

Introduction to Image Processing



Lecturer:
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Week 5 – Frequency Filters

December 3, 2024

Recap on signal theory principles

Classification of image operations

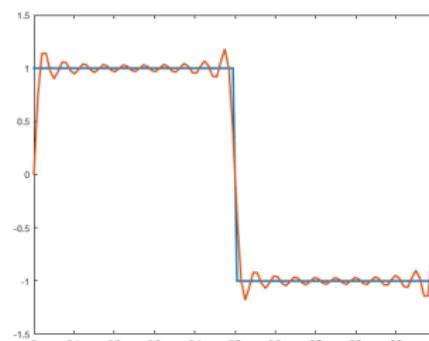
- *Point operation*: output pixel value depends only on corresponding input pixel (e.g. thresholding):
 $f_{out}(x, y) = \mathcal{O}(\{f_{in}(x, y)\})$ where \mathcal{O} denotes the *grayscale transformation* function.
 - *Local operation*: output pixel value depends on neighborhood $\mathcal{B}(x, y)$ of input pixel:
 $f_{out}(x, y) = \mathcal{O}(\{f_{in}(x', y') : (x', y') \in \mathcal{B}(x, y)\}).$
 - *Global operation*: output pixel value depends on all input pixels. Example: (discrete) Fourier transform.
 - *Geometric operation*: spatial transformation (scaling, translation, rotation)

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Basics

- signal: superposition of primary frequencies of harmonic function (e.g. *sin* and *cosine*)
 - full signal reconstruction: frequency coefficients required → Fourier transform
 - Fourier transform requires signal sampling at $\omega = \frac{1}{\Delta T}$
 - *perfect* (i.e. non-aliased) reconstruction: sampling width $\Delta T \leq \frac{1}{2\mu_{max}}$, with μ_{max} being the maximum frequency in the signal (*Nyquist frequency*)
 - smallest detail in signal f relates to highest primary frequency μ_{max} → small features require higher sampling frequency ω



Basics

- digital samples → finite spectrum → **DFT**: from finite set of samples $\{f_n\}$ to finite set of frequency coefficients $\{F_\mu\}$

$$F_\mu = \sum_{n=0}^{M-1} f_n e^{-i2\pi\mu\frac{n}{M}} \quad \forall \mu \in \{0..M-1\}$$

- DFT of a convolved signal $f \star h$ equals product of DFTs of individual signals f (input) and h (Dirac sample):

$$\mathcal{F}\{f \star h\} = F_\mu \ H_\mu \quad \forall \mu \in \{0..M-1\}$$

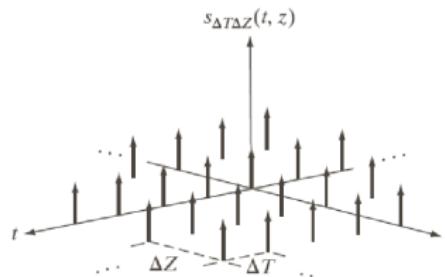
Symbolics & rules

$f(t)$: signal
$t \in T$: signal size
$\Delta T \leq \frac{1}{2\mu_{max}}$: sample spacing
μ_{max}	: maximum primary frequency in f
$\Delta\mu = \frac{1}{T}$: frequency spacing
$\Omega = \frac{1}{\Delta T}$: size of frequency space
$F_\mu = F(\mu)$: frequency space of $f(t)$
$s_{\Delta T}(t) = \sum_{n=-\infty}^{\infty} \delta(t - n\Delta T)$: sampling function
$H_\mu = S(\mu)$: frequency space of $s_{\Delta T}(t)$

2D sampling

A signal $f(x, y)$ can be reconstructed from its frequency space $F(u, v)$ if:

- the amount of sampled frequencies are larger than the highest primary frequencies of the image features:
 $|u| \geq \mu_{max}, |v| \geq v_{max}$
- the sampling distance adheres to the Nyquist criterion:
 $\Delta T \leq \frac{1}{2\mu_{max}}, \Delta Z \leq \frac{1}{2v_{max}}$



2D frequency space

Remember that $F(u, v)$ is a *complex* function:

$$F(u, v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y) e^{-i2\pi(u \frac{x}{M} + v \frac{y}{N})} \quad \forall u \in \{0..M-1\}, v \in \{0..N-1\}$$

Result: $F(u, v) = R(u, v) + iI(u, v) = |F(u, v)|e^{i\phi(u, v)}$, with:

- $|F(u, v)| = \sqrt{R^2(u, v) + I^2(u, v)}$ (frequency spectrum)
- $\phi(u, v) = \arctan(\frac{I(u, v)}{R(u, v)})$ (frequency phase angle)

