

# Image Enhancement

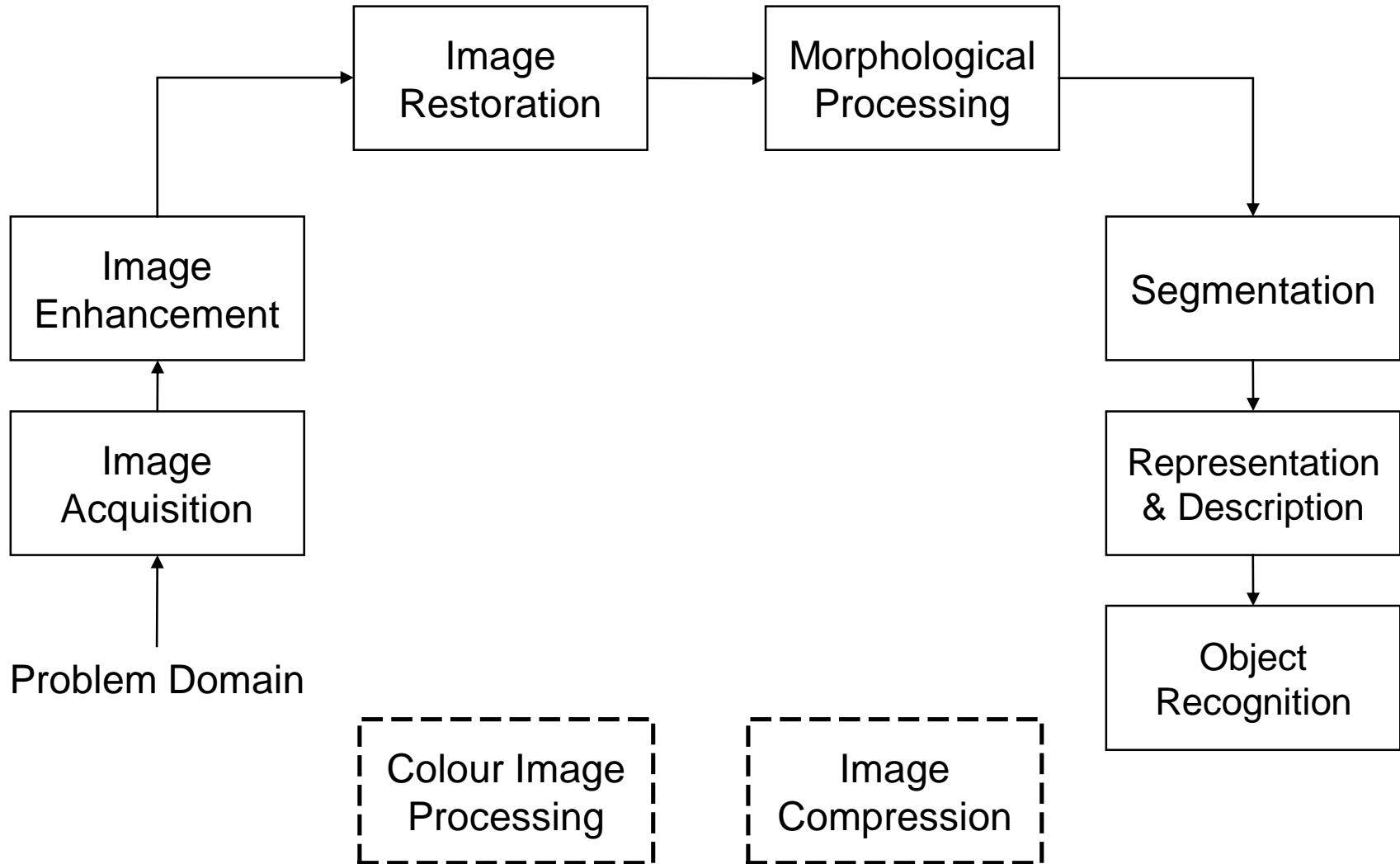
## In Frequency Domain

*Partially Adopted from Brian Mac Namee*

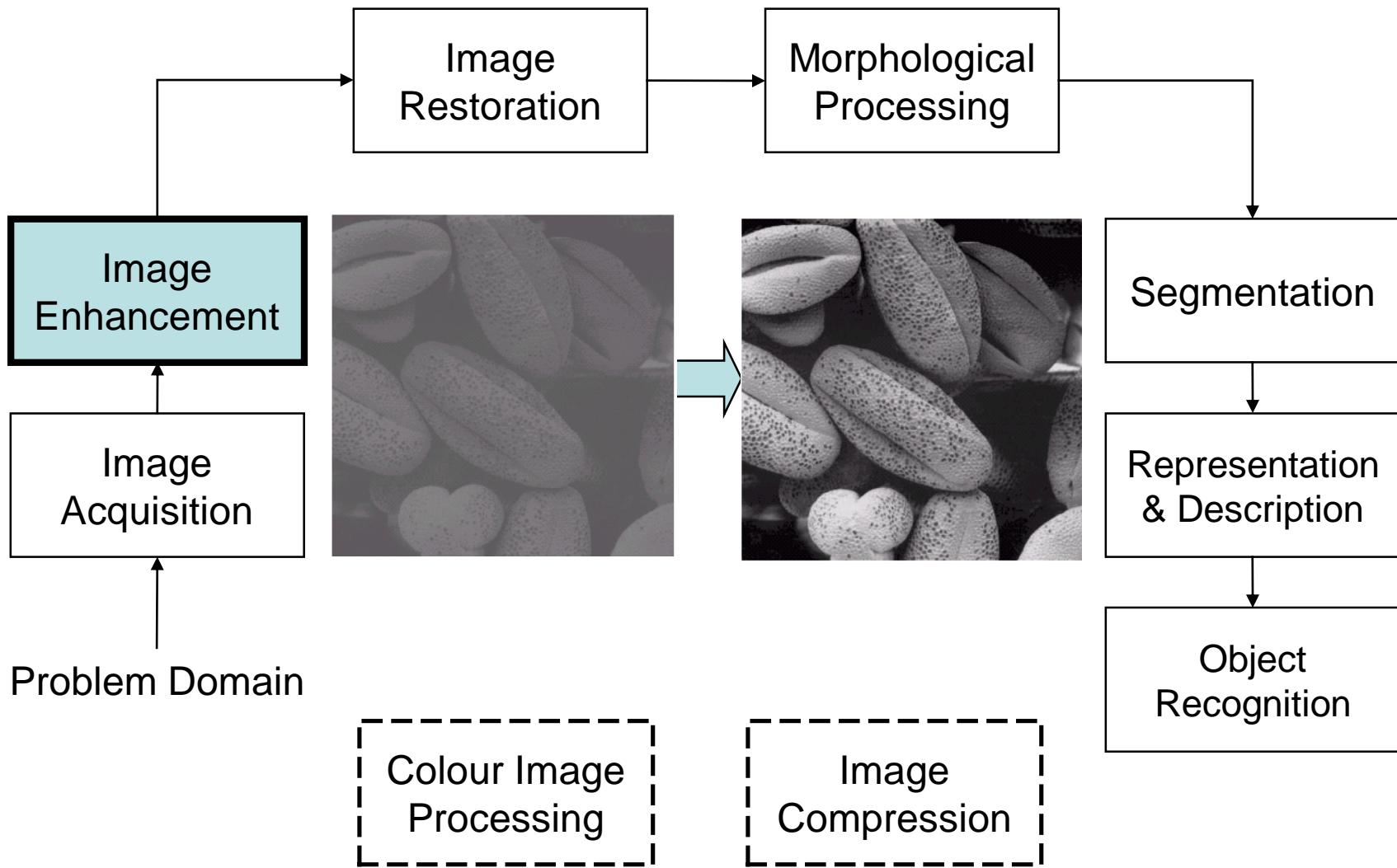
# Contents

- ✓ Jean Baptiste Joseph Fourier
- ✓ The Fourier Series & the Fourier Transform
- ✓ Image Processing in the Frequency Domain
  - Image smoothing
  - Image sharpening
- ✓ Fast Fourier Transform

# Phases of Digital Image Processing



# Phases of Digital Image Processing: Image Enhancement



# Jean Baptiste Joseph Fourier



Fourier was born in Auxerre, France in 1768.

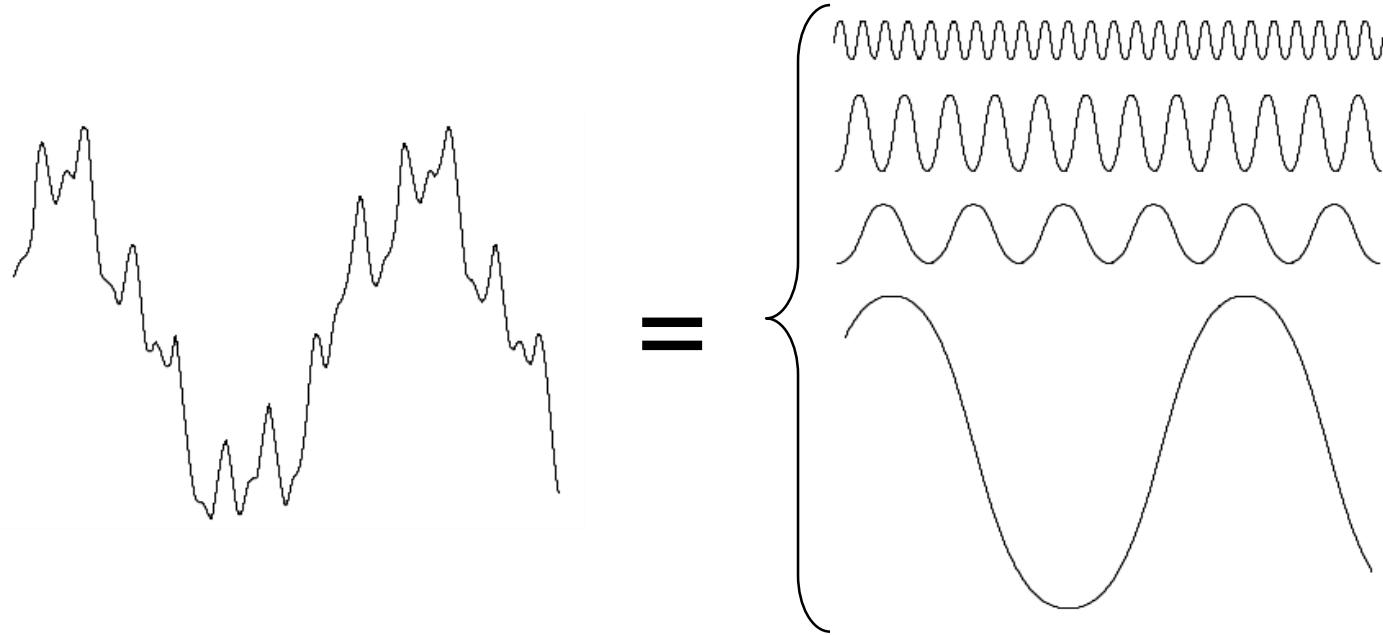
Most famous for his work “*La Théorie Analytique de la Chaleur*” published in 1822.

