



# **Fundamentals of Multimedia**

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**Authored by Ze-Nian Li Mark S. Drew**

**Lecturer: Lu Dongming**



## 7、 Lossy Compression Algorithms

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# Content

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- ❑ **Distortion Measures**
- ❑ **The Rate-Distortion Theory**
- ❑ **Quantization**
- ❑ **Transform Coding**
- ❑ **Wavelet-Based Coding**



# 1. Distortion Measures

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# 1.1 Concept of Distortion

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## □ Distortion Measure

- A mathematical quantity: specifies **how close an approximation to its original**
- It's nature to think of the numerical difference
- When it comes to **image** data, difference may **not yield the intended result**
- Measures of **perceptual distortion**



## 1.2 Numerical Distortion Measures

- Many numerical distortion measures -- the most commonly used distortion measures are presented: **MSE**、**SNR**、**PSNR**

- Mean Square Error (MSE) :

$$\sigma^2 = \frac{1}{N} \sum_{n=1}^N (x_n - y_n)^2$$

- Average pixel difference

- Signal-to-Noise Ratio (SNR):

$$SNR = 10 \log_{10} \frac{\sigma_x^2}{\sigma_d^2}$$

- The size of the error relative to the signal

- Peak-Signal-to-Noise Ratio (PSNR):

$$PSNR = 10 \log_{10} \frac{x_{peak}^2}{\sigma_d^2}$$

- The size of the error relative to the peak value of the signal

## 1.2 Numerical Distortion Measures

- Examples of PSNR and corresponding images



original image



polluted by noise

PSNR=18.24



processed by noise filter

PSNR=39.5



## 2.The Rate-Distortion Theory

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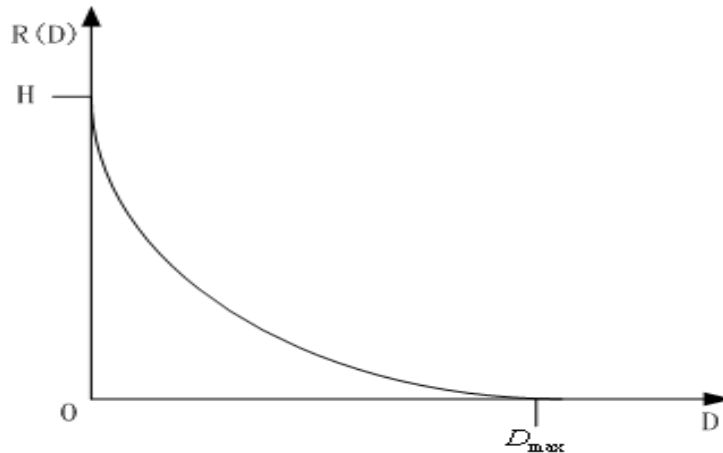


# 2.1 Concept

- Lossy compression always involves a **tradeoff between rate and distortion**
  - **Rate** -- the average number of bits required to represent each source symbol;
  - **$R(D)$**  note rate-distortion function;
- What is  $R(D)$ ?
  - $R(D)$  specifies **the lowest rate** at which the source data can be encoded while **keeping the distortion bounded above by  $D$**
  - **At  $D=0$** , no loss, so is the entropy of the source data
  - Describe **a fundamental limit** for the performance of a coding algorithm
  - Can be used to **evaluate the performance of different algorithm**

## 2.2 A Typical R-D Function

- A figure of a typical rate-distortion function



- $D=0$ , the entropy of the source data
- $R(D)=0$ , nothing coded
- For a given source, it's **difficult to find a closed-form analytic description** of the rate-distortion function



# 3. Quantization

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# 3.1 Functions of Quantization

- **Quantization: the heart** of any lossy scheme
  - Without quantization, **almost** no losing information
  - Reduce the number of distinct values via quantization
- Each quantizer has its unique **partition of the input range** and the set of output values.
  - Scalar quantizer
  - Vector quantizer



## 3.2 Uniform Scalar Quantization

- **Uniform scalar quantizer**
  - Partitions the input domain into **equally spaced intervals**
  - **Decision boundaries**: the **end points** of partition intervals
  - Output value: **midpoint** of the interval
  - Step size: the length of each interval
- **Two types of uniform scalar quantizer**
  - **midrise**: with an even number of output levels, one partition interval brackets zero;
  - **midtread**: odd number of output levels, zero is an output value.
- **The goal of a successful uniform quantizer**
  - **Minimize the distortion** for a given source input with a desired number of output values

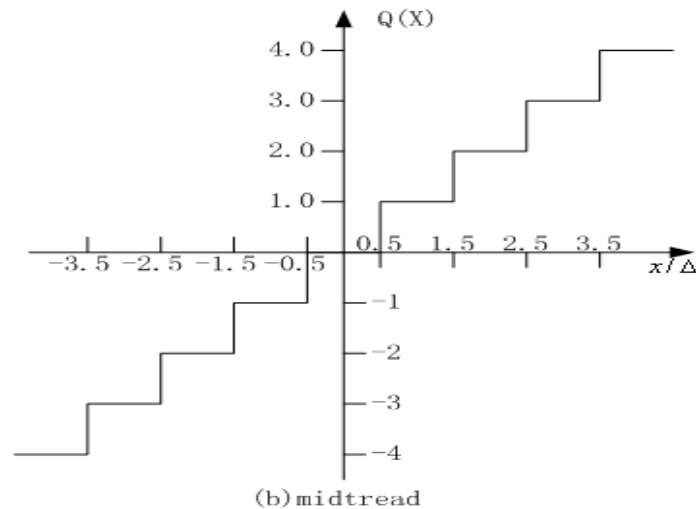
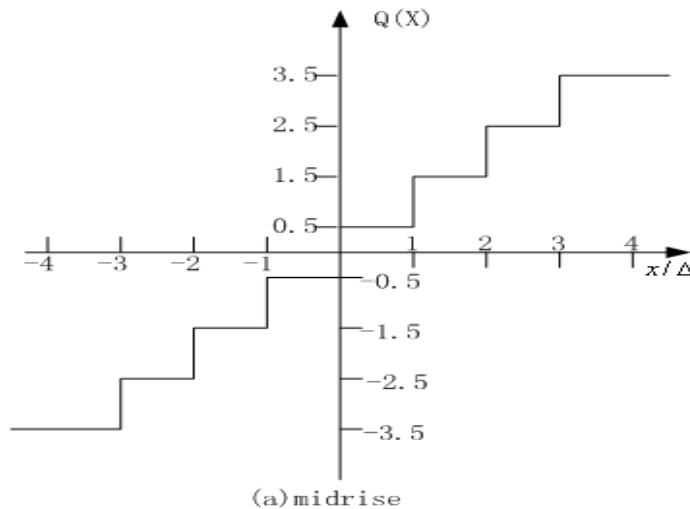
## 3.2 Uniform Scalar Quantization