If f is real (i.e. no imaginary part), then the complex conjugate $F^*(u, v)$ is symmetric:

- $F^*(u, v) = F(-u, -v)$
- $|F(u, v)| = |F(-u, -v)|$
- $\phi(u, v) = -\phi(-u, -v)$

2D convolution theorem

For $M \times N$ digital images f and h the 2-D discrete circular convolution is defined by

$$(f \star h)(x, y) = \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} f(m, n) h(x-m, y-n) \quad \forall x \in \{0..M-1\}, y \in \{0..N-1\}$$

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To compute $f \star h$:

- ① compute the Fourier transforms $F(u, v)$ and $H(u, v)$
- ② multiply the transforms (pointwise)
- ③ compute the inverse 2D Fourier transform of the product

Design and application of frequency filters

Frequency domain filtering

1. For $M \times N$ input $f(x, y)$ and $C \times D$ filter $h(x, y)$ apply zero padding to get padded image $f_p(x, y)$ of size $P \times Q$ ($P \geq M + C - 1$, $Q \geq N + D - 1$).

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6. Do the 2-D inverse DFT, take the real part (to remove small imaginary parts due to computational inaccuracy) and undo the centering:

$$g_p(x, y) = \{real[IDFT[G(u, v)]]\}(-1)^{(x+y)}$$

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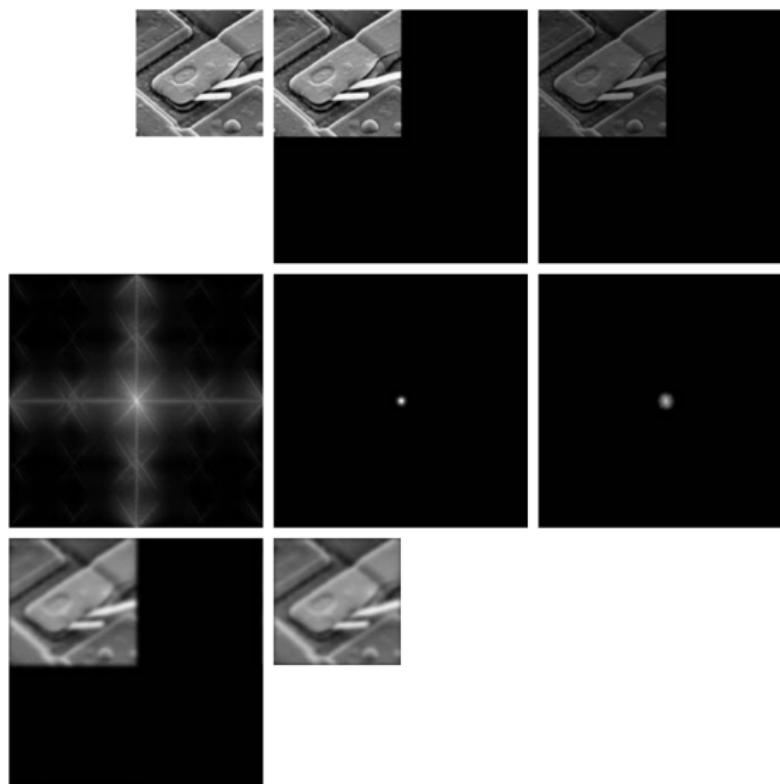
7. Obtain $g(x, y)$ by extracting the top left $M \times N$ quadrant from $g_p(x, y)$.

Frequency domain filtering

a b c
d e f
g h

FIGURE 4.36

- (a) An $M \times N$ image, f .
- (b) Padded image, f_p of size $P \times Q$.
- (c) Result of multiplying f_p by $(-1)^{x+y}$.
- (d) Spectrum of F_p .
- (e) Centered Gaussian lowpass filter, H , of size $P \times Q$.
- (f) Spectrum of the product HF_p .
- (g) g_p , the product of $(-1)^{x+y}$ and the real part of the IDFT of HF_p .
- (h) Final result, g , obtained by cropping the first M rows and N columns of g_p .