Translated into English in 1878 by Freeman: “*The Analytic Theory of Heat*”.

Nobody gave much attention when the work was first published.

One of the most important mathematical theories in Modern Engineering / Science.

# Theoretical View



Any function that periodically repeats itself can be expressed as the sum of sines and/or cosines of different frequencies, each multiplied by a different coefficient – a Fourier series.

# Theoretical View

Even function that are not periodic (but whose area under the curve is finite) can be expressed as the integral of sines and/or cosines multiplied by a weighing function – a Fourier transform.

Its utility is even greater than the Fourier series in most practical problems.

A function, expressed in either Fourier series or Fourier transform, can be reconstructed completely via an inverse function, with no loss of information.

# The One-Dimensional Fourier Transform and its Inverse

The Fourier transform,  $F(u)$ , of a single variable, continuous function,  $f(x)$  is defined by:

$$F(u) = \int_{-\infty}^{\infty} f(x) e^{-j2\pi ux} dx, \text{ where } j = \sqrt{-1}.$$

Conversely, given  $F(u)$ , we can obtain  $f(x)$  by means of the inverse Fourier transform

$$f(x) = \int_{-\infty}^{\infty} F(u) e^{j2\pi ux} du, \quad \text{where } j = \sqrt{-1}.$$

# The two-Dimensional Fourier Transform and its Inverse

$$F(u, v) = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} f(x, y) e^{-j2\pi(ux+vy)} dx dy$$

$$f(x, y) = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} F(u, v) e^{j2\pi(ux+vy)} du dv$$

# The Discrete Fourier Transform (DFT)

The *Discrete Fourier Transform* of  $f(x, y)$ , for  $x = 0, 1, 2 \dots M-1$  and  $y = 0, 1, 2 \dots N-1$ , denoted by  $F(u, v)$ , is given by the equation:

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

for  $u = 0, 1, 2 \dots M-1$  and  $v = 0, 1, 2 \dots N-1$ .

# The Discrete Fourier Transform (DFT)

- Fourier Spectrum:

$$|F(u, v)| = [R^2(u, v) + I^2(u, v)]^{1/2}$$

- Phase Angle / Phase Spectrum:

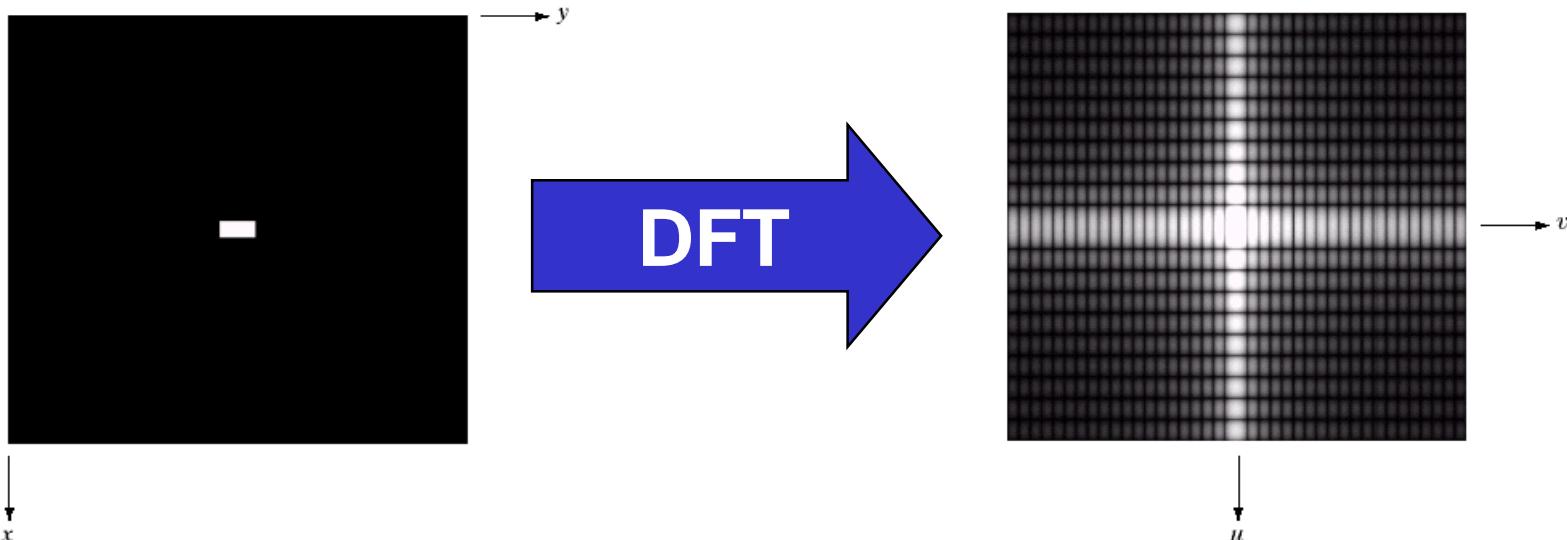
$$\phi(u, v) = \tan^{-1} \left[ \frac{I(u, v)}{R(u, v)} \right]$$

- Power Spectrum:

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

# DFT of an Image

The DFT of a two dimensional image can be visualised by showing the spectrum of the image component frequencies.

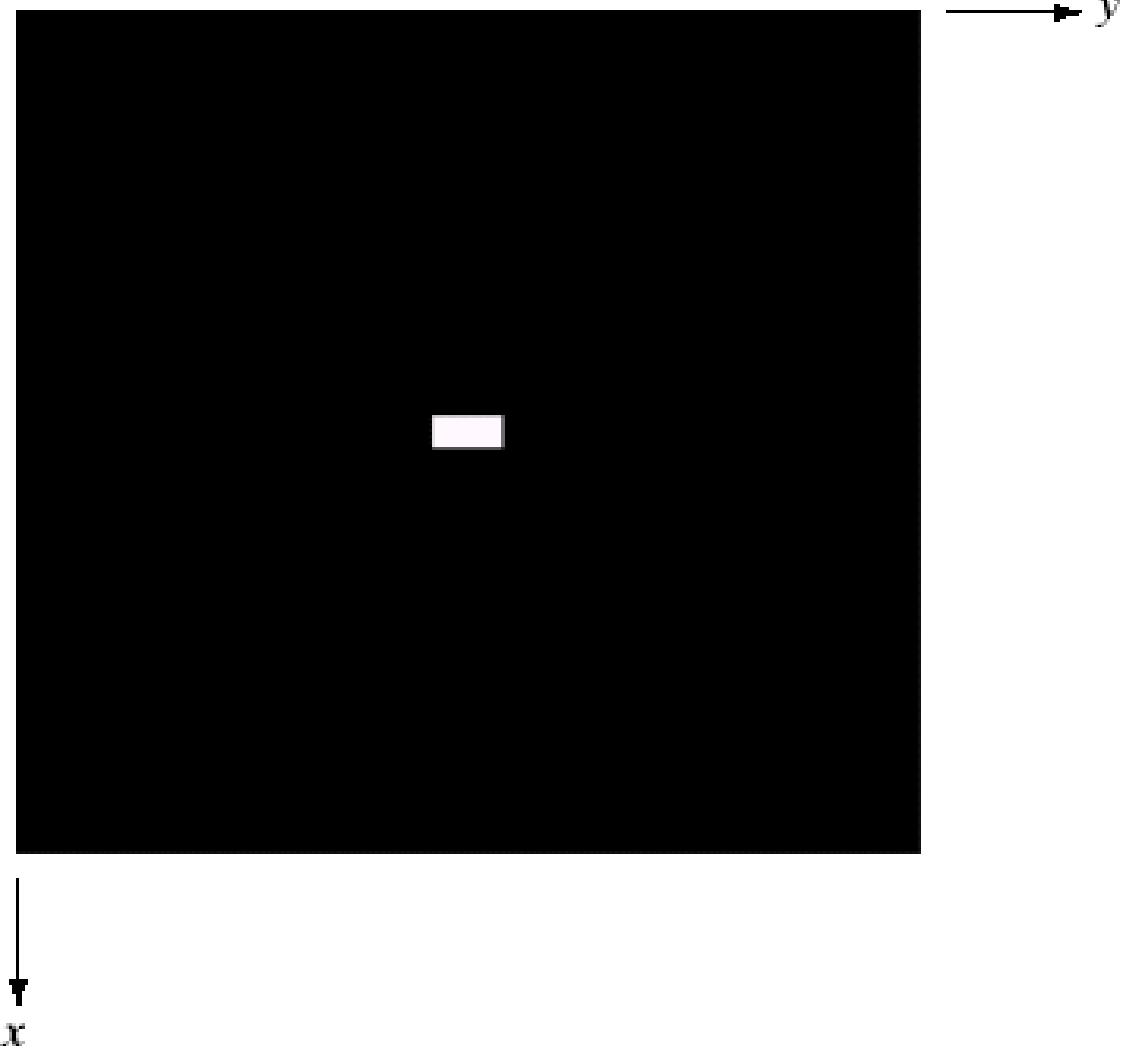


It is common practice to multiply the input image function by  $(-1)^{x+y}$  prior to computing Fourier transform.

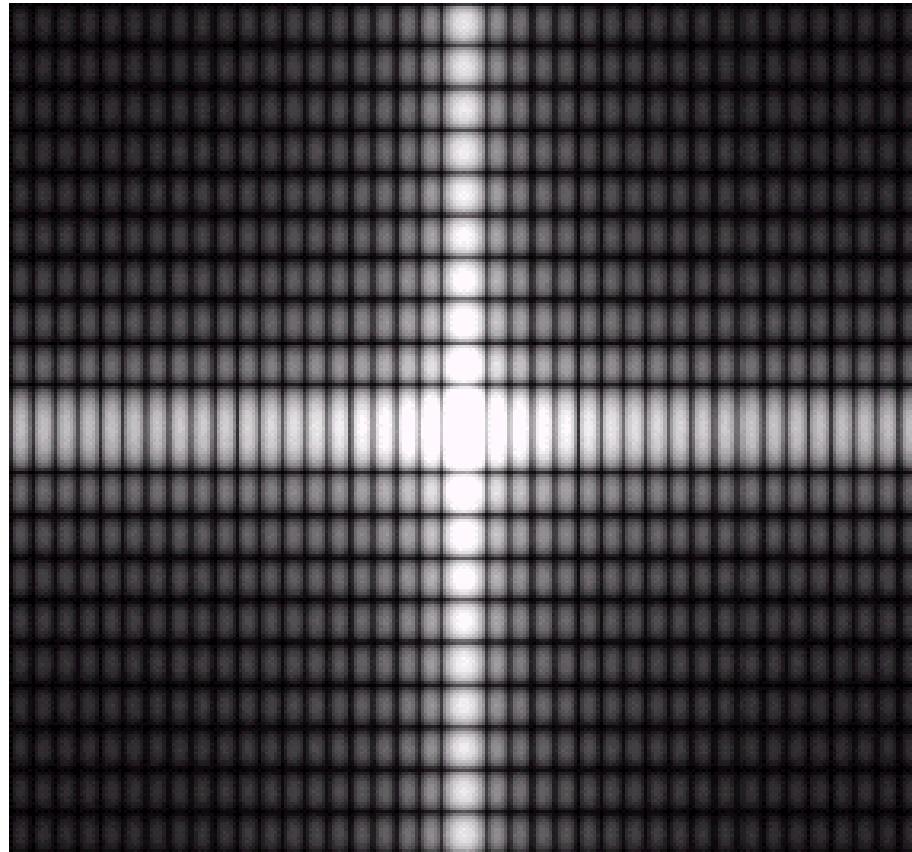
This operation shifts the origin of FT of  $f(x,y)(-1)^{x+y}$  (i.e.  $F(0,0)$ ) to  $[M/2, N/2]$ .



