



Video-Coding Basics



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LVA: 389.134 [Video and Multimedia Transmissions over cellular Networks](#)

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LVA 389.168 [Advanced Wireless Communications 1](#)

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Outline

- Basics on Video Sampling
 - Video Standards
- Data Rate Reduction
 - A brief overview of image compression
 - Video compression techniques
- Quality Improvements:
 - Deblocking, Error concealment
- Video over wireless

Video Sampling

- 3D Sampling:
 - 2D spatial domain (pixels)
 - 1D temporal domain (frame rate)
- Standards (Analogue):
 - National Television Systems Committee (NTSC) in use in Canada, Japan, South Korea, USA, and some other places in South America, working with 29.97 f/s (denoted commonly as 30 f/s)
 - In the rest of the world Phase Alternation by Line (PAL) and Sequentiel couleur a memoire (SECAM) are used, operating at a frame rate of 25 f/s.

Digital Television (DTV)

(not yet for cellular)

- is the transmission of audio and video by [digital signals](#), in contrast to the [analog signals](#) used by [analog TV](#). Many countries are replacing broadcast [analog television](#) with digital television to allow other uses of the television [radio spectrum](#).
- With DTV broadcasting, the range of formats can be broadly divided into two categories: [high definition television](#) (HDTV) for the transmission of [high-definition video](#) and [standard-definition television](#) (SDTV). These terms by themselves are not very precise, and many subtle intermediate cases exist.
- One of several different HDTV formats that can be transmitted over DTV is: 1280×720 [pixels](#) in [progressive scan](#) mode (abbreviated [720p](#)) or 1920×1080 pixels in [interlaced video](#) mode ([1080i](#)). Each of these utilizes a [16:9 aspect ratio](#). (Some televisions are capable of receiving an HD resolution of 1920×1080 at a 60 Hz progressive scan frame rate — known as [1080p](#).) HDTV cannot be transmitted over current analog [television channels](#) because of [channel capacity](#) issues.
- SDTV may use one of several different formats taking the form of various aspect ratios depending on the technology used in the country of broadcast. For [4:3](#) aspect-ratio broadcasts, the 640×480 format is used in [NTSC](#) countries, while 720×576 is used in [PAL](#) countries. For [16:9](#) broadcasts, the 704×480 format is used in NTSC countries, while 720×576 is used in PAL countries.

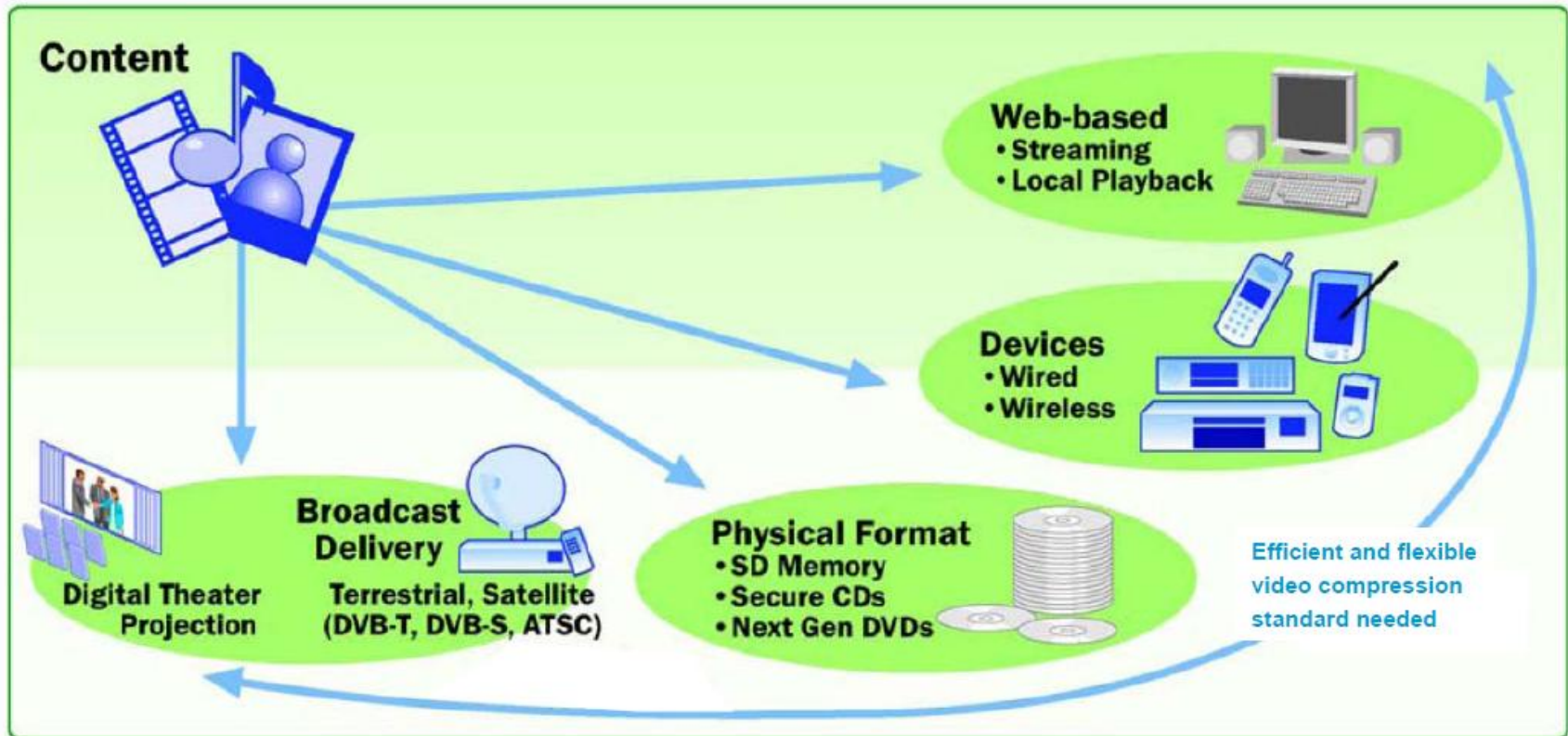
Digital Video coding standards also suited for cellular: applications and common structure

- ITU-T Rec. H.261 1988
- (ITU-T Rec. H.263) 1995
- ISO/IEC MPEG-1 1991
- ISO/IEC MPEG-2 1994
- (ISO/IEC MPEG-4) 1998
- State-of-the-art: H.264/AVC 2001
- New state of the art: H.265 2013

Applications of Video Compression

Digital television broadcasting	2 . . . 6 Mbps (10...20 Mbps for HD)	MPEG-2 (H264/AVC)
DVD video	5 . . . 8 Mbps	MPEG-2
Internet video streaming	20 . . . 300 kbps	MPEG-1, H.264/AVC, VC-1, or similar proprietary
Videoconferencing, videotelephony	20 . . . 2000 kbps	H.261, H.263, H.264/AVC
Video over 3G wireless	100 . . . 500 kbps	H.263, MPEG-4, H.264/AVC

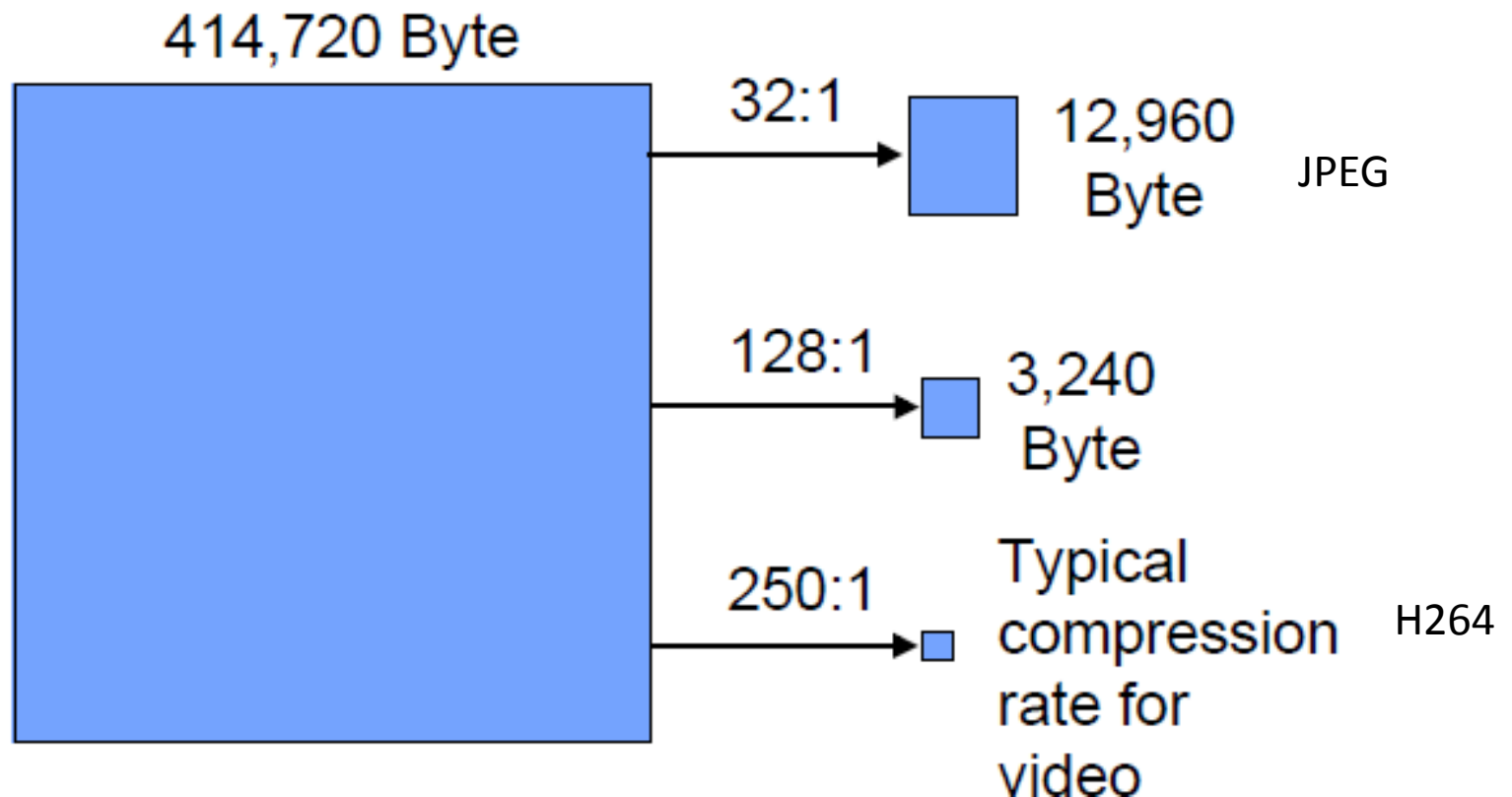
Applications of Video Compression



- Adapted from [Srinivasan et al., 2004]

