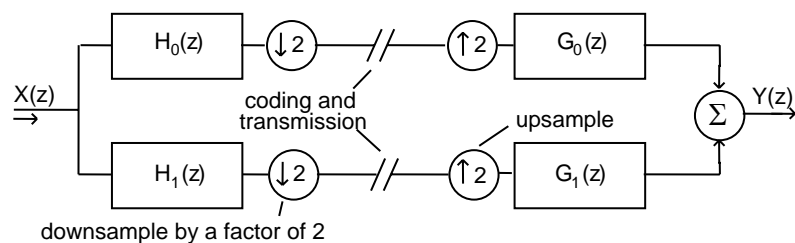


Two-band Analysis Filter



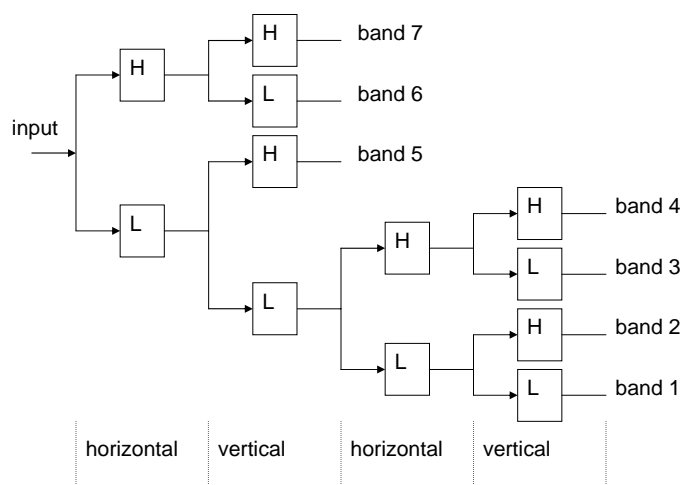
- To eliminate aliasing distortion, the synthesis and analysis filters must have certain relationships

Two-band Wavelet Transform Codec

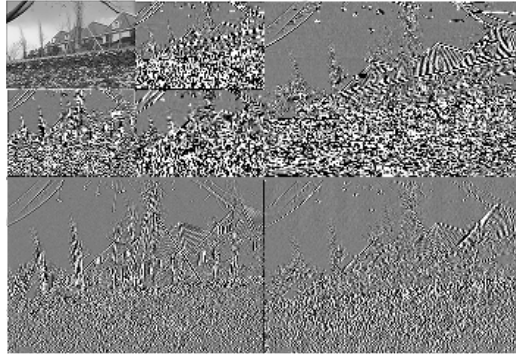


- $H_0(z)$: the z-transform of the low-pass analysis filter
- $H_1(z)$: the z-transform of high-pass analysis filter
- $G_0(z)$ and $G_1(z)$ are the corresponding synthesis filters
- The downsampling factor is 2, so as the upsampling

Tree-Structured Multiband Wavelet Transform



Tree-Structured Multiband Wavelet Transform



LL		HL2	
1	2	5 HL1	
3	4 HH2		
6 LH1		7 HH1	

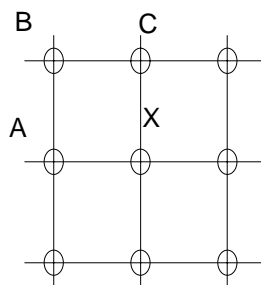
Daubechies (9,3) filter bank

Low-pass = {0.033, -0.066, -0.177, 0.420, 0.994, 0.420, -0.177, -0.066, 0.033}

High-pass = {-0.354, 0.707, -0.354}

Prediction for Coding The Lowest Band Coefficients

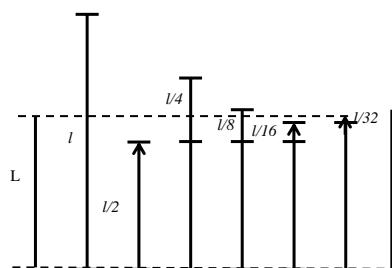
- $w_{prd} = w_C$, if $|w_A - w_B| < |w_A - w_C|$, otherwise $w_{prd} = w_A$



Zero-Tree Coding of Higher Bands

- The higher order wavelet coefficients are coded with the embedded zero-tree wavelet (EZW)
- The method
 - based on the concept of quantization by *successive approximation*,
 - exploits the similarities of the bands of the same orientation.

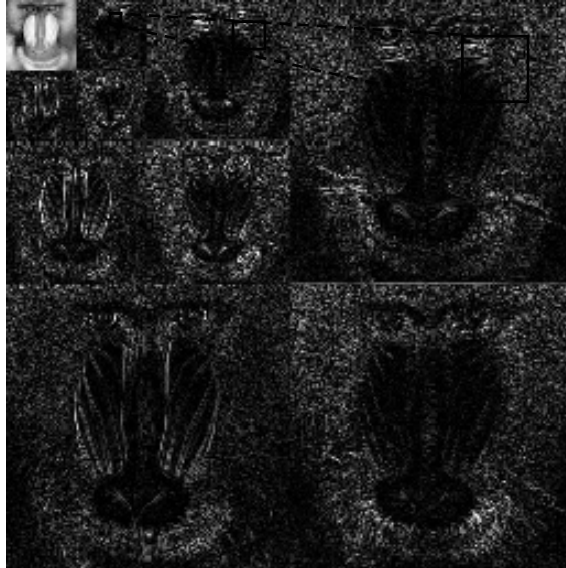
Principle of Successive Approximation



- As shown above, the quantized length, can be expressed as

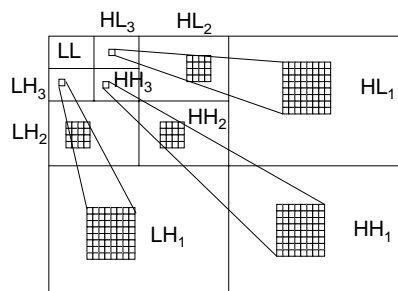
$$\hat{L} = 0 \times l + 1 \times \frac{l}{2} + 0 \times \frac{l}{4} + 0 \times \frac{l}{8} + 1 \times \frac{l}{16} + 1 \times \frac{l}{32} \dots = \frac{l}{2} + \frac{l}{16} + \frac{l}{32}$$