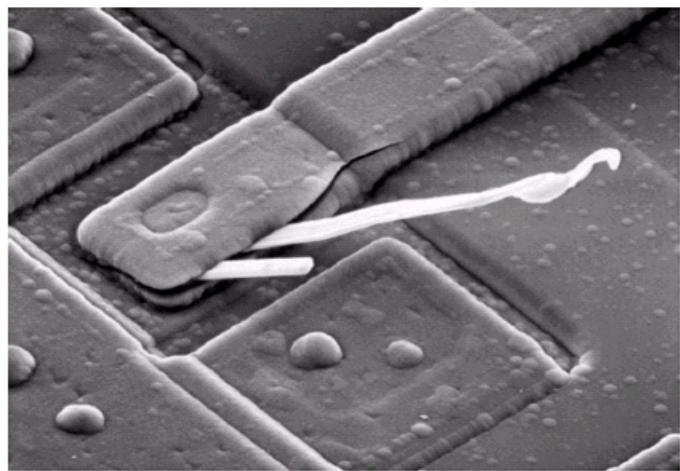
# DFT of an Images



# DFT of an Image

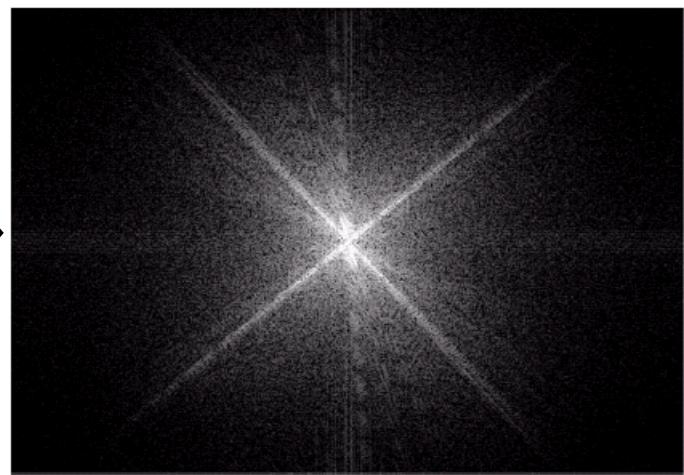


# DFT of an Image



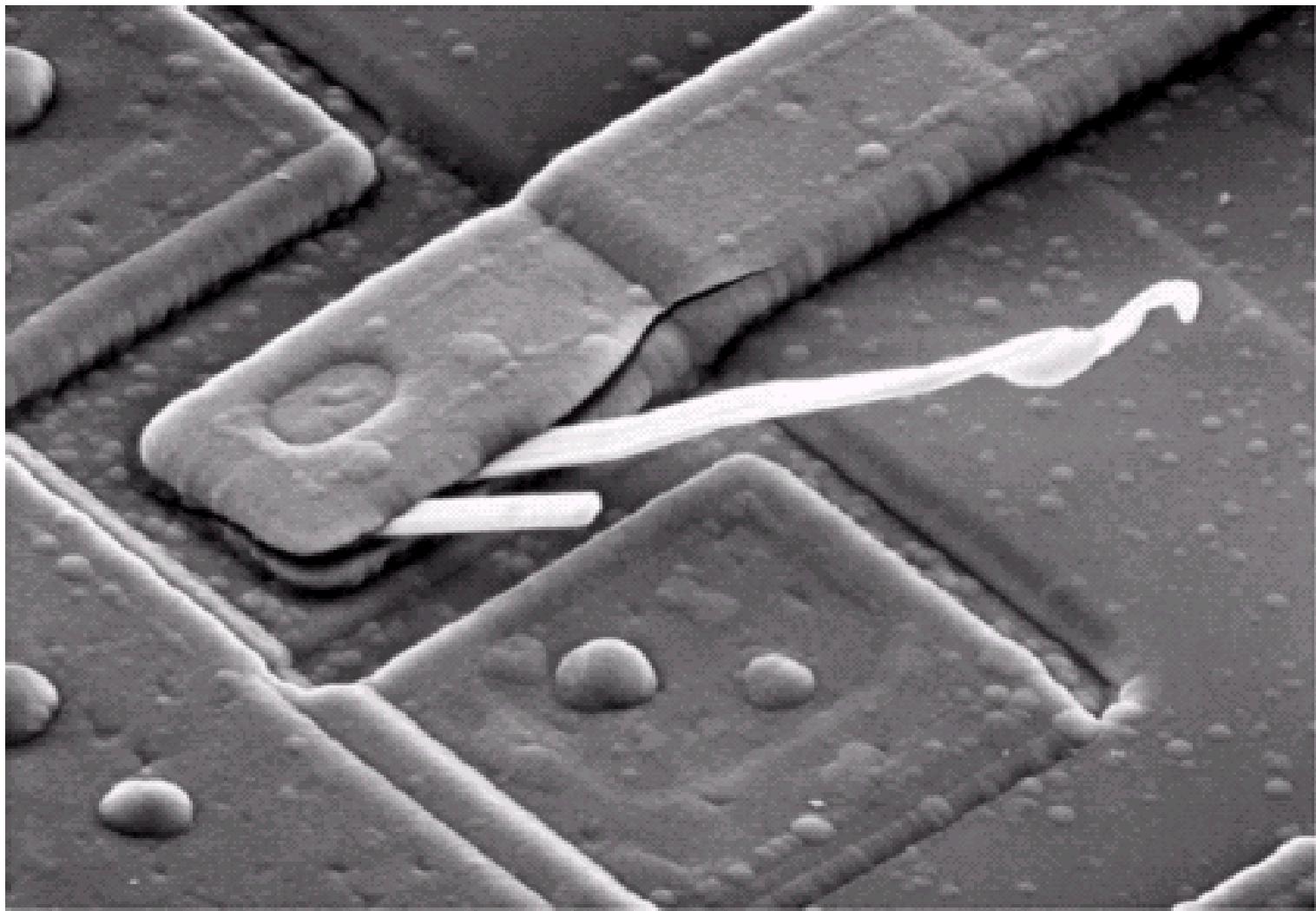
Scanning electron microscope  
image of an integrated circuit  
magnified ~2500 times

DFT

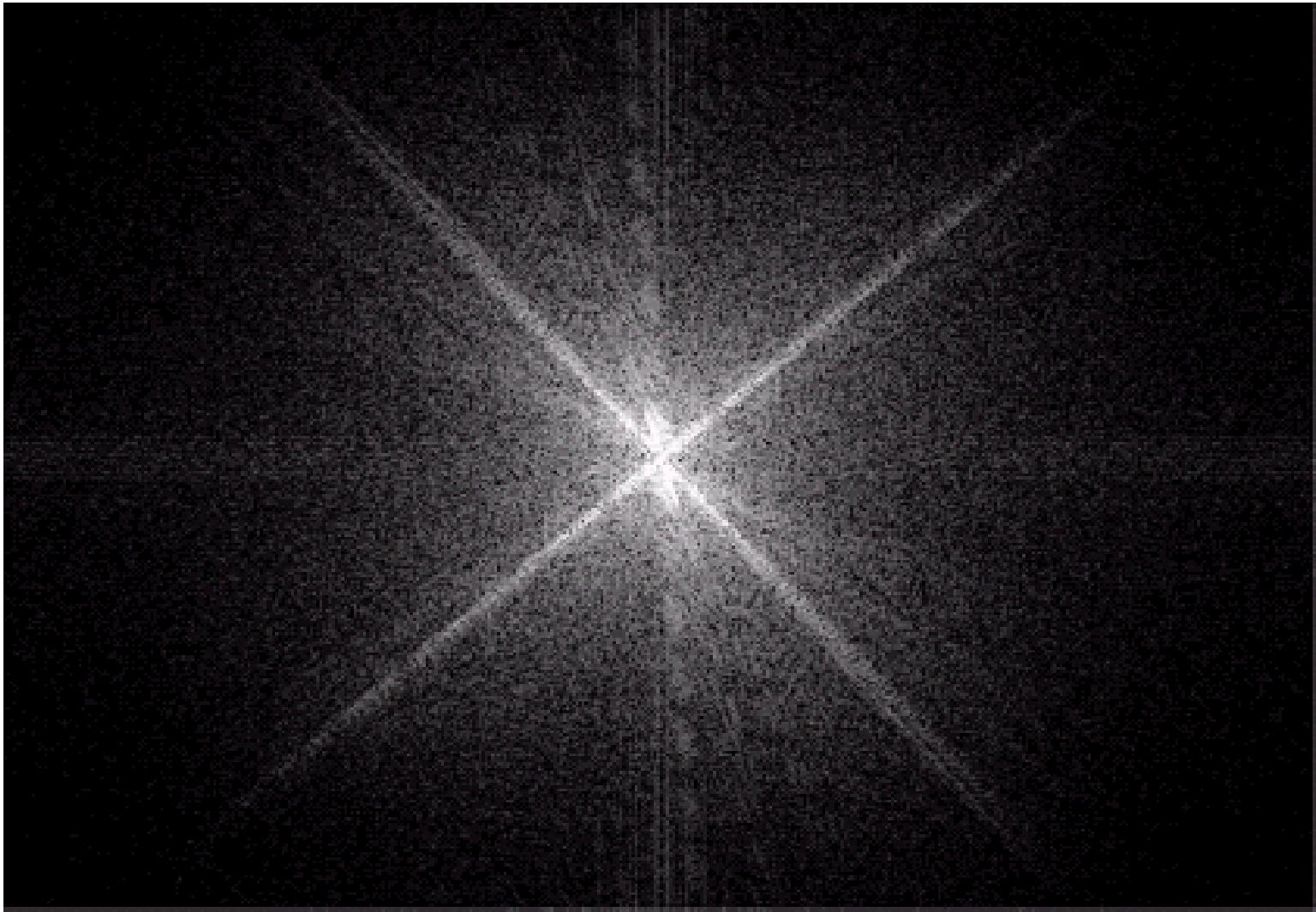


Fourier spectrum of the image

# DFT of an Image



# DFT of an Image



# The Inverse DFT

It is really important to note that the Fourier transform is completely **reversible**.

