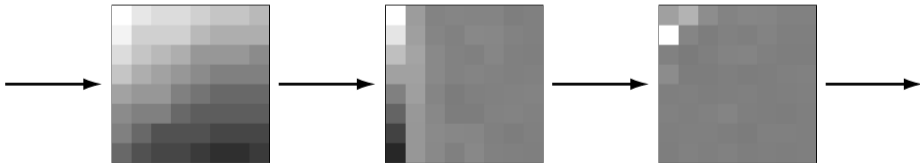
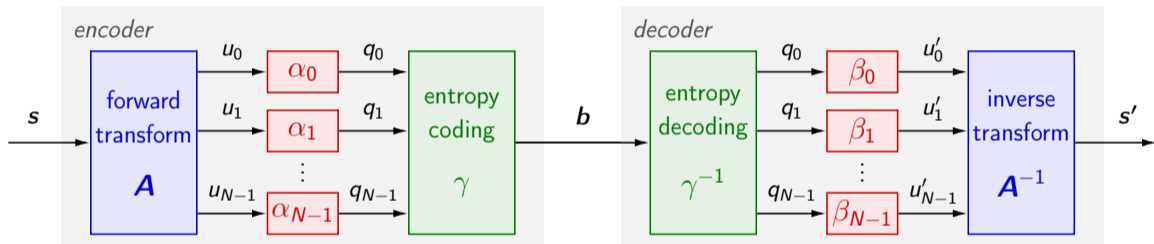


Transform Coding in Practice



Last Lectures: Basic Concept Transform Coding

- Transform reduces linear dependencies (correlation) between samples before scalar quantization
- For correlated sources: Scalar quantization in transform domain is more efficient



Encoder (block-wise)

- ➔ Forward transform: $\mathbf{u} = \mathbf{A} \cdot \mathbf{s}$
- ➔ Scalar quantization: $q_k = \alpha_k(u_k)$
- ➔ Entropy coding: $\mathbf{b} = \gamma(\{q_k\})$

Decoder (block-wise)

- ➔ Entropy decoding: $\{q_k\} = \gamma^{-1}(\mathbf{b})$
- ➔ Inverse quantization: $u'_k = \beta_k(q_k)$
- ➔ Inverse transform: $\mathbf{s}' = \mathbf{A}^{-1} \cdot \mathbf{u}'$

Last Lectures: Orthogonal Block Transforms

- Transform matrix has property: $\mathbf{A}^{-1} = \mathbf{A}^T$ (special case of unitary matrix: $\mathbf{A}^{-1} = (\mathbf{A}^*)^T$)

$$\mathbf{A} = \begin{bmatrix} \text{---} & b_0 & \text{---} \\ \text{---} & b_1 & \text{---} \\ \text{---} & b_2 & \text{---} \\ & \vdots & \\ \text{---} & b_{N-1} & \text{---} \end{bmatrix} \quad \mathbf{A}^{-1} = \mathbf{A}^T = \begin{bmatrix} | & | & | & & | \\ b_0 & b_1 & b_2 & \cdots & b_{N-1} \\ | & | & | & & | \end{bmatrix}$$

- ➔ Basis vectors \mathbf{b}_k (rows of \mathbf{A} , columns of $\mathbf{A}^{-1} = \mathbf{A}^T$) form an orthonormal basis
- ➔ Geometric interpretation: Rotation (and potential reflection) in N -dimensional signal space

Why Orthogonal Transforms ?

- Same MSE distortion in sample and transform space: $\|\mathbf{u}' - \mathbf{u}\|_2^2 = \|\mathbf{s}' - \mathbf{s}\|_2^2$
- ➔ **Minimum MSE in signal space can be achieved by minimization of MSE for each individual transform coefficient**

Last Lectures: Bit Allocation and High-Rate Approximations

Bit Allocation of Transform Coefficients

- Optimal bit allocation: Pareto condition

$$\frac{\partial}{\partial R_k} D_k(R_k) = -\lambda = \text{const} \quad \implies \quad \text{high rates: } D_k(R_k) = \text{const}$$

High-Rate Approximation

- High-rate distortion rate function for transform coding with optimal bit allocation

$$D(R) = \tilde{\varepsilon}^2 \cdot \tilde{\sigma}^2 \cdot 2^{-2R} \quad \text{with} \quad \tilde{\varepsilon}^2 = \left(\prod_k \varepsilon_k^2 \right)^{\frac{1}{N}}, \quad \tilde{\sigma}^2 = \left(\prod_k \sigma_k^2 \right)^{\frac{1}{N}}$$

- High-rate transform coding gain G_T and energy compaction measure G_{EC}

$$G_T = \frac{D_{SQ}(R)}{D_{TC}(R)} = \frac{\varepsilon_S^2 \cdot \sigma_S^2}{\tilde{\varepsilon}^2 \cdot \tilde{\sigma}^2}, \quad G_{EC} = \frac{\sigma_S^2}{\tilde{\sigma}^2} = \frac{\frac{1}{N} \sum_{k=0}^{N-1} \sigma_k^2}{\sqrt[N]{\prod_{k=0}^{N-1} \sigma_k^2}}$$

Last Lectures: Karhunen Loève Transform (KLT)

- Design criterion: Orthogonal transform \mathbf{A} that yields uncorrelated transform coefficients

$$\mathbf{C}_{UU} = \mathbf{A} \cdot \mathbf{C}_{SS} \cdot \mathbf{A}^T = \begin{bmatrix} \sigma_0^2 & 0 & \cdots & 0 \\ 0 & \sigma_1^2 & \cdots & 0 \\ \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & \cdots & \sigma_{N-1}^2 \end{bmatrix} \implies \mathbf{C}_{SS} \cdot \mathbf{b}_k = \sigma_k^2 \cdot \mathbf{b}_k$$

- Eigenvector equation for all basis vectors \mathbf{b}_k (rows of transform matrix \mathbf{A})
- Rows of KLT matrix \mathbf{A} are the unit-norm eigenvectors of \mathbf{C}_{SS}
- Transform coefficient variances σ_k^2 are the eigenvalues of \mathbf{C}_{SS}

$$\mathbf{A} = \begin{bmatrix} \text{---} \mathbf{b}_0 \text{---} \\ \text{---} \mathbf{b}_1 \text{---} \\ \vdots \\ \text{---} \mathbf{b}_{N-1} \text{---} \end{bmatrix} \quad \mathbf{C}_{UU} = \begin{bmatrix} \sigma_0^2 & 0 & \cdots & 0 \\ 0 & \sigma_1^2 & \cdots & 0 \\ \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & \cdots & \sigma_{N-1}^2 \end{bmatrix}$$

Last Lectures: Maximum Energy Compaction and Optimality

High-Rate Approximation for KLT and Gauss-Markov

- High-rate operational distortion-rate function

$$D_N(R) = \varepsilon^2 \cdot \sigma_S^2 \cdot (1 - \rho^2)^{\frac{N-1}{N}} \cdot 2^{-2R}$$

- High-rate transform coding gain: Increases with transform size N

$$G_T^N = G_{EC}^N = (1 - \rho^2)^{\frac{1-N}{N}} \implies G_T^\infty = \frac{1}{1 - \rho^2}$$

- For $N \rightarrow \infty$, gap to fundamental lower bound reduces to space-filling gain (1.53 dB)

On Optimality of KLT

- KLT yields uncorrelated transform coefficients and maximizes energy compaction G_{EC}
- KLT is the optimal transform for stationary Gaussian sources
- Other sources: Optimal transform is hard to find (iterative algorithm)

Transform Selection in Practice

Optimal Unitary Transform

- Stationary Gaussian sources: KLT
- General sources: Not straightforward to determine (typically KLT close to optimal)
- ➔ Signal dependent (may change due to signal instationarities)

Adaptive Transform Selection

- Determine transform in encoder, include transform specification in bitstream
- ➔ Increased side information may lead to sub-optimal overall coding efficiency
- ➔ Simple variant: Switched transforms (e.g., in H.266/VVC)