- Given step size  $\Delta=1$ , output values for the two type of Quantizers be computed as:

$$Q_{midrise}(x) = \lceil x \rceil - 0.5$$

$$Q_{midtread}(x) = \lfloor x + 0.5 \rfloor$$

- Two types quantizers:



## 3.2 Uniform Scalar Quantization

- Performance of a **M level quantizer**:
  - Decision Boundaries:  $B = \{b_0, b_1, \dots, b_M\}$
  - The set of output values:  $Y = \{y_1, y_2, \dots, y_M\}$
  - The input is **uniformly distributed**:  $[-X_{\max}, X_{\max}]$
  - The **rate** of quantizer:  $R = \log_2 M$  is the number of bits required to code M things;
  - Step size is given by:  $\Delta = 2X_{\max}/M$
  - **Granular distortion**: error caused by the quantizer for bounded input
  - **Overload distortion**: error caused by quantizer for input values larger than  $X_{\max}$  or smaller than  $-X_{\max}$

## 3.2 Uniform Scalar Quantization

- **Granular distortion for a midrise quantizer**
  - **Decision boundaries  $b_i: [(i-1)\Delta, i\Delta]$ ,  $i=1..M/2$ , covering positive data  $X$  (another for native  $X$  values)**
  - **Output values  $y_i: i\Delta - \Delta/2$ ,  $i=1..M/2$**
  - **The total distortion: **twice the sum over the positive data:****

$$D_{gran} = 2 \sum_{i=1}^{\frac{M}{2}} \int_{(i-1)\Delta}^{i\Delta} \left(x - \frac{2i-1}{2} \Delta\right)^2 \frac{1}{2X_{\max}} dx$$

- **The error value at  $X$  is  $e(x)=x-\Delta/2$ , **variance of errors:****

$$\sigma_d^2 = \frac{1}{\Delta} \int_0^{\Delta} (e(x) - \bar{e})^2 dx = \frac{1}{\Delta} \int_0^{\Delta} \left(x - \frac{\Delta}{2} - 0\right)^2 dx = \frac{\Delta^2}{12}$$



## 3.2 Uniform Scalar Quantization

- **Signal variance**  $\sigma_x^2 = (2X_{\max})^2 / 12$  ; if the quantizer is  $n$  bits,  $M=2^n$
- **SQNR can be calculated as:**

$$\begin{aligned} SQNR &= 10 \log_{10} \left( \frac{\sigma_x^2}{\sigma_d^2} \right) \\ &= 10 \log_{10} \left( \frac{(2X_{\max})^2}{12} \cdot \frac{12}{\Delta^2} \right) \\ &= 10 \log_{10} \left( \frac{(2X_{\max})^2}{12} \cdot \frac{12}{\left( \frac{2X_{\max}}{M} \right)^2} \right) \\ &= 10 \log_{10} M^2 = 20n \cdot \log_{10} 2 \\ &= 6.02n(dB) \end{aligned}$$



## 3.3 Nonuniform Scalar Quantization

- If the input source is not uniformly distributed, a uniform quantizer may be inefficient.
- **Increasing the number of decision levels** within the **densely distributed region** can lower granular distortion
- **Enlarge the region** where **the source is sparsely distributed** can keep the total number of decision levels
- So nonuniform quantizers have **nonuniformly defined decision boundaries**.
- **Two common approaches** for nonuniform quantization:
  - The Lloyd-Max Quantizer
  - The companded quantizer



## 4. Transform Coding

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# 4.1 Basic Idea

- According principles of information theory
  - Coding vectors is more efficient than coding scalars
  - Need to group consecutive samples from input into vectors
- Let  $X = \{x_1, x_2, \dots, x_k\}$  be vector of samples, there's an amount **correlation among neighboring**.
- If  $Y$  is the result of a linear transform  $T$  of the input vector and its components have **much less correlation**, then  $Y$  can be coded **more efficiently** than  $X$ .
  - The transform  $T$  itself does not compress any data.
  - The compression comes from the processing and quantization of the components of  $Y$ .
- **DCT is a widely used transform**, it can perform de-correlation of the input signal.



## 4.2 Discrete Cosine Transform (DCT)

### □ 1D Discrete Cosine Transform:

$$F(u) = \frac{C(u)}{2} \sum_{i=0}^7 \cos \frac{(2i+1)u\pi}{16} f(i)$$

### □ 1D Inverse Discrete Cosine Transform:

$$\tilde{f}_i = \sum_{u=0}^7 \frac{C(u)}{2} \cos \frac{(2i+1)u\pi}{16} F(u)$$

$$C(u) = \begin{cases} \frac{\sqrt{2}}{2} & \text{if } u = 0 \\ 1 & \text{else} \end{cases}$$

### □ 2D transform can be used to process 2D signals such as digital images



## 4.2 Discrete Cosine Transform (DCT)

### DCT (2D) Definition:

- Given a function  $f(i, j)$  over an image, the 2D DCT transforms it into a new function  $F(u, v)$ , integer  $u$  and  $v$  running over the same range as  $i$  and  $j$ .
- The general definition of the DCT transform is:

$$F(u, v) = \frac{2C(u)C(v)}{\sqrt{MN}} \sum_{i=0}^{M-1} \sum_{j=0}^{N-1} \cos \frac{(2i+1)u\pi}{2M} \cos \frac{(2j+1)v\pi}{2N} f(i, j)$$



## 4.2 Discrete Cosine Transform (DCT)

- In the JPEG image compression standard
  - An image block is defined to have dimension  $M=N=8$ ;
  - The definition of 2D DCT and its inverse IDCT are as follows:

- **2D Discrete Cosine Transform(2D DCT):**

$$F(u, v) = \frac{C(u)C(v)}{4} \sum_{i=0}^7 \cos \frac{(2i+1)u\pi}{16} \cos \frac{(2j+1)v\pi}{16} f(i, j)$$

- **2D Inverse Discrete Cosine Transform(2D IDCT):**

$$\tilde{f}(i, j) = \sum_{u=0}^7 \sum_{v=0}^7 \frac{C(u)C(v)}{4} \cos \frac{(2i+1)u\pi}{16} \cos \frac{(2j+1)v\pi}{16} F(u, v)$$



## 4.2 Discrete Cosine Transform (DCT)

### □ DCT related concepts

#### ■ **Direct current (DC) and alternating current (AC)**

- Represent constant and variable magnitude respectively;

#### ■ **Fourier analysis**

- Any signal can be expressed as a sum of multiple signals that are sine or cosine wave forms.
- An signal usually composed of one DC and several AC components;

#### ■ **Cosine Transform**

- The process used to determine the amplitude of the AC and DC components of the signal.