The inverse DFT is given by:

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

for  $x = 0, 1, 2 \dots M-1$  and  $y = 0, 1, 2 \dots N-1$

# Basics of Filtering in Frequency Domain

To filter an image in the frequency domain:

1. Multiply the input image by  $(-1)^{x+y}$  to center the transform, as indicated in

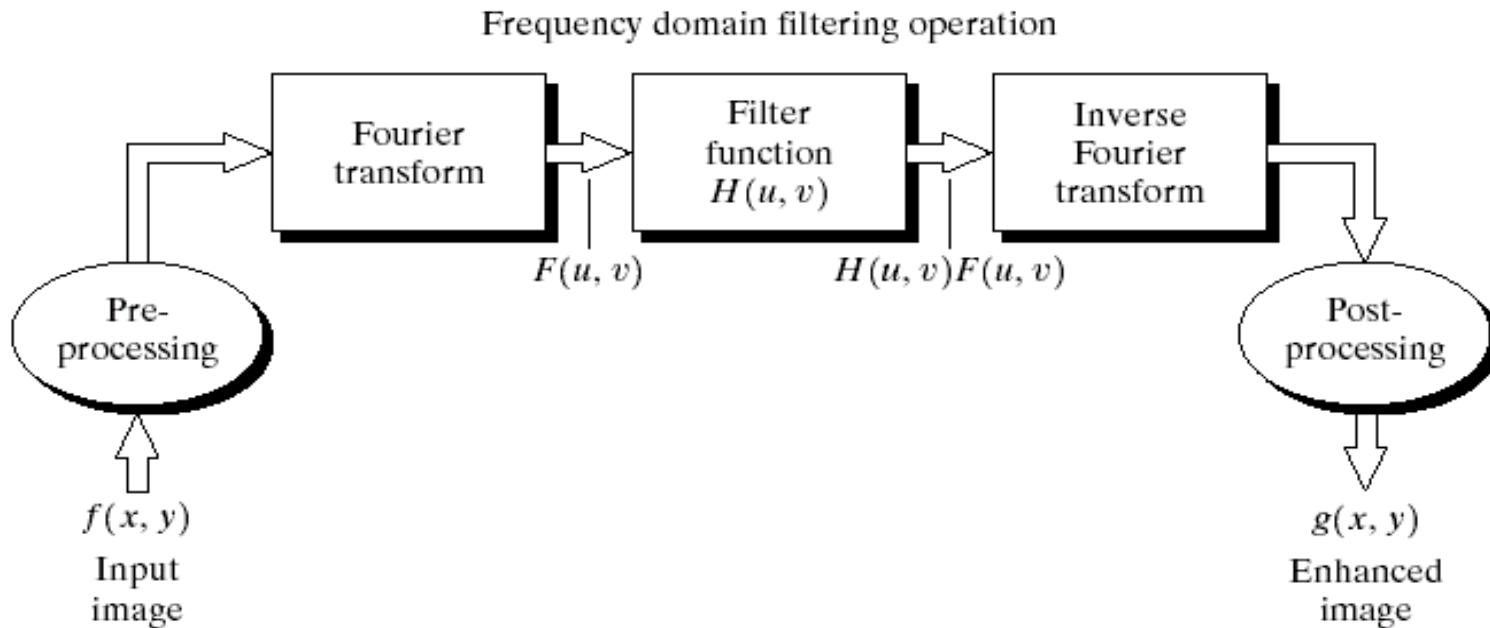
$$\Im[f(x, y)(-1)^{x+y}] = F(u - M/2, v - N/2).$$

2. Compute  $F(u, v)$ , the DFT of the image from (1).

3. Multiply  $F(u, v)$  by a filter function  $H(u, v)$ .

# Basics of Filtering in Frequency Domain

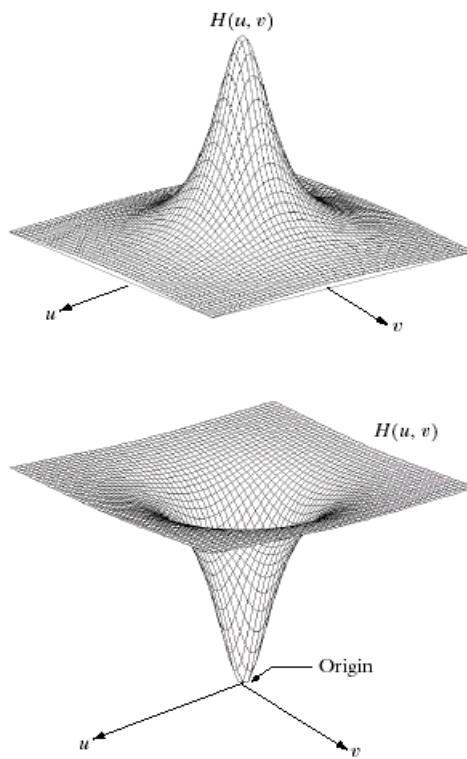
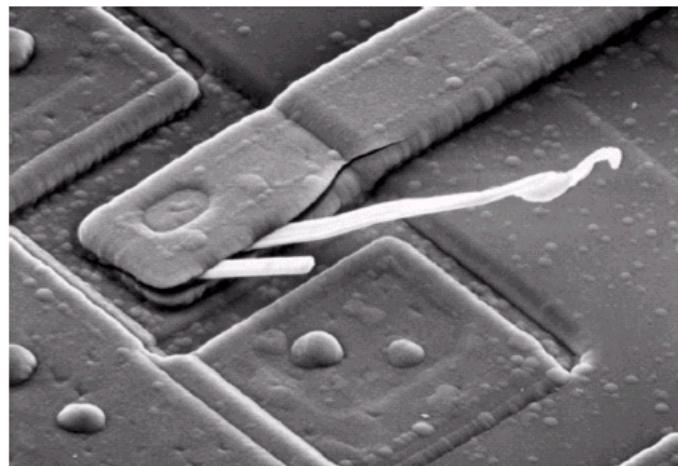
4. Compute the inverse DFT of the result in (3).
5. Obtain the real part of the result in (4).
6. Multiply the result in (5) by  $(-1)^{x+y}$ .



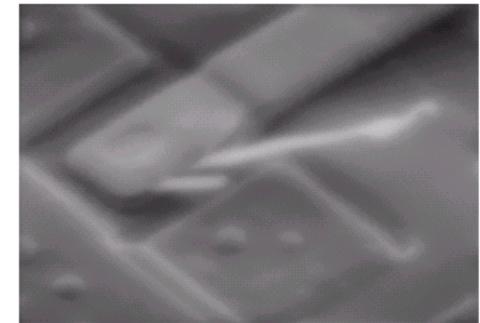
# Basics of Filtering in Frequency Domain

- Each component of  $H$  multiplies both the real and imaginary parts of the corresponding components in  $F$ .
- This type of filters are called *zero-phase-shift* filters.