Signal-Independent Transforms

- Choose transform that provides good performance for variety of signals
- ➔ Not optimal, but often close to optimal for typical signal
- ➔ Most often used design in practice

Walsh-Hadamard Transform

- For transform sizes N that are positive integer powers of 2

$$\mathbf{A}_N = \frac{1}{\sqrt{2}} \begin{bmatrix} \mathbf{A}_{N/2} & \mathbf{A}_{N/2} \\ \mathbf{A}_{N/2} & -\mathbf{A}_{N/2} \end{bmatrix} \quad \text{with} \quad \mathbf{A}_1 = \begin{bmatrix} 1 \end{bmatrix}.$$

- Examples: Transform matrices for $N = 2$, $N = 4$, and $N = 8$

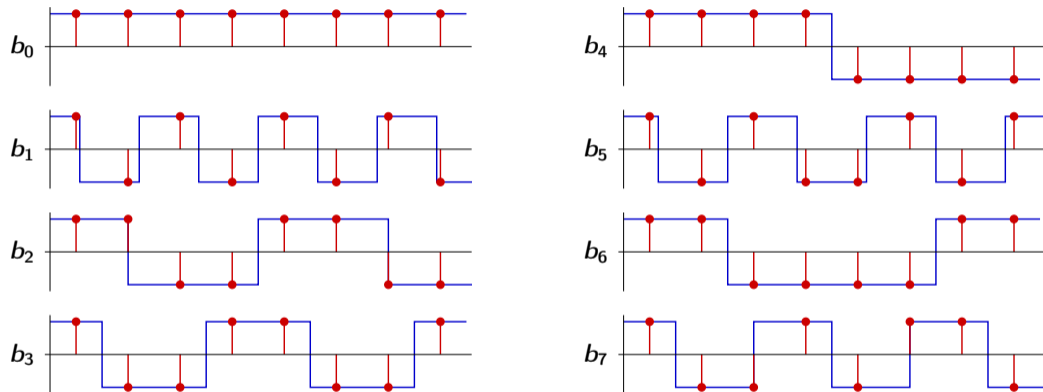
$$\mathbf{A}_2 = \frac{1}{\sqrt{2}} \begin{bmatrix} 1 & 1 \\ 1 & -1 \end{bmatrix}$$

$$\mathbf{A}_4 = \frac{1}{\sqrt{4}} \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -1 & 1 & -1 \\ 1 & 1 & -1 & -1 \\ 1 & -1 & -1 & 1 \end{bmatrix}$$

$$\mathbf{A}_8 = \frac{1}{\sqrt{8}} \begin{bmatrix} 1 & 1 & 1 & 1 & 1 & 1 & 1 & 1 \\ 1 & -1 & 1 & -1 & 1 & -1 & 1 & -1 \\ 1 & 1 & -1 & -1 & 1 & 1 & -1 & -1 \\ 1 & -1 & -1 & 1 & 1 & -1 & -1 & 1 \\ 1 & 1 & 1 & 1 & -1 & -1 & -1 & -1 \\ 1 & -1 & 1 & -1 & -1 & 1 & -1 & 1 \\ 1 & 1 & -1 & -1 & -1 & -1 & 1 & 1 \\ 1 & -1 & -1 & 1 & -1 & 1 & 1 & -1 \end{bmatrix}$$

→ Very simple orthogonal transform (only additions, subtractions, and final scaling)

Basis Functions of the WHT (Example for $N = 8$)



Media coding: Walsh-Hadamard transform with strong quantization

→ Piece-wise constant basis vectors yield subjectively disturbing artifacts

Discrete Version of the Fourier Transform

The Fourier Transform

- Fundamental transform used in mathematics, physics, signal processing, communications, ...
- Integral transform representing signal as integral of frequency components
- Forward and inverse transform are given by

$$\boxed{X(f) = \mathcal{F}\{x(t)\} = \int_{-\infty}^{\infty} x(t) \cdot e^{-2\pi i f t} dt} \quad \iff \quad \boxed{x(t) = \mathcal{F}^{-1}\{X(f)\} = \int_{-\infty}^{\infty} X(f) \cdot e^{2\pi i f t} df}$$

→ Basis functions are complex exponentials $b_f(t) = e^{2\pi i f t}$

Discrete Version of the Fourier Transform

- Fourier transform for finite discrete signals
- Could also be useful for coding of discrete signals
- Can be derived using sampling and windowing

Important Properties of the Fourier Transform

■ Linearity:

$$\mathcal{F}\{a \cdot h(t) + b \cdot g(t)\} = a \cdot H(f) + b \cdot G(f)$$

■ Scaling:

$$\mathcal{F}\{h(a \cdot t)\} = \frac{1}{|a|} \cdot H\left(\frac{f}{a}\right)$$

■ Translation:

$$\mathcal{F}\{h(t - t_0)\} = e^{-2\pi i t_0 f} \cdot H(f)$$

■ Modulation:

$$\mathcal{F}\{e^{2\pi i t f_0} \cdot h(t)\} = H(f - f_0)$$

■ Duality:

$$\mathcal{F}\{H(t)\} = h(-f)$$

■ Convolution:

$$\mathcal{F}\{h(t) * g(t)\} = \mathcal{F}\left\{\int_{-\infty}^{\infty} g(\tau) h(t - \tau) d\tau\right\} = H(f) \cdot G(f)$$

■ Multiplication:

$$\mathcal{F}\{h(t) \cdot g(t)\} = H(f) * G(f)$$

The Dirac Delta Function

Dirac Delta Function

- Not a function in traditional sense → **Dirac delta distribution**
- Can be thought of function with the following properties

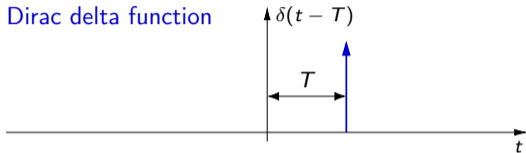
$$\delta(x) = \begin{cases} +\infty & : x = 0 \\ 0 & : x \neq 0 \end{cases} \quad \text{and} \quad \int_{-\infty}^{\infty} \delta(x) dx = 1$$

Important Properties

- Sifting:
$$\int_{-\infty}^{\infty} h(t) \delta(t - t_0) dt = h(t_0)$$
- Convolution:
$$h(t) * \delta(t - t_0) = \int_{-\infty}^{\infty} h(\tau) \delta(t - t_0 - \tau) d\tau = h(t - t_0)$$
- Sampling:
$$\int_{-\infty}^{\infty} h(t) \left(\sum_{k=-\infty}^{\infty} \delta(t - k \cdot t_0) \right) dt = \sum_{k=-\infty}^{\infty} h(k \cdot t_0)$$

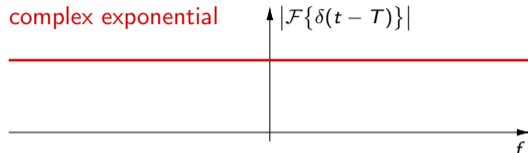
Selected Fourier Transform Pairs

Dirac delta function



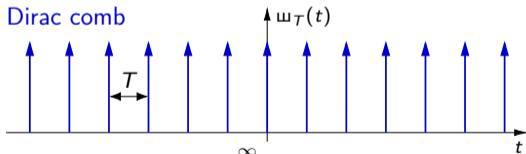
$$x(t) = \delta(t - T)$$

complex exponential



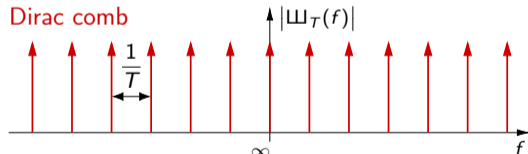
$$X(f) = e^{-2\pi ifT} = \cos(2\pi fT) + i \sin(2\pi fT)$$