Similarities among Image Subbands

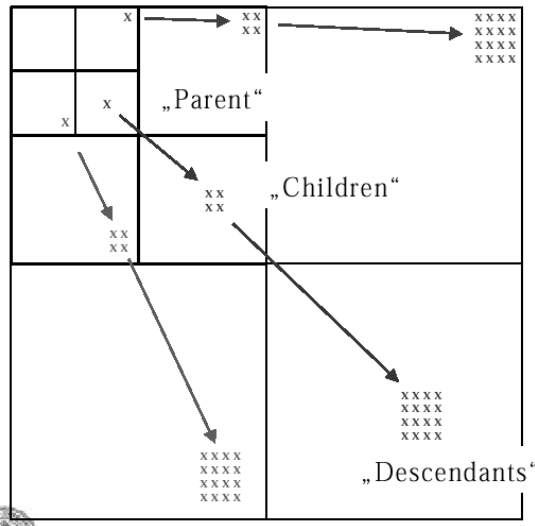


Zero-tree Coding: Quadtree Representation of Higher Bands

- Subimages of lower bands have quarter dimensions of their higher bands
- A quad-tree representation of the bands of the same orientation for a 10-band splitting is shown below (three-stage wavelet transform)
- If a coefficient in LH_3 is zero, it's more likely that its children in higher bands of LH_2 and LH_1 will also be zero => "zero tree"



Embedded Zerotree Wavelet (EZW) Algorithm



- Idea: Conditional coding of all descendants (incl. children)
- Coefficient magnitude > threshold: significant coefficients
- Four cases
 - ZTR: zero-tree, coefficient and all descendants are **not** significant
 - IZ: coefficient is not significant, but some descendants are significant
 - POS: positive significant
 - NEG: negative significant

Embedded Zerotree Wavelet (EZW) Algorithm

- For the highest bands, ZTR and IZ symbols are merged into one symbol Z
- Successive approximation quantization and encoding
 - Initial "dominant" pass
 - Set initial threshold T, determine significant coefficients
 - Arithmetic coding of symbols ZTR, IZ, POS, NEG
 - Subordinate pass
 - Refine magnitude of coefficients **found significant so far** by one bit (subdivide magnitude bin by two)
 - Arithmetic coding of sequence of zeros and ones.
 - Repeat dominant pass
 - Set previously found significant coefficients to zero
 - Decrease threshold by factor of 2, determine new significant coefficients
 - Arithmetic coding of symbols ZTR, IZ, POS, NEG
 - Repeat subordinate and dominate passes, until bit budget is exhausted.

Embedded Zerotree Wavelet (EZW) Algorithm

- Decoding: bitstream can be truncated to yield a coarser approximation: “embedded” representation
- Further details: *J. M. Shapiro, “Embedded image coding using zerotrees of wavelet coefficients,” IEEE Transactions on Signal Processing, vol. 41, no. 12, pp. 3445-3462, December 1993.*

Summary

- Layered coding is a means of facilitating unequal protection of image/video information at various important levels
- Three general layered coding schemes are discussed
 - Pyramidal coding:
 - only has a historical importance
 - DCT pyramid has proven to be very efficient in image condensation
 - Layered coding based on standard DCT-Based codec
 - only three methods of scalability have been recognized (spatial, SNR, and temporal)
 - supported in H.263+ and MPEG-2
 - Wavelet transform
 - has been adopted in JPEG-2000 and MPEG-4
 - generates more layers than DCT-based codec => very attractive in video networking

MPEG-4 Fine Granularity Scalability

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MPEG-4 Fine Granularity Scalability

- **Internet applications**
- **broadcast applications over packet networks**
 - Low complexity
 - Supports both unicast & multicasting capabilities
 - Supports various layers of SNR enhancements
 - Covers a “range” of bitrates instead of a few discrete bitrates
 - Base-layer compatible to MPEG-4
 - Error robustness

Challenges for Internet Video

- Challenges
 - No QOS guarantees (bandwidth, delay, packet loss)
 - Bandwidth differences of heterogeneous networks
 - Bandwidth variation with time
- Conventional video coding techniques
 - Optimizing perceived quality at a given bitrate