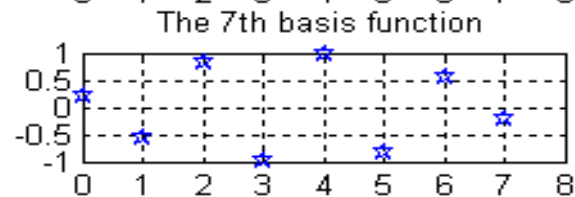
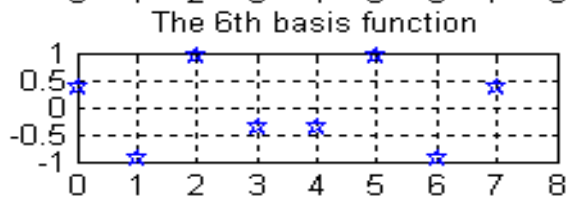
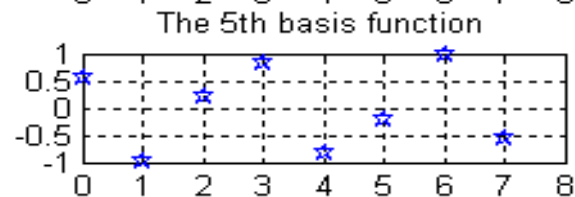
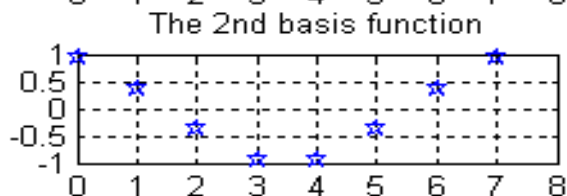
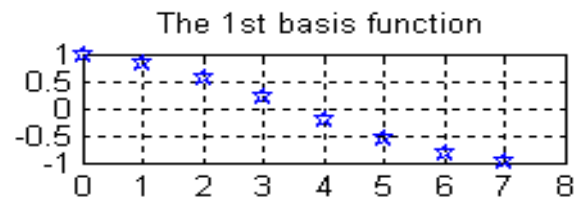
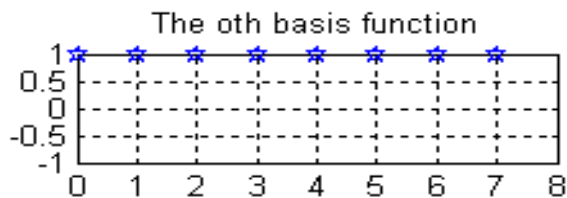


## 4.2 Discrete Cosine Transform (DCT)

- **DCT related concepts (Continue)**
  - **Discrete Cosine Transform: integer indices**
    - $U=0$ , we get the DC coefficient;
    - $U=1, 2, \dots, 7$ , we get the first up to seventh AC coefficients.
  - **Invert Discrete Cosine transform: using DC, AC and cosine functions to reconstruct the signal**
  - **DCT and IDCT adopt the same set of cosine functions which are know as **basis functions****

## 4.2 Discrete Cosine Transform (DCT)

### □ 1D DCT basis functions





## 4.2 Discrete Cosine Transform (DCT)

- DCT enable to process or analyze the signal in frequency domain
- Suppose  $f(i)$  represents a signal changes with time  $i$ 
  - 1D DCT transforms  $f(i)$  in time domain to  $F(U)$  in frequency domain.
  - $F(u)$  are known as frequency response, form the **frequency spectrum** of  $f(i)$

## 4.2 Discrete Cosine Transform (DCT)

□ **Example (1):**  $f_1(i)=100$ , a signal with magnitude of 100

$$\left\{ F(u) = \frac{C(u)}{2} \sum_{i=0}^7 \cos \frac{(2i+1)u\pi}{16} f(i) \right\}$$

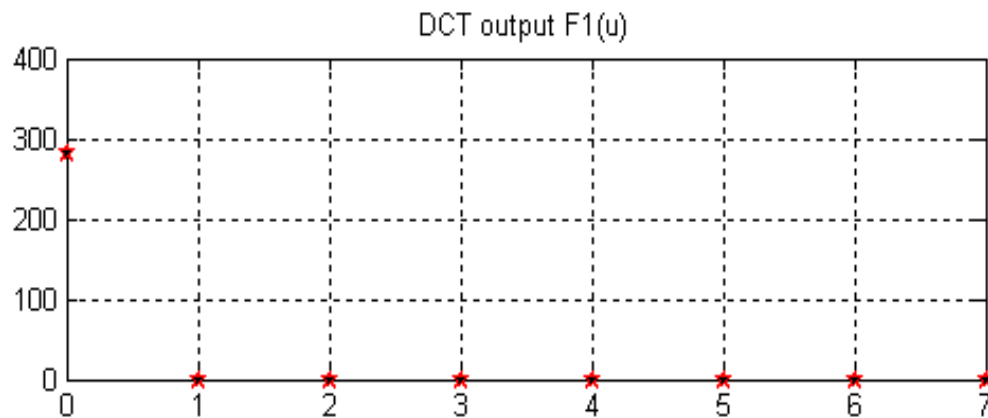
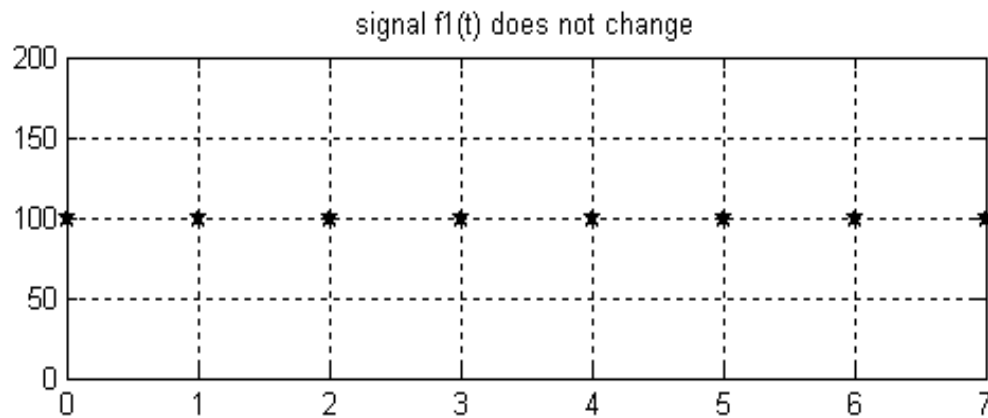
■  $F_1(0)=C(0)/2 * (1 \cdot 100 + 1 \cdot 100 + 1 \cdot 100 + 1 \cdot 100 + 1 \cdot 100 + 1 \cdot 100 + 1 \cdot 100 + 1 \cdot 100)$  noticed that  $C(0) = \frac{\sqrt{2}}{2}$