Filter design

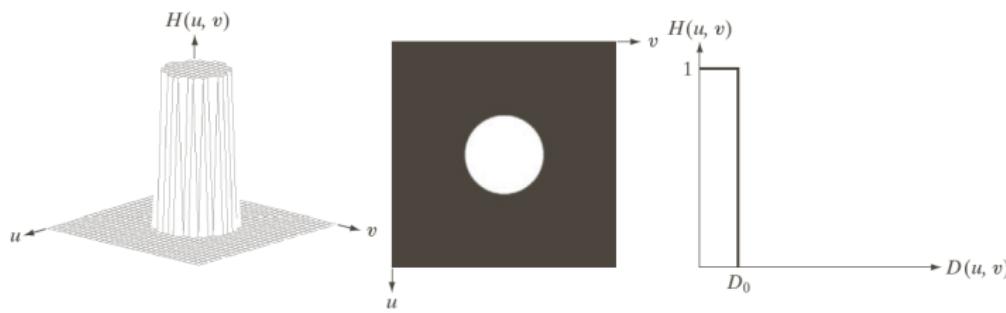
Lowpass filter : suppress high frequencies. Example: box filter ($H(\mu) = 1$ for $0 \leq \mu \leq \mu_M$, μ_M is the cutoff frequency).

Bandpass filter : pass energy within a given frequency window.

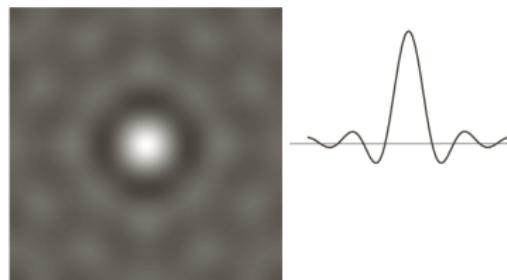
Bandstop filter : suppress energy within a given frequency window.

Highpass filter : suppress low frequencies. (Useful for edge detection).

Ideal lowpass filter (ILPF)

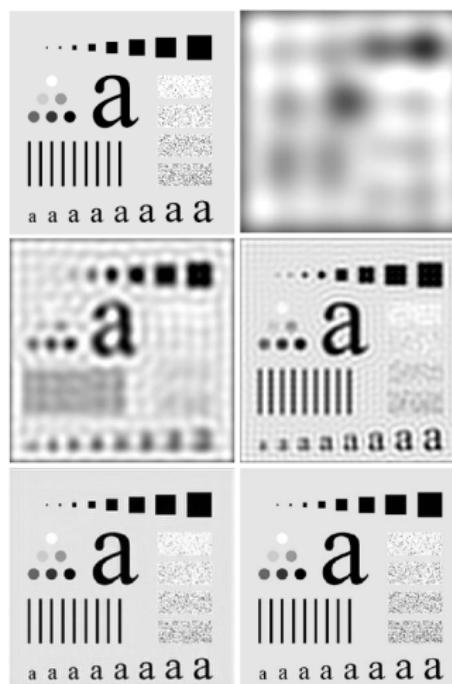


Frequency domain



Spatial domain

Ideal lowpass filter (ILPF)



ILPF with increasing frequency cutoff radius. Note: **ringing artefacts**

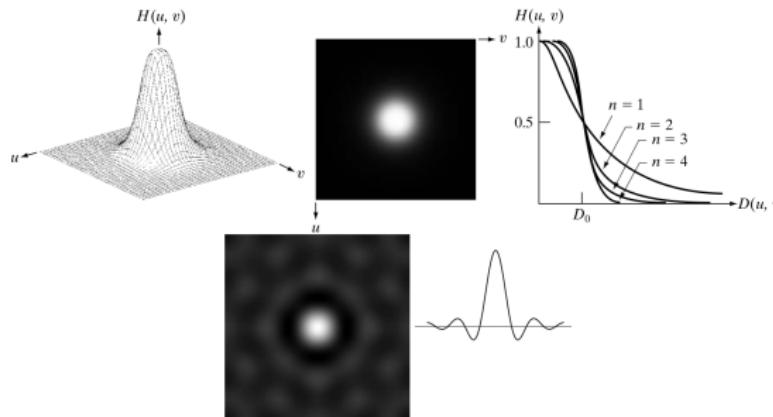
Lowpass frequency filters - Butterworth filter

Ringing artefact → non-smooth spatial transform of filter → **filter design**

Low pass : $H(u, v) = \frac{1}{1 + [D(u, v)/D_0]^{2n}}$

Kernel : $D(u, v) = [(u - P/2)^2 + (v - Q/2)^2]^{1/2}$, $D(u, 0) = D(u) = u$

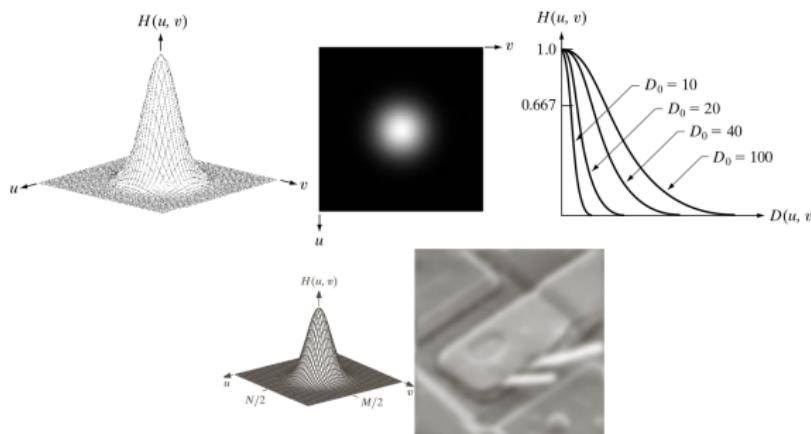
For $\{P, Q\}$, see explanation for image padding before.