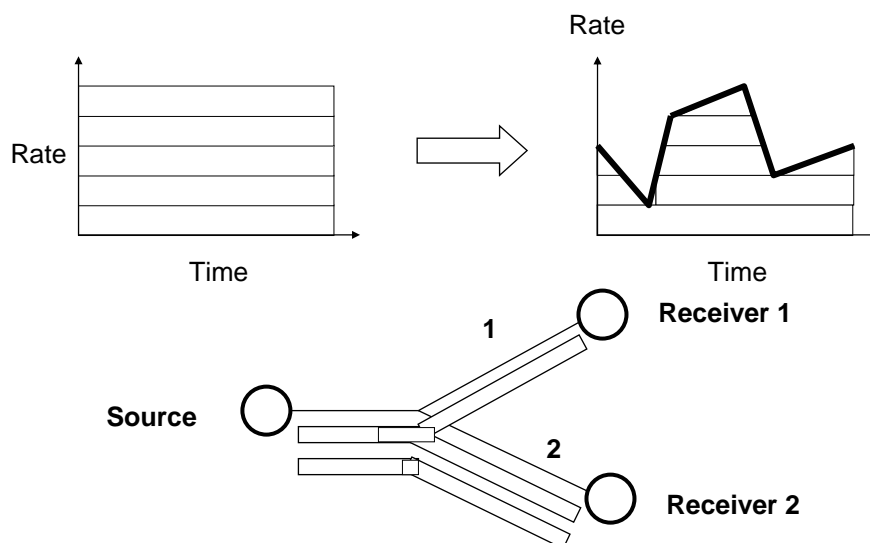
Bandwidth Variation

- “Broadband” Internet access has wider variation:
 - Cable modem: from < 100 to > 1000 Kbit/sec
 - DSL: from < 600 to > 6000 Kbit/sec

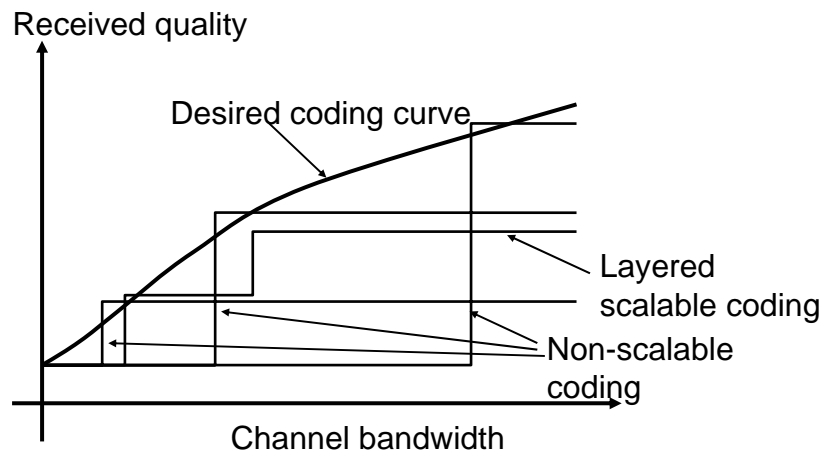
Bandwidth Variation - Solutions

- Multiple Bit-streams pre-encoded with different bit-rates
 - Large storage
 - Complexity in bit-streams management and switching
- Real-time Transcoder
 - High complexity in the streaming sever
- **Scalable video**
 - **Degree of scalability**

Why Scalable Video – Rate Control



Performance Comparison of Various Coding Schemes



FGS Standard

- July '99 **FGS Working Draft issued**
- Oct '99 **MPEG adopted FGS as "MPEG-4-Version4"**
First proposals on FGS profiles presented
- Dec '99 **Large number of proposals in support of FGS**
- Mar '00 FGS was issued as MPEG-4-V4 (PDAM)

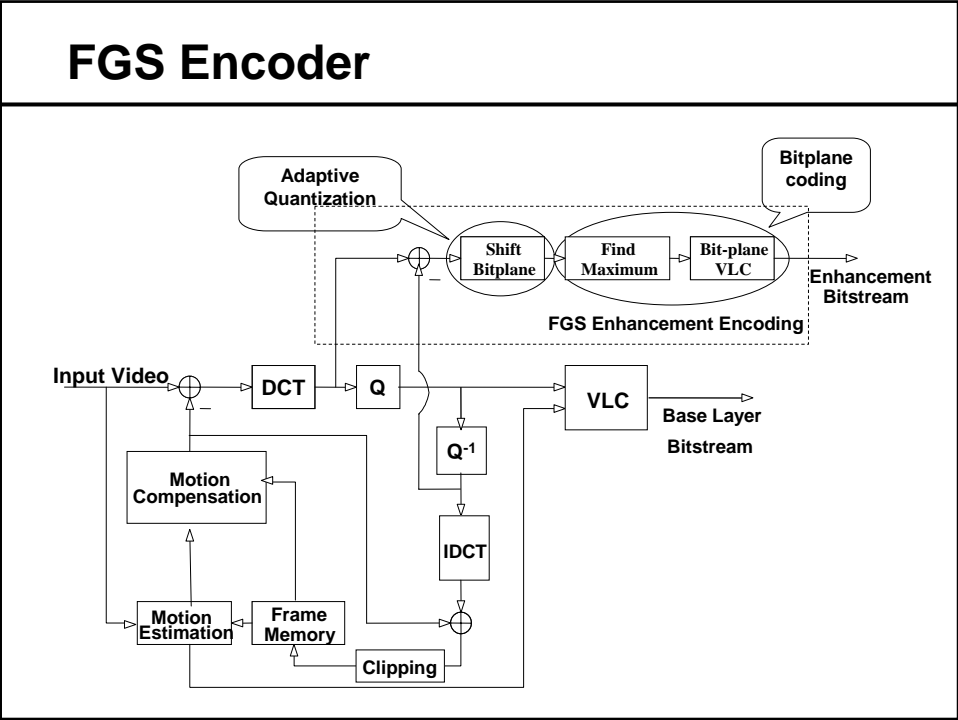
Rate Adaptation with MPEG-4 FGS

The diagram illustrates the MPEG-4 FGS architecture, showing a Base Layer and a Single Enhancement Layer.