■  $=C(0) \cdot 400 \approx 283$

■  $F_1(1) = \frac{1}{2} \left( \cos \frac{\pi}{16} \cdot 100 + \cos \frac{3\pi}{16} \cdot 100 + \cos \frac{5\pi}{16} \cdot 100 + \cos \frac{7\pi}{16} \cdot 100 + \cos \frac{9\pi}{16} \cdot 100 \right.$   
 $\left. + \cos \frac{11\pi}{16} \cdot 100 + \cos \frac{13\pi}{16} \cdot 100 + \cos \frac{15\pi}{16} \cdot 100 \right)$   
 $= 0$

■  $F_1(2)=F_1(3)=F_1(4)=F_1(5)=F_1(6)=F_1(7)=0$

## 4.2 Discrete Cosine Transform (DCT)



## 4.2 Discrete Cosine Transform (DCT)

- **Example 2: a signal  $f_2(i)$ , has the same frequency and phase as the second cosine basis function, amplitude is 100**

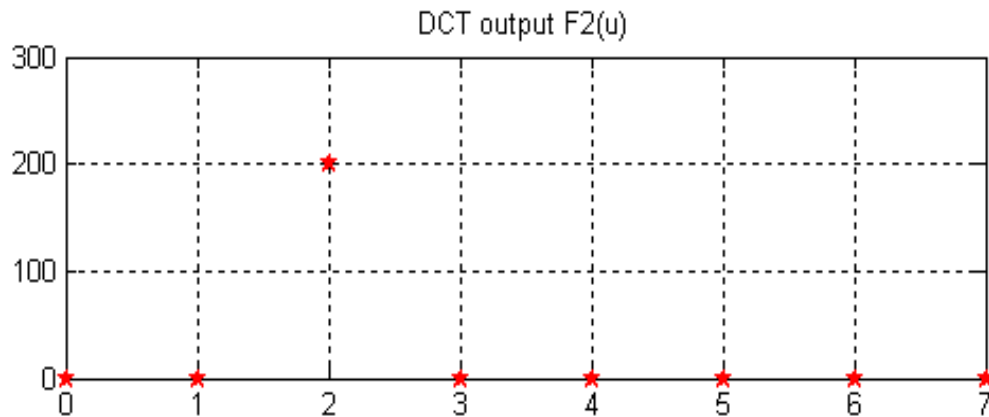
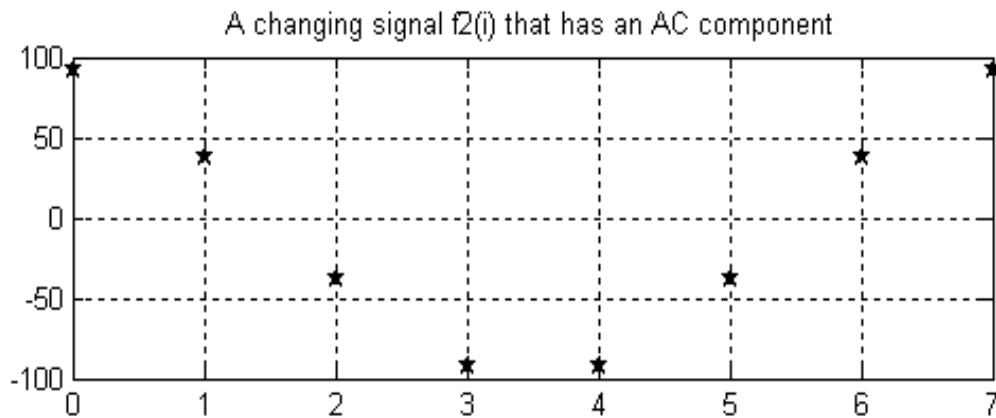
$$\begin{aligned} \blacksquare \quad F_2(0) &= \frac{\sqrt{2}}{2 \cdot 2} \cdot 1 \cdot (100 \cos \frac{\pi}{8} + 100 \cos \frac{3\pi}{8} + 100 \cos \frac{5\pi}{8} + 100 \cos \frac{7\pi}{8} \\ &\quad + 100 \cos \frac{9\pi}{8} + 100 \cos \frac{11\pi}{8} + 100 \cos \frac{13\pi}{8} + 100 \cos \frac{15\pi}{8}) \\ &= 0 \end{aligned}$$

$$\begin{aligned} \blacksquare \quad F_2(2) &= \frac{1}{2} \cdot (\cos \frac{\pi}{8} \cdot \cos \frac{\pi}{8} + \cos \frac{3\pi}{8} \cdot \cos \frac{3\pi}{8} + \cos \frac{5\pi}{8} \cdot \cos \frac{5\pi}{8} \\ &\quad + \cos \frac{7\pi}{8} \cdot \cos \frac{7\pi}{8} + \cos \frac{9\pi}{8} \cdot \cos \frac{9\pi}{8} + \cos \frac{11\pi}{8} \cdot \cos \frac{11\pi}{8} \\ &\quad + \cos \frac{13\pi}{8} \cdot \cos \frac{13\pi}{8} + \cos \frac{15\pi}{8} \cdot \cos \frac{15\pi}{8}) \cdot 100 \\ &= 200 \end{aligned}$$

- **We can get other values by similar way**

$$\blacksquare \quad F_2(1) = F_2(3) = F_2(4) = \dots = F_2(7) = 0$$

## 4.2 Discrete Cosine Transform (DCT)





## 4.2 Discrete Cosine Transform (DCT)

### Properties of DCT transform

- **DCT produces the frequency spectrum  $F(u)$  of signal  $f(i)$** 
  - The 0th DCT coefficient  $F(0)$  is the DC component of the signal  $f(i)$ ;
  - The other seven DCT coefficients reflect the various changing components of signal  $f(i)$  at different frequencies;
  - If **DC is a negative value**, this means that **the average of  $f(i)$  is less than zero**;
  - if **AC is a negative value**, this means that  $f(i)$  and some basis function have the **same frequency** but one of them happens **to be half a cycle behind**.
- **DCT is a linear transform**

$$T(\alpha p + \beta q) = \alpha T(p) + \beta T(q)$$



## 4.2 Discrete Cosine Transform (DCT)

### □ One-Dimensional IDCT

- If  $F(u)$  contains ( $u=0\dots7$ ): 69 -49 74 11 16 117 44 -5
- IDCT can be implemented by eight iterations:

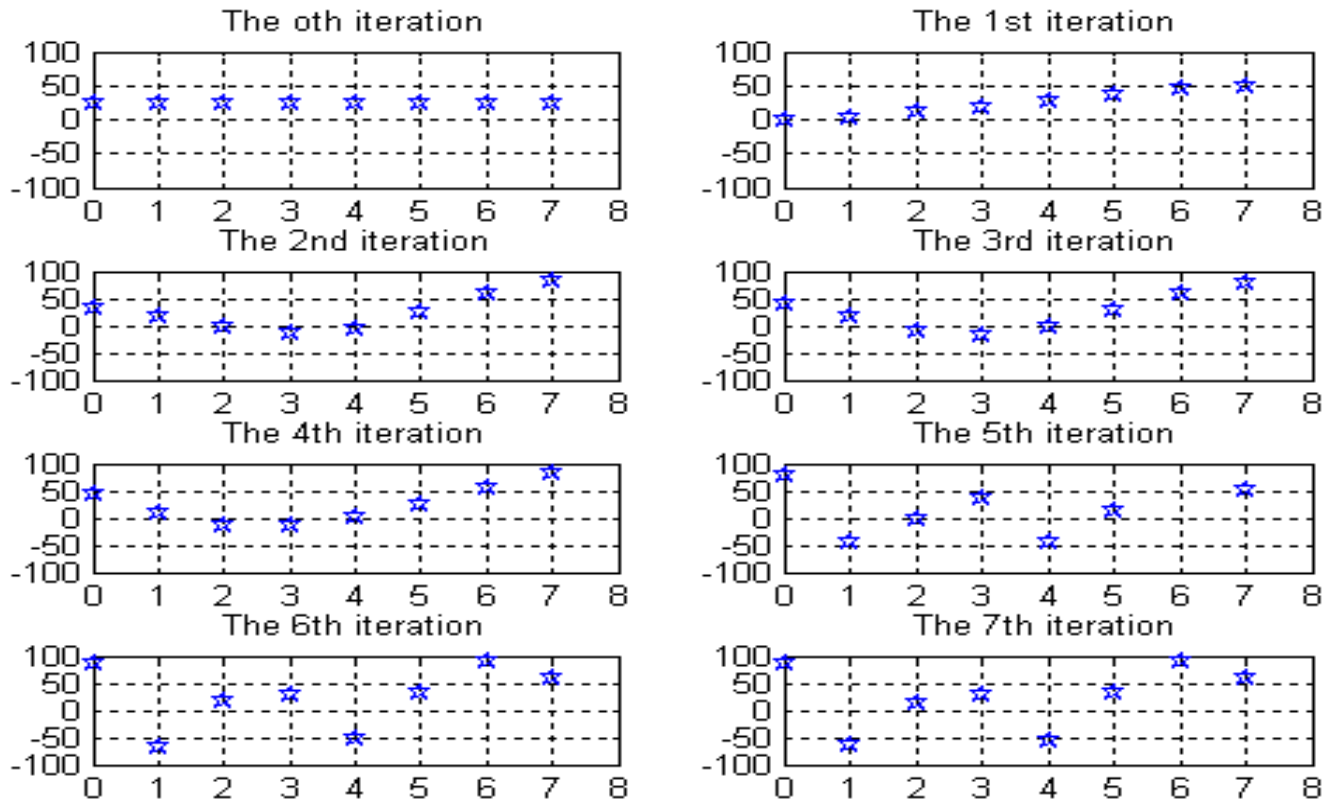
$$\text{Iteration } 0: \tilde{f}_i = \frac{C(0)}{2} \cos 0 \cdot F(0) \approx 24.3$$

$$\begin{aligned} \text{Iteration } 1: \tilde{f}_i &= \frac{C(0)}{2} \cos 0 \cdot F(0) + \frac{C(1)}{2} \cos \frac{(2i+1)\pi}{16} \cdot F(1) \\ &\approx 24.3 - 24.5 \cdot \cos \frac{(2i+1)\pi}{16} \end{aligned}$$

$$\begin{aligned} \text{Iteration } 2: \tilde{f}_i &= \frac{C(0)}{2} \cos 0 \cdot F(0) + \frac{C(1)}{2} \cos \frac{(2i+1)\pi}{16} \cdot F(1) + \frac{C(2)}{2} \cos \frac{(2i+1)\pi}{8} \cdot F(2) \\ &\approx 24.3 - 24.5 \cdot \cos \frac{(2i+1)\pi}{16} + 37 \cdot \cos \frac{(2i+1)\pi}{8} \end{aligned}$$

## 4.2 Discrete Cosine Transform (DCT)

### □ An example of IDCT





## 4.2 Discrete Cosine Transform (DCT)

- Cosine basis functions are orthogonal

$$\sum_{i=0}^7 \left[ \cos \frac{(2i+1) \cdot p \pi}{16} \cdot \cos \frac{(2i+1) \cdot q \pi}{16} \right] = 0 \quad \text{if } p \neq q$$
$$\sum_{i=0}^7 \left[ \frac{C(p)}{2} \cos \frac{(2i+1) \cdot p \pi}{16} \cdot \frac{C(q)}{2} \cos \frac{(2i+1) \cdot q \pi}{16} \right] = 1 \quad \text{if } p = q$$