# Some Basic Frequency Domain Filters

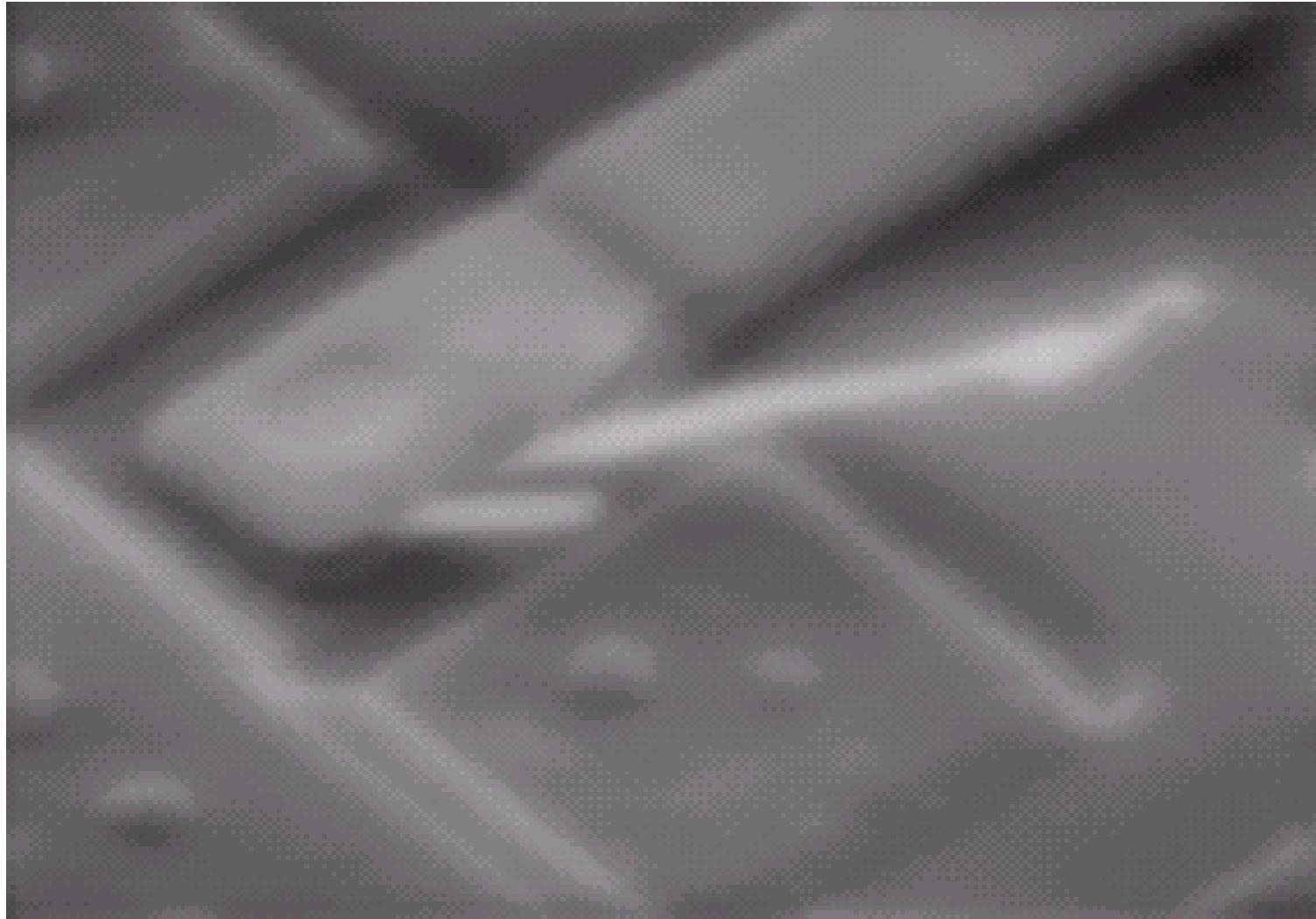


Low Pass Filter

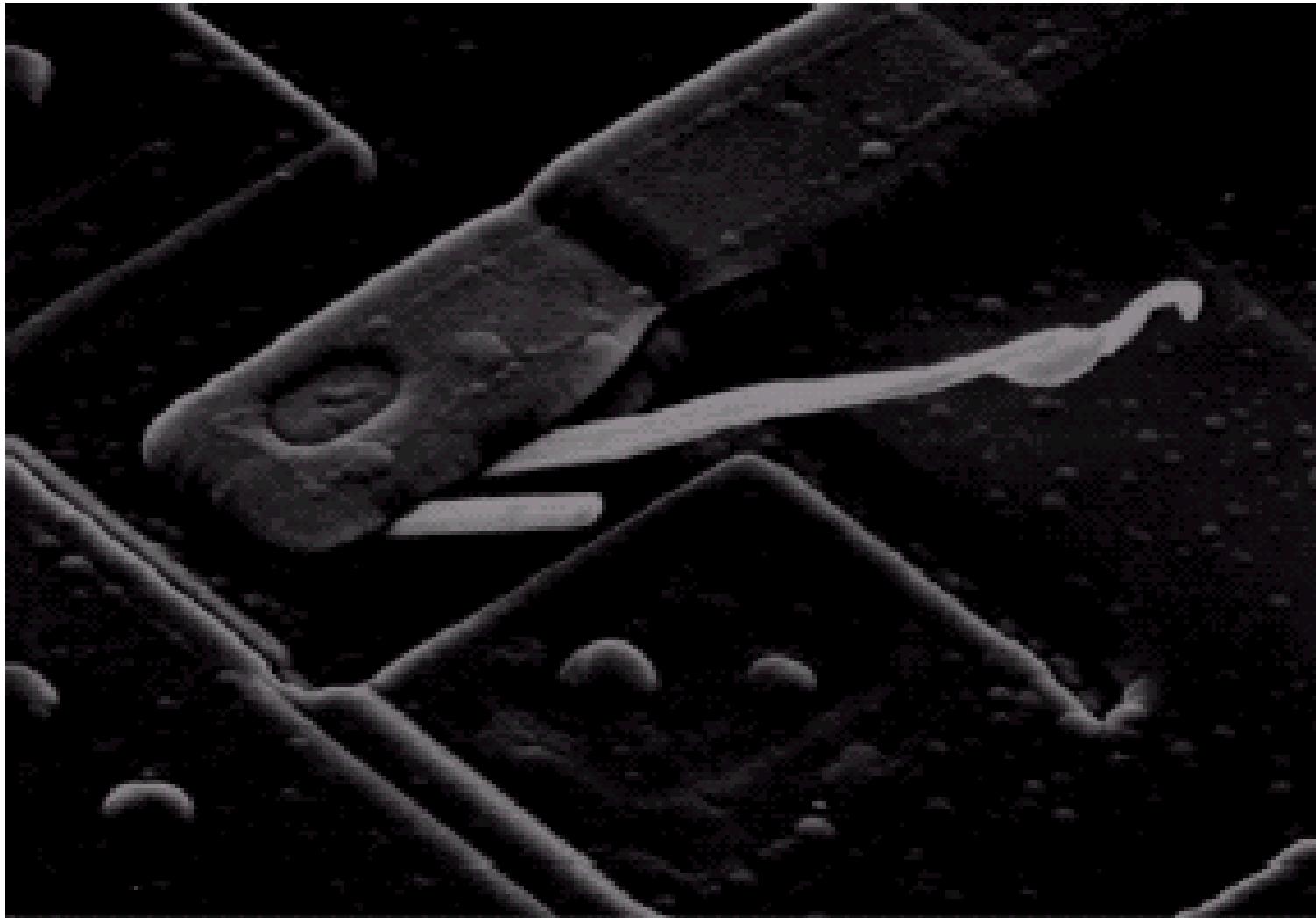


High Pass Filter

# Some Basic Frequency Domain Filters



# Some Basic Frequency Domain Filters



# Filters in Frequency Domain

## ✓ Image Smoothing Filters

- Ideal Low Pass Filter
- Butterworth Low Pass Filter
- Gaussian Low Pass Filter

## ✓ Image Sharpening Filters

- Ideal High Pass Filter
- Butterworth High Pass Filter
- Gaussian High Pass Filter

# Smoothing Frequency Domain Filters

Smoothing is achieved in the frequency domain by dropping out the high frequency components.

The basic model for filtering is:

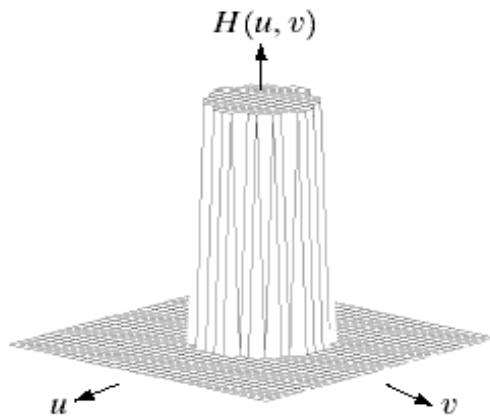
$$G(u,v) = H(u,v)F(u,v)$$

where  $F(u,v)$  is the Fourier transform of the image being filtered and  $H(u,v)$  is the filter transform function.

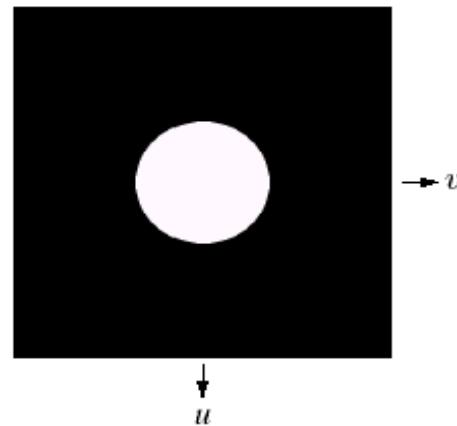
*Low pass filters – only pass the low frequencies, drop the high ones.*

# Ideal Low Pass Filter

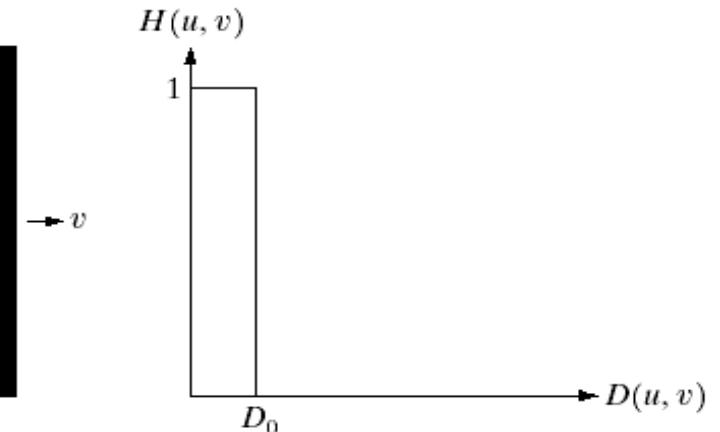
Simply cut off all high frequency components that are a specified distance  $D_0$  from the origin of the transform.



Perspective plot  
of the transfer function



Displayed as an  
image



Radial cross section

Changing the distance changes the behaviour of the filter.

# Ideal Low Pass Filter

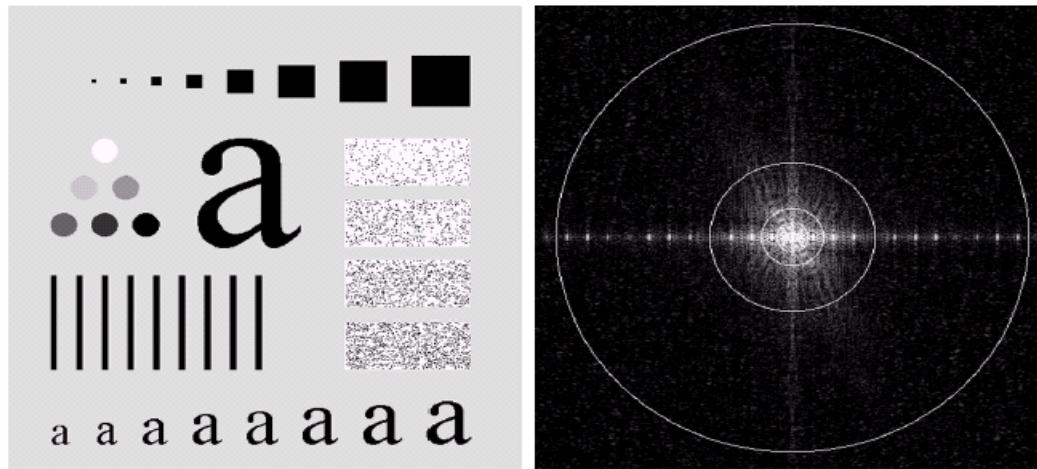
The transfer function for the ideal low pass filter can be given as:

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

where  $D(u, v)$  is given as:

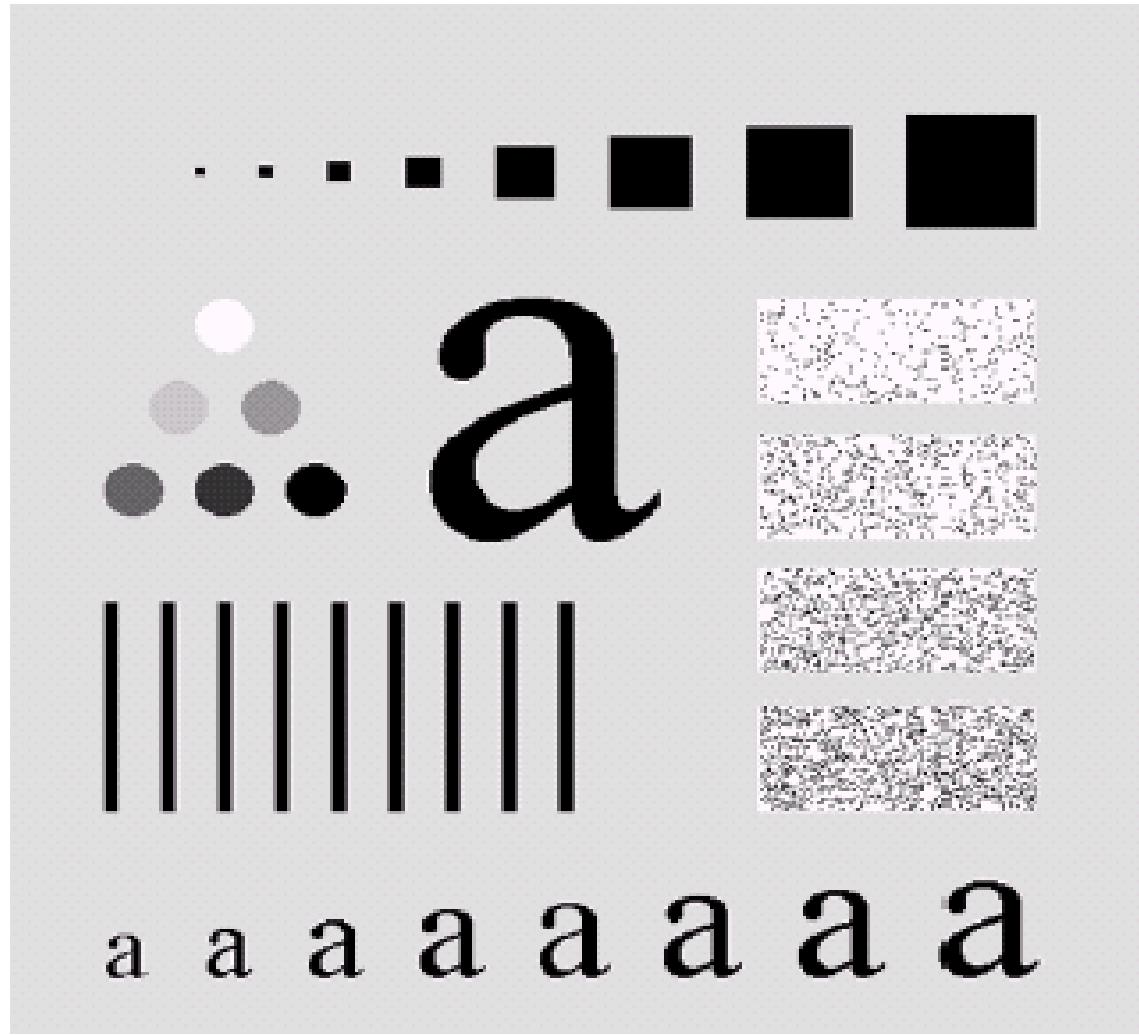
$$D(u, v) = [(u - M/2)^2 + (v - N/2)^2]^{1/2}$$

# Ideal Low Pass Filter

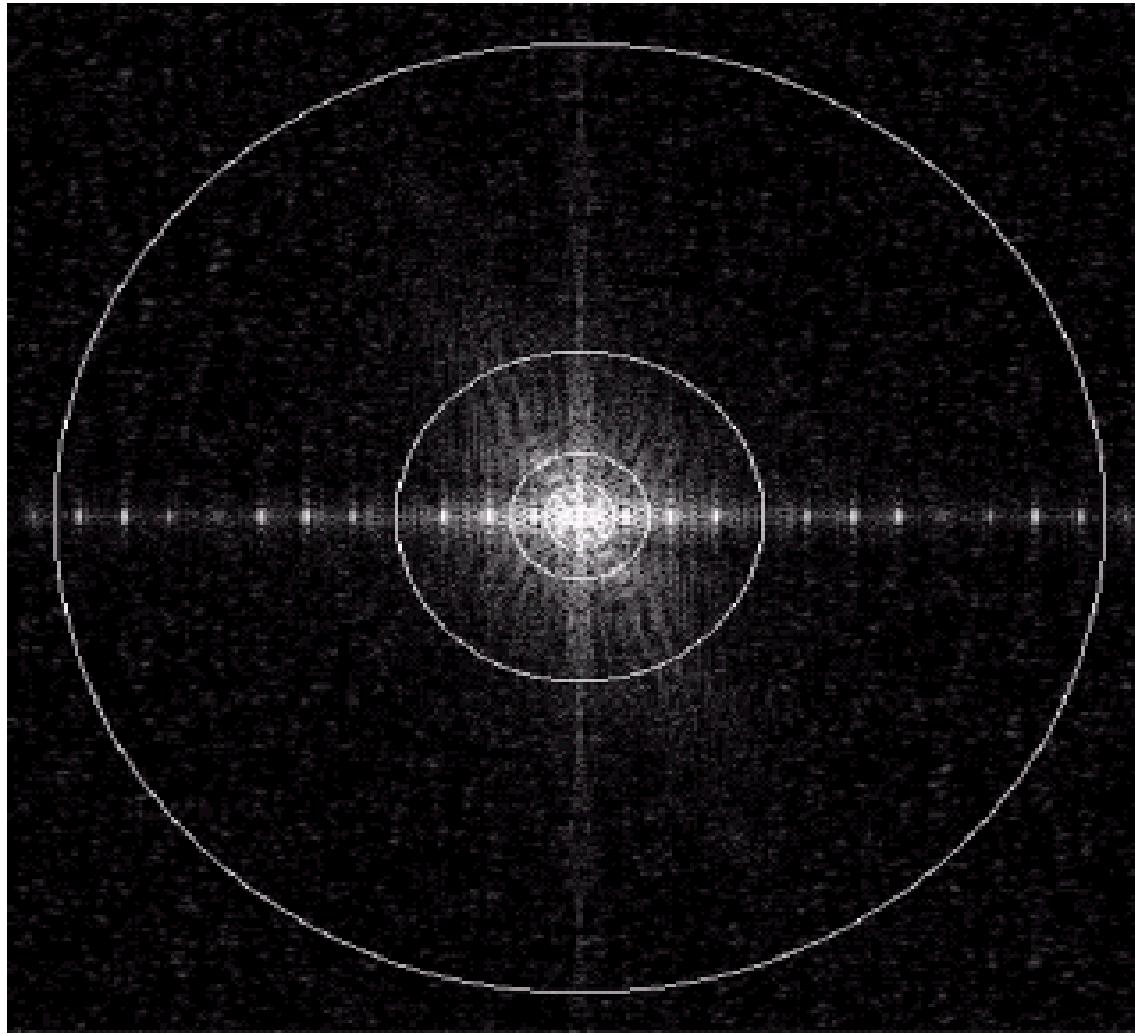


An image, its Fourier spectrum and a series of ideal low pass filters of radius 5, 15, 30, 80, and 230 superimposed on top of it.

# Ideal Low Pass Filter

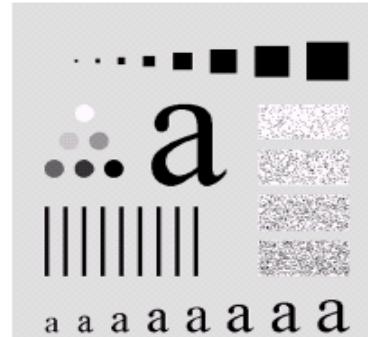


# Ideal Low Pass Filter

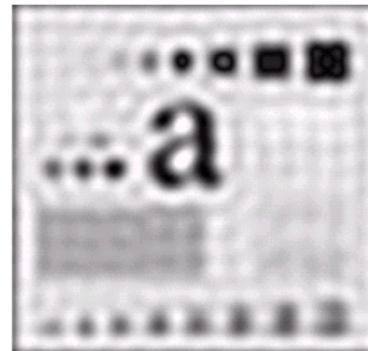


# Ideal Low Pass Filter

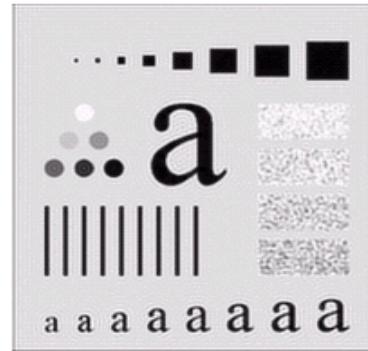
Original image



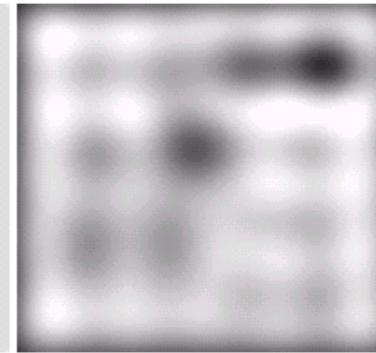
Result of filtering  
with ideal low pass  
filter of radius 15



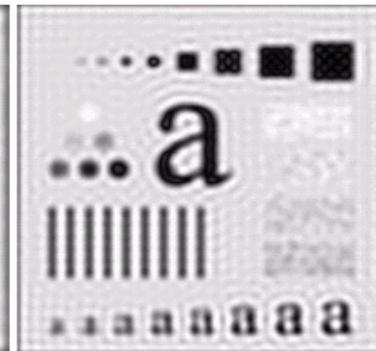
Result of filtering  
with ideal low pass  
filter of radius 80



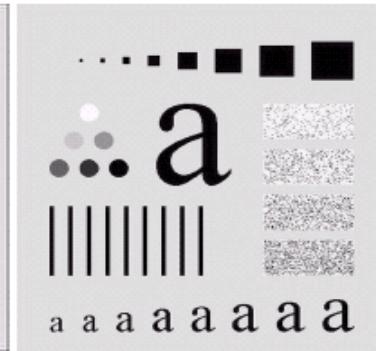
Result of filtering  
with ideal low pass  
filter of radius 5



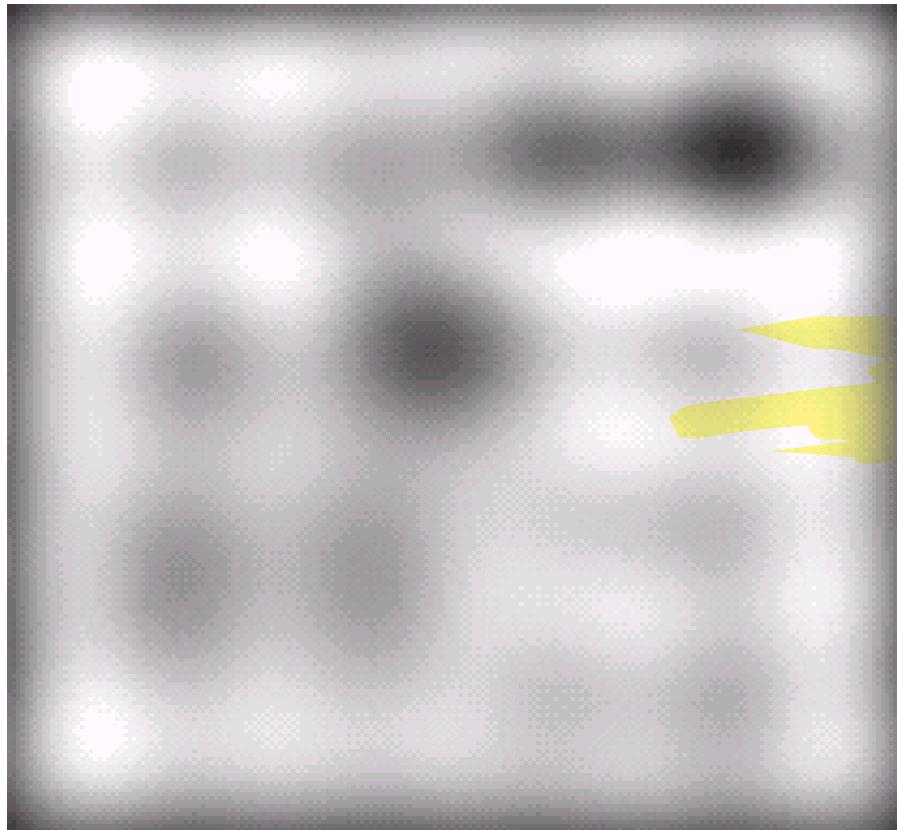
Result of filtering  
with ideal low pass  
filter of radius 30