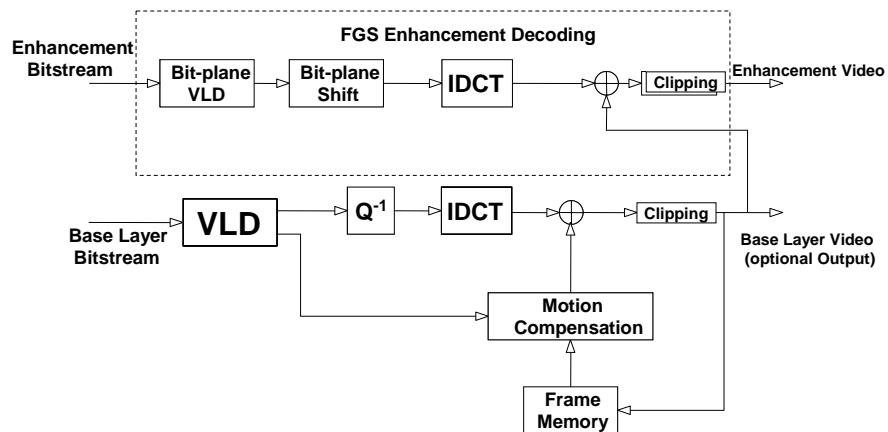
Base Layer: This layer consists of five blocks labeled I, P, P, P, and P, connected sequentially by arrows. The first block is labeled 'I' (Intra-coded), and the subsequent four blocks are labeled 'P' (Predictor-coded).

Single Enhancement Layer: This layer is positioned above the Base Layer and consists of five corresponding blocks. Each block in the Enhancement Layer is connected to its corresponding block in the Base Layer by an upward-pointing arrow. The Enhancement Layer blocks are represented as rectangles divided into two horizontal sections, with the bottom section being smaller than the top section.

The overall structure shows that the Base Layer provides the primary video data, while the Single Enhancement Layer provides additional detail for rate adaptation.



FGS Decoder



FGS Bit-plane Coding

- Bitplane coding considers each DCT coefficients as a *binary number of several bits* instead of a decimal integer
- Coding the DCT coefficients from MSB plane to LSB plane
- Find maximum number of bitplanes

FGS Bit-plane Coding

- Symbols
 - (RUN , EOP)
 - VLC
 - Escape, FLC
 - All-Zero
- Different VLC tables for different bitplanes
 - 4 VLC tables (MSB, MSB-1, MSB-2, others)
- At the first 2 bitplanes, All-Zero in macroblock layer (special pattern)

FGS Bit-plane Coding Example

❖ The absolute residue values after zigzag ordering are given as follows:

10,0,6,0,3,0,...,0,0

$(10)_{10} = (1010)_2$

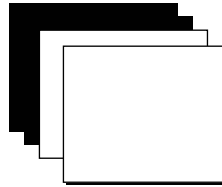
1,0,0,0,0,0,0,0,0,0,0,0,0,0,0,...0,0 (MSB)

0,0,1,0,0,0,0,0,0,0,0,0,0,0,0,...0,0 (MSB-1)

1,0,1,0,0,1,0,1,1,0,0,1,0,0,0,...0,0 (MSB-2)

0,0,0,0,0,1,0,0,0,0,0,0,0,0,1,0,...0,0 (MSB-3)

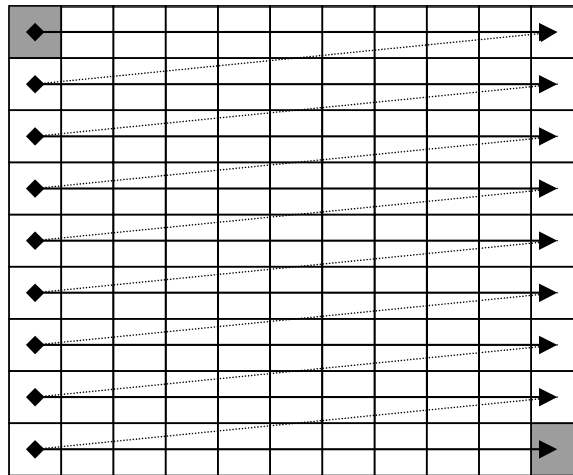
FGS Bit-plane Coding Example



- Converting the four bit-planes into (RUN,EOP)symbols:

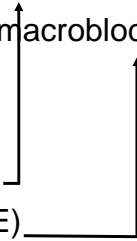
(0,1)	(MSB)
(2,1)	(MSB-1)
(0,0),(1,0),(2,0)(1,0),(0,0),(2,1)	(MSB-2)
(5,0),(8,1)	(MSB-3)

