





Image, Video, and Multimedia Communications Laboratory

Digital Image Processing

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> > 2016



Topic

Simple review of Fourier transform

Filtering in frequency domain





Simple review of Fourier transform





The Fourier Transform

• The Fourier transform converts a signal from spatial domain to the frequency domain.

• The Continuous Fourier Transform of a one-dimensional continuous function *f*(*t*) is defined as

$$\mathbf{F}{f(t)} = F(s) = \int_{-\infty}^{\infty} f(t)e^{-j2\pi st}dt$$





The inverse Fourier transform

$$\mathbf{F}^{-1}{F(s)} = \int_{-\infty}^{\infty} F(s)e^{j2\pi st} ds$$

$$F(s)$$
 Inverse Fourier Transform $\longrightarrow f(t)$



The Discrete Fourier Transform

• If $\{f_i\}$ is a sequence of length N, then its *discrete* Fourier transform (DFT) is given by

$$F_n = \frac{1}{N} \sum_{i=0}^{N-1} f_i e^{-j2\pi \frac{n}{N}i}, \quad n = 0, 1, \dots N-1$$

And the inverse DFT is given by

$$f_i = \sum_{n=0}^{N-1} F_n e^{-j2\pi \frac{i}{N}n}, \quad i = 0, 1, \dots, N-1$$

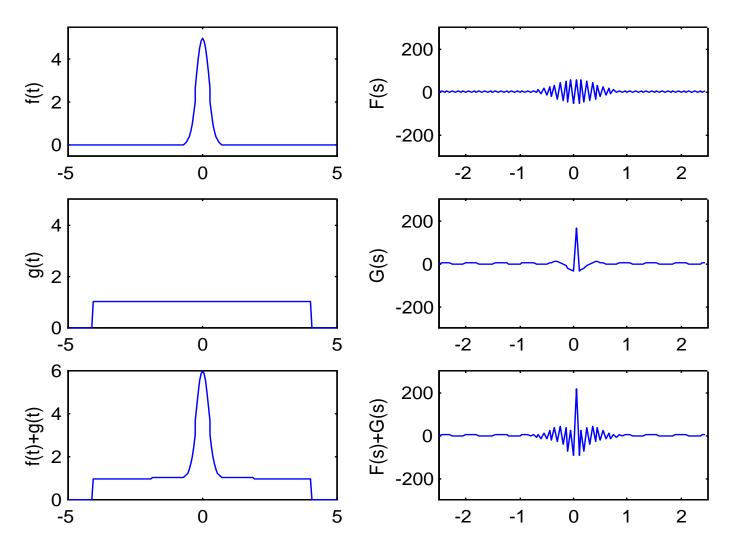
Where $0 \le i, n \le N-1$ are indices.



Properties of the Fourier Transform

- The Addition Theorem
 - If $\mathbf{F}{f(t)} = F(s)$ and $\mathbf{F}{g(t)} = G(s)$, then $\mathbf{F}{f(t) + g(t)} = F(s) + G(s)$
 - And take it as an axiom that for any real number c $\mathbf{F}\{cf(t)\} = cF(s)$
 - This implies that Fourier transform is a linear transform.







The Shift Theorem

• Time shift:

$$\mathbf{F}\{f(t-a)\} = e^{-j2\pi as}F(s)$$

• Frequency shift:

$$\mathbf{F}\{f(t)e^{j2\pi s_0 t/N}\} = F(s-s_0)$$

When
$$s_0 = N/2$$
,

$$\mathbf{F}{f(t)(-1)^t} = F(s-N/2)$$



The Convolution Theorem

 The Fourier transform of the convolution of two functions is the product of the Fourier transforms of these two functions

$$\mathbf{F}\{f(t) * g(t)\} = F(s)G(s)$$



The Two-Dimensional DFT

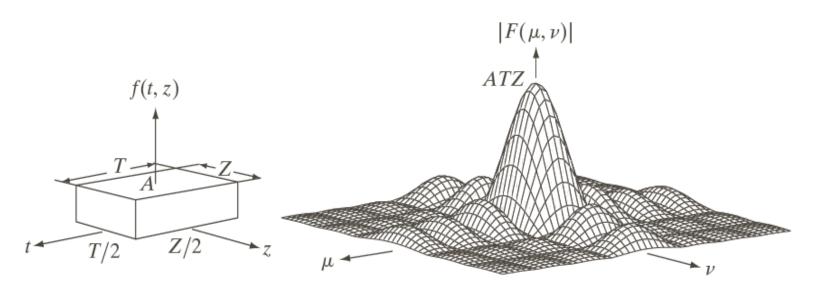
• The discrete Fourier transform of an image of size M*N is given by

$$F(u,v) = \frac{1}{MN} \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x,y) e^{-j2\pi(ux/M + vy/N)}$$