Issue: higher-order n reintroduce ringing.

Lowpass frequency filters - Gaussian filter

Low pass : $H(u, v) = e^{-D(u,v)^2/2\sigma^2}$, $h(x, y) = \sqrt{2\pi}\sigma e^{-2\pi^2\sigma^2 D(x,y)^2}$



No ring artefacts, yet also not fully *blocking* high frequencies.

Lowpass frequency filters - Gaussian filter

Challenge: determine σ

$$\sigma = D_0$$

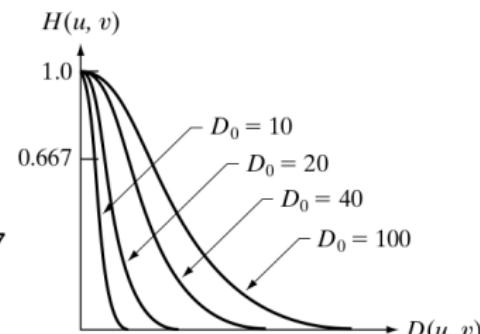
$$H(u, v) = e^{-D(u, v)^2/2D_0^2}$$

for $D(u, v) = D_0$:

$$H(u, v) = e^{-D_0^2/2D_0^2} = e^{-\frac{1}{2}} = 0.607$$

For practical purpose:

$\sigma = D_0$ (cutoff frequency)



Summary of lowpass filters

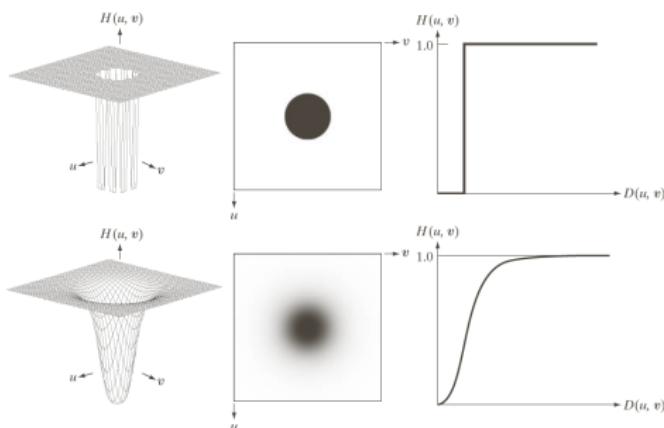
TABLE 4.4

Lowpass filters, D_0 is the cutoff frequency and n is the order of the Butterworth filter.

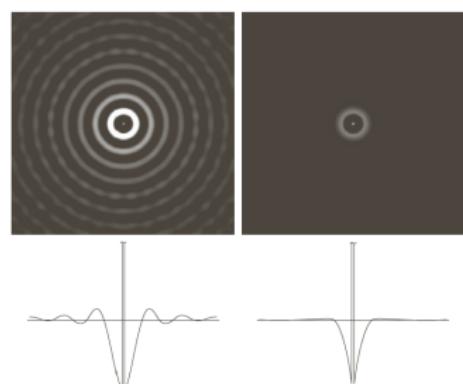
Ideal	Butterworth	Gaussian
$H(u, v) = \begin{cases} 1 & \text{if } D(u, v) \leq D_0 \\ 0 & \text{if } D(u, v) > D_0 \end{cases}$	$H(u, v) = \frac{1}{1 + [D(u, v)/D_0]^{2n}}$	$H(u, v) = e^{-D^2(u, v)/2D_0^2}$

Overview of common lowpass filters.

Highpass frequency filters



Frequency domain.
Top: ideal (IHPF); bottom: Butterworth.



Spatial domain.
Left: ideal (IHPF);
right: Butterworth.