Dirac comb



$$w_T(t) = \sum_{k=-\infty}^{\infty} \delta(t - kT)$$

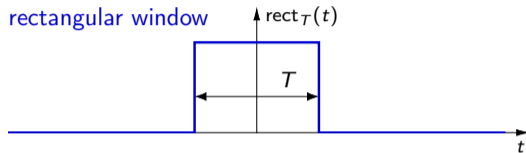
Dirac comb



$$W_T(f) = \sum_{k=-\infty}^{\infty} \delta(f - k/T)$$

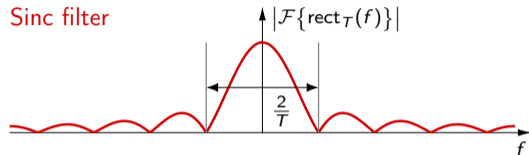
Selected Fourier Transform Pairs

rectangular window



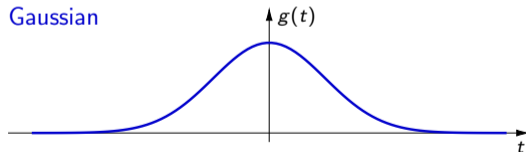
$$\text{rect}_T(t) = \begin{cases} 1 & : |t| \leq T/2 \\ 0 & : |t| > T/2 \end{cases}$$

Sinc filter



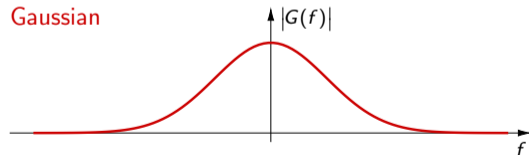
$$\mathcal{F}\{\text{rect}_T(f)\} = \frac{1}{\pi f} \sin(\pi f T) = T \text{sinc}(fT)$$

Gaussian



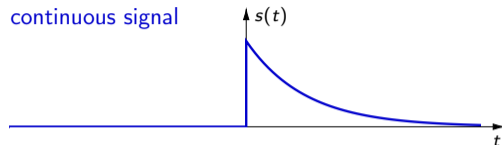
$$g(t) = e^{-\pi \cdot t^2} \quad \text{with} \quad \sigma_t^2 = \frac{1}{2\pi}$$

Gaussian

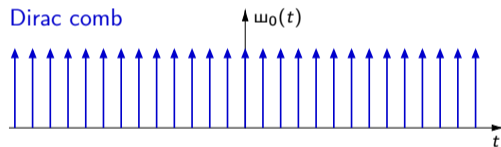


$$G(f) = e^{-\pi \cdot f^2} = g(f)$$

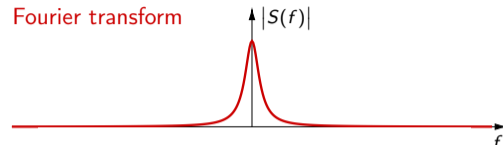
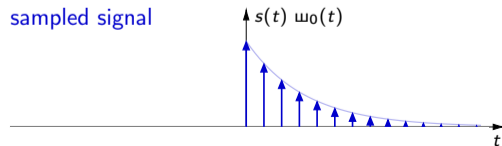
Derivation of Discrete Fourier Transform: (1) Sampling of Signal



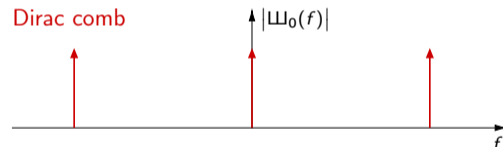
× (multiplication)



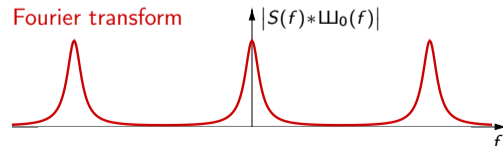
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* (convolution)

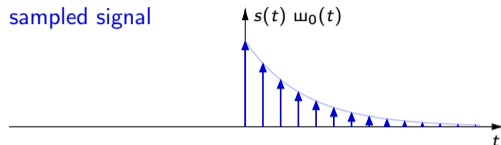


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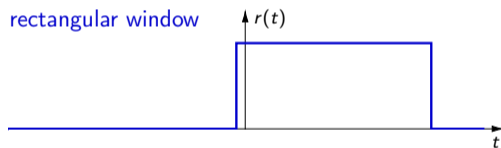
Derivation of Discrete Fourier Transform: (2) Time Restriction

sampled signal



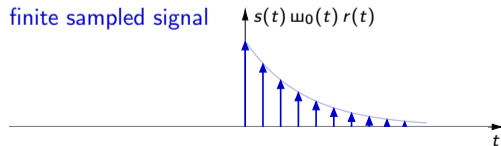
× (multiplication)

rectangular window

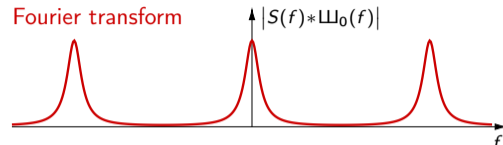


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finite sampled signal

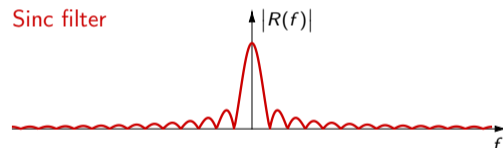


Fourier transform



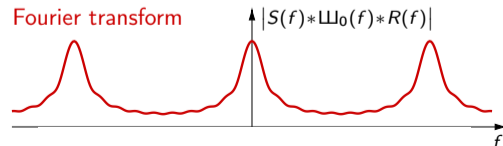
* (convolution)

Sinc filter

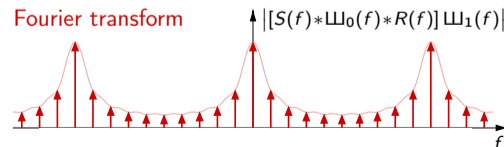
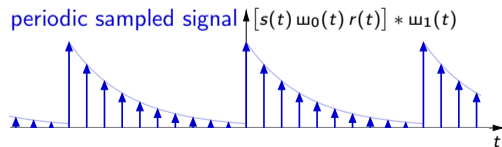
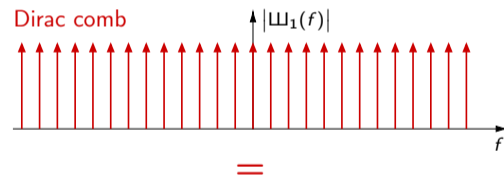
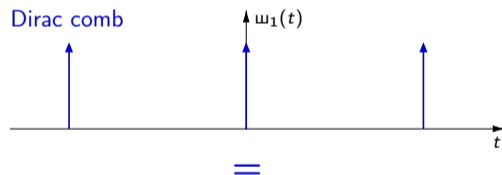
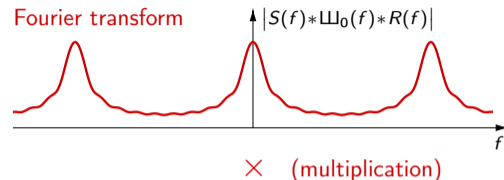
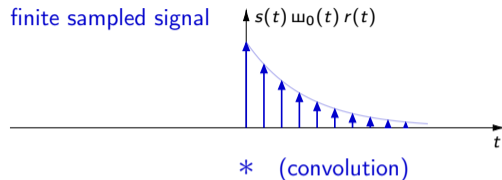


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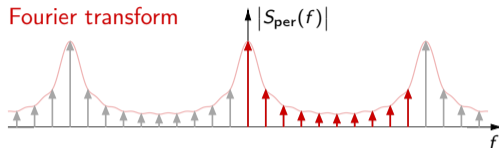
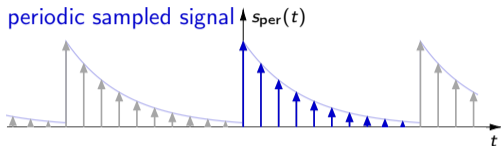
Fourier transform