Macroblock Scan Order in FGS

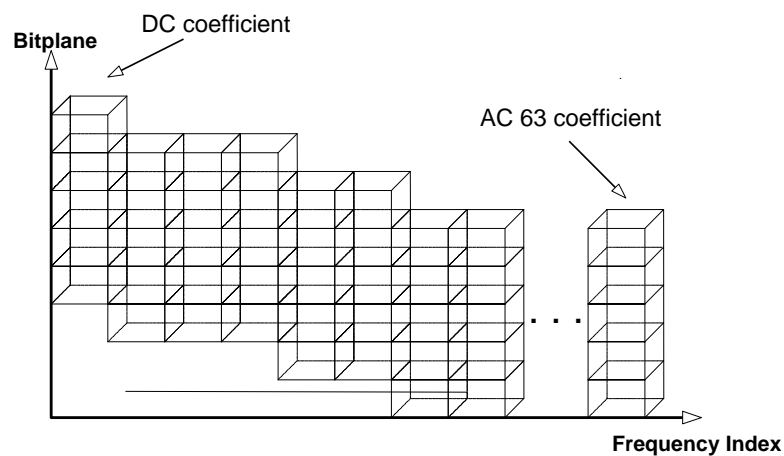


Selective Enhancement Tools in FGS

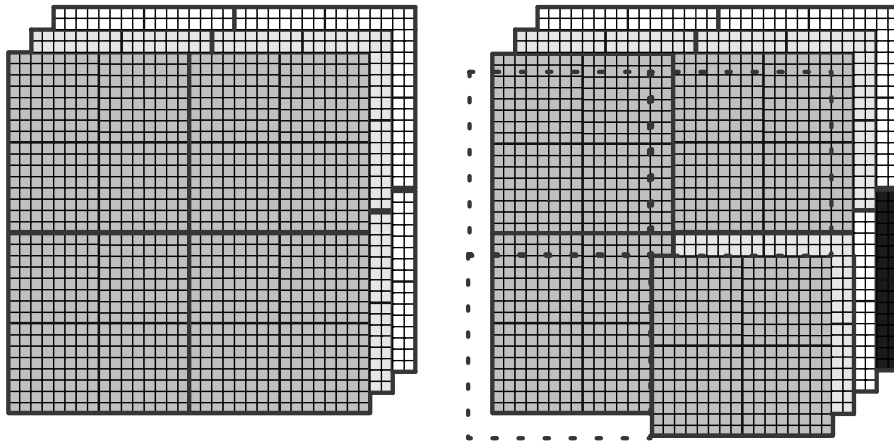
- Improving perceived visual-quality
- Base layer
 - Quantization matrix for different coefficients
 - Quantization factor varies on macroblocks
- Enhancement layer
 - Bitplane shifting
 - Frequency weighting (FW)
 - Selective enhancement (SE)



Frequency Weighting



Selective Enhancement



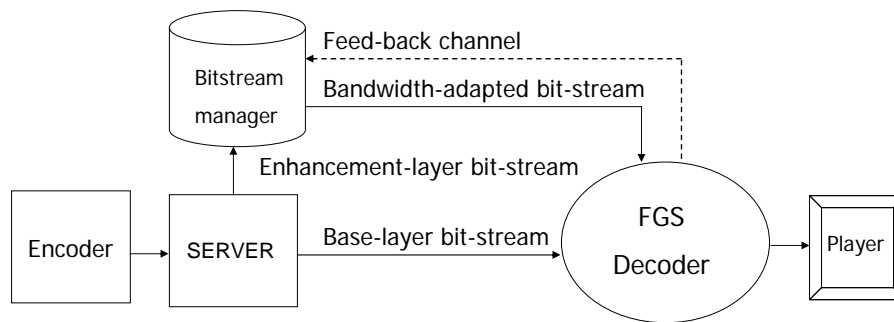
Selective Enhancement on Face Area



without SE on face

with SE on face

FGS-based Bandwidth Adaptation

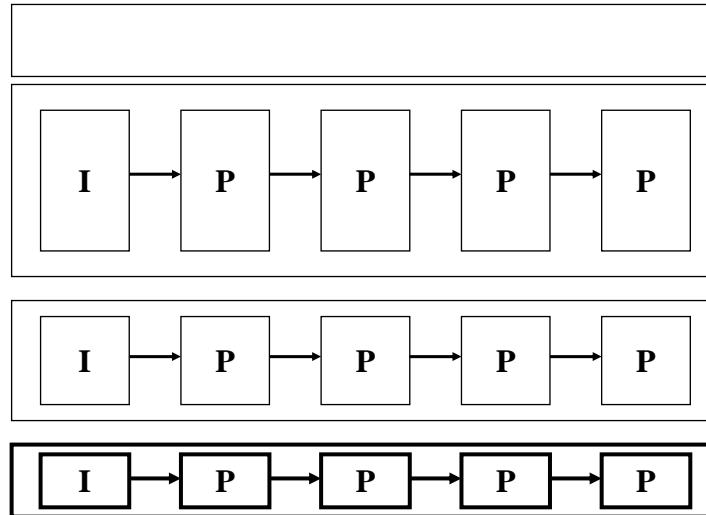


Packet-Loss Resilience

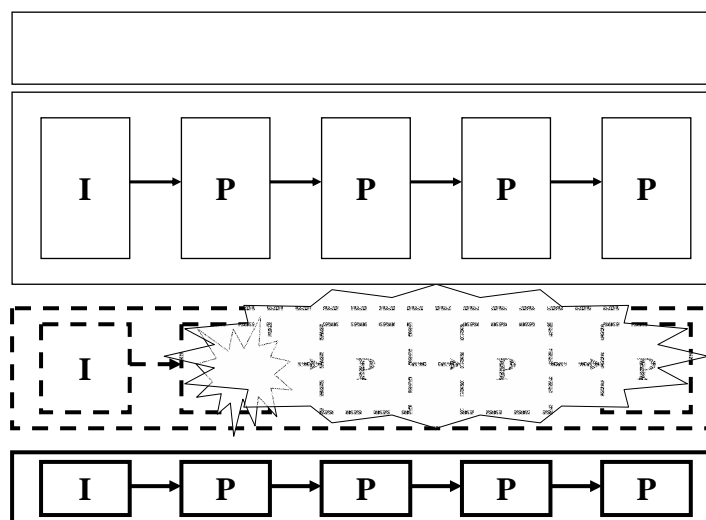
- Retransmission- and FEC-based recovery methods have been shown to be viable for real-time applications
- HOWEVER, there are NO guarantees for *on-time* delivery