• Inverse Fourier transform

$$f(x,y) = \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} F(u,v) e^{j2\pi(ux/M + vy/N)}$$





a b

FIGURE 4.13 (a) A 2-D function, and (b) a section of its spectrum (not to scale). The block is longer along the *t*-axis, so the spectrum is more "contracted" along the μ -axis. Compare with Fig. 4.4.





Fourier spectrum

$$|F(u,v)| = [R^2(u,v)+I^2(u,v)]$$

Phase angle

$$\phi(u,v) = \tan^{-1} \left[\frac{I(u,v)}{R(u,v)} \right]$$

Power spectrum

$$P(u,v) = |F(u,v)|^{2}$$
$$= R^{2}(u,v) + I^{2}(u,v)$$

where R(u, v) and I(u, v) are the real and imaginary parts of F(u, v), respectively



• For display, it is common practice to shift the original point to the center (M/2, N/2). According to the shift theorem, we need to multiply the input image function by $(-1)^{x+y}$ prior to computing the Fourier transform.:

$$\mathcal{F}\left[f(x,y)(-1)^{x+y}\right] = F(u-M/2,v-N/2)$$



• What's the value of the transform at (u, v)=(0, 0)?

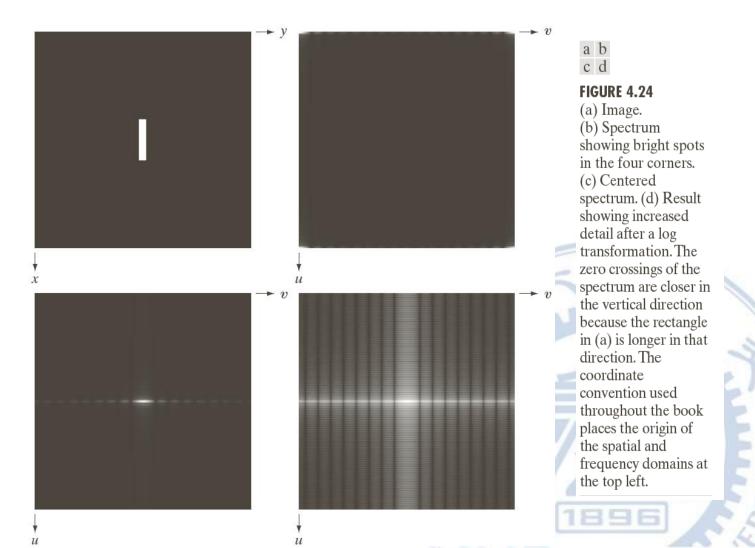
$$F(0,0) = \frac{1}{MN} \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x,y)$$

• The average of f(x, y).



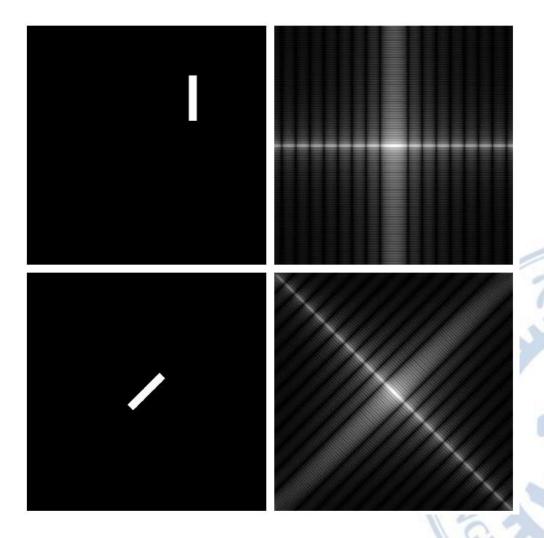


A simple example





Translation and Rotation



a b c d

FIGURE 4.25

(a) The rectangle in Fig. 4.24(a) translated, and (b) the corresponding spectrum. (c) Rotated rectangle, and (d) the corresponding spectrum. The spectrum corresponding to the translated rectangle is identical to the spectrum corresponding to the original image in Fig. 4.24(a).