Derivation of Discrete Fourier Transform: (3) Sampling of Spectrum



The Discrete Fourier Transform



→ N samples are represented by N complex Fourier coefficients

Discrete Fourier Transform

■ Forward and inverse transform are given by

$$u[k] = \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{2\pi kn}{N}}$$

and

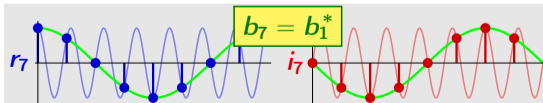
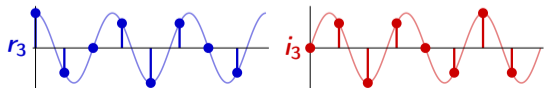
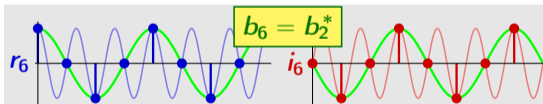
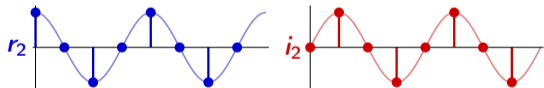
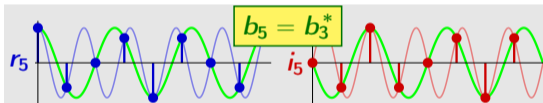
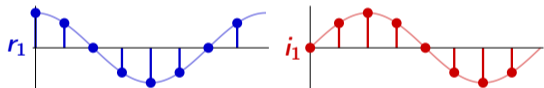
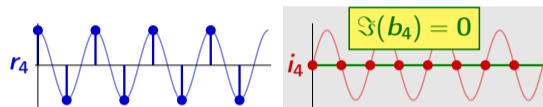
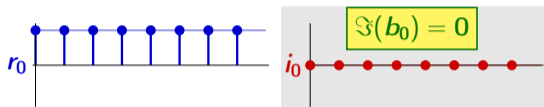
$$s[n] = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} u[k] \cdot e^{i \frac{2\pi kn}{N}}$$

→ Unitary transform that produces complex transform coefficients

→ Basis vectors are sampled complex exponentials

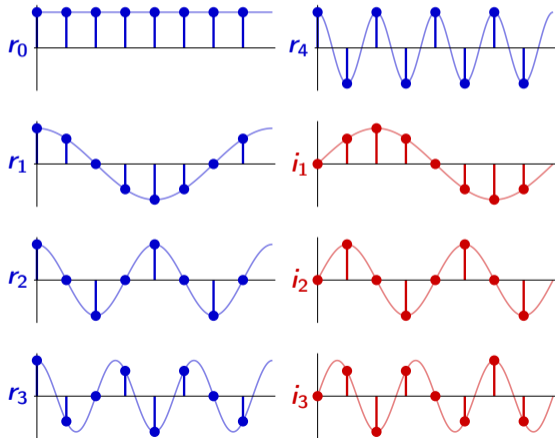
Complex Basis Functions of the DFT (Example for $N = 8$)

$$b_k[n] = \frac{1}{\sqrt{N}} e^{i\frac{2\pi k}{N}n} = \frac{1}{\sqrt{N}} \cos\left(\frac{2\pi k}{N}n\right) + i \cdot \frac{1}{\sqrt{N}} \sin\left(\frac{2\pi k}{N}n\right) = r_k[n] + i \cdot i_k[n]$$



Complex Basis Functions of the DFT (Example for $N = 8$)

$$\mathbf{b}_k[n] = \frac{1}{\sqrt{N}} e^{i\frac{2\pi k}{N}n} = \frac{1}{\sqrt{N}} \cos\left(\frac{2\pi k}{N}n\right) + i \cdot \frac{1}{\sqrt{N}} \sin\left(\frac{2\pi k}{N}n\right) = \mathbf{r}_k[n] + i \cdot \mathbf{i}_k[n]$$



DFT for Real Signals

- Symmetry of complex coefficients

$$u[k] = u^*[N - k]$$

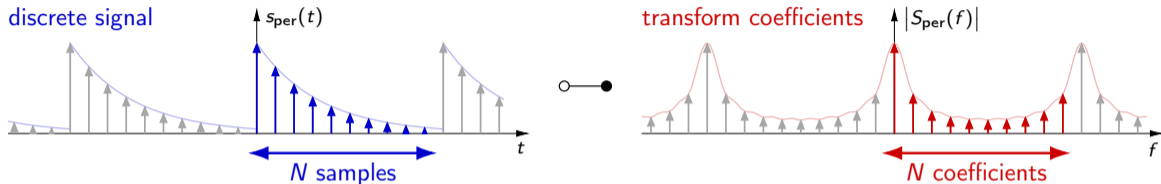
- Vanishing imaginary parts

$$k \in \{0, \frac{N}{2}\} : \Im\{u[k]\} = 0$$

➔ N real samples are mapped to N real coefficients

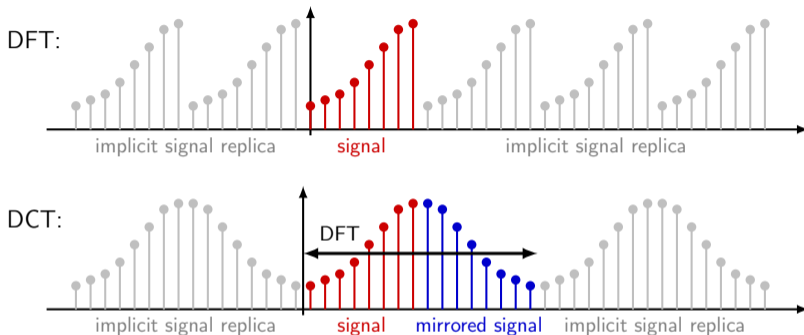
- Fast algorithm:
Fast Fourier transform (FFT)

Disadvantage of DFT for Transform Coding



- Sampling of frequency spectrum causes **implicit periodic signal extension**
- Often: Large differences between left and right signal boundary
- Large difference reduces rate of convergence of Fourier series
- **Strong quantization yields significant high-frequency artefacts**

Overcome DFT Disadvantage: Discrete Cosine Transform



Idea of Discrete Cosine Transform (DCT)

- Introduce mirror symmetry (different possibilities)
- Apply DFT of approximately double size (or four times the size)
- ➔ **No discontinuities in periodic signal extension**
- ➔ Ensure symmetry around zero: **Only cosine terms**

Discrete Trigonometric Transforms (DTTs)

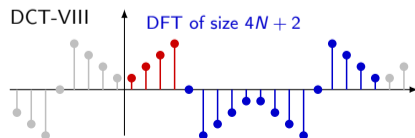
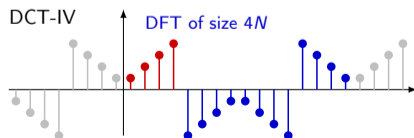
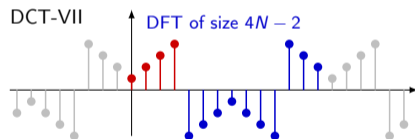
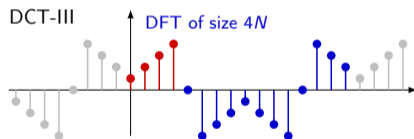
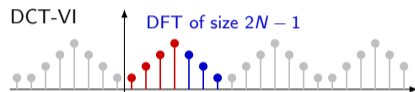
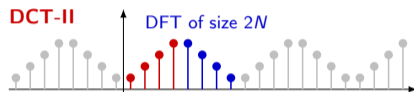
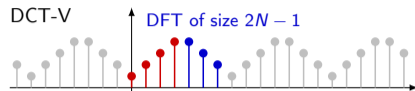
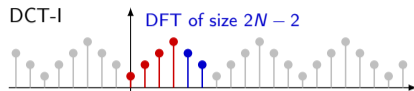
Discrete Cosine Transforms (DCTs)

- Introduce mirror symmetry around zero and apply DFT of larger size
 - Imaginary sine terms get eliminated
 - Only cosine terms remain
- 8 possibilities: DCT-I to DCT-VIII
 - 2 cases for left side: Whole sample or half-sample symmetry
 - 4 cases for right side: Whole sample or half-sample symmetry or anti-symmetry
- Most relevant case: **DCT-II** (half-sample symmetry at both sides)