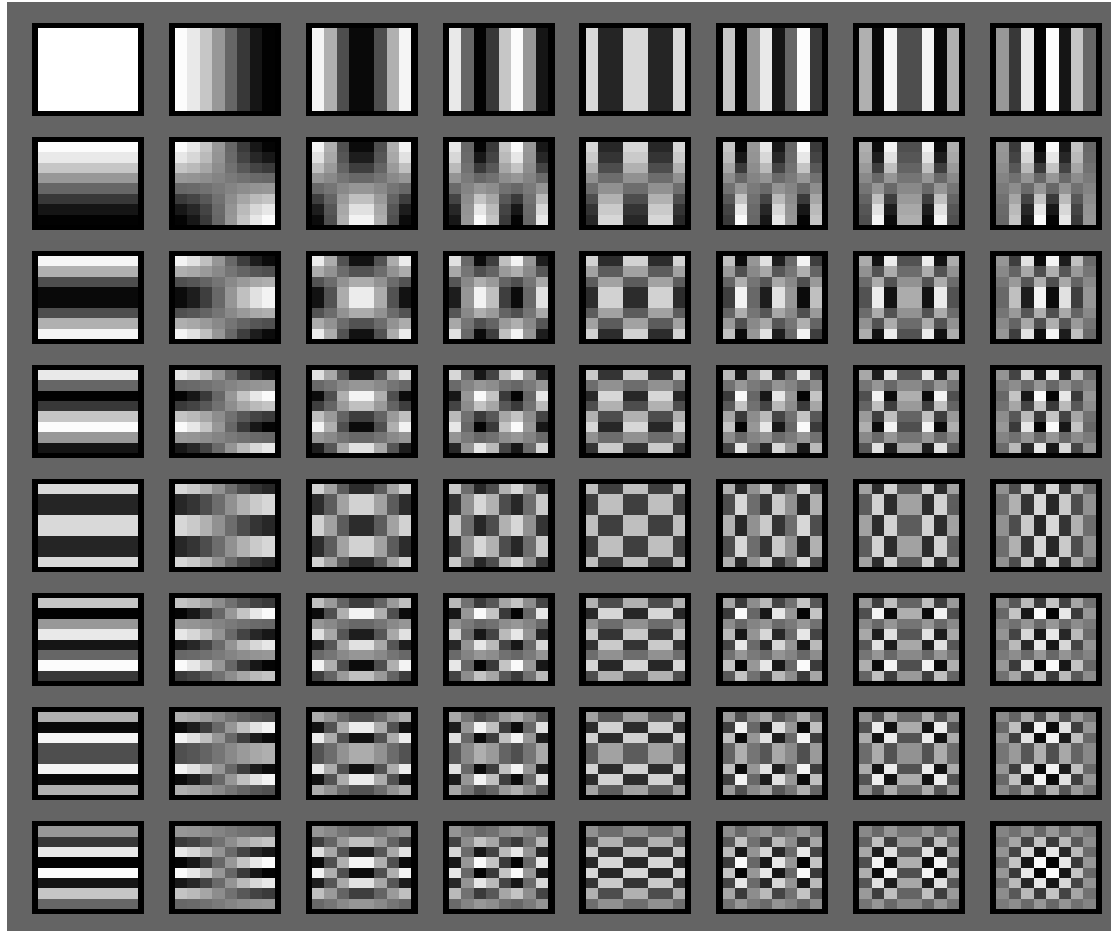
- 2D separable basis

- Factorization 2D DCT into two 1D DCT transforms

$$G(i, v) = \frac{1}{2} C(v) \sum_{j=0}^7 \cos \frac{(2j+1)v\pi}{16} f(i, j)$$
$$F(u, v) = \frac{1}{2} C(u) \sum_{j=0}^7 \cos \frac{(2i+1)u\pi}{16} G(i, v)$$

## 4.2 Discrete Cosine Transform (DCT)

2D Basis Functions





## 4.3 Comparison of DCT and DFT

### □ Fourier Transform

$$F(\omega) = \int_{-\infty}^{+\infty} f(t) e^{-j\omega t} dt$$

### □ Discrete Fourier Transform

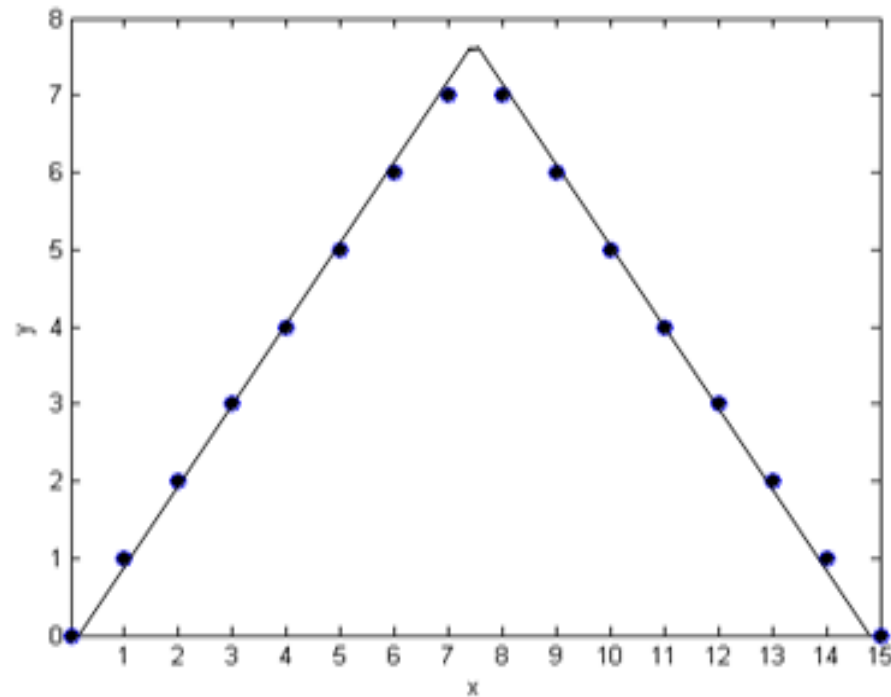
$$F_{\omega} = \sum_{x=0}^7 f_x \cdot e^{-j \frac{2\pi \omega x}{8}}$$

$$F_{\omega} = \sum_{x=0}^7 f_x \cdot \cos\left(\frac{2\pi \omega x}{8}\right) - j \sum_{x=0}^7 f_x \cdot \sin\left(\frac{2\pi \omega x}{8}\right)$$

- DCT is a close counterpart of the DFT, in signal processing, DCT is more popular.

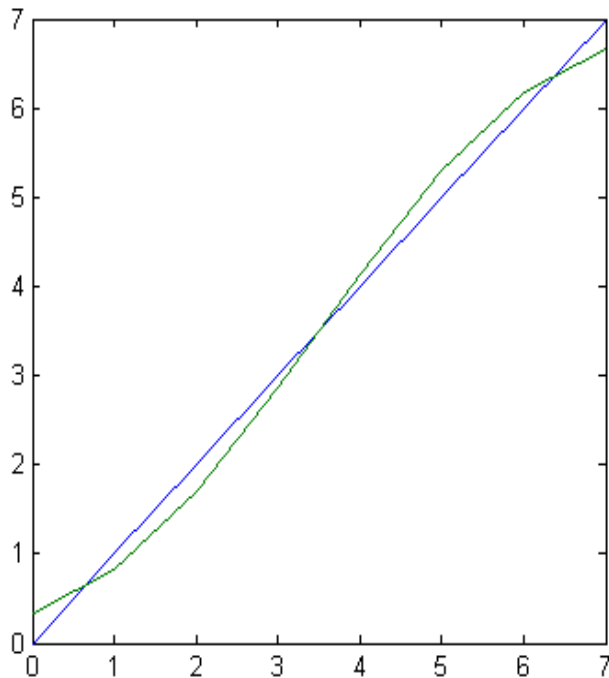
## 4.3 Comparison of DCT and DFT

| Ramp | DCT   | DFT   |
|------|-------|-------|
| 0    | 9.90  | 28.00 |
| 1    | -6.44 | -4.00 |
| 2    | 0.00  | 9.66  |
| 3    | -0.67 | -4.00 |
| 4    | 0.00  | 4.00  |
| 5    | -0.20 | -4.00 |
| 6    | 0.00  | 1.66  |
| 7    | -0.51 | -4.00 |