Result of filtering  
with ideal low pass  
filter of radius 230



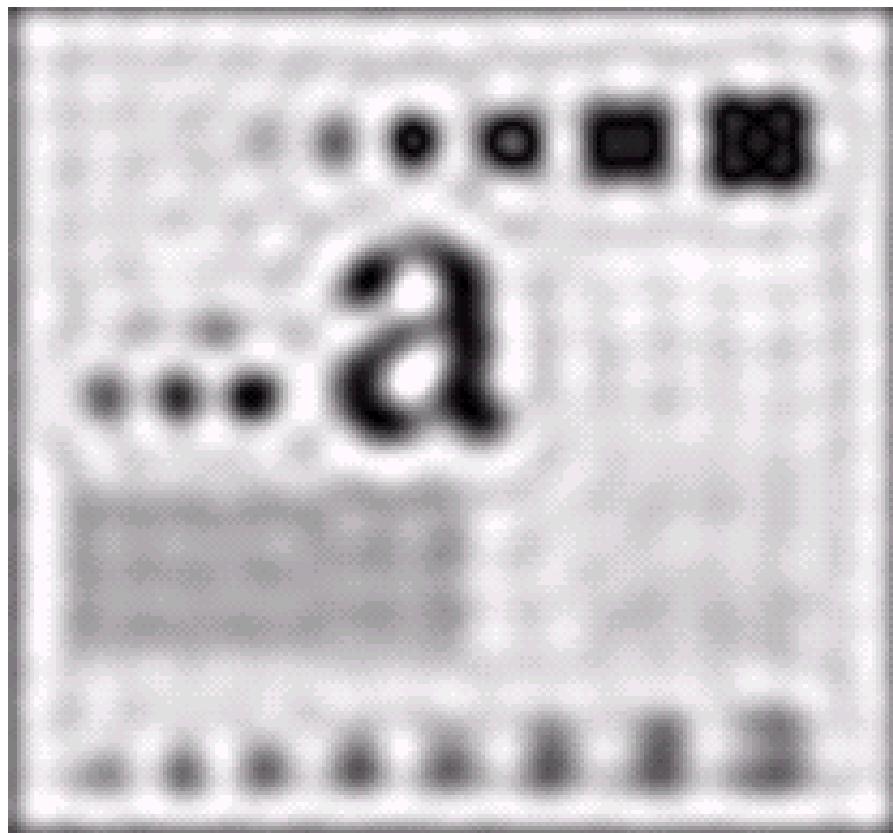
# Ideal Low Pass Filter



Result of filtering  
with ideal low pass  
filter of radius 5

Suffered by Ringing  
artifact.

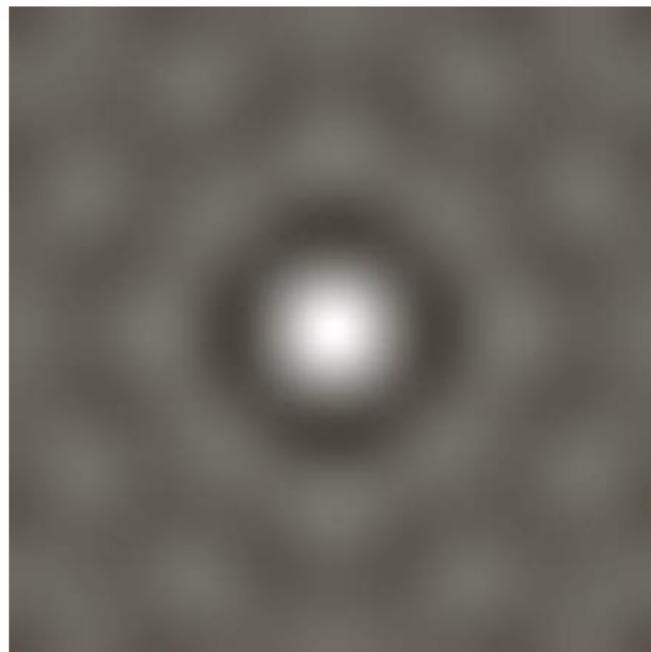
# Ideal Low Pass Filter



Result of filtering  
with ideal low pass  
filter of radius 15

Suffered by Ringing  
artifact.

# Ideal Low Pass Filter



a b

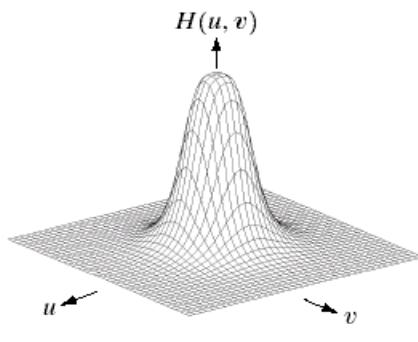
**FIGURE 4.43**

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

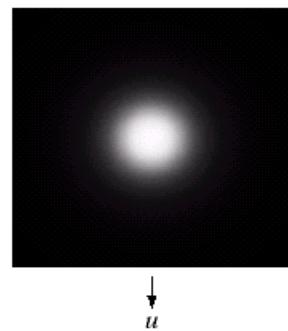
# Butterworth Low Pass Filter

The transfer function of a Butterworth lowpass filter of order  $n$  with cutoff frequency at distance  $D_0$  from the origin is defined as:

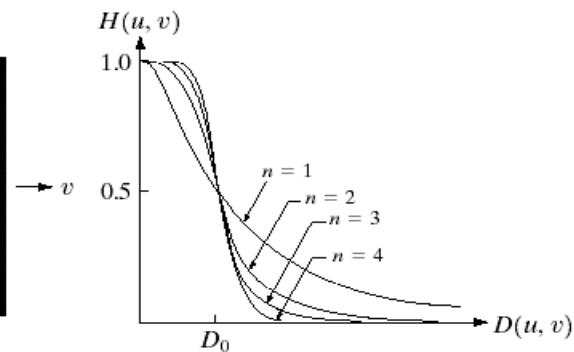
$$H(u, v) = \frac{1}{1 + [D(u, v)/D_0]^{2n}}$$



Perspective plot  
of the transfer function



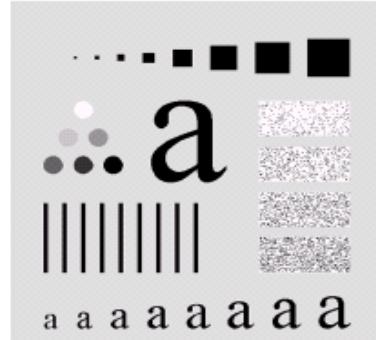
Displayed as an  
image



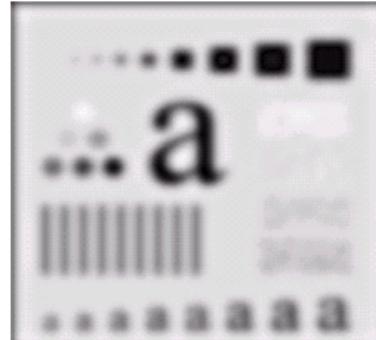
Radial cross section

# Butterworth Low Pass Filter

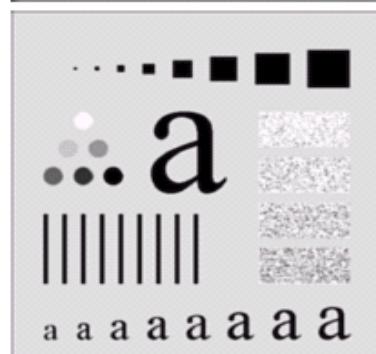
Original image



Result of filtering with Butterworth filter of order 2 and cutoff radius 15



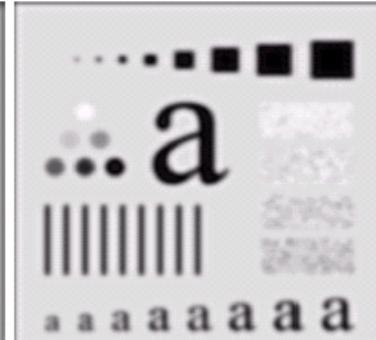
Result of filtering with Butterworth filter of order 2 and cutoff radius 80



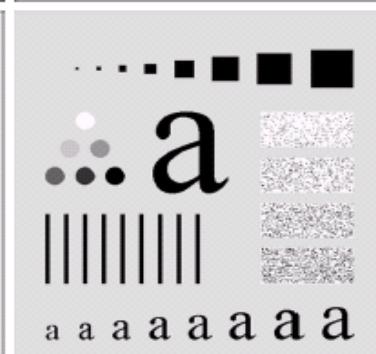
Result of filtering with Butterworth filter of order 2 and cutoff radius 5