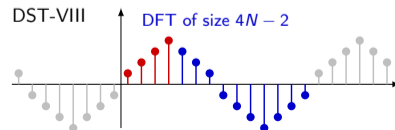
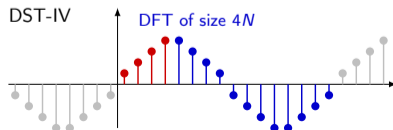
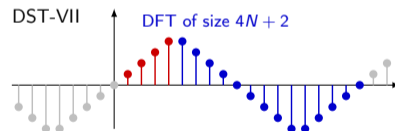
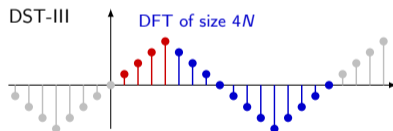
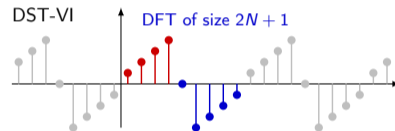
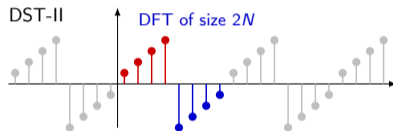
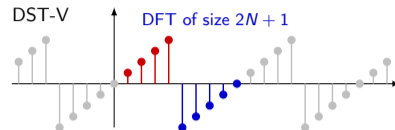
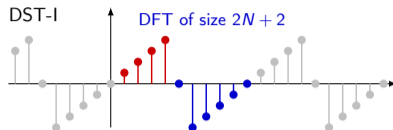
Discrete Sine Transforms (DSTs)

- Introduce anti-symmetry around zero and apply DFT of larger size
 - Real cosine terms get eliminated
 - Only imaginary sine terms remain
- Similarly as for DCT: 8 possibilities (DST-I to DST-VIII)

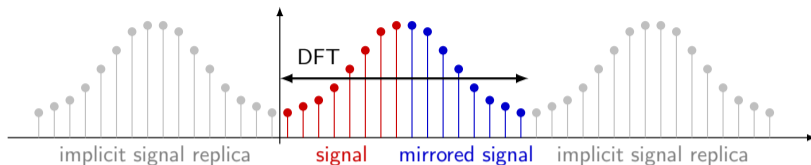
The Discrete Cosine Transform (DCT) Family



The Discrete Sine Transform (DST) Family



Derivation of the Discrete Cosine Transform of Type II (DCT-II)



Signal for applying the DFT

- Given: Discrete signal $s[n]$ of size N (i.e., $0 \leq n < N$)
- Mirror signal with sample repetition at both sides (size $2N$)

$$s^m[n] = \begin{cases} s[n] & : 0 \leq n < N \\ s[2N - n - 1] & : N \leq n < 2N \end{cases}$$

- Ensure symmetry around zero by adding half-sample shift

$$s^+[n] = s^m[n - 1/2] = \begin{cases} s[n - 1/2] & : 0 \leq n < N \\ s[2N - n - 3/2] & : N \leq n < 2N \end{cases}$$

→ Apply DFT of size $2N$ to new signal $s^+[n]$

Derivation of the Discrete Cosine Transform of Type II (DCT-II)

$$s^+[n] = \begin{cases} s[n - 1/2] & : 0 \leq n < N \\ s[2N - n - 3/2] & : N \leq n < 2N \end{cases}$$

→ DFT of size $2N$:

$$u^+[k] = \frac{1}{\sqrt{(2N)}} \sum_{n=0}^{(2N)-1} s^+[n] \cdot e^{-i \frac{2\pi kn}{(2N)}} \quad \left(\begin{array}{l} s^+ \text{ only known at half-sample} \\ \text{positions} \rightarrow \text{use } m = n - 1/2 \end{array} \right)$$

$$= \frac{1}{\sqrt{2N}} \sum_{m=0}^{2N-1} s^+ \left[m + \frac{1}{2} \right] \cdot e^{-i \frac{\pi k}{N} \left(m + \frac{1}{2} \right)}$$

$$= \frac{1}{\sqrt{2N}} \left(\sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{\pi k}{N} \left(n + \frac{1}{2} \right)} + \sum_{m=N}^{2N-1} s[2N - m - 1] \cdot e^{-i \frac{\pi k}{N} \left(m + \frac{1}{2} \right)} \right)$$

$\downarrow n=2N-m-1$

$$= \frac{1}{\sqrt{2N}} \left(\sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{\pi k}{N} \left(n + \frac{1}{2} \right)} + \sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{\pi k}{N} \left(2N - n - \frac{1}{2} \right)} \right)$$

Derivation of the Discrete Cosine Transform of Type II (DCT-II)

■ Continue derivation

$$\begin{aligned}
 u^+[k] &= \frac{1}{\sqrt{2N}} \left(\sum_{n=0}^{N-1} s[n] \cdot e^{-i\frac{\pi k}{N}(n+\frac{1}{2})} + \sum_{n=0}^{N-1} s[n] \cdot e^{-i\frac{\pi k}{N}(2N-n-\frac{1}{2})} \right) \\
 &= \frac{1}{\sqrt{2N}} \left(\sum_{n=0}^{N-1} s[n] \cdot e^{-i\frac{\pi k}{N}(n+\frac{1}{2})} + \sum_{n=0}^{N-1} s[n] \cdot \underbrace{e^{-i2\pi k}}_1 \cdot e^{i\frac{\pi k}{N}(n+\frac{1}{2})} \right) \\
 &= \frac{1}{\sqrt{2N}} \sum_{n=0}^{N-1} s[n] \cdot \underbrace{\left(e^{-i\frac{\pi k}{N}(n+\frac{1}{2})} + e^{i\frac{\pi k}{N}(n+\frac{1}{2})} \right)}_{2 \cos\left(\frac{\pi k}{N}\left(n+\frac{1}{2}\right)\right)}
 \end{aligned}$$

➔ DFT of extended signal

$$u^+[k] = \sqrt{\frac{2}{N}} \cdot \sum_{n=0}^{N-1} s[n] \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

Derivation of the Discrete Cosine Transform of Type II (DCT-II)

- DFT of extended signal ($2N$ real samples) has $2N$ real transform coefficients

$$k = 0, \dots, 2N - 1 : \quad u^+[k] = \sqrt{\frac{2}{N}} \cdot \sum_{n=0}^{N-1} s[n] \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

- 1 Signal $s[n]$ is completely described by first N transform coefficients

$$k = 0, \dots, N - 1 : \quad u^+[k] = \sqrt{\frac{2}{N}} \cdot \sum_{n=0}^{N-1} s[n] \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

- 2 Basis functions of derived transform are orthogonal to each other, but don't have the same norm

→ Introduce factors α_k so that transform matrix becomes orthogonal

$$k = 0, \dots, N - 1 : \quad u[k] = \alpha_k \cdot \sum_{n=0}^{N-1} s[n] \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

Discrete Cosine Transform of Type II (DCT-II)

Specification of DCT-II

- Forward transform (DCT-II) and inverse transform (IDCT-II) are given by

$$u[k] = \alpha_k \sum_{n=0}^{N-1} s[n] \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right) \quad \text{and} \quad s[n] = \sum_{k=0}^{N-1} \alpha_k \cdot u[k] \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

with scaling factors

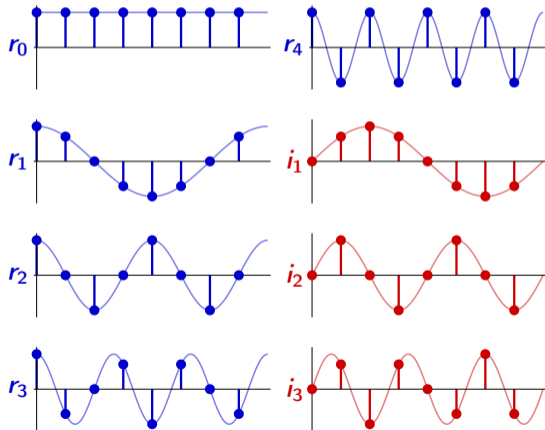
$$\alpha_k = \begin{cases} \sqrt{1/N} & : k = 0 \\ \sqrt{2/N} & : k \neq 0 \end{cases}$$