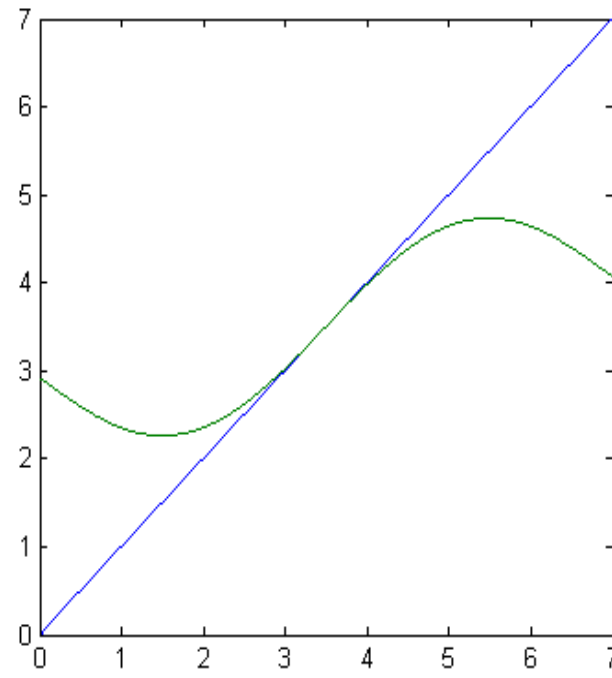


DCT and DFT coefficient of the ramp function

## 4.3 Comparison of DCT and DFT



(a) three-term DCT approximation



(b) three-term DFT approximation

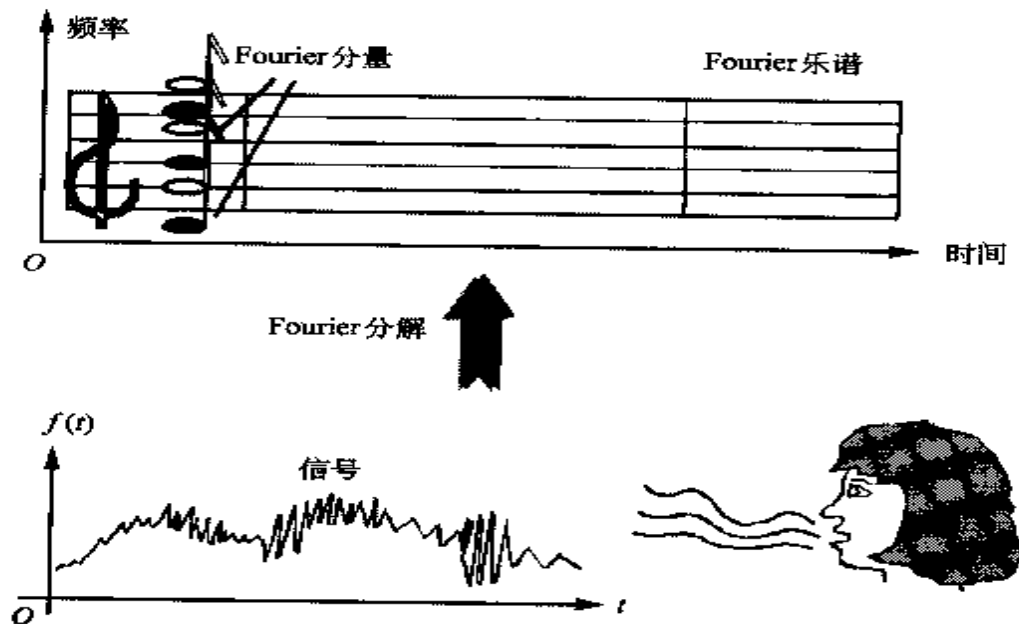


## 5. Wavelet-Based Coding

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# 5.1 Introduction

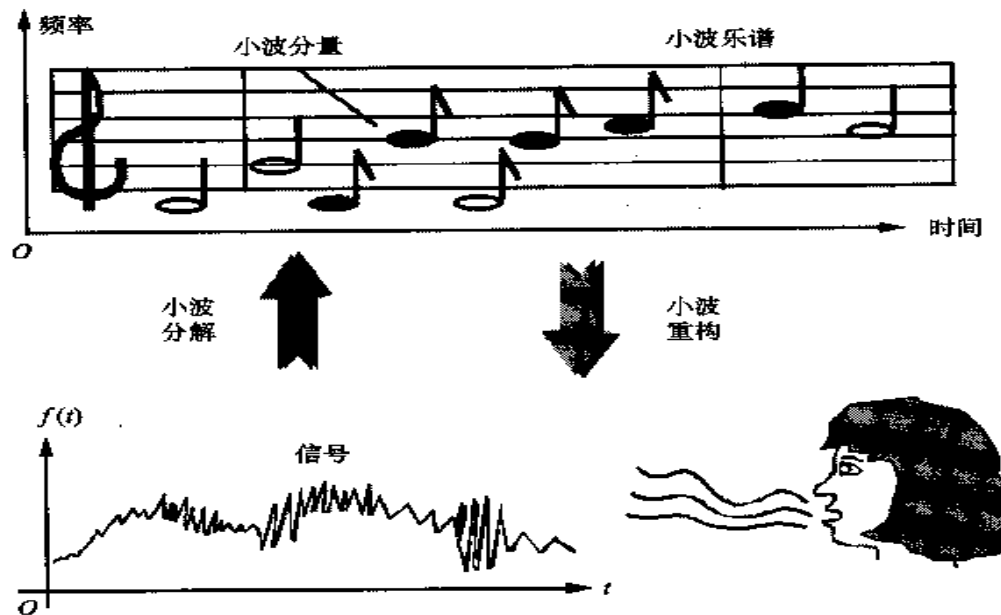
- DFT and DCT can give very fine resolution in the frequency domain, but no temporal resolution.



Fourier Decomposition of A Song Signal

# 5.1 Introduction

- Wavelet transform seeks to represent a signal with good resolution in both time and frequency.