The 2-D Convolution Theorem

• 2-D circular convolution

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

The 2-D convolution theorem is given by

$$f(x,y)\star h(x,y) \Leftrightarrow F(u,v)H(u,v)$$
 and, conversely,

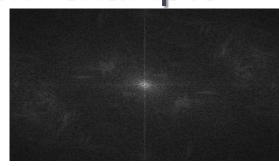
$$f(x,y)h(x,y) \Leftrightarrow F(u,v) \star H(u,v)$$



2D convolution theorem example

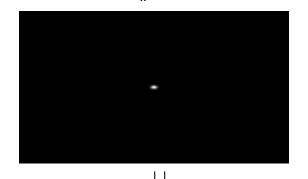






 $|F(s_x,s_y)|$

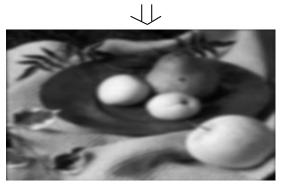
h(x,y)





 $|H(s_x, s_y)|$

g(x,y)





 $|G(s_x,s_y)|$



Example: cheetah pic



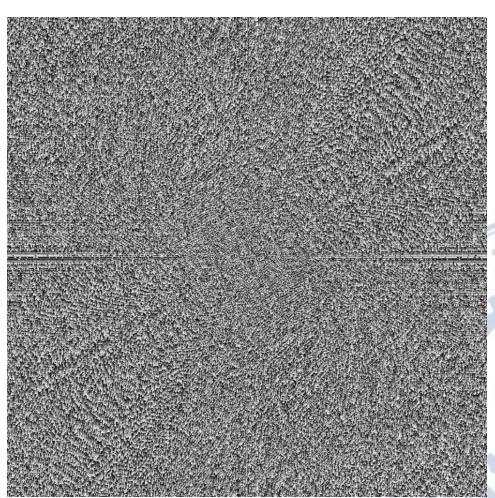


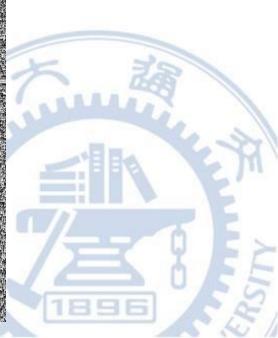
This is the magnitude transform of the cheetah pic