Example

Geometric Interpretation



Luminance and Chrominance vs RGB

- Camera samples three colors: RGB

$$Y = k_r R + (1 - k_b - k_r)G + k_b B$$

$$C_b = \frac{0.5}{1 - k_b} (B - Y)$$

$$C_r = \frac{0.5}{1 - k_r} (R - Y)$$

$$k_b = 0.114, k_r = 0.229,$$

- Reason is better coding effect. For 4Y (luma) pixels, typically 1B and 1R are sufficient (YCrCb=4:2:0)

Bit Rates

- Typically an $N \times M$ image is sampled with $3q$ bits each pixel, resulting in $3qNM$ pixels per image
- Example: $N=M=1000$, $q=6\text{bits}$, $FR=30\text{f/s}$
 - 560 Mbit/s
- Rate reduction techniques:
 - Frame rate decimation (2,3,4,5) for low quality
 - Interlaced vs progressive (=non interlaced) scan

Mobile Video Standards

- Due to low quality and small screens, NxM can be selected a lot smaller:

Abbreviation	Size	Description
VGA	640×480	Video Graphics Array
QVGA	320×240	Quarter Video Graphics Array, called also Standard Interchange Format (SIF)
Q2VGA	160×120	
CIF	352×288	Common Intermediate Format (quarter of resolution 704×576 used in PAL)
QCIF	176×144	Quarter Common Intermediate Format

Smart phone
(2011)

Cheap, low
cost (2011)

- Mobile Video Example (4:2:0, QCIF):
 $1.5 \times 25 \times 8 \times 176 \times 144 = 7.6 \text{ Mbit/s}$

Smart phone 2013:
2.560 x 1.440

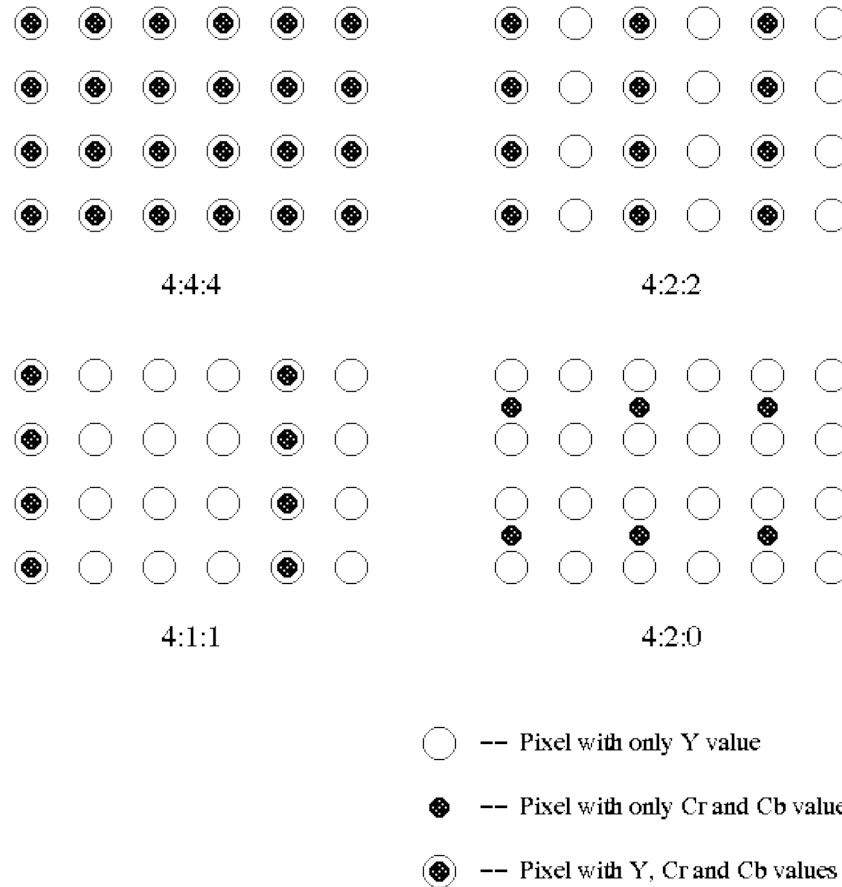
Chroma Subsampling

- Because of storage and transmission limitations, there is always a desire to reduce (or compress) the signal. Since the human visual system is much more sensitive to variations in brightness than color, a video system can be optimized by devoting more bandwidth to the luma component (usually denoted Y'), than to the color difference components Cb and Cr .
- The 4:2:2 $Y'CbCr$ scheme for example requires two-thirds the bandwidth of (4:4:4) $R'G'B'$. This reduction results in almost no visual difference as perceived by the viewer.

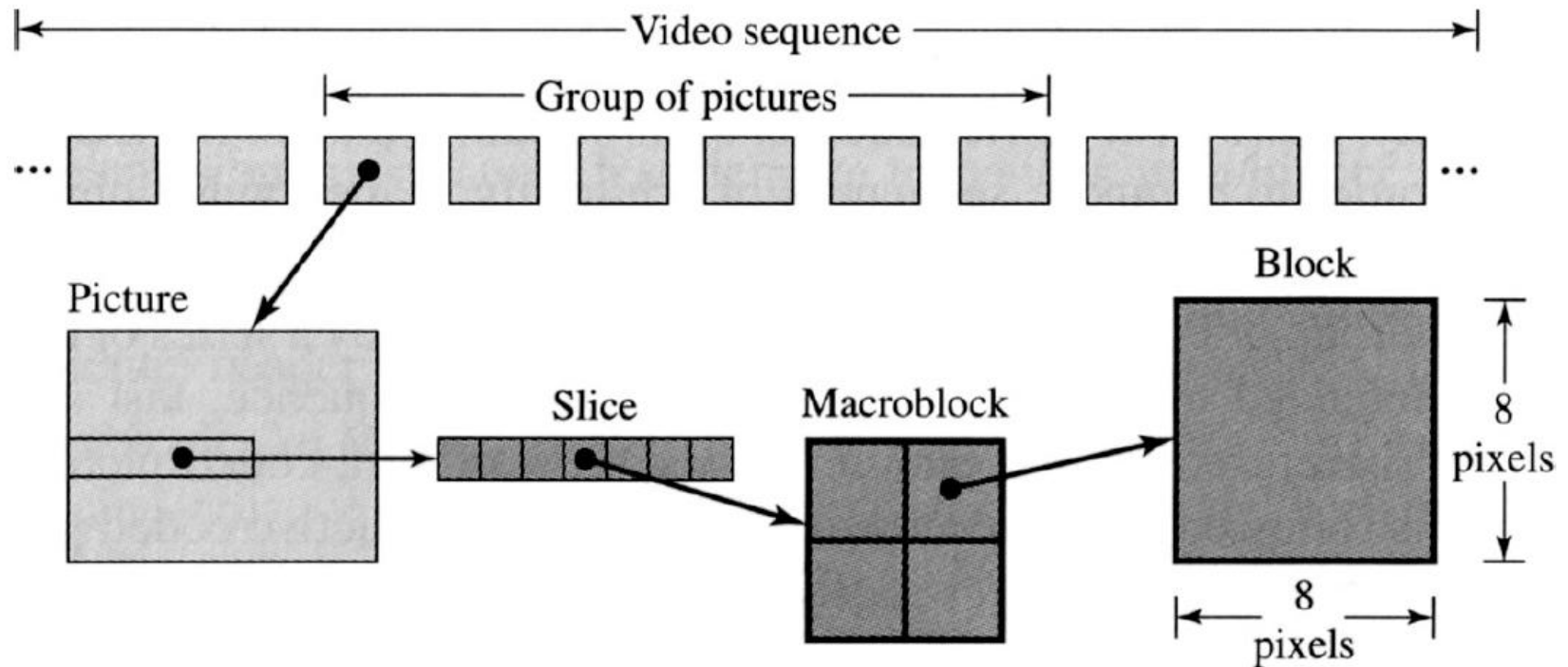
Chroma Subsampling

- The subsampling scheme is commonly expressed as a three part ratio $J:a:b$ (e.g. 4:2:2), that describe the number of luminance and chrominance samples in a conceptual region that is J pixels wide, and 2 pixels high. The parts are (in their respective order):
 - J horizontal sampling reference (width of the conceptual region). Usually, 4.
 - a number of chrominance samples (Cr, Cb) in the first row of J pixels.
 - b number of (additional) chrominance samples (Cr, Cb) in the second row of J pixels.
- See also <http://lea.hamradio.si/~s51kq/V-BAS.HTM>

Subsampling YCrCB schemes



Video Compression Standards: Hierarchical Syntax

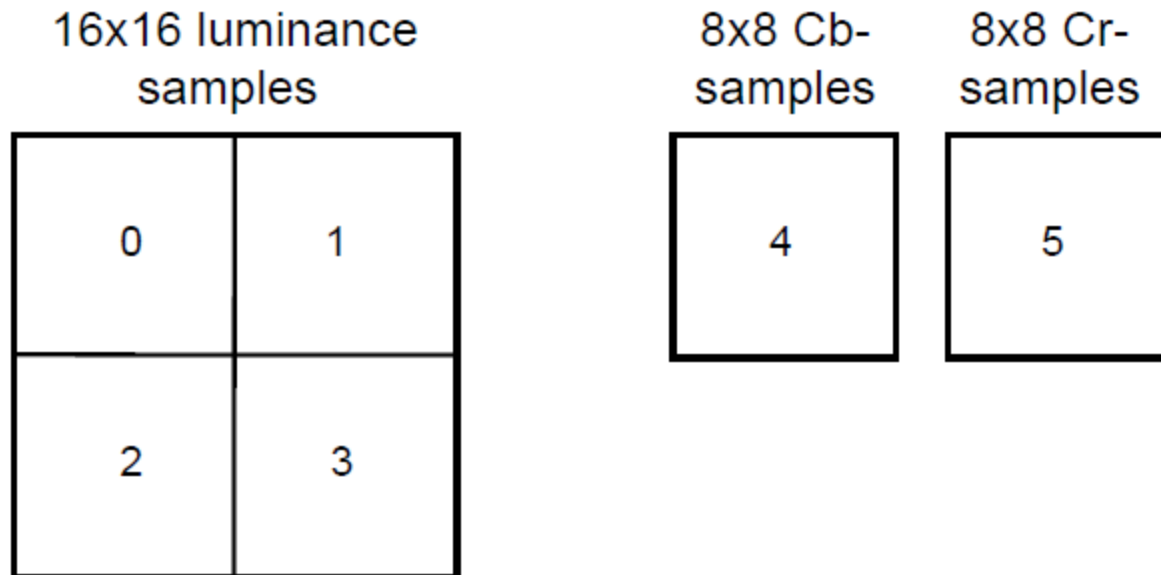


ITU-T Rec. H.261

- International standard for ISDN picture phones and for video conferencing systems (1990)
- Image format: CIF (352 x 288 Y samples) or
- QCIF (176 x 144 Y samples), frame rate 7.5 ... 30 f/s
- Bit-rate: multiple of 64 kbit/s (= ISDN-channel), typically 128 kbit/s including audio.
- Picture quality: for 128 kbit/s acceptable with limited motion in the scene
- Stand-alone videoconferencing system or
- desk-top videoconferencing system, integrated with PC