- The orthogonal transform matrix $\mathbf{A} = \{a_{kn}\}$ has the elements

$$a_{kn} = \alpha_k \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

Comparisons of DFT and DCT-II Basis Functions (Example for $N = 8$)

$$\text{DFT: } b_k[n] = \frac{1}{\sqrt{N}} e^{i \frac{2\pi k}{N} n} = r_k[n] + i \cdot i_k[n]$$



$$\text{DCT-II: } b_k[n] = \alpha_k \cdot \cos\left(\frac{\pi}{N} k \left(n + \frac{1}{2}\right)\right)$$

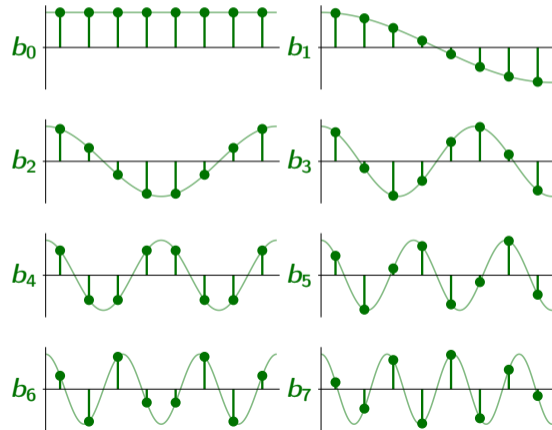


Image & Video Coding: 2D Transforms

Separable Transforms

- Successive 1D transforms of rows and columns of image block
- Separable forward and inverse transforms

$$\mathbf{u} = \mathbf{A} \cdot \mathbf{s} \cdot \mathbf{B}^T$$

and

$$\mathbf{s} = \mathbf{A}^T \cdot \mathbf{u} \cdot \mathbf{B}$$

with \mathbf{s} — $N \times M$ block of image samples

\mathbf{A} — $N \times N$ transform matrix (typically DCT-II)

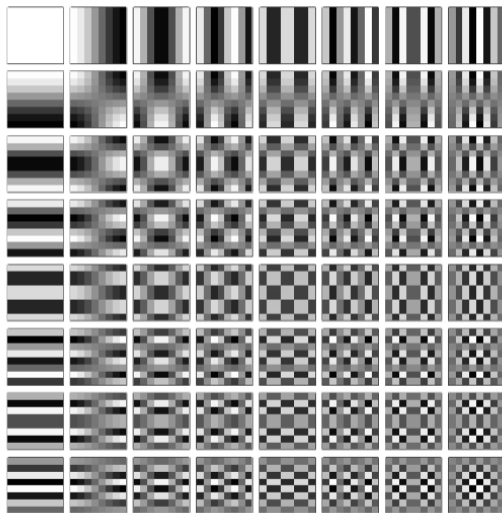
\mathbf{B} — $M \times M$ transform matrix (typically DCT-II)

\mathbf{u} — $N \times M$ block of transform coefficients

Great practical importance:

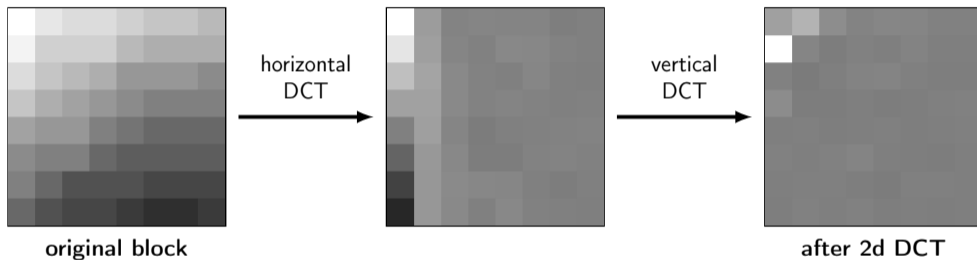
- Two matrix multiplications of size $N \times N$ instead of one multiplication of a vector of size $1 \times N^2$ with a matrix of size $N^2 \times N^2$
- Complexity reduction from $\mathcal{O}(N^4)$ to $\mathcal{O}(N^3)$ [also fast algorithms for DCT-II]

Example: Basis Images of Separable 8×8 DCT-II



Example: Separable DCT-II for 8×8 Image Block

Forward transform for 8×8 block of samples: $\mathbf{u} = \mathbf{A} \cdot \mathbf{s} \cdot \mathbf{A}^T$



Example calculation of 2d DCT-II:

- 1 Horizontal DCT of input block: $\mathbf{u}^* = \mathbf{s} \cdot \mathbf{A}^T$
- 2 Vertical DCT of intermediate result: $\mathbf{u} = \mathbf{A} \cdot \mathbf{u}^* = \mathbf{A} \cdot \mathbf{s} \cdot \mathbf{A}^T$

Practical Importance of DCT-II

Justification for usage of DCT-II

- Represents signal as weighted sum of frequency components
- Similar to KLT for highly correlated sources ($\rho \rightarrow 1$)
- Independent of source characteristics
- Fast algorithms for computing forward and inverse transform

DCT-II of size 8×8 is used in

- Image coding standard: JPEG
- Video coding standards: H.261, H.262/MPEG-2, H.263, MPEG-4 Visual

Integer approximation of DCT-II is used in

- Video coding standard H.264/AVC (4 × 4 and 8 × 8)
- Video coding standard H.265/HEVC (4 × 4, 8 × 8, 16 × 16, 32 × 32)
- New standardization project H.266/VVC (from 4 × 4 to 64 × 64, including non-square blocks)

Transform Coding in Practice

Orthogonal Transform

- Typically: DCT-II or integer approximation thereof (separable transform for blocks)
- Potential extension in H.266/VVC:
 - Switched transform of DCT/DST families (DCT-II, DST-VII, ...)
 - Non-separable transforms

Scalar Quantization

- Uniform reconstruction quantizers (or very similar designs)
- Bit allocation by using **same quantization step size** for all coefficients
- Usage of advanced quantization algorithms in encoder
- May use quantization weighting matrices for perceptual optimization

Entropy Coding of Quantization Indexes

- Zig-zag scan (or similar scan) for 2D transforms
- Simple: Run-level coding, run-level-last coding, or similar approach
- Better coding efficiency: Adaptive arithmetic coding

Bit Allocation in Practice (for Uniform Reconstruction Quantizers)

- Remember: Optimal bit allocation: Pareto condition

$$\frac{\partial D_k(R_k)}{\partial R_k} = \text{const}$$

- Pareto condition for high rates

$$D_k = \varepsilon_k^2 \cdot \sigma_k^2 \cdot 2^{-2R_k} \quad \Longrightarrow \quad D_k(R_k) = \text{const}$$

- High rate distortion approximation for URQs

$$D_k = \frac{1}{12} \Delta_k^2$$

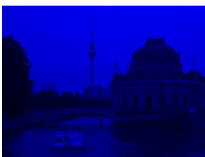
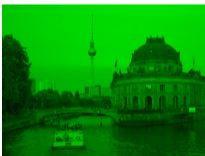
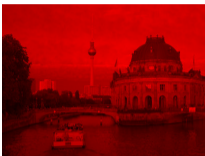
- Quantization step sizes for optimal bit allocation at high rates

$$D_k = \frac{1}{12} \Delta_k^2 = \text{const} \quad \Longrightarrow \quad \Delta_k = \text{const} = \Delta$$

- In practice, (nearly) optimal bit allocation is typically achieved by using the same quantization step size Δ for all transform coefficients**