This is the phase transform of the cheetah pic





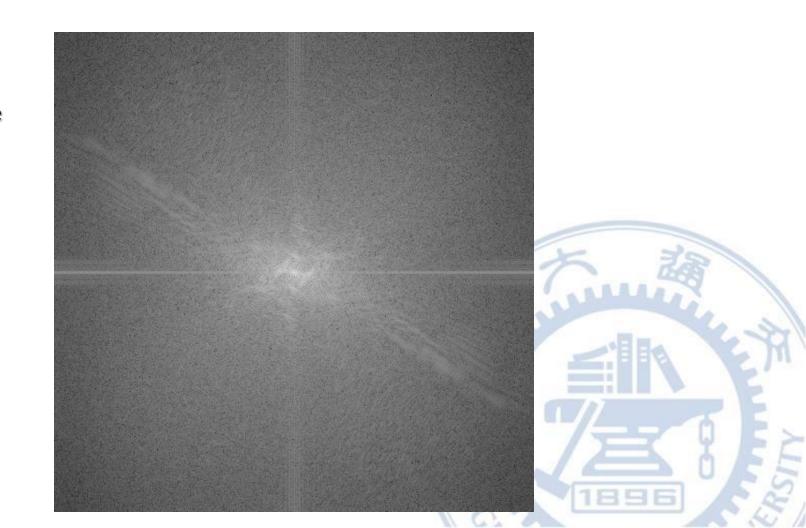


Example: zebra pic





This is the magnitude transform of the zebra pic



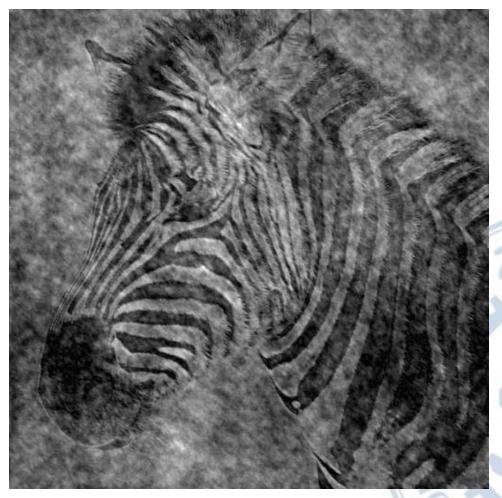


This is the phase transform of the zebra pic





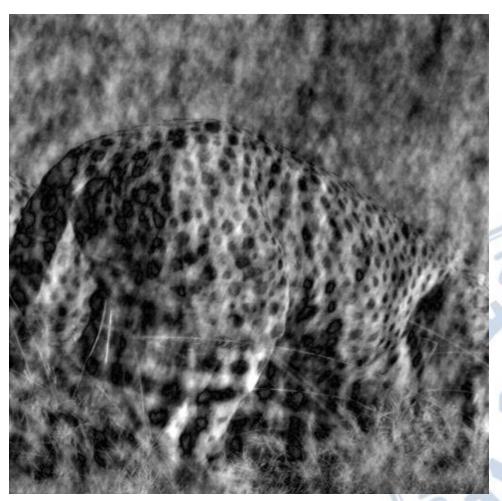
Reconstru ction with zebra phase, cheetah magnitude







Reconstru ction with cheetah phase, zebra magnitude



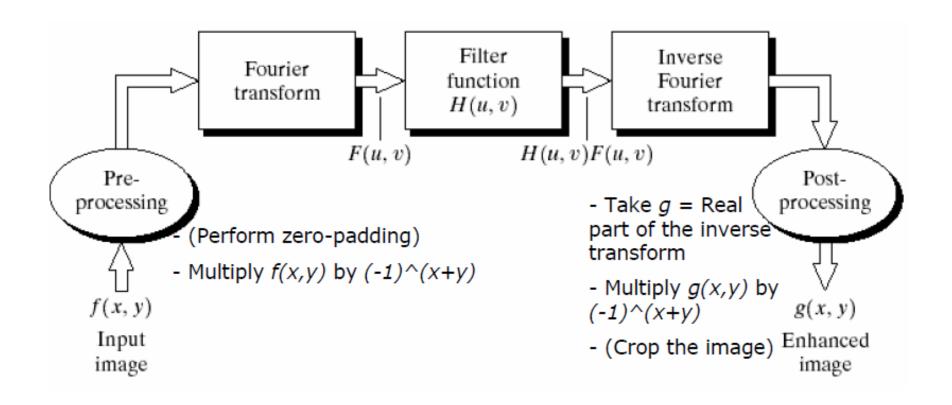






Filtering in frequency domain







Steps for filtering in frequency domain

- 1) Multiply the input image by (-1)^{x+y} to center the transform
- 2) Zero padding in case of aliasing
- 3) Compute F(u, v), the DFT of the image from (2)
- 4) Multiply F(u, v) by a filter function H(u, v)
- 5) Compute the inverse DFT of the result in (4)
- 6) Obtain the real part of the result in (5)
- 7) Multiply the result in (6) by $(-1)^{x+y}$
- 8) Crop the image

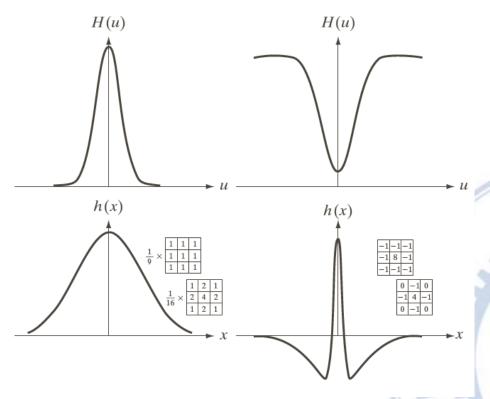


• h(x, y) is a spatial filter, and can be obtained from the response of a frequency domain filter to an impulse.

 h(x, y) sometimes is referred to as the impulse response of H(u, v)



Gaussian low-pass filter and high-pass filter in the frequency and spatial domain



a c b d

FIGURE 4.37

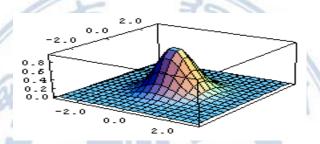
(a) A 1-D Gaussian lowpass filter in the frequency domain. (b) Spatial lowpass filter corresponding to (a). (c) Gaussian highpass filter in the frequency domain. (d) Spatial highpass filter corresponding to (c). The small 2-D masks shown are spatial filters we used in Chapter 3.