H.261 Macroblocks

- Macroblock (MB) of 16x16 pixels
- Sampling format: 4:2:0
- MB consists of 4 luminance and 2 chrominance blocks



A bit of image compression basics

- Orthogonal Transform
- 2D DCT
- Laplacian Densities
- Coding: Zigzag, runlength, entropy

Coding gain of orthonormal transform

- Assume distortion rate functions for image samples

$$d(R) \cong \varepsilon^2 \sigma_X^2 2^{-2R}$$

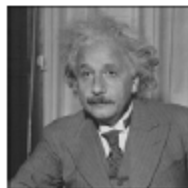
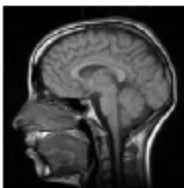
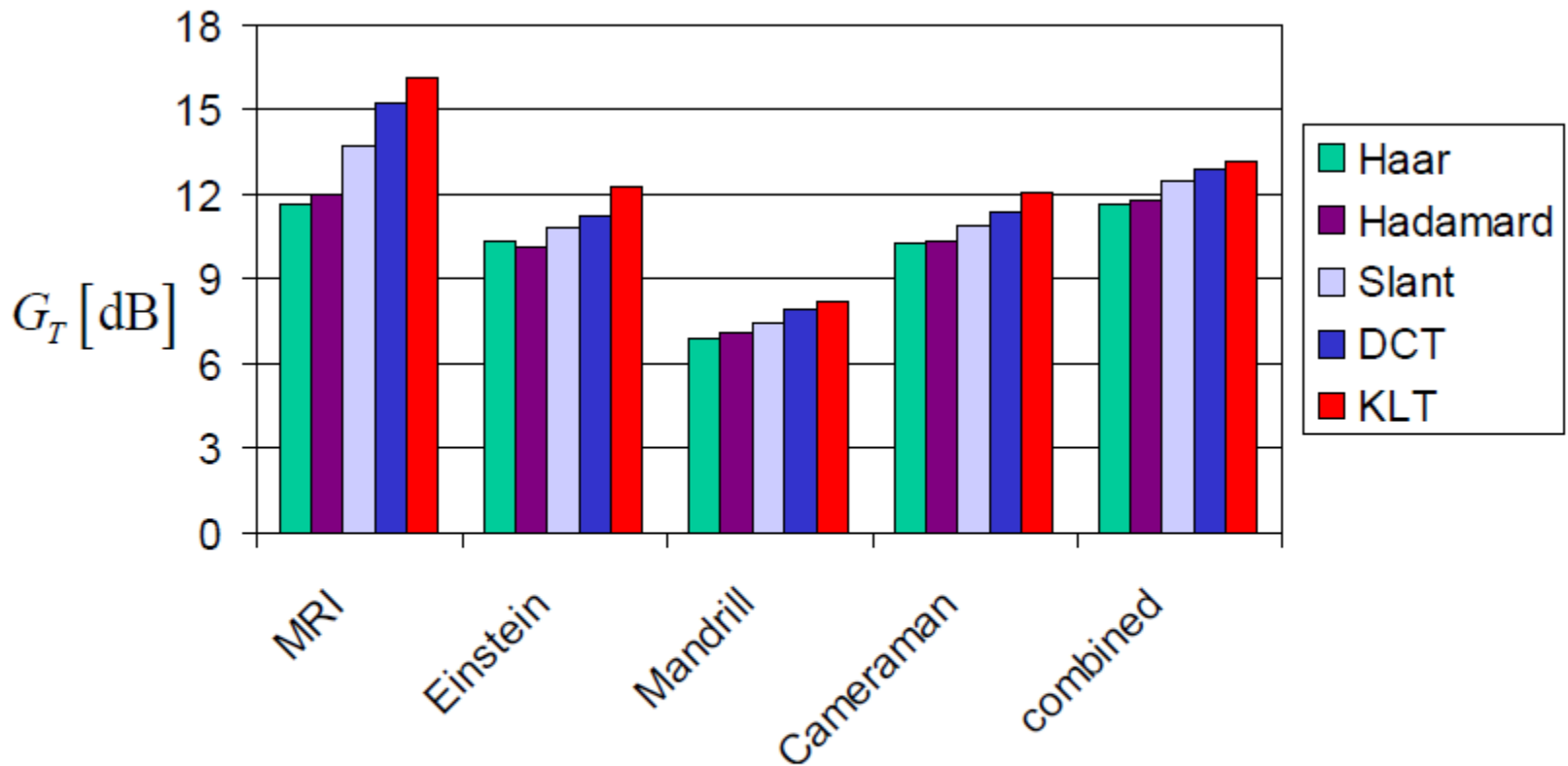
... and for encoding transform coefficients

$$d^{XFORM}(R) = \frac{1}{N} \sum_{n=0}^{N-1} d_n(R_n) \cong \frac{1}{N} \sum_{n=0}^{N-1} \varepsilon^2 \sigma_{Y_n}^2 2^{-2R_n}; \quad R = \frac{1}{N} \sum_{n=0}^{N-1} R_n$$

- Transform coding gain

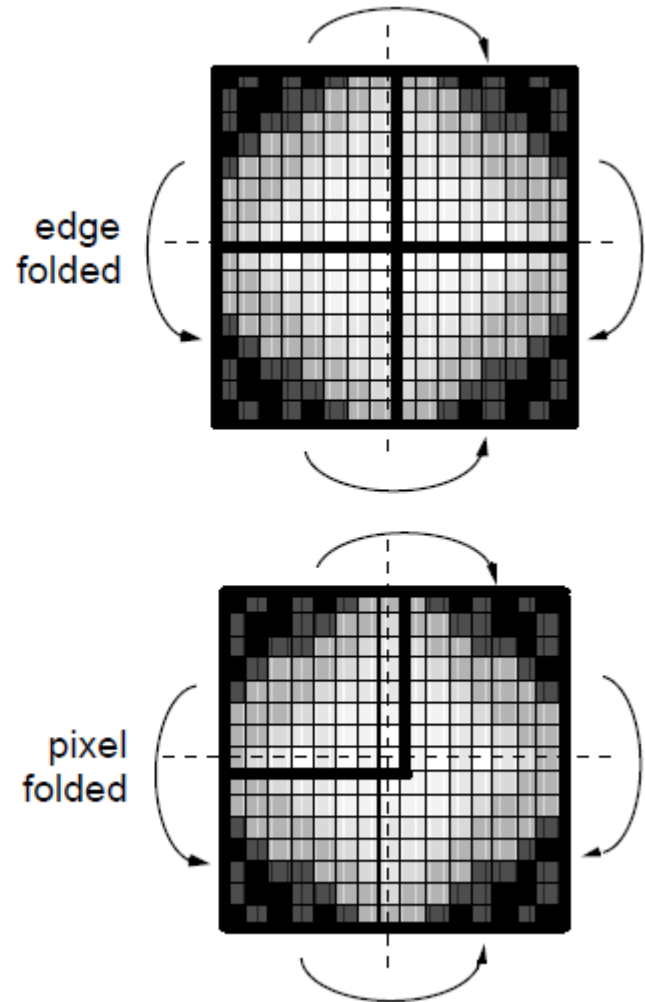
$$G_T = \frac{d(R)}{d^{XFORM}(R)}$$

Coding gain with 8x8 transforms



DCT vs DFT

- Transform coding of images using the Discrete Fourier Transform (DFT):
 - For stationary image statistics, the energy concentration properties of the DFT converge against those of the KLT for large block sizes.
 - Problem of blockwise DFT coding: blocking effects due to circular topology of the DFT and Gibbs phenomena.
 - Remedy: reflect image at block boundaries, DFT of larger symmetric block -> "DCT"



2D DCT

Type II-DCT of blocksize $N \times N$ is defined by transform matrix A containing elements

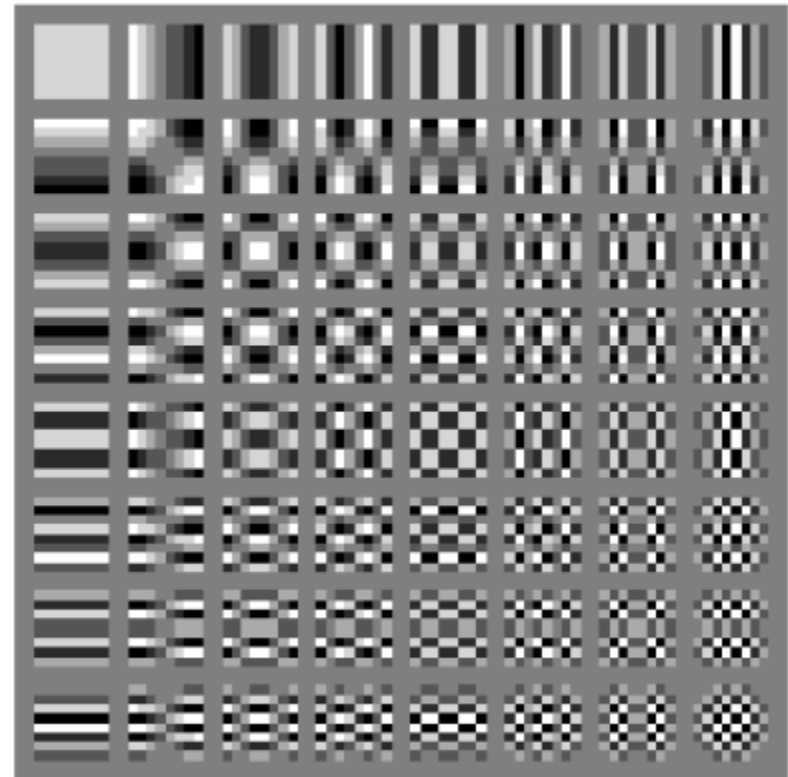
$$a_{ik} = \alpha_i \cos \frac{\pi(2k+1)i}{2N}$$