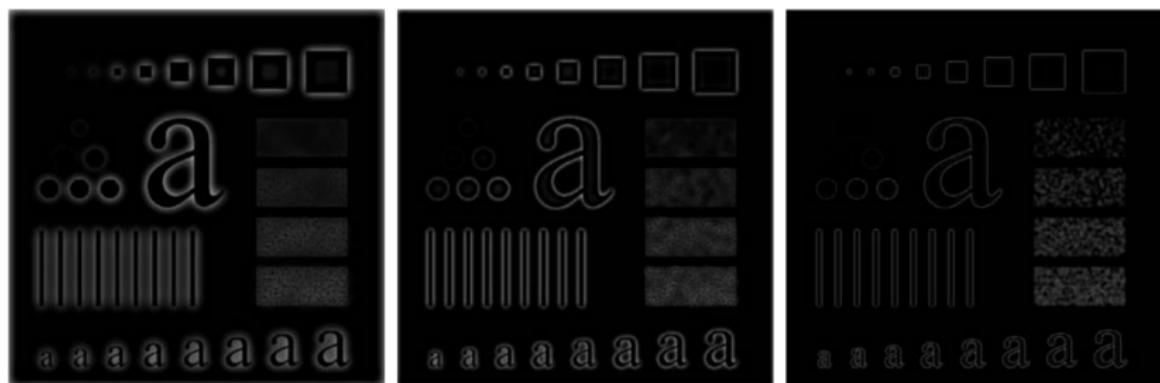
Ideal highpass filter (IHPF)



IHPF with increasing frequency radius.

$$H(u, v) = \begin{cases} 0 & \text{if } D(u, v) \leq D_0 \\ 1 & \text{if } D(u, v) > D_0 \end{cases}$$

Highpass frequency filters - Butterworth



Butterworth filter with increasing frequency radius.

$$H(u, v) = \frac{1}{1 + [D_0/D(u, v)]^{2n}}$$

Highpass frequency filters - Laplacian



Laplacian edge filter in the frequency domain.

$$H(u, v) = -4\pi^2(u^2 + v^2) = -4\pi^2 D^2(u, v)$$

Highpass frequency filters - Gaussians

Common Gaussian: $H(u, v) = 1 - e^{-D^2(u, v)/2D_0^2}$

Highpass frequency filters - Gaussians

Common Gaussian: $H(u, v) = 1 - e^{-D^2(u, v)/2D_0^2}$

Laplace-of-Gaussian (i.e. *Marr-Hildreth* filter):

$$K_\sigma(x, y) = -\frac{1}{\pi\sigma^4} \left(1 - \frac{r^2}{2\sigma^2}\right) e^{-\frac{r^2}{2\sigma^2}}$$

(spatial description)



Summary of highpass filters

TABLE 4.5

Highpass filters. D_0 is the cutoff frequency and n is the order of the Butterworth filter.

Ideal	Butterworth	Gaussian
$H(u, v) = \begin{cases} 1 & \text{if } D(u, v) \leq D_0 \\ 0 & \text{if } D(u, v) > D_0 \end{cases}$	$H(u, v) = \frac{1}{1 + [D_0/D(u, v)]^{2n}}$	$H(u, v) = 1 - e^{-D^2(u,v)/2D_0^2}$

Overview of common highpass filters.