A Gaussian kernel gives less weight to pixels further from the center of the window

0	0	0	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	90	0	90	90	90	0	0
0	0	0	90	90	90	90	90	0	0
0	0	0	0	0	0	0	0	0	0
0	0	90	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0	0	0

1	1	2	1
<u>-</u>	2	4	2
10	1	2	1

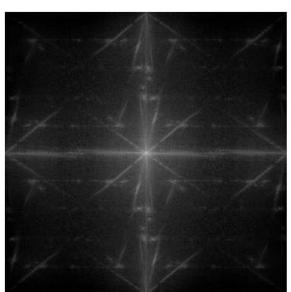


$$h(u,v) = \frac{1}{2\pi\sigma^2}e^{-\frac{u-v}{\sigma^2}}$$

This kernel is an approximation of a Gaussian function:







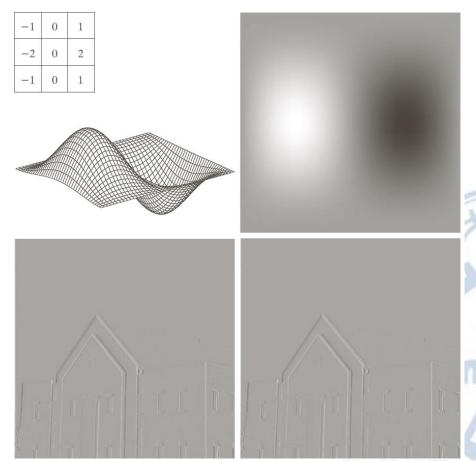
a b

FIGURE 4.38

(a) Image of a building, and (b) its spectrum.







a b

FIGURE 4.39

(a) A spatial mask and perspective plot of its corresponding frequency domain filter. (b) Filter shown as an image. (c) Result of filtering Fig. 4.38(a) in the frequency domain with the filter in (b). (d) Result of filtering the same image with the spatial filter in (a). The results are identical.



Image Smoothing

Ideal low-pass filters

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

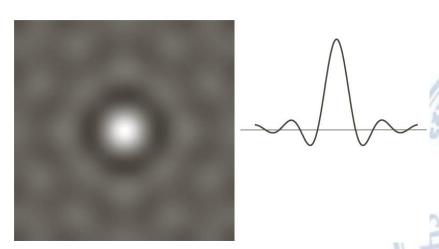
$$v = \begin{cases} 1 & \text{if } D(u,v) \leq D_0 \\ 0 & \text{if } D(u,v) \leq D_0 \end{cases}$$
a b c

FIGURE 4.40 (a) Perspective plot of an ideal lowpass-filter transfer function. (b) Filter displayed as an image. (c) Filter radial cross section.



Ideal low-pass filters

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



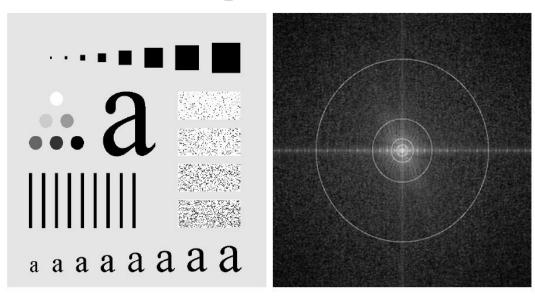
a b

FIGURE 4.43

(a) Representation in the spatial domain of an ILPF of radius 5 and size
1000 × 1000.
(b) Intensity profile of a horizontal line passing through the center of the image.

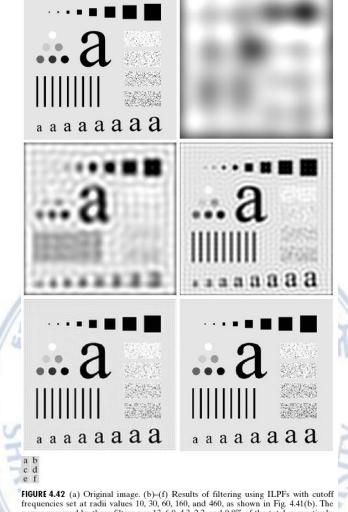


Ideal low-pass filters



a b

FIGURE 4.41 (a) Test pattern of size 688×688 pixels, and (b) its Fourier spectrum. The spectrum is double the image size due to padding but is shown in half size so that it fits in the page. The superimposed circles have radii equal to 10, 30, 60, 160, and 460 with respect to the full-size spectrum image. These radii enclose 87.0, 93.1, 95.7, 97.8, and 99.2% of the padded image power, respectively.



power removed by these filters was 13, 6.9, 4.3, 2.2, and 0.8% of the total, respectively.