Wavelet Decomposition of A Song Signal



## 5.2 1D Haar Transform

$$\{x_{n,i}\} = \{10, 13, 25, 26, 29, 21, 7, 15\}$$

$$x_{n-1,i} = \frac{x_{n,2i} + x_{n,2i+1}}{2}$$

$$d_{n-1,i} = \frac{x_{n,2i} - x_{n,2i+1}}{2}$$

$$\{x_{n-1,i}, d_{n-1,i}\} = \{11.5, 25.5, 25, 11, -1.5, -0.5, 4, -4\}$$

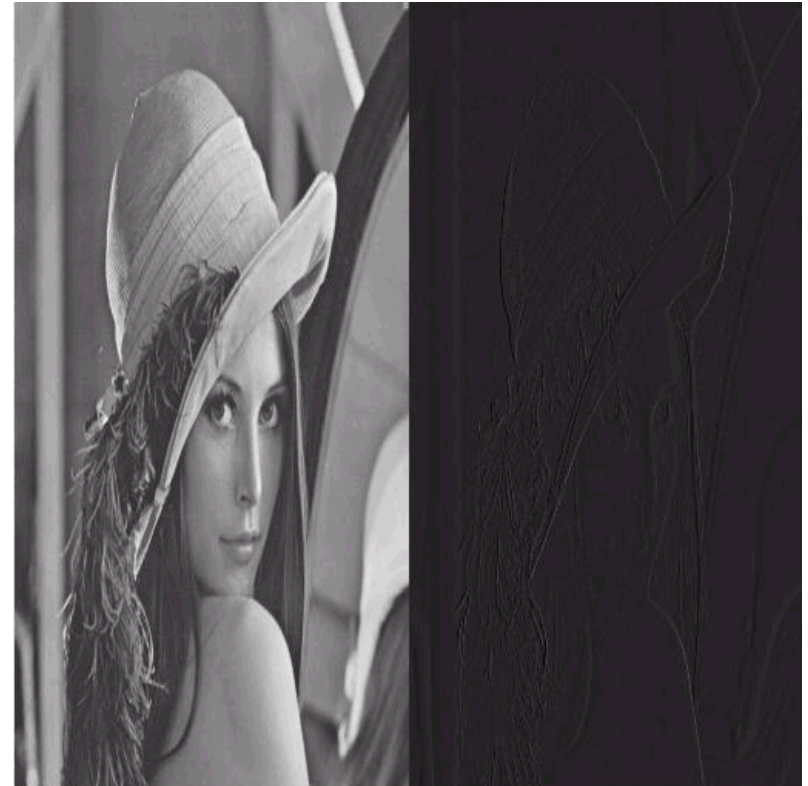
$$\{x_{n-2,i}, d_{n-2,i}, d_{n-1,i}\} = \{18.5, 18, -7, 7, -1.5, -0.5, 4, -4\}$$

$$\{x_{n-3,i}, d_{n-3,i}, d_{n-2,i}, d_{n-1,i}\} = \{18.25, 0.25, -7, 7, -1.5, -0.5, 4, -4\}$$

## 5.3 2D Haar Transform

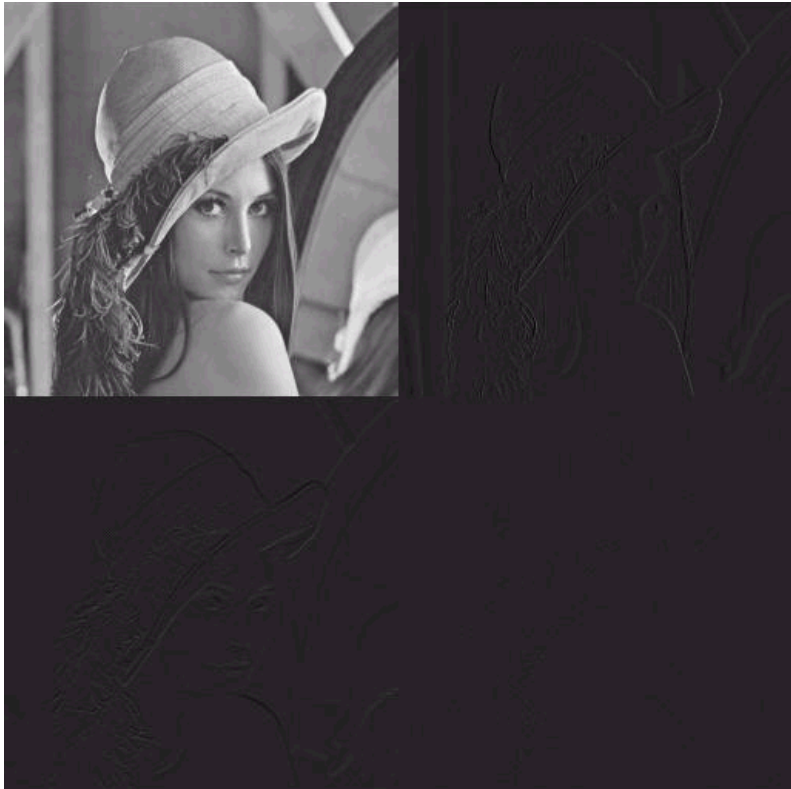


**original image**

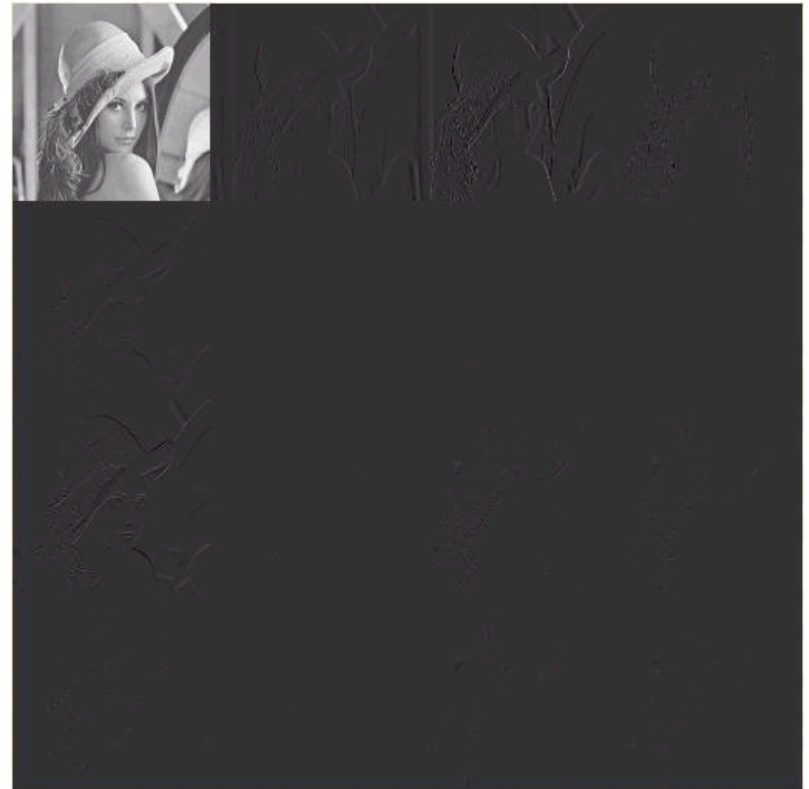


**wavelet horizontally transform**

## 5.3 2D Haar Transform



**wavelet horizontally and vertically transform (one level)**



**wavelet transformation (2 levels)**



# The End!

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## Thanks !