Bandreject (or Bandstop) filter

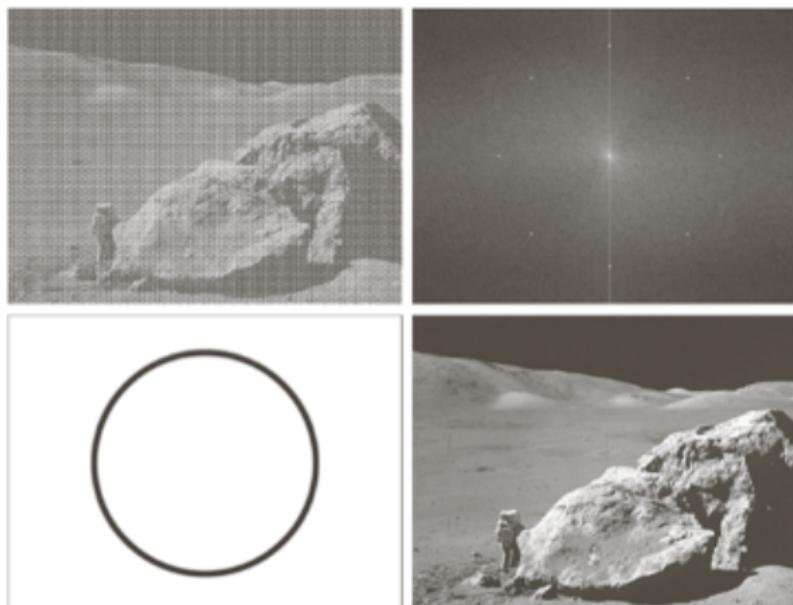
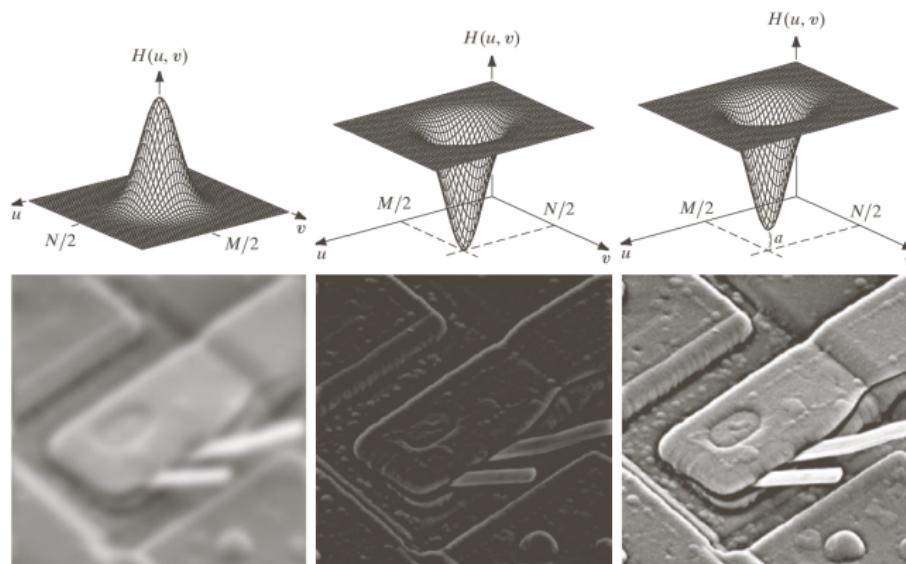


Figure: (a) Image corrupted by sinusoidal noise. (b) Spectrum.
(c) Butterworth bandreject filter. (d) Filtered image.

Bandpass filter - Difference-of-Gaussian

$$H(u, v) = Ae^{-D(u,v)^2/2\sigma_1^2} - Be^{-Du,v)^2/2\sigma_2^2} \quad \text{for } a = A - B, A > B$$



Other frequency transformations

Observation 1: We can represent an image as accumulated frequency coefficients.

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- basic wave function $r(x, y, u, v) = f(x, y) \rightarrow F(u, v): e^{-i2\pi(u \frac{x}{M} + v \frac{y}{N})}$

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Transformations & basic wave functions

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- Fourier
- Walsh-Hadamard

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- Wavelets → Gabor, Daubechies, Haar, Shannon, Spline, etc.

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Question: How can we use those base functions ?

Frequency transformations - Base functions

Fourier example:

$$F(u, v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y) e^{-i2\pi(u \frac{x}{M} + v \frac{y}{N})} \quad \forall u \in \{0..M-1\}, v \in \{0..N-1\}$$

Reformulation of the *forward, discrete transform*

$$T(u, v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y) e^{-i2\pi(u \frac{x}{M} + v \frac{y}{N})} \quad \forall u \in \{0..M-1\}, v \in \{0..N-1\}$$

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Frequency transformations - Base functions

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The *inverse, discrete transform* respectively:

$$g(x, y) = \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} T(u, v) s(x, y, u, v) \quad \forall x \in \{0..M-1\}, y \in \{0..N-1\}$$

Frequency transformations - Base functions

Fourier example:

$$\begin{aligned} r(x, y, u, v) &= e^{-j2\pi(u \frac{x}{M} + v \frac{y}{N})} \\ &= e^{-j2\pi u \frac{x}{M}} e^{-j2\pi v \frac{y}{N}} \\ &= r_1(x, u) r_2(y, v) \\ s(x, y, u, v) &= \frac{1}{M} e^{j2\pi x \frac{u}{M}} \frac{1}{N} e^{j2\pi y \frac{v}{N}} = \frac{1}{NM} e^{j2\pi(x \frac{u}{M} + y \frac{v}{N})} \end{aligned}$$

Frequency transformations - Base functions

Fourier example:

$$\begin{aligned} r(x, y, u, v) &= e^{-j2\pi(u \frac{x}{M} + v \frac{y}{N})} \\ &= e^{-j2\pi u \frac{x}{M}} e^{-j2\pi v \frac{y}{N}} \\ &= r_1(x, u) r_2(y, v) \\ s(x, y, u, v) &= \frac{1}{M} e^{j2\pi x \frac{u}{M}} \frac{1}{N} e^{j2\pi y \frac{v}{N}} = \frac{1}{NM} e^{j2\pi(x \frac{u}{M} + y \frac{v}{N})} \end{aligned}$$

Next step: let us replace $r(x, y, u, v)$ and $s(x, y, u, v)$ with another base function.