Butterworth low-pass filters

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

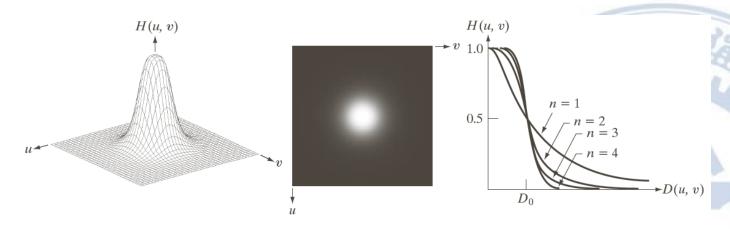


FIGURE 4.44 (a) Perspective plot of a Butterworth lowpass-filter transfer function. (b) Filter displayed as an image. (c) Filter radial cross sections of orders 1 through 4.



Butterworth low-pass filters

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

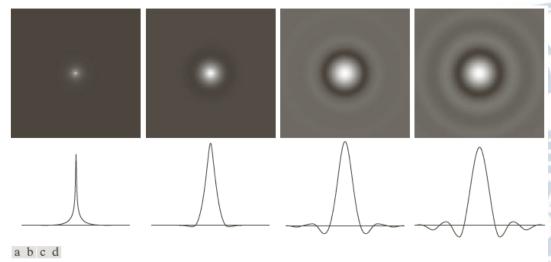


FIGURE 4.46 (a)–(d) Spatial representation of BLPFs of order 1, 2, 5, and 20, and corresponding intensity profiles through the center of the filters (the size in all cases is 1000×1000 and the cutoff frequency is 5). Observe how ringing increases as a function of filter order.





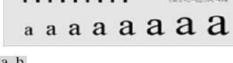






FIGURE 4.45 (a) Original image. (b)-(f) Results of filtering using BLPFs of order 2, with cutoff frequencies at the radii shown in Fig. 4.41. Compare with Fig. 4.42.



Gaussian low-pass filters

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

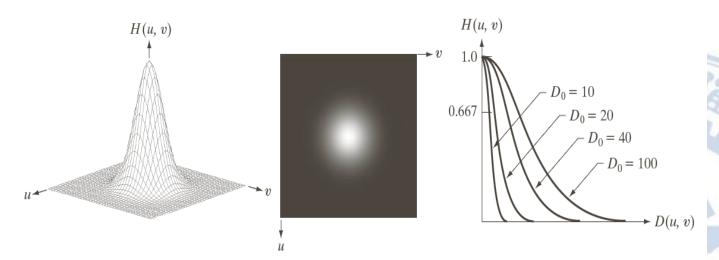
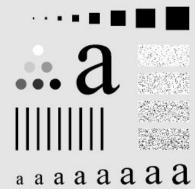
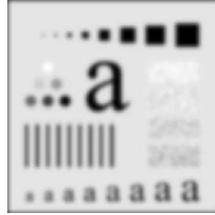


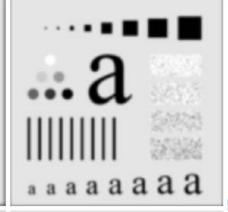
FIGURE 4.47 (a) Perspective plot of a GLPF transfer function. (b) Filter displayed as an image. (c) Filter radial cross sections for various values of D_0 .

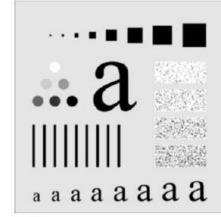












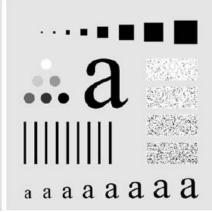




FIGURE 4.48 (a) Original image. (b)-(f) Results of filtering using GLPFs with cutoff frequencies at the radii shown in Fig. 4.41 Compare with Figs. 4.42 and 4.45



Gaussian low-pass filters

Historically, certain computer programs were written using only two digits rather than four to define the applicable year. Accordingly, the company's software may recognize a date using "00" as 1900 rather than the year 2000.

eа

Historically, certain computer programs were written using only two digits rather than four to define the applicable year. Accordingly, the company's software may recognize a date using "00" as 1900 rather than the year 2000.

a b

FIGURE 4.49

(a) Sample text of low resolution (note broken characters in magnified view). (b) Result of filtering with a GLPF (broken character segments were joined).