⇒ Video has to be packet-loss resilient

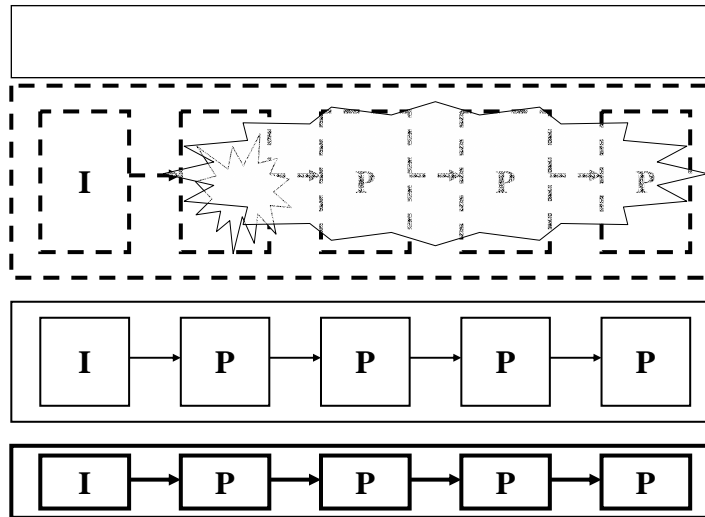
Packet-Loss Resilience – Non-Scalable Option



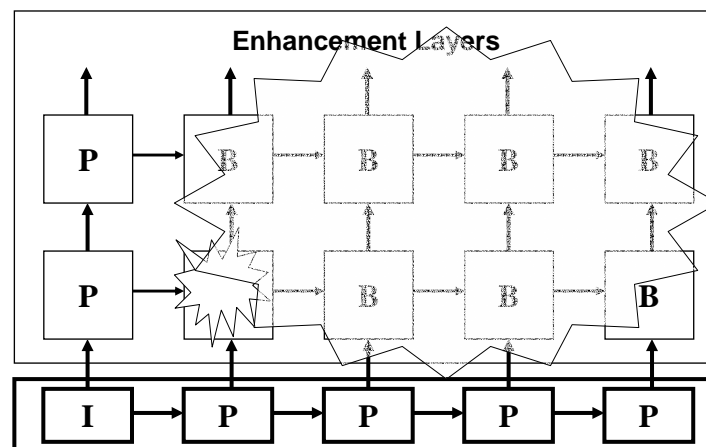
Packet-Loss Resilience – Non-Scalable Option



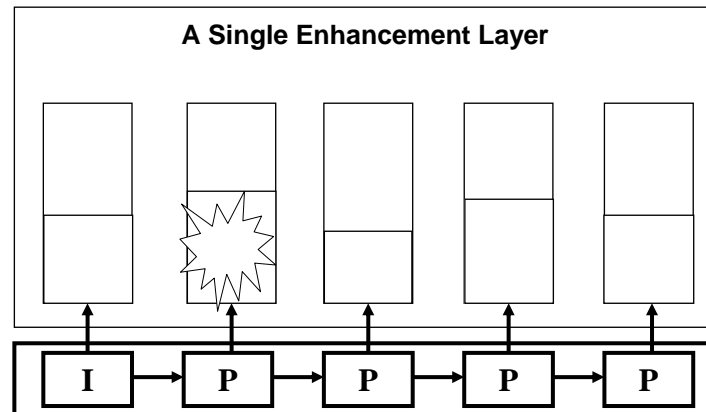
Packet-Loss Resilience – Non-Scalable Option