Color Transform for Image & Video Coding

RGB



Color Transform for Compression

- Many versions (also depends on RGB color space)
- ➔ **Example:** RGB → YCbCr transform used in JPEG

$$\begin{bmatrix} Y \\ Cb - 128 \\ Cr - 128 \end{bmatrix} = \begin{bmatrix} 0.2990 & 0.5870 & 0.1140 \\ -0.1687 & -0.3313 & 0.5000 \\ 0.5000 & -0.4187 & -0.0813 \end{bmatrix} \cdot \begin{bmatrix} R \\ G \\ B \end{bmatrix}$$

$$\begin{bmatrix} R \\ G \\ B \end{bmatrix} = \begin{bmatrix} 1 & 0 & 1.4020 \\ 1 & -0.3441 & -0.7141 \\ 1 & 1.7720 & 0 \end{bmatrix} \cdot \begin{bmatrix} Y \\ Cb - 128 \\ Cr - 128 \end{bmatrix}$$

Energy Compaction for Example Image

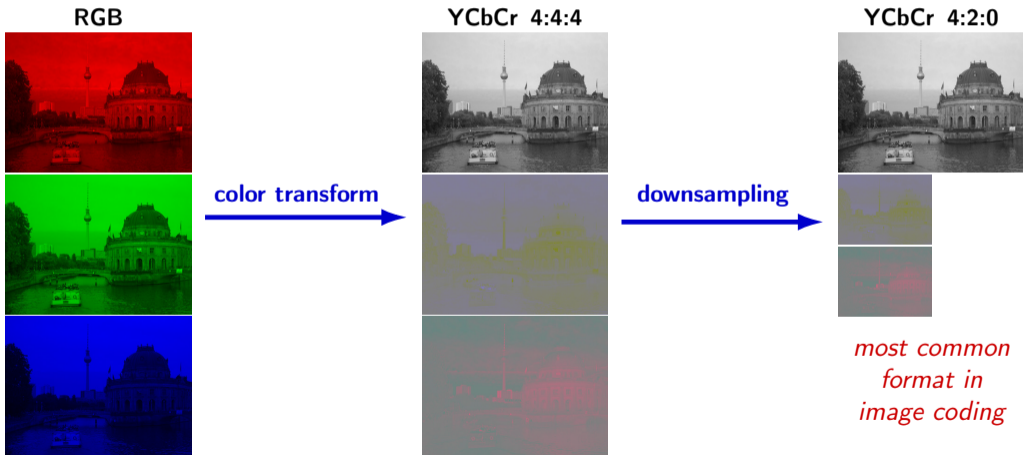
$$\begin{array}{ll} \sigma_R^2 = 3862.28 & \sigma_Y^2 = 3099.67 \\ \sigma_G^2 = 4250.44 & \sigma_{Cb}^2 = 83.94 \\ \sigma_B^2 = 5869.39 & \sigma_{Cr}^2 = 70.10 \end{array} \quad \rightarrow$$

YCbCr



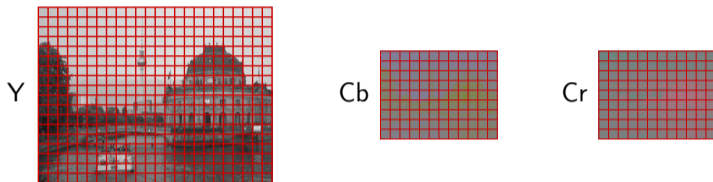
The YCbCr Chroma Sampling Format

- Human being are less sensitive to color differences (at same luminance)
- ➔ In most applications: Color difference components are downsampled

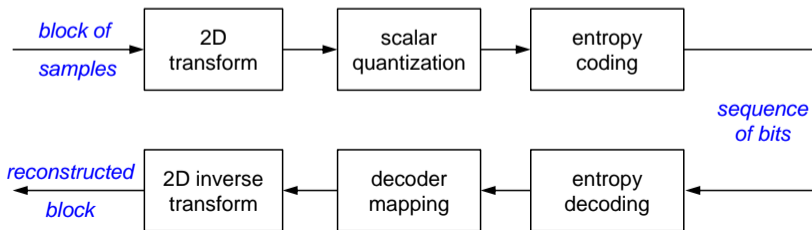


The Image Compression Standard JPEG

- Partition color components (Y, Cb, Cr) into blocks of 8×8 samples

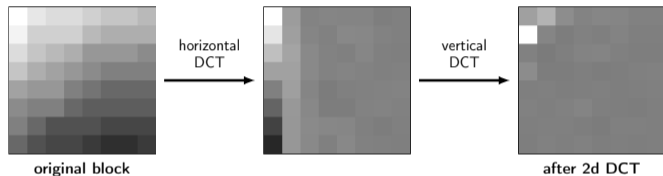


- Transform coding of 8×8 blocks of samples

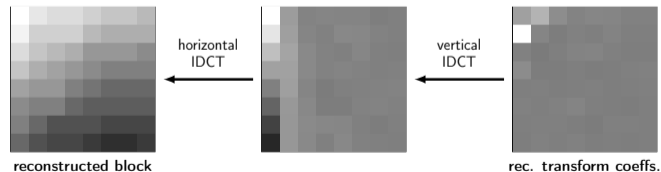


JPEG: Transform of Sample Blocks

- Separable DCT-II of size 8×8 (fast implementation possible)
- Forward transform (in encoder)

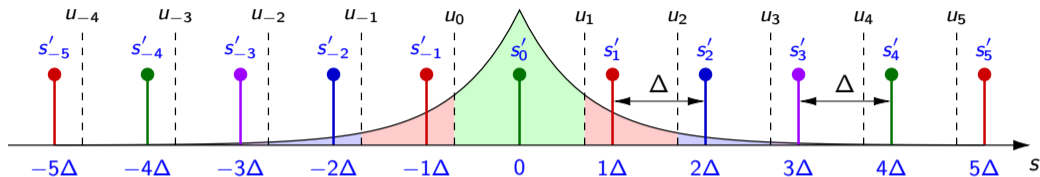


- Inverse transform (in decoder)



➔ Effect of transform: Compaction of signal energy (for typical blocks)

JPEG: Quantization



Uniform Reconstruction Quantizers

- Equally spaced reconstruction levels (indicated by step size Δ)
- Simple decoder mapping

$$t' = \Delta \cdot q$$

- Simplest (but not best) encoder:

$$q = \text{round}(t/\Delta)$$

- Better encoders use Lagrangian optimization (minimization of $D + \lambda R$)

→ **Quantization step size Δ determines tradeoff between quality and bit rate**

JPEG: Entropy Coding

2 Entropy Coding of Sequences of Quantization Indexes

- Often long sequences of zeros (in particular at end of sequence)
- Entropy coding should exploit this property

JPEG: Run-Level Coding (V2V code)

- Map sequence of symbols (transform coefficients) into (run,level) pairs, including a special end-of-block (eob) symbol

level: value of next non-zero symbol

run: number of zero symbols that precede next non-zero symbol

eob: all following symbols are equal to zero (end-of-block)

- Assign codewords to (run,level) pairs (including eob symbol)

- **Example:**

64 symbols:	5 3 0 0 0 1 0 1 0 0 1 0 0 0 0 0 0 0 0 0 ...
(run,level) pairs:	(0,5) (0,3) (3,1) (1,1) (2,1) (eob)

JPEG Compression Example

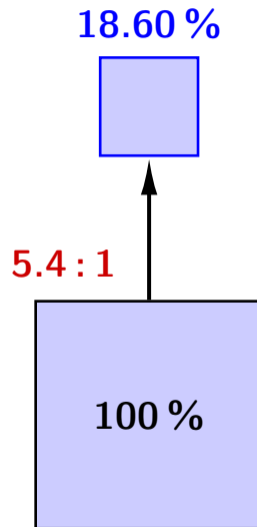
Original Image (960×720 image points, RGB: 2 MByte)