for $i, k = 0, \dots, N-1$

$$\text{with } \alpha_0 = \sqrt{\frac{1}{N}}$$

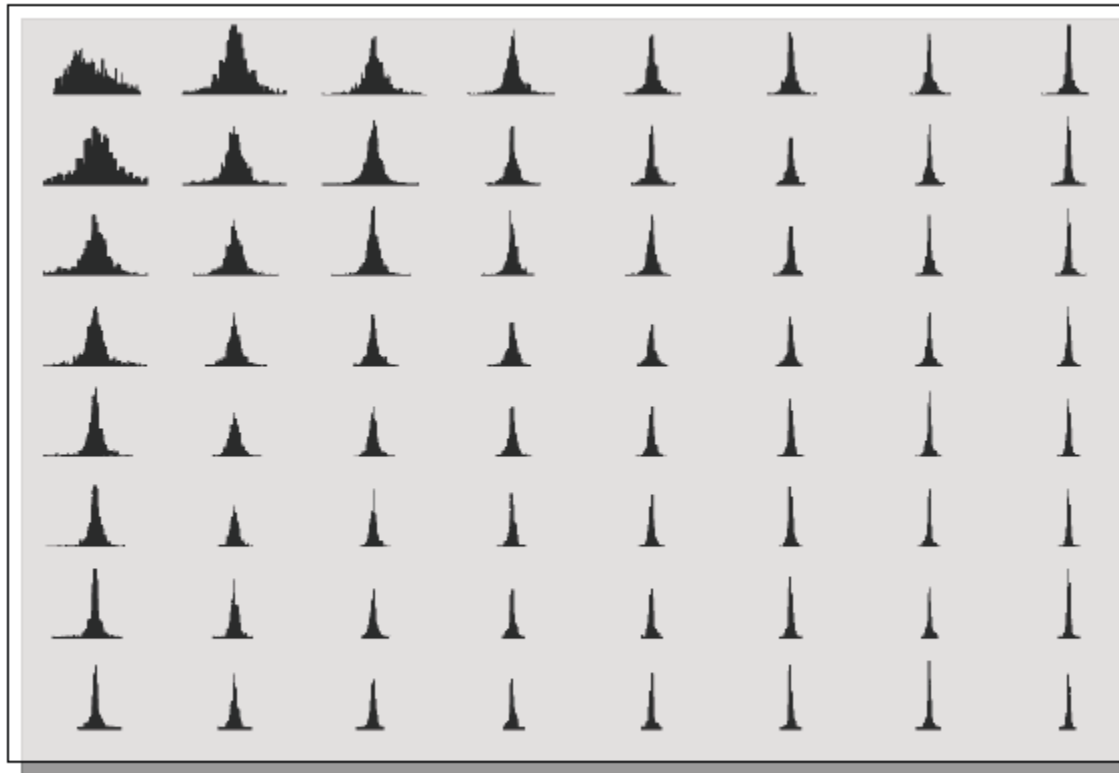
$$\alpha_i = \sqrt{\frac{2}{N}} \quad \forall i \neq 0$$

- 2D basis functions of the DCT:



Orthonormal Transforms results in Laplacian Densities

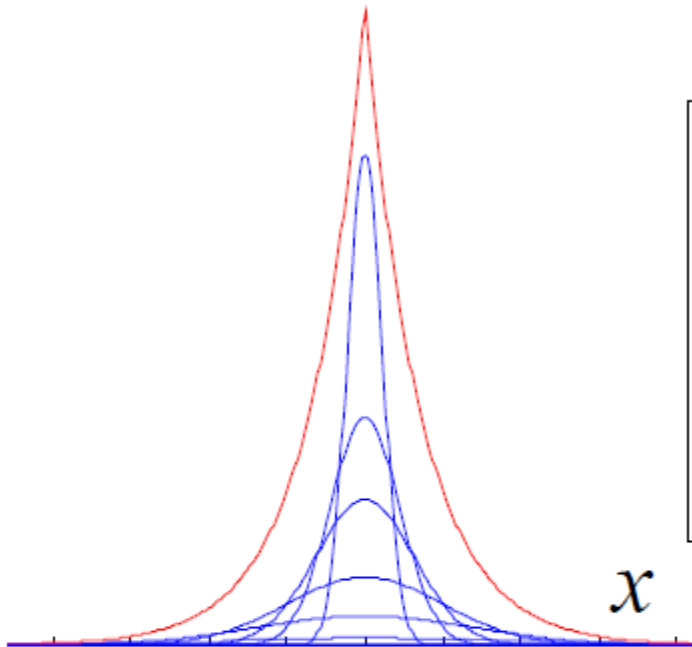
- Histograms for 8x8 DCT coefficient amplitudes measured for test image
[Lam, Goodman, 2000]



Test image
Bridge

- AC coefficients: Laplacian PDF
- DC coefficient distribution similar to the original image

Laplacian Density

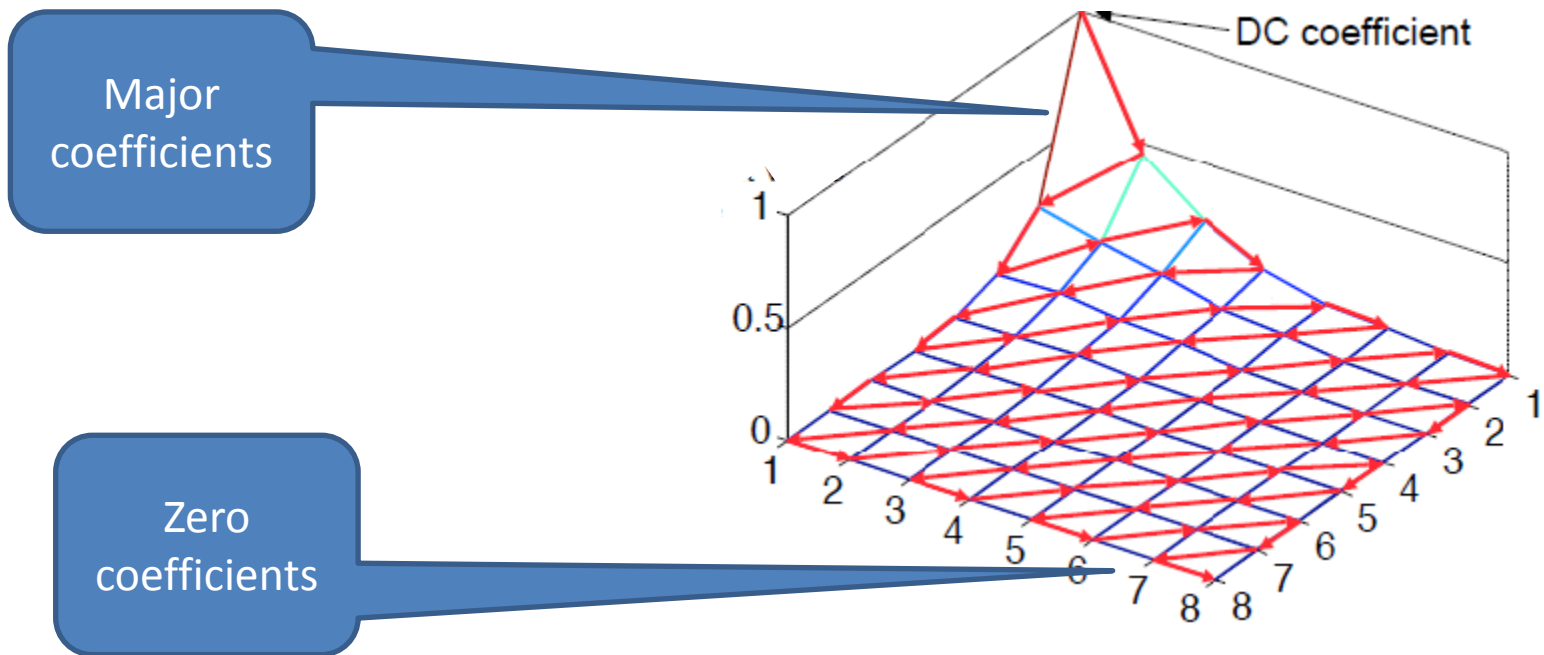


$$\begin{aligned} p_{Y_n}(y) &= \int_0^{\infty} \frac{1}{\sqrt{2\pi v}} \cdot e^{-y^2/2v} \frac{1}{\sigma^2} e^{-v/\sigma_{y_n}^2} dv \\ &= \sqrt{\frac{1}{2\sigma_{y_n}^2}} \cdot e^{-\sqrt{2} \cdot |y| / \sigma_{y_n}} \end{aligned}$$

- For a given block variance, coefficient pdfs are Gaussian
- Gaussian mixture w/ exponential variance distribution yields a Laplacian
- Gaussian mixture w/ half-Gaussian variance distribution yields pdf very close to Laplacian [*Lam, Goodman, 2000*]
- Elegant explanation of Laplacian pdfs of DCT coefficients

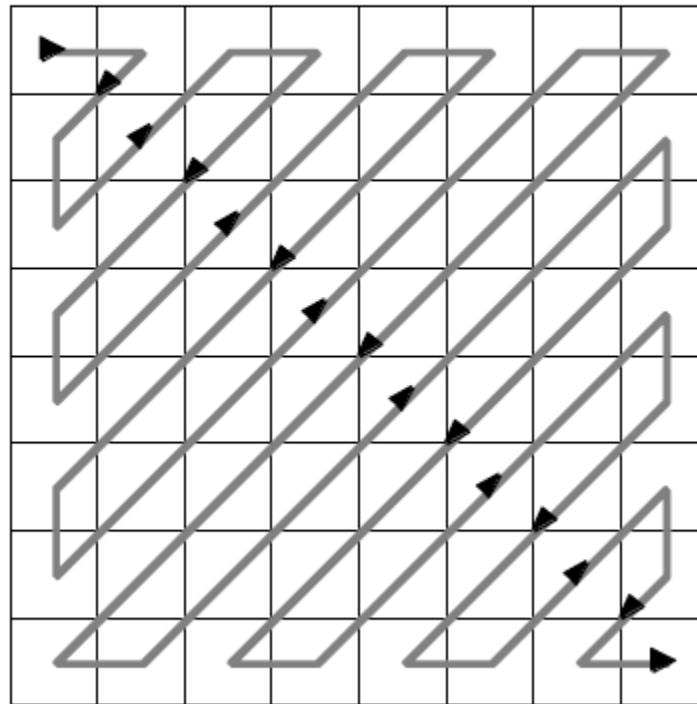
2D DCT

- 2D Discrete Cosine Transform maps an 8x8 block onto:



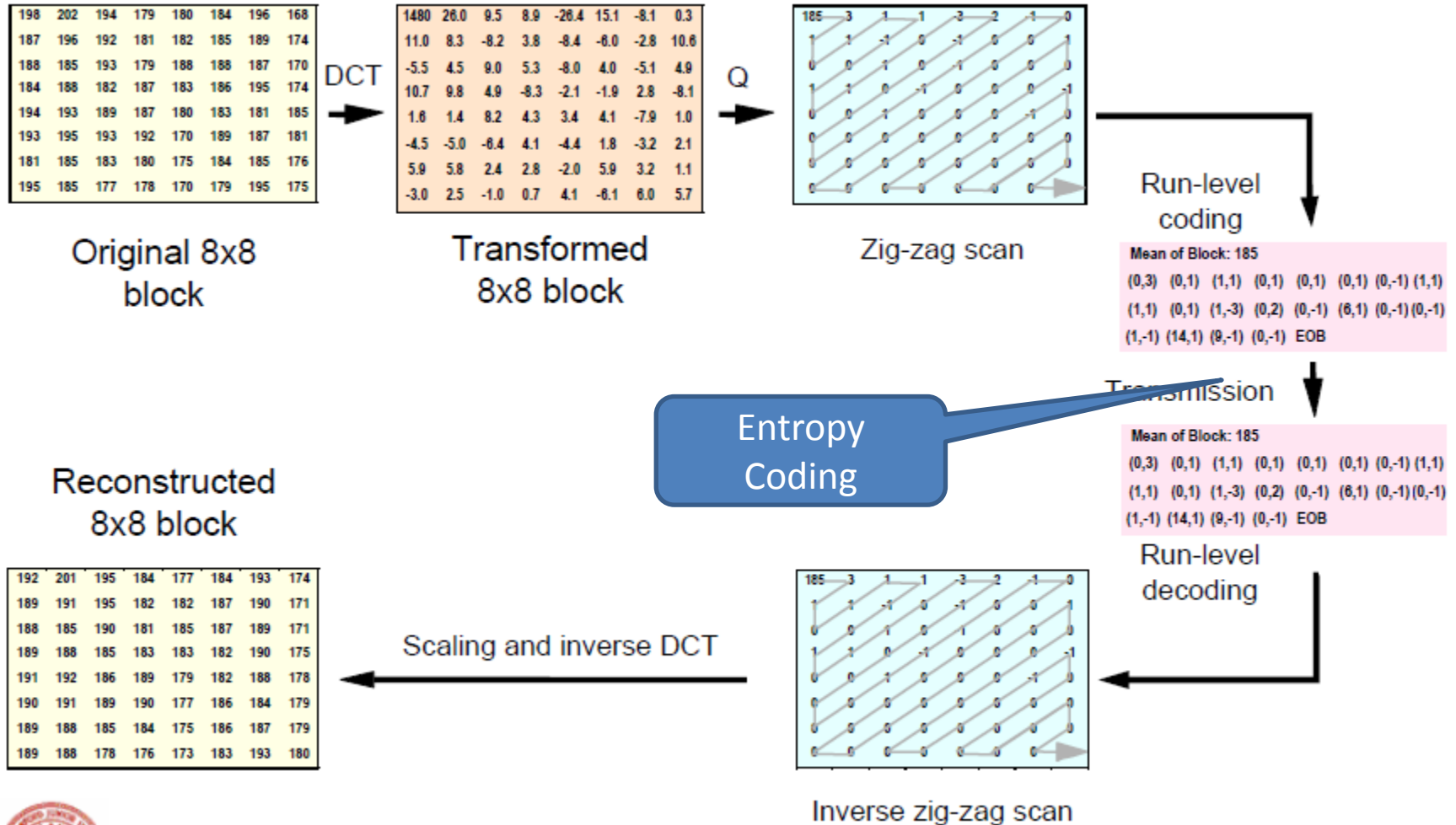
Zig-Zag-Scan

Efficient encoding of the position of non-zero transform coefficients: zig-zag-scan + run-level-coding



ordering of the transform coefficients by zig-zag-scan

Threshold Coding



Entropy of a Memoryless Source

- Let a memoryless source be characterized by an ensemble U_0 with:

Alphabet $\{ a_0, a_1, a_2, \dots, a_{K-1} \}$

Probabilities $\{ P(a_0), P(a_1), P(a_2), \dots, P(a_{K-1}) \}$

- Shannon: information conveyed by message " a_k ":

$$I(a_k) = -\log(P(a_k))$$

- "Entropy of the source" is the average information contents:

$$H(U_0) = E\{I(a_k)\} = - \sum_{k=0}^{K-1} P(a_k) * \log(P(a_k))$$

- For „log“ = „log₂“ the unit is bits/symbol

Redundant Codes: Example

a_i	$P(a_i)$	redundant code	optimum code
a_1	0.500	00	0
a_2	0.250	01	10
a_3	0.125	10	110
a_4	0.125	11	111

$$H(U_0) = 1.75 \text{ bits}$$

$$\lambda_{av} = 2 \text{ bits}$$

$$\rho = 0.25 \text{ bits}$$

$$\lambda_{av} = 1.75 \text{ bits}$$

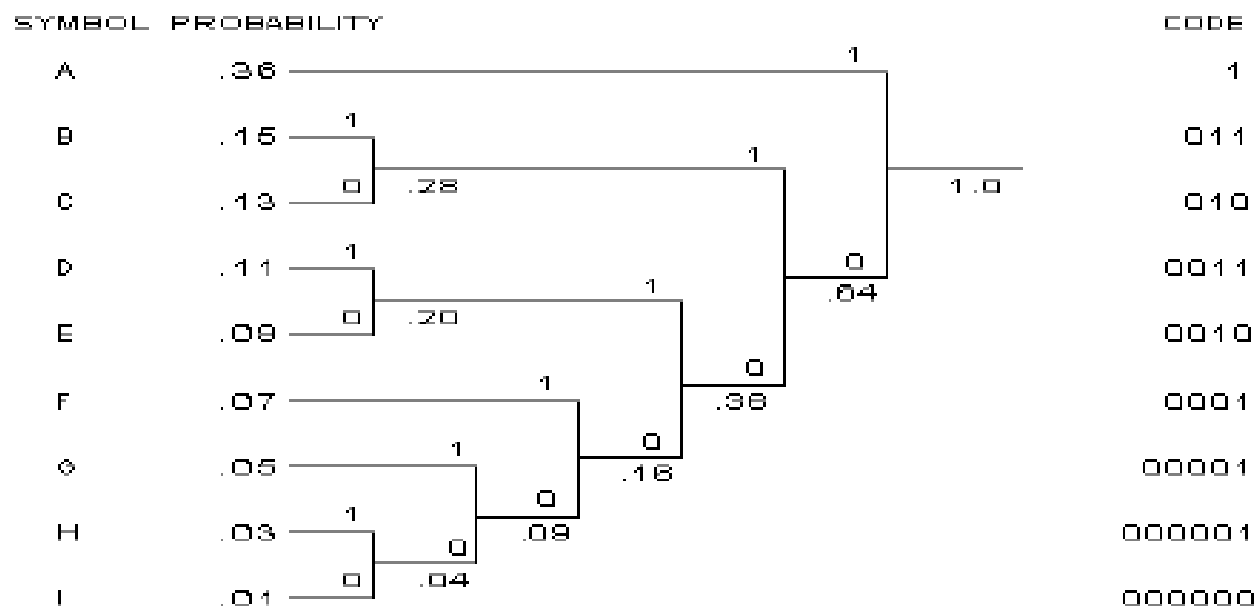
$$\rho = 0 \text{ bits}$$

Huffman Code

- Design algorithm for variable length codes proposed by D. A. Huffman (1952) always finds a code with minimum redundancy.
- Obtain code tree as follows:

- 1 Pick the two symbols with lowest probabilities and merge them into a new auxiliary symbol.
- 2 Calculate the probability of the auxiliary symbol.
- 3 If more than one symbol remains, repeat steps 1 and 2 for the new auxiliary alphabet.
- 4 Convert the code tree into a prefix code.

Huffman Code Example



Fixed length coding:	$R_{fixed} = 4$ bits/symbol
Huffman code:	$R_{Huffman} = 2.77$ bits/symbol
Entropy	$H(X) = 2.69$ bits/symbol
Redundancy of the Huffman code:	$\rho = 0.08$ bits/symbol

Example: Morse vs. Huffman

	%	Morse Code	Huffman Code
A	6.22	.-	1011
B	1.32	-...	010100
C	3.11	-.-.	10101
D	2.97	-..	01011
E	10.53	.	001
F	1.68	...-	110001
G	1.65	-.	110000
H	3.63	11001
I	6.14	..	1001
J	0.06	.—	01010111011
K	0.31	-.-	01010110
L	3.07	.-..	10100
M	2.48	—	00011

N	5.73	-.	0100
O	6.06	—	1000
P	1.87	.-.	00000
Q	0.10	-.-	0101011100
R	5.87	.-.	0111
S	5.81	...	0110
T	7.68	-	1101
U	2.27	..-	00010
V	0.70	...-	0101010
W	1.13	.-	000011
X	0.25	-..-	010101111
Y	1.07	-.—	000010
Z	0.06	—..	0101011101011

Figure 1: Morse and Huffman Codes for American-Roman Alphabet. The % column indicates the average probability (expressed in percent) of the letter occurring in English. The entropy $H(A)$ of this source is 4.14 bits. The average Morse codeword length is 2.5 symbols. Adding one more symbol for the letter separator and converting to bits yields an average codeword length of 5.56 bits. The average Huffman codeword length is 4.35 bits.