Packet-Loss Resilience – Multilayer Scalable Option



Packet-Loss Resilience - FGS



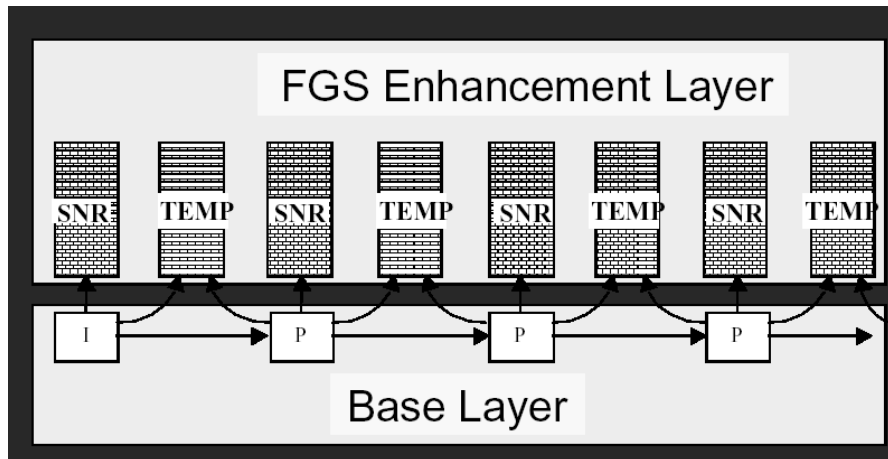
Packet-Loss Resilience



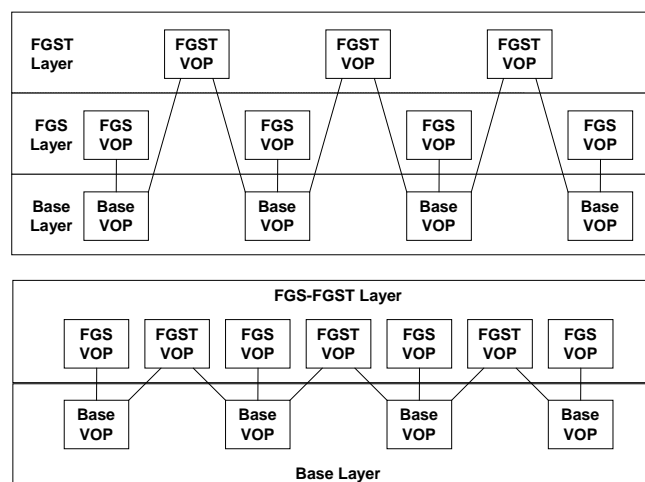
MPEG-4 Single-Layer

FGS

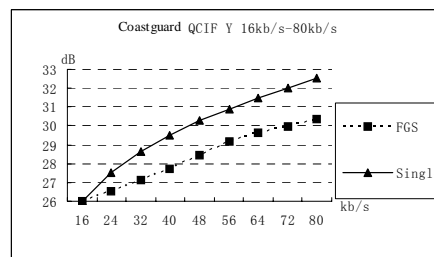
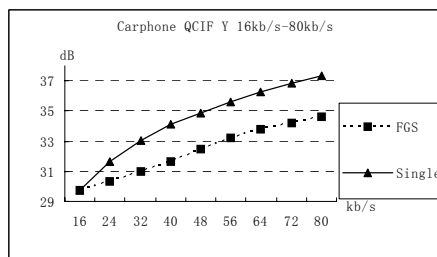
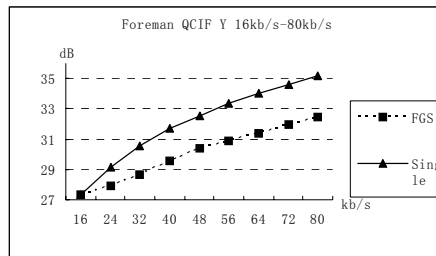
Hybrid SNR/Temporal Scalability



Hybrid SNR/Temporal Scalability



Performance Comparison of FGS Coder and Single-Layer Coder



1.5~2.5 dB degradation was observed

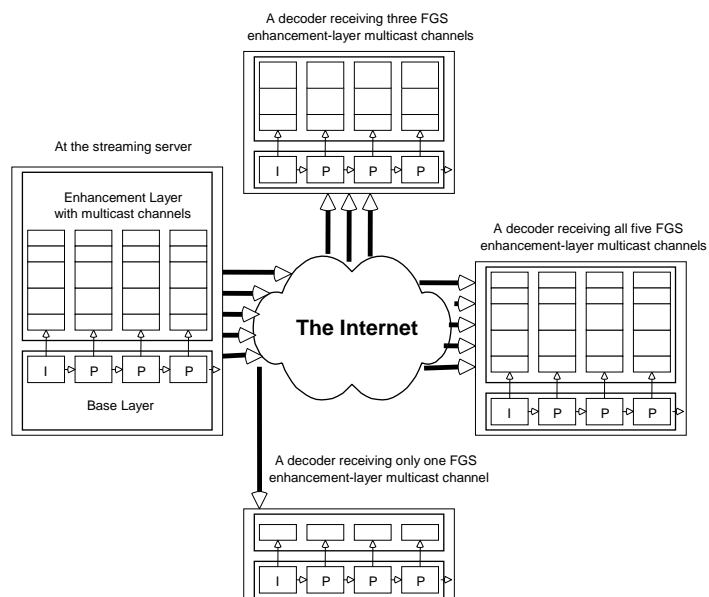
Other Proposals for Improving FGS

- Progressive FGS (PFGS)
 - Proposed by Microsoft Research Asia
- Adaptive Motion-Compensated FGS (AMC-FGS)
 - Proposed by Philips Research
- Reliable FGS (RFGS)
 - Proposed by NCTU
 - Uses leaky prediction

FGS Multicasting

- MPEG-4 FGS method
 - consists of layered video coding that supports SNR, temporal, and hybrid temporal-SNR scalabilities
 - simplicity and flexibility in supporting multicast streaming applications
 - Base layer and one or more enhancement layer send into different multicasting group individually
- Problems in multimedia transmission
 - Network heterogeneous
 - Different codec, resource, network conditions, user requirements, etc.
 - Feedback implosion