100 %

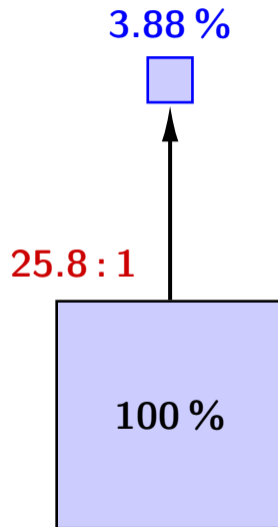
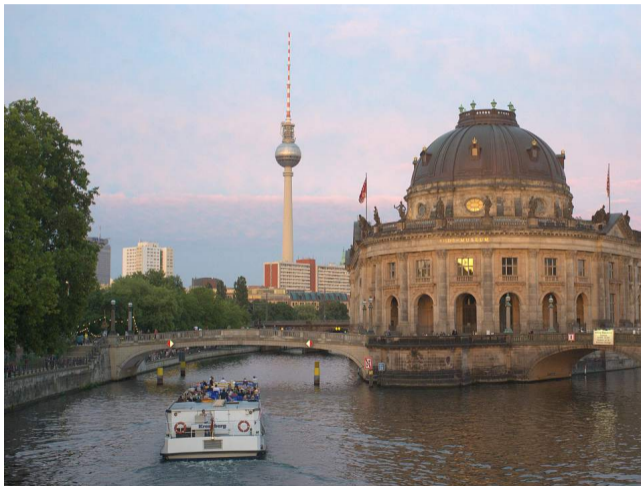
JPEG Compression Example

Lossy Compressed: JPEG (Quality 94)



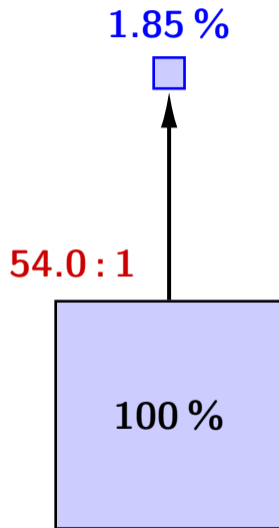
JPEG Compression Example

Lossy Compressed: JPEG (Quality 66)



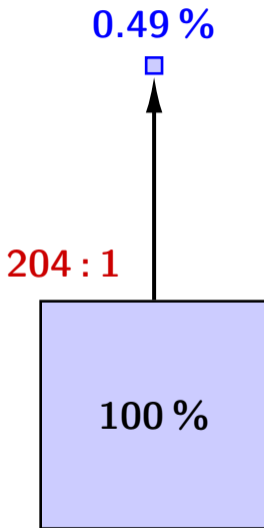
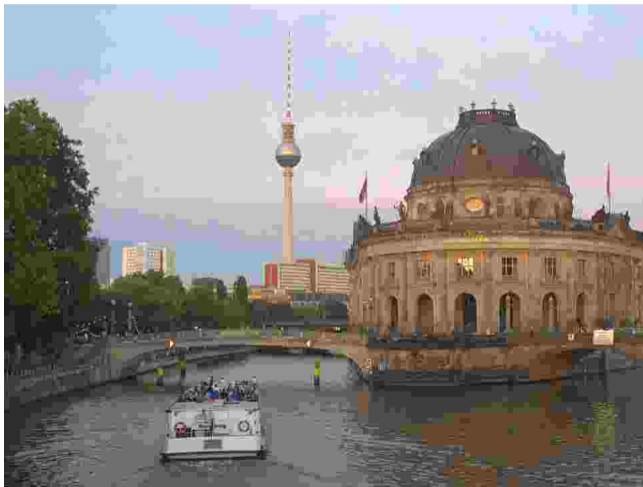
JPEG Compression Example

Lossy Compressed: JPEG (Quality 27)



JPEG Compression Example

Lossy Compressed: JPEG (Quality 6)



Audio Compression Example: MPEG-2 Advanced Audio Coding (AAC)

Main Component: Transform Coding of Sample Blocks

- Transform: Modified DCT for overlapping blocks
- Quantization: Scalar quantization with psycho-acoustic model
- Entropy Coding: Variant of Huffman coding

Linear Transform

- Audio signal is coded based on overlapping blocks of samples
- Transform: Modified discrete cosine transform (MDCT)

Quantization of Transform Coefficients

- Scalar quantization of transform coefficients (spectral coefficients)
- Utilization of psycho-acoustic models by noise shaping

Entropy Coding of Quantization Indexes

- Grouping and interleaving
- Huffman coding for tuples of n quantization indexes (n is variable)

Modified Discrete Cosine Transform (MDCT)

Forward Transform (MDCT)

- The forward transform maps $2N$ samples to N transform coefficients

$$u[k] = \frac{1}{\sqrt{N}} \sum_{n=0}^{2N-1} s[n] \cdot \cos \left(\frac{\pi}{N} \left(n + \frac{N+1}{2} \right) \left(k + \frac{1}{2} \right) \right)$$

Inverse Transform (IMDCT)

- The inverse transform maps N transform coefficients to $2N$ samples

$$x[n] = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} u[k] \cdot \cos \left(\frac{\pi}{N} \left(n + \frac{N+1}{2} \right) \left(k + \frac{1}{2} \right) \right)$$

Perfect Reconstruction

- Neighboring blocks of samples $s[n]$ overlap by 50% (at each side)
 - Perfect reconstruction of $s[n]$ is achieved by adding the inverse transformed blocks $x[n]$
- Property of time-domain aliasing cancellation

Summary of Lecture

Signal-Independent Transforms

- Walsh-Hadamard Transform (WHT): Perceptual disturbing artefacts
- Discrete Fourier Transform (DFT): Problem due to implicit periodic signal extension
- Discrete Trigonometric Transforms: Family of Sine and Cosine transforms

Discrete Cosine Transform of Type II (DCT-II)

- DFT of mirrored signal with half-sample symmetry at both sides
- Reduced blocking artifacts compared to DFT
- Good approximation of KLT for highly-correlated signals

Transform Coding in Practice

- Color transforms in image and video coding: RGB to YCbCr conversion
- JPEG image compression: 2D DCT-II + URQ + Run-level coding
- AAC audio compression: MDCT for overlapped blocks + scalar quantization + Huffman coding

Exercise 1: Correlation of Transform Coefficients

Given is a zero-mean AR(1) sources with a variance σ^2 and a correlation coefficient $\rho = 0.9$

Consider transform coding of blocks of 2 samples using the transform

$$\begin{bmatrix} u_{k,0} \\ u_{k,1} \end{bmatrix} = \frac{1}{\sqrt{2}} \begin{bmatrix} 1 & 1 \\ -1 & 1 \end{bmatrix} \cdot \begin{bmatrix} s_{2k} \\ s_{2k+1} \end{bmatrix},$$

where k represents the index of the transform block

- Determine the following variances and covariances of the transform coefficients (inside a block and between neighbouring blocks):

$$E\{ U_{k,0}^2 \} = ?$$

$$E\{ U_{k,0} U_{k+1,0} \} = ?$$

$$E\{ U_{k,1}^2 \} = ?$$

$$E\{ U_{k,1} U_{k+1,1} \} = ?$$

$$E\{ U_{k,0} U_{k,1} \} = ?$$

$$E\{ U_{k,0} U_{k+1,1} \} = ?$$

- Is it worth to exploit the correlations between the transform coefficients of neighboring block (e.g., for typical correlation factors of $\rho \approx 0.9$)?

Exercise 2: First Version of Lossy Image Codec (Implementation)

Implement a first lossy image codec for PPM images:

- 1** Use the source code of last weeks exercise as basis (see KVV)
- 2** Add some variant of entropy coding for the quantization indexes, for example:
 - Simple Rice coding or Exp-Golomb coding (see lossless codec example in KVV)
 - Adaptive binary arithmetic coding using a unary binarization (see lossless coding example in KVV)
 - ...
- 3** Implement an encoder that converts a PPM image into a bitstream file
- 4** Implement a corresponding decoder that converts a bitstream file into a PPM image
- 5** Test your encoder with some example images and multiple quantization step sizes
- 6** (Optional) Try to improve your codec by using the YCbCr color format
 - Implement an RGB-to-YCbCr transform before the actual encoding
 - Implement the inverse YCbCr-to-RGB transform after the actual decoding
 - Possible extension: Sub-sampling of chroma components