Frequency transforms - Cosines

$$T(u, v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} g(x, y) r(x, y, u, v) \quad \forall u \in \{0..M-1\}, v \in \{0..N-1\}$$

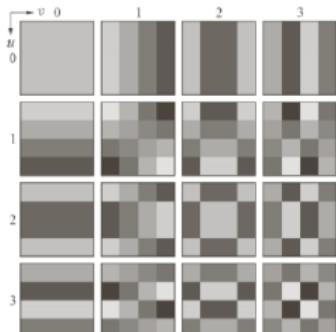
$$g(x, y) = \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} T(u, v) s(x, y, u, v) \quad \forall x \in \{0..M-1\}, y \in \{0..N-1\}$$

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Cosine transform: $r(x, y, u, v) = s(x, y, u, v)$



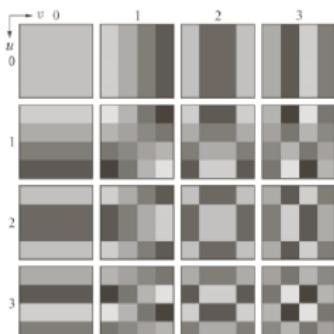
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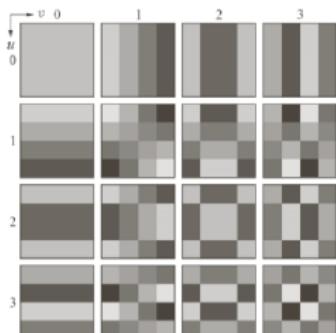
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$$\alpha(u) = \begin{cases} \sqrt{1/M} & \text{for } u = 0 \\ \sqrt{2/M} & \text{for } u > 0 \end{cases}$$

($\alpha(v)$ analogous)

Discrete Cosine Transform - Applications

Where can we use those transformations ?

- ① high- and lowpass filtering → $T(u, v)$ are still all *frequency* transforms with coefficients.

Discrete Cosine Transform - Applications

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Discrete Cosine Transform - Applications

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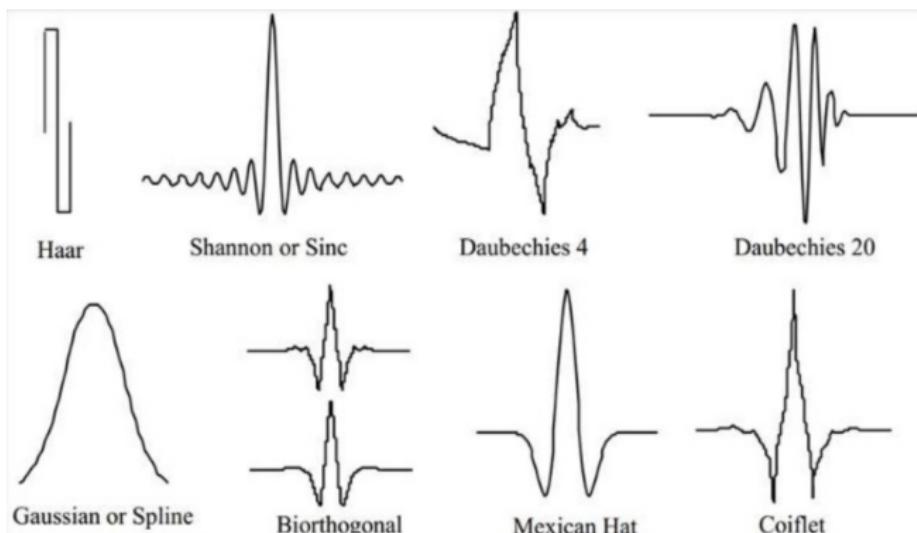
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- ③ image *compression* → see this next time.