Typical DCT Coding Artifacts

DCT coding with increasingly coarse quantization, block size 8x8



quantizer stepsize
for AC coefficients: 25

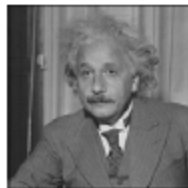
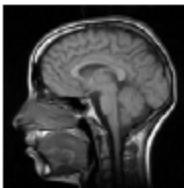
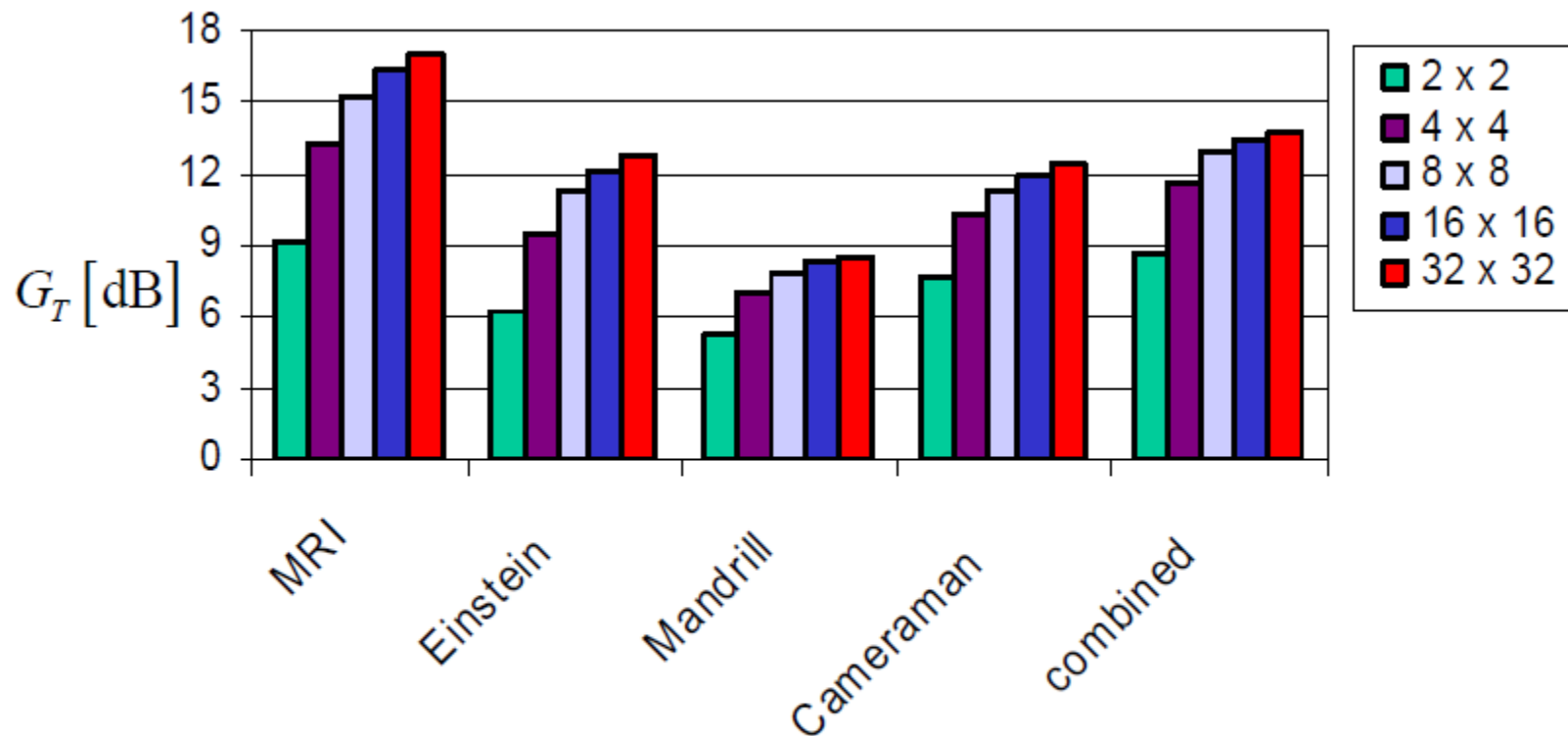


quantizer stepsize
for AC coefficients: 100



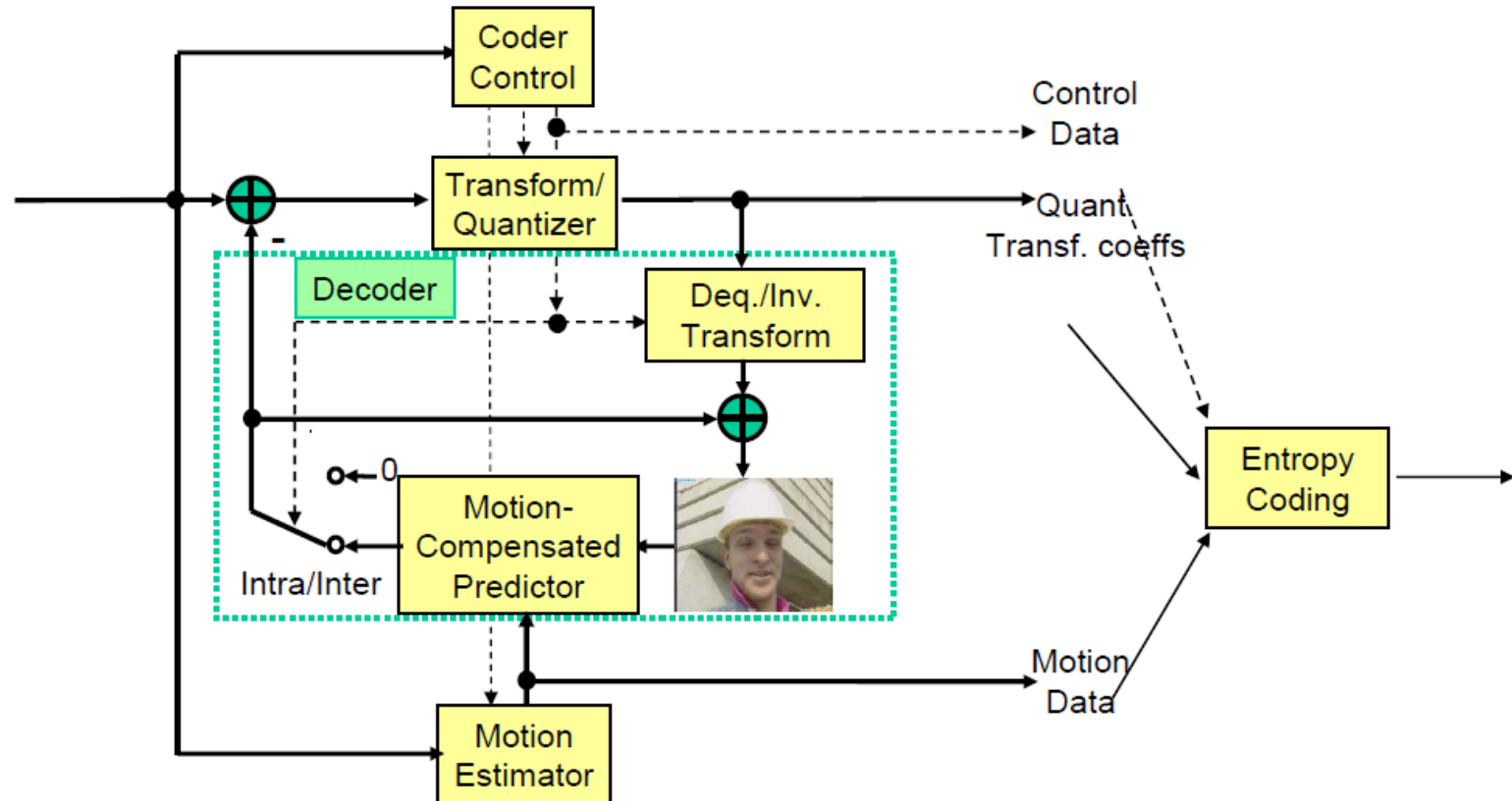
quantizer stepsize
for AC coefficients: 200

Influence of DCT Blocksize



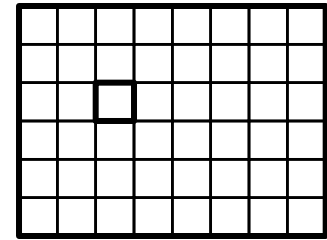
Motion-compensated Hybrid Coding

H.261, MPEG-1, MPEG-2, H.263, MPEG-4, H.264/AVC

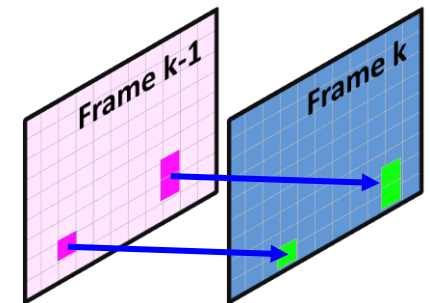
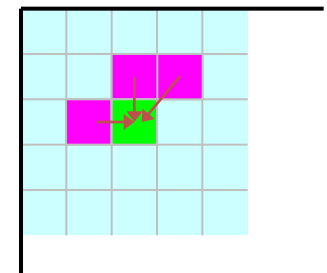


Prediction

- One sequence is encoded exploiting its **spatial** and **temporal** correlation
- As a first step, the picture is segmented into **macroblocks**
- A prediction is built for each macroblock
- The INTRA (spatial) encoding uses the neighboring macroblocks as source of prediction.
- The INTER (temporal) encoding uses the macroblocks belonging to the previous pictures as a source of prediction
- The INTER encoding is much more performant than the INTRA encoding but...
 - Scene changes



Frame k

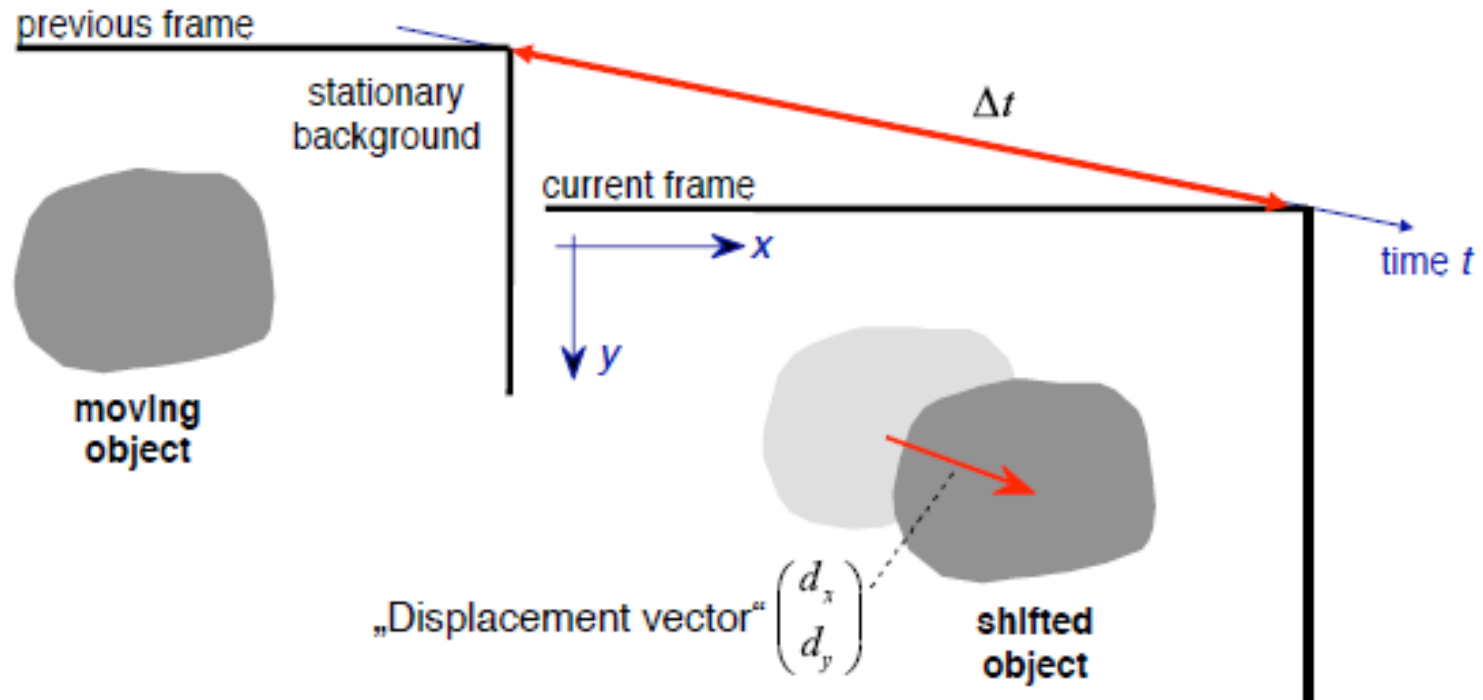


Interframe Coding of Video Signals

... exploits similarity of successive pictures



Motion Compensated Prediction Block Matching Algorithm



Prediction for the luminance signal $S(x, y, t)$ within the moving object:

$$\hat{S}(x, y, t) = S(x - d_x, y - d_y, t - \Delta t)$$

Performance Indicators

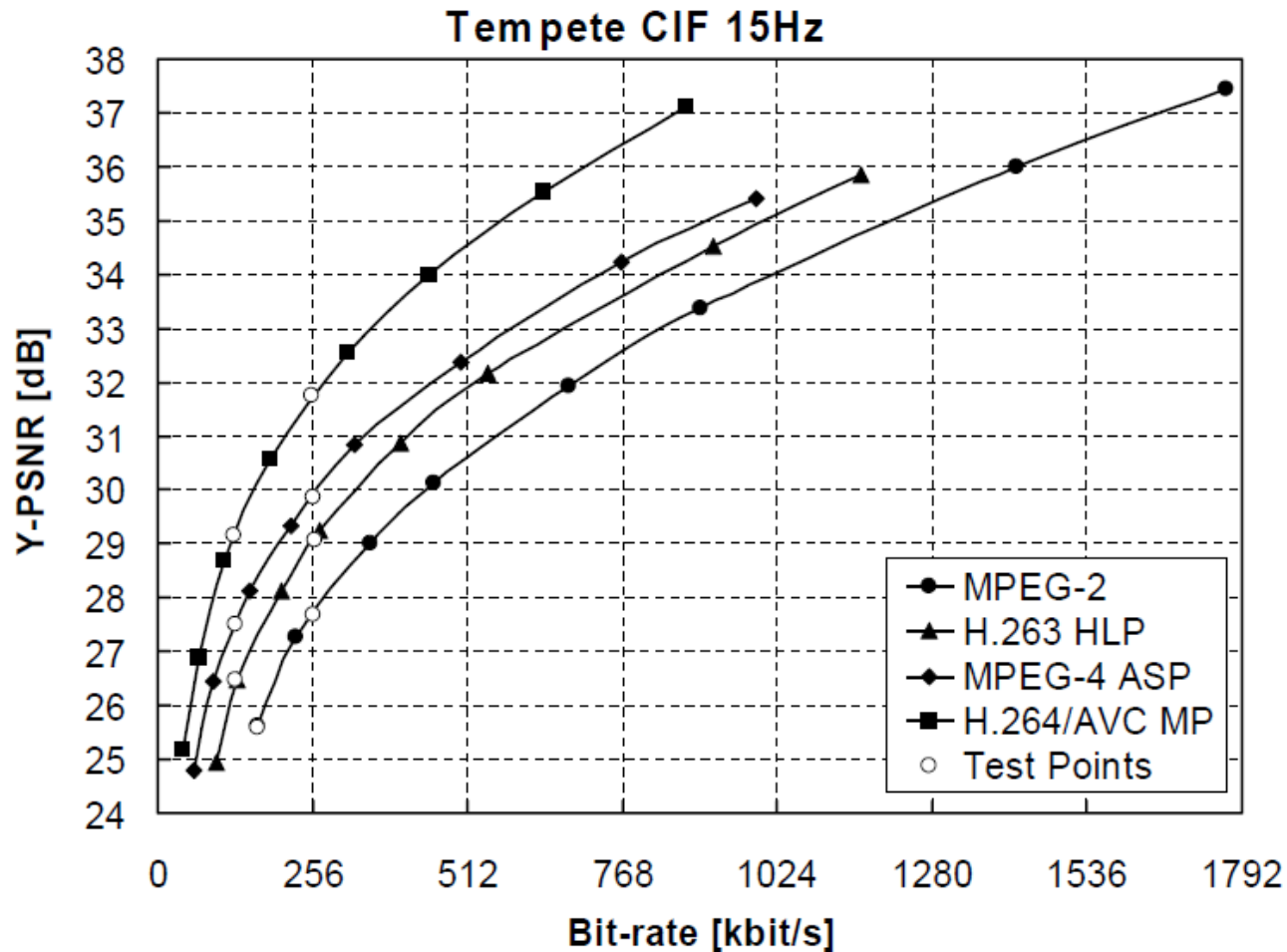
- Minimum Mean Square Error (MMSE)

$$\text{MSE}[n] = \frac{1}{M \cdot N \cdot |\mathcal{C}|} \sum_{c \in \mathcal{C}} \sum_{i=1}^N \sum_{j=1}^M \left[\mathbf{F}_n^{(c)}(i, j) - \mathbf{R}_n^{(c)}(i, j) \right]^2$$

- Peak Signal to Noise Ratio (PSNR)

$$\text{PSNR}[n] = 10 \cdot \log_{10} \frac{(2^q - 1)^2}{\text{MSE}[n]} \text{ [dB]}$$

Video Coding Test Results



[Wiegand, et al. 2003]

Soccer Video Sequences

- Soccer video streaming is one of the preferred contents
- The quality (as appreciated by the users) suffers from
 - Resolution downsampling (to fit the mobile device display)
 - High compression ratio (to match the available data rate)



Original uncompressed
sequence



Compressed
sequence

Quality Improvement: Deblocking Filter



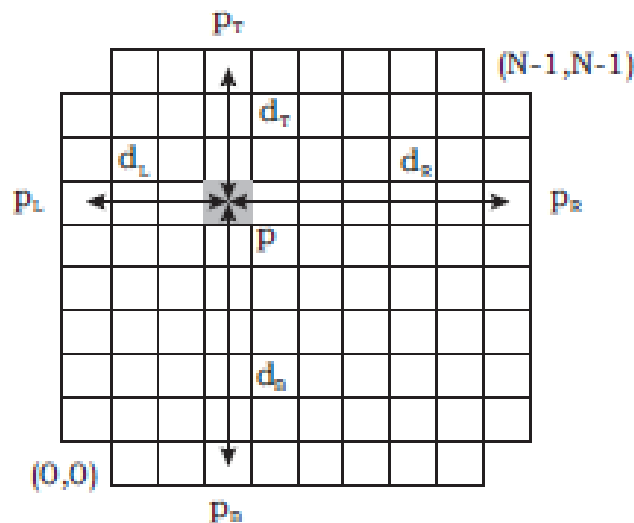
Without Filter



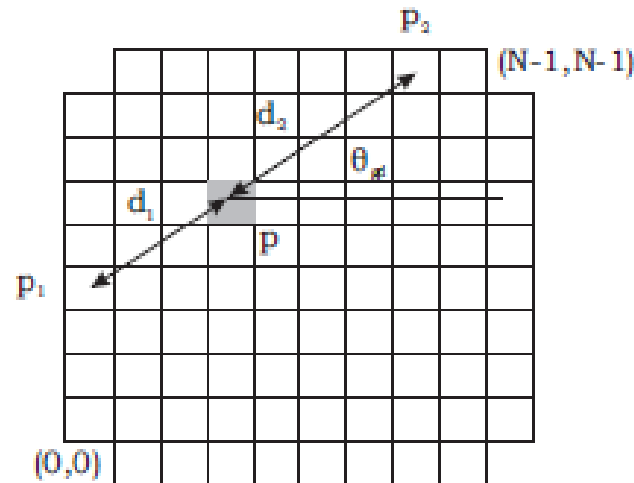
With H264/AVC Deblocking

Error Concealment Methods (not standardized)

- Spatial Concealment by interpolation



• weighted averaging



directional interpolation

Dissertation O.Nemethova

$$f_{i,j} = \frac{1}{d_1 + d_2} [d_2 f_{i_1,j_1} + d_1 f_{i_2,j_2}] ,$$

Spatial Interpolation



weighted averaging



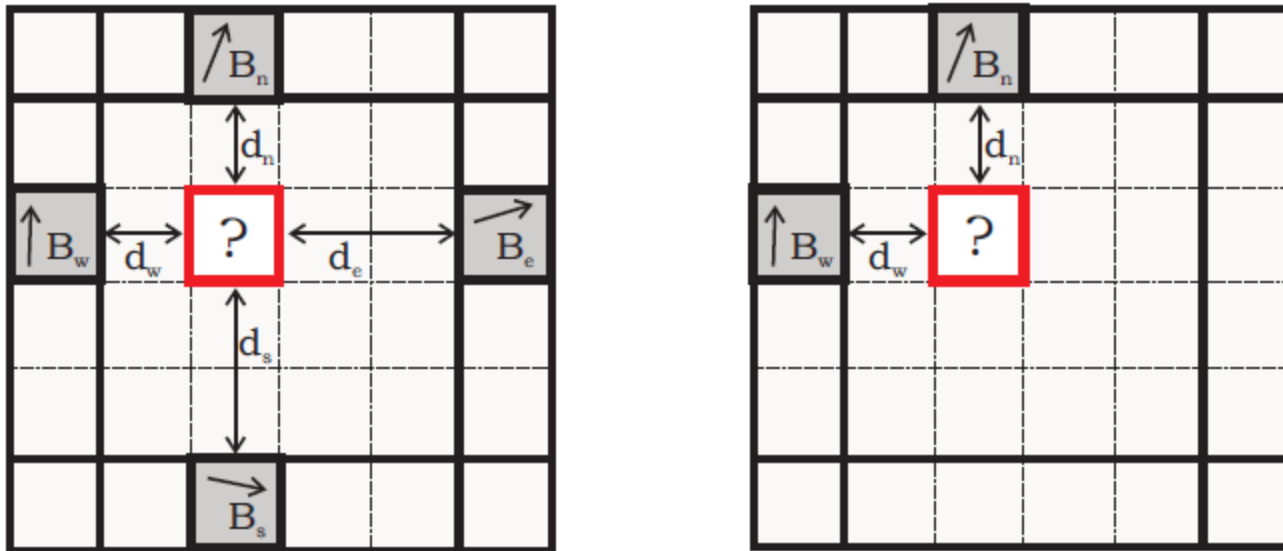
Directional interpolation



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Temporal Concealment

- Motion Vector estimation:



$$\widehat{mv}^{(i,j)} = \frac{d_e \underline{mv}_w^{(j)} + d_w \underline{mv}_e^{(j)} + d_n \underline{mv}_s^{(i)} + d_s \underline{mv}_n^{(i)}}{d_e + d_w + d_n + d_s}$$