 A high-pass filter is obtained from a given lowpass filter

$$H_{HP}(u,v) = 1 - H_{LP}(u,v)$$



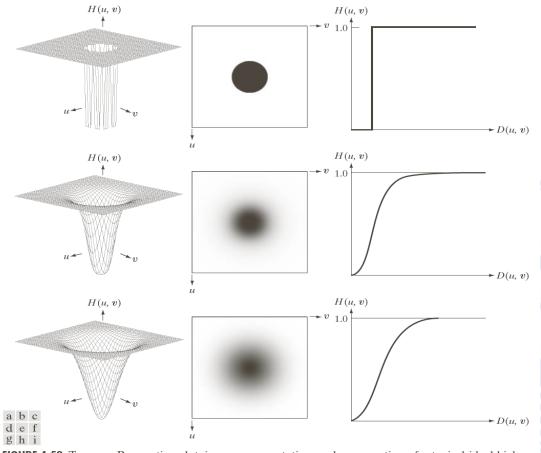


FIGURE 4.52 Top row: Perspective plot, image representation, and cross section of a typical ideal highpass filter. Middle and bottom rows: The same sequence for typical Butterworth and Gaussian highpass filters.



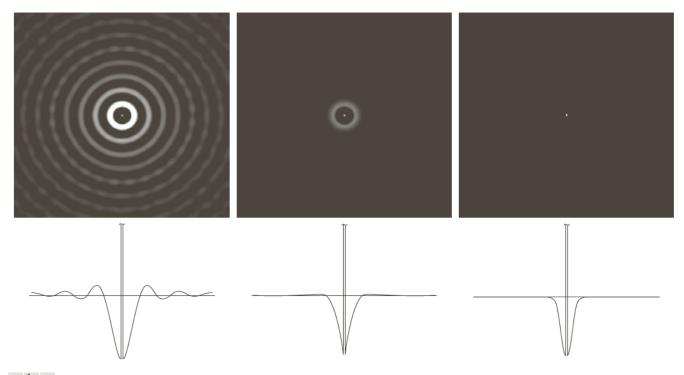
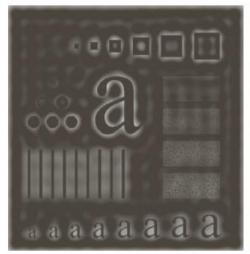


FIGURE 4.53 Spatial representation of typical (a) ideal, (b) Butterworth, and (c) Gaussian frequency domain highpass filters, and corresponding intensity profiles through their centers.



Ideal high-pass filters

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



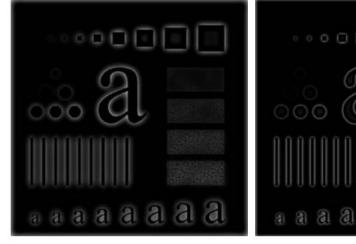






Butterworth high-pass filters

$$H(u,v) = \frac{1}{1 + \left\lceil D_0 / D(u,v) \right\rceil^{2n}}$$



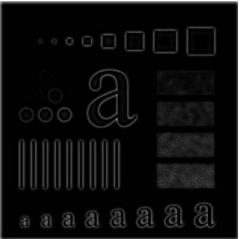


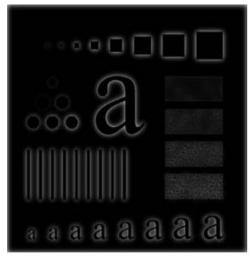


FIGURE 4.55 Results of highpass filtering the image in Fig. 4.41(a) using a BHPF of order 2 with $D_0 = 30, 60$, and 160, corresponding to the circles in Fig. 4.41(b). These results are much smoother than those obtained with an IHPF.



Gaussian high-pass filters

$$H(u,v) = 1 - e^{-D^2(u,v)/2\sigma^2}$$





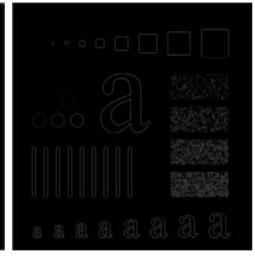


FIGURE 4.56 Results of highpass filtering the image in Fig. 4.41(a) using a GHPF with $D_0 = 30, 60, \text{ and } 160, \text{ corresponding to the circles in Fig. 4.41(b)}$. Compare with Figs. 4.54 and 4.55.