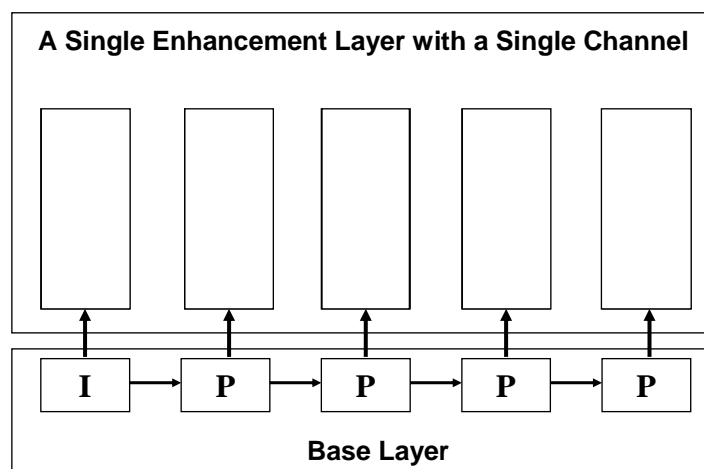
FGS Multicasting



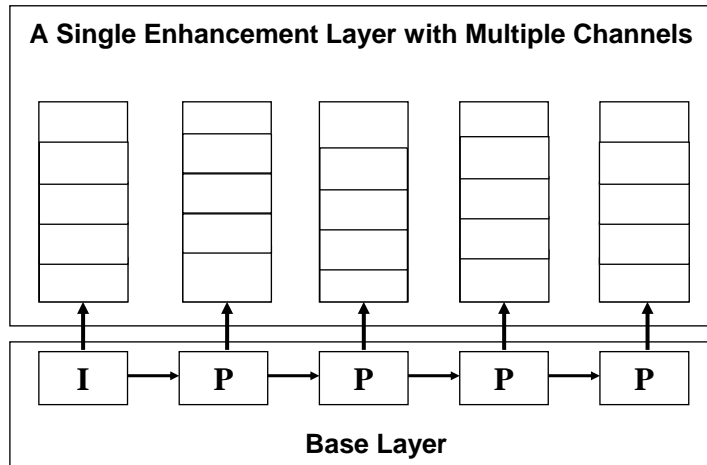
Current FGS Multicasting Approaches

- Receiver-driven multicast
 - Multicasting the base-layer over one MC group
 - Multicasting the enhancement-layer over one or more MC groups
 - Total flexibility in creating “customized MC channels”
- Sender Adaptive & Receiver-driven multicast
 - Better layer arrangement and resource allocation

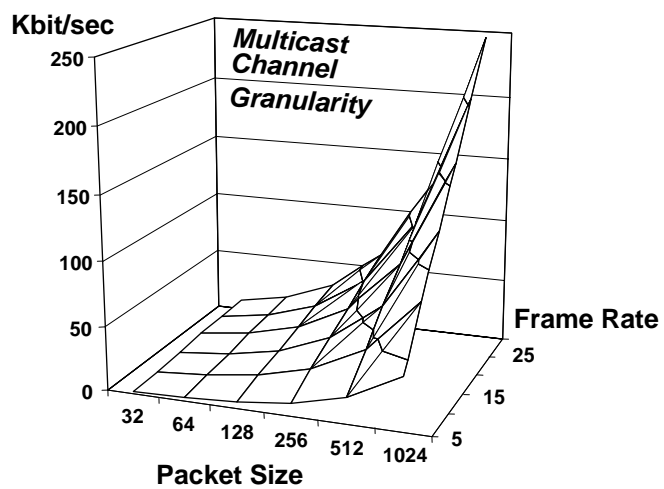
FGS Multicast Channels



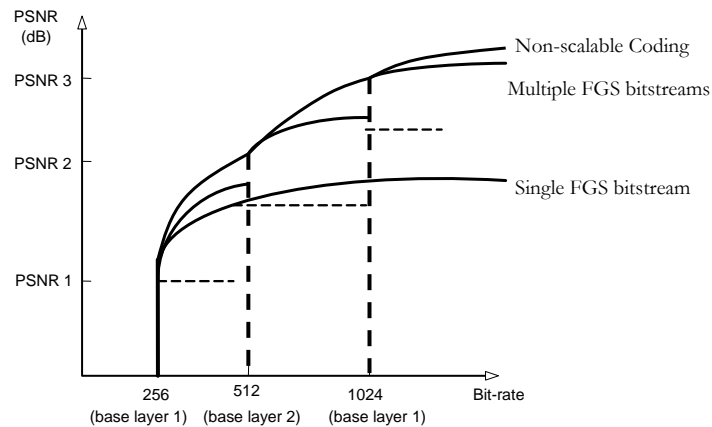
FGS Multicast Channels



Multicasting with FGS

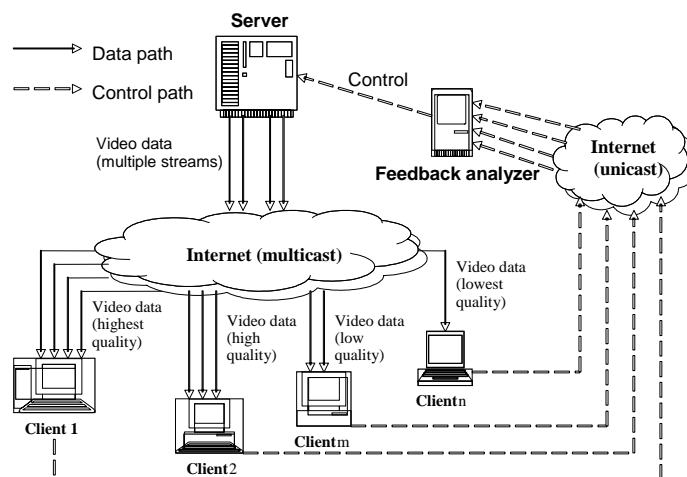


Combination of Scalable Coding & Bistream Switching



The coding efficiency of MPEG-4 FGS with “large” bit-rate range is usually not good
 ⇒ **FGS + multiple-bitstream switching**
 (integrated with receiver-driven multicasting)

Receiver-Driven Multicasting



Summary

- MPEG-4 FGS solves the bandwidth-variation problem over the Internet
 - A single enhancement-layer stream
- Totally flexible, efficient, and simple solution
 - For both unicast and multicast
- Packet loss resilient
- Open standard

MPEG-21 Scalable Video Coding

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