Concealment Methods

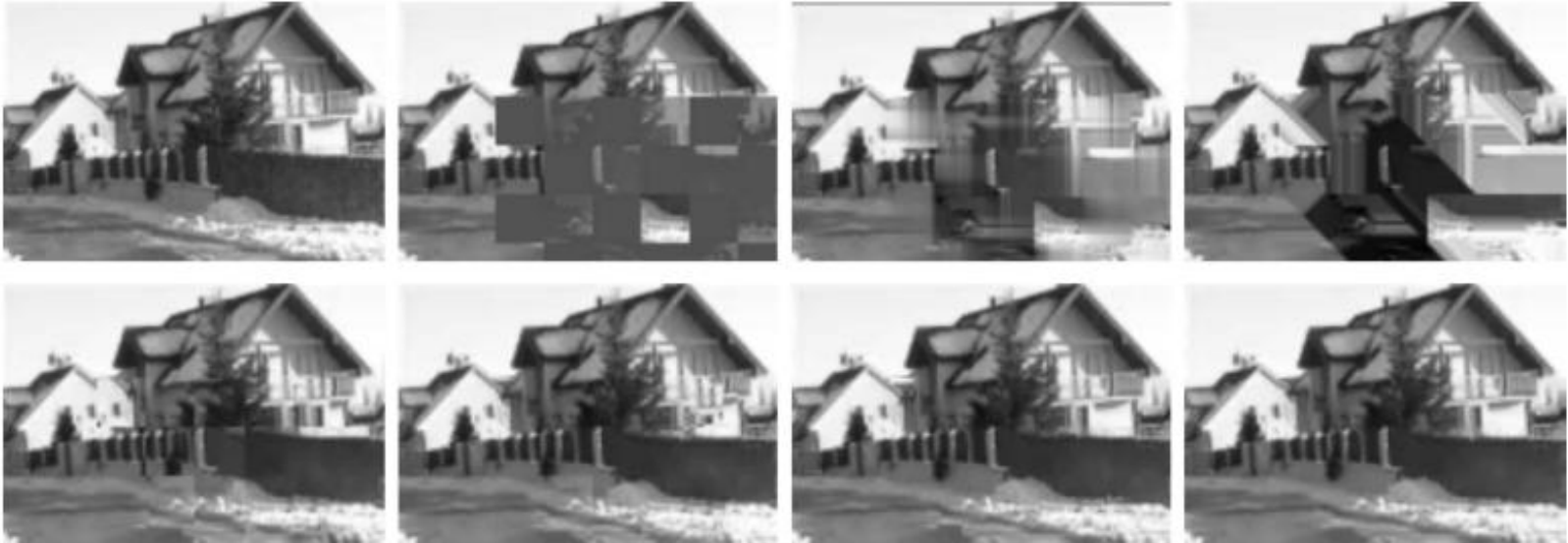


Figure 5.6 Screenshots of a part of an I frame in the 'panorama' sequence: compressed original (Y-PSNR= 35.86 dB), error pattern (Y-PSNR= 10.45 dB), weighted averaging (Y-PSNR= 18.09 dB), directional interpolation (Y-PSNR= 16.57 dB), copy-paste (Y-PSNR= 22.76 dB), boundary matching (Y-PSNR= 26.27 dB), 8×8 block matching with (Y-PSNR= 30.27 dB), 2×2 block matching with (Y-PSNR= 30.74 dB).

Concealment Methods

Method/case	1	2	3	4	5	6	7
Weighted averaging	31.71	41.52	36.86	36.03	37.63	40.08	37.52
Maximal smoothness FOD-US	41.90	41.66	36.71	36.81	37.61	38.41	37.83
Maximal smoothness FOD-EA	43.46	40.81	37.40	37.16	38.28	40.42	37.53
Maximal smoothness SOD-US	42.21	40.43	36.26	36.79	37.74	39.60	37.49
Directional interpolation	42.60	30.26	27.32	22.30	21.27	41.55	35.51
Segmented dir. int.	42.60	30.26	38.12	28.87	23.31	43.28	34.87
POCS, WA, 5 it.	42.85	33.28	28.33	21.25	23.39	40.86	35.95
POCS, dir. int. initial, 5 it.	42.86	31.12	38.39	20.30	22.41	40.92	35.03
Copy-paste	40.41	38.17	36.93	32.67	39.09	53.62	30.86
MV interpolation	43.56	43.94	35.58	28.16	44.32	48.06	30.12
MV interpolation SM	43.92	44.72	35.77	29.82	47.74	57.46	30.75
Boundary matching	43.92	44.28	37.02	33.21	47.86	57.46	30.68
Block matching	44.14	47.65	36.86	33.21	48.39	57.46	30.75
Model based (PCA)	45.24	41.11	37.72	25.53	43.21	59.88	31.41

Adaptive Method Selection

```
if scene change
    if clear edges AND enough neighbours
        method = directional interpolation
    else
        method = weighted averaging
else
    if I frame
        method = block matching
    else
        if MV correct
            method = decode without residuals
        else
            method = MV interpolation
```

Dissertation O.Nemethova

Video Transmission

- All the information needed for reconstructing the frame are stored
 - Type of encoding
 - Element used for prediction
 - Coefficients
 - ...

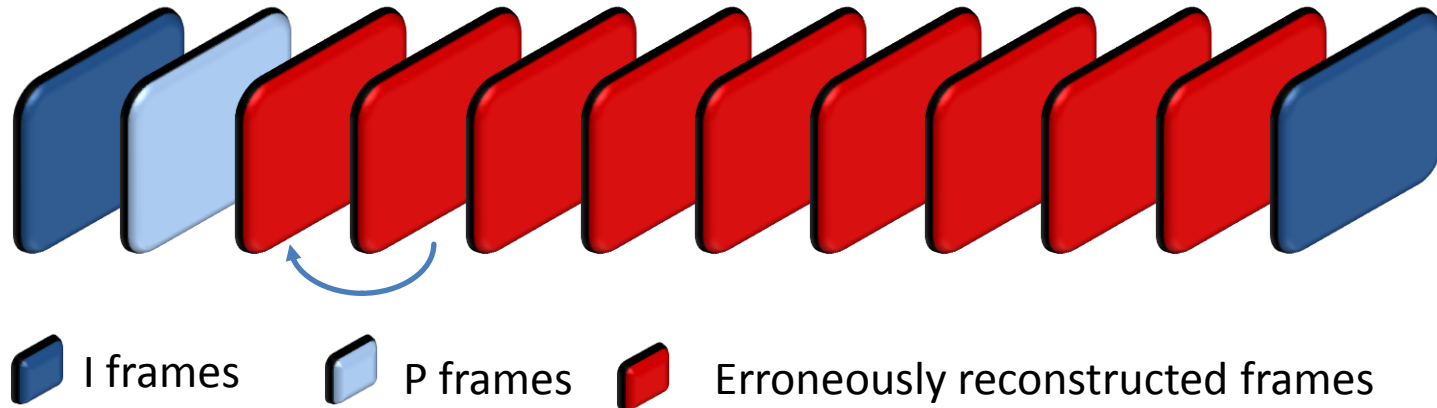


- The bitstream has to be segmented into smaller chunks (packets)



- Each data chunk is further encapsulated into a protocol stack

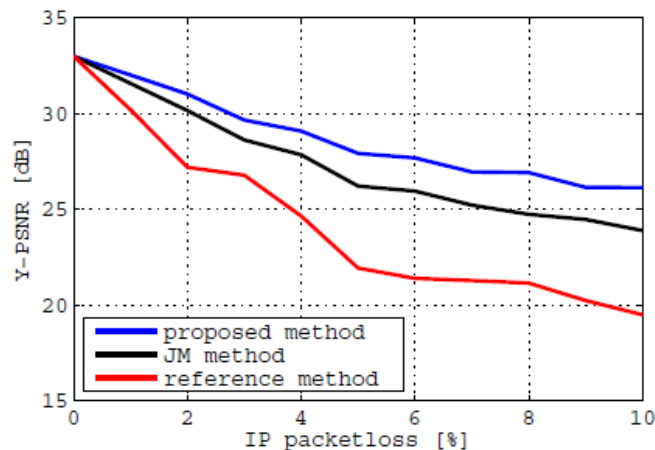
Effect of Errors at Sequence Level



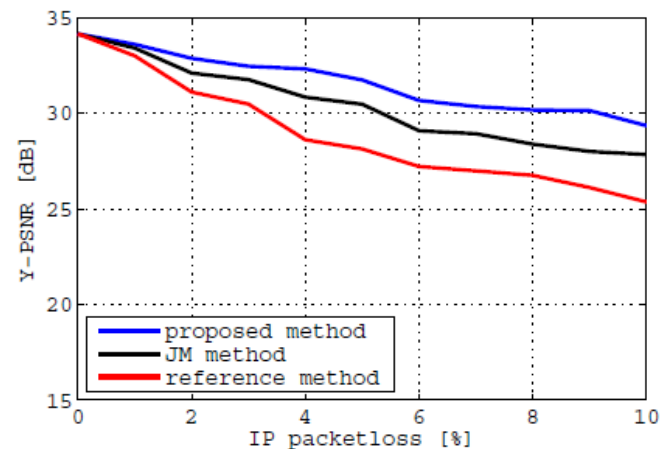
- If one packet is corrupted, the picture is incorrectly reconstructed
- The following pictures are using the damaged frame for temporal prediction
 - Even though their packets are correctly received, the corresponding frames are incorrectly reconstructed
- This effect, temporal error propagation, lasts until the following Intra frame

Comparison

- Reference: weighted averaging (spatial) and copy paste (temporal)
- JM: implemented in Joint Model software



(a) 'mobile'



(b) 'soccer'

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Further Reading

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