• The Laplacian in the frequency domain

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

$$\nabla^2 f(x,y) = \mathcal{F}^{-1} \left\{ H(u,v) F(u,v) \right\}$$



Sharpening Spatial Filters

- The Laplacian
 - Laplacian operator

$$\nabla^2 f = \frac{\partial^2 f}{\partial x^2} + \frac{\partial^2 f}{\partial y^2}$$

X-direction

$$\frac{\partial^2 f}{\partial x^2} = f(x+1,y) + f(x-1,y) - 2f(x,y)$$

Y-direction

$$\frac{\partial^2 f}{\partial y^2} = f(x, y+1) + f(x, y-1) - 2f(x, y)$$



Sharpening Spatial Filters

The discrete Laplacian of two variables

$$\nabla^{2} f(x, y) = f(x+1, y) + f(x-1, y) + f(x, y+1) + f(x, y-1) - 4f(x, y)$$

0	1	0	1	1	1
1	-4	1	1	-8	1
0	1	0	1	1	1
0	-1	0	-1	-1	-1
-1	4	-1	-1	8	-1
0	-1	0	-1	-1	-1

a b c d

FIGURE 3.37

(a) Filter mask used to implement Eq. (3.6-6). (b) Mask used to implement an extension of this equation that includes the diagonal terms. (c) and (d) Two other implementations of the Laplacian found frequently in practice.



Enhancement is achieved using the equation

$$g(x,y) = f(x,y) + c \left[\nabla^2 f(x,y)\right]$$

In frequency domain

$$g(x,y) = \mathcal{F}^{-1} \left\{ F(u,v) - H(u,v) F(u,v) \right\}$$
$$= \mathcal{F}^{-1} \left\{ \left[1 - H(u,v) \right] F(u,v) \right\}$$
$$= \mathcal{F}^{-1} \left\{ \left[1 + 4\pi^2 D^2(u,v) \right] F(u,v) \right\}$$



- High-boost filtering
 - Spatial domain:

$$g(x,y) = f(x,y) + k \cdot g_{mask}(x,y)$$
$$g_{mask}(x,y) = f(x,y) - f_{LP}(x,y)$$

Frequency domain

$$g(x,y) = \mathcal{F}^{-1} \left\{ 1 + k * \left[1 - H_{LP}(u,v) \right] F(u,v) \right\}$$
$$f_{LP}(x,y) = \mathcal{F}^{-1} \left[H_{LP}(u,v) F(u,v) \right]$$



High-frequency-emphasis filtering:

$$g(x,y) = \mathcal{F}^{-1} \left\{ 1 + k * H_{HP}(u,v) F(u,v) \right\}$$

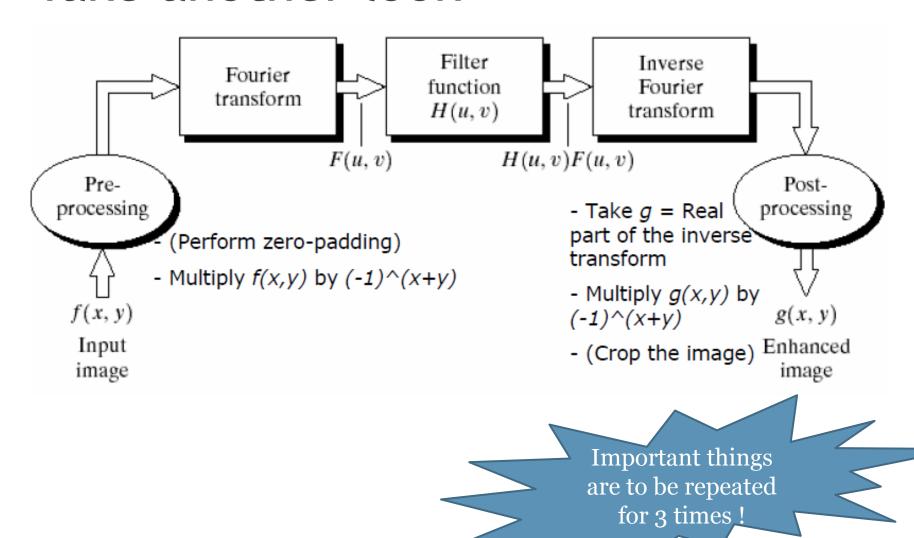


 A slightly more general formulation of high-frequencyemphasis filtering

$$g(x,y) = \mathcal{F}^{-1}\left\{k_1 + k_2 * H_{HP}(u,v)F(u,v)\right\}$$



Take another look





Homework One

Page327: problem 4.8(b), 4.22, 4.28, 4.33, 4.35





Thank You!