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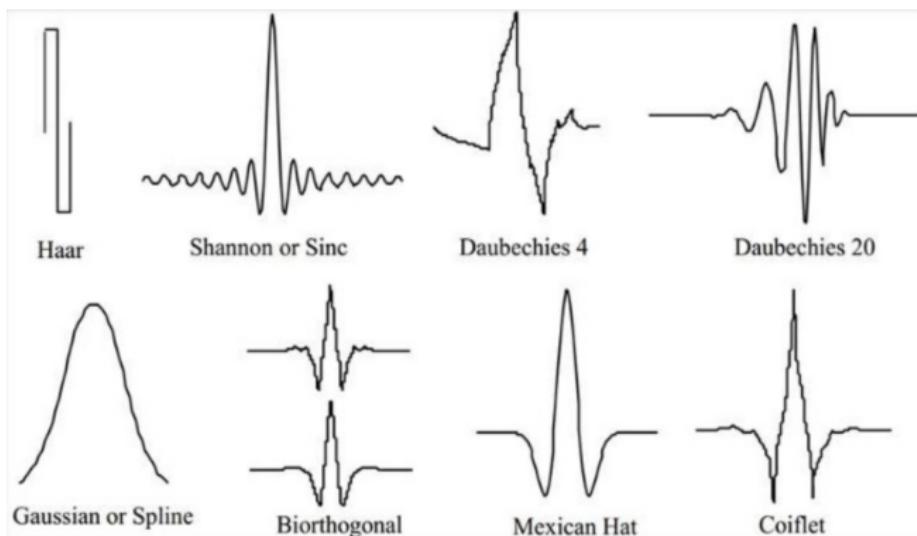
Frequency transformations - Wavelet bases



application depends on:

- the use of the coefficients (filter, store, reconstruct)

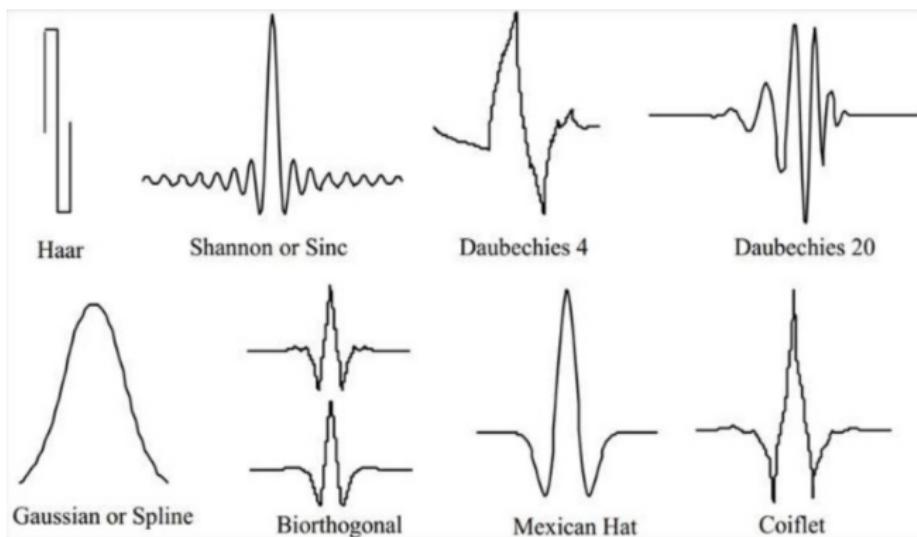
Frequency transformations - Wavelet bases



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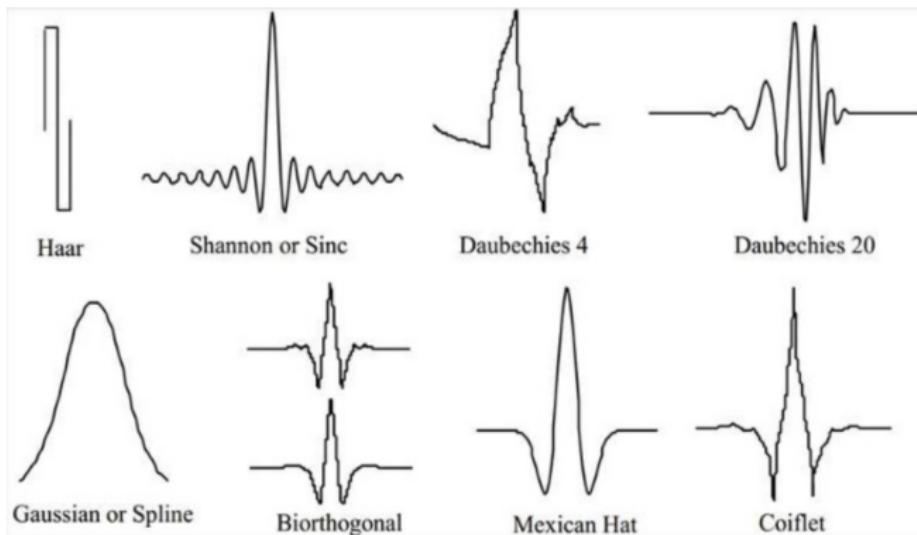
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Frequency transformations - Wavelet bases



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- the use of the coefficients (filter, store, reconstruct)
- required precision: double, float, half-float, integer, byte
- the need for a *symmetric* base function
- the ease of function simplification (i.e. kernelization)

That's it for this week!