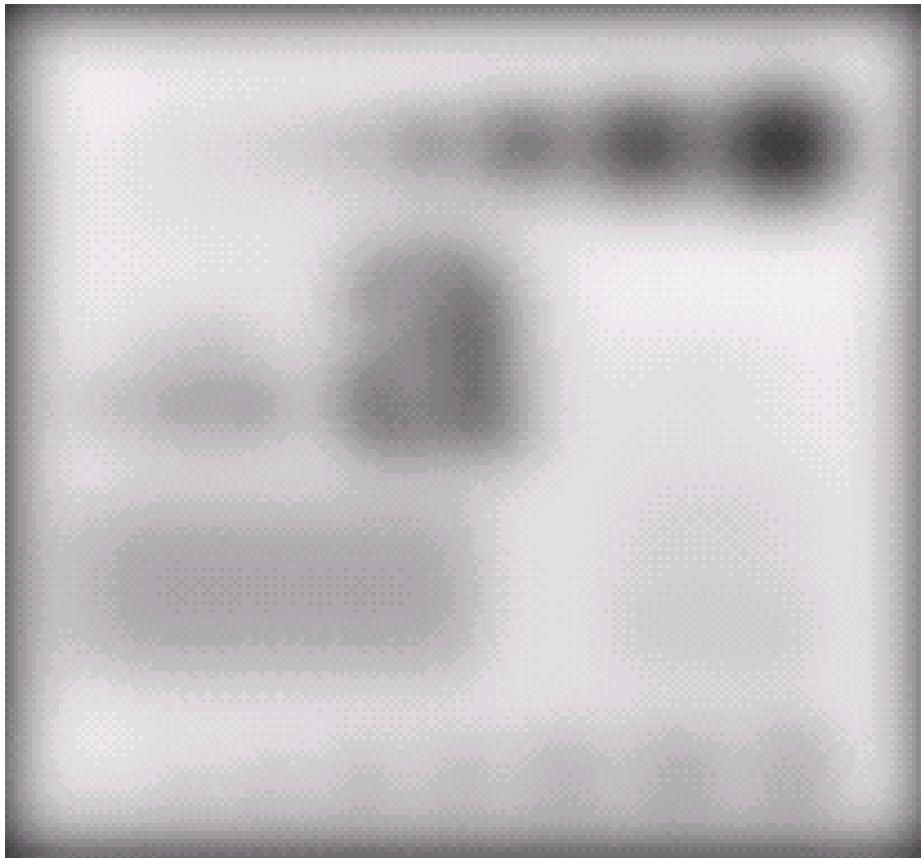
Result of filtering with Butterworth filter of order 2 and cutoff radius 30



Result of filtering with Butterworth filter of order 2 and cutoff radius 230



# Butterworth Low Pass Filter



Result of filtering  
with Butterworth  
filter of order 2 and  
cutoff radius 5

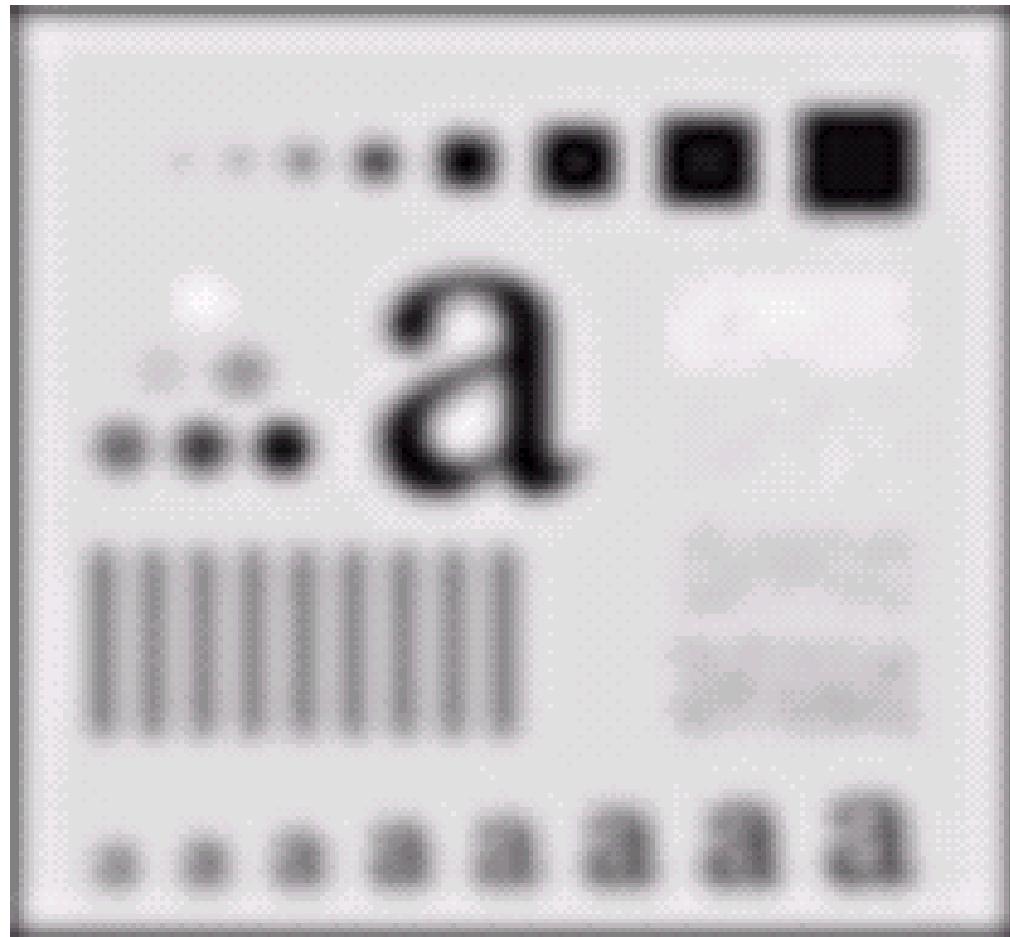
No Ringing artifact.



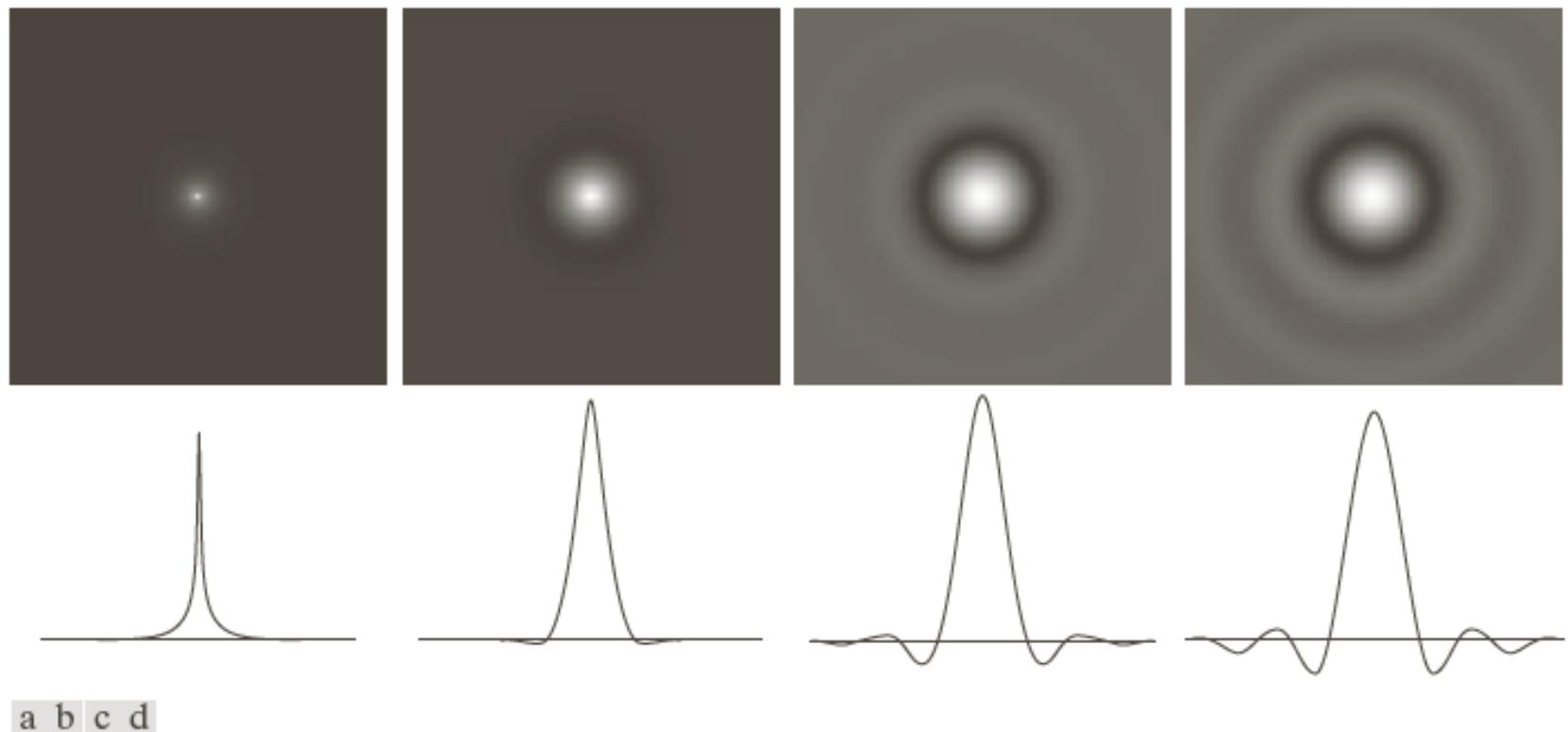
# Butterworth Low Pass Filter

Result of filtering with  
Butterworth filter of  
order 2 and cutoff  
radius 15

No Ringing artifact.



# Butterworth Low Pass Filter

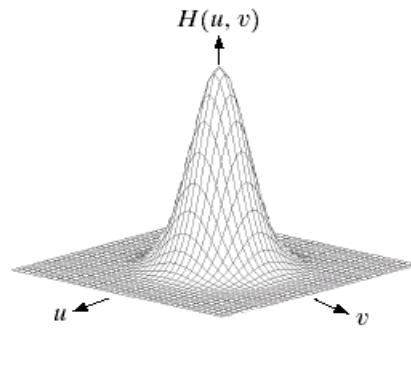


**FIGURE 4.46** (a)–(d) Spatial representation of BLPFs of order 1, 2, 5, and 20, and corresponding intensity profiles through the center of the filters (the size in all cases is  $1000 \times 1000$  and the cutoff frequency is 5). Observe how ringing increases as a function of filter order.

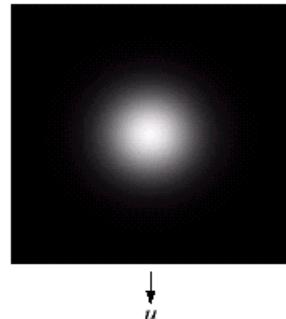
# Gaussian Low Pass Filter

The transfer function of a Gaussian lowpass filter is defined as:

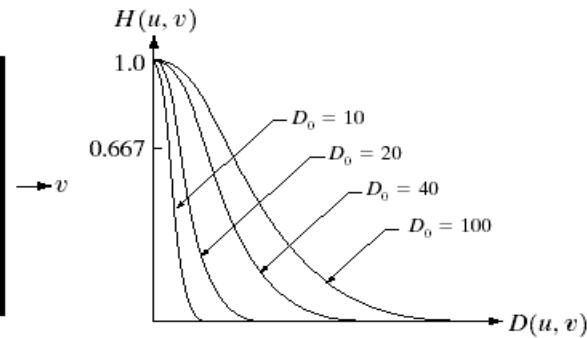
$$H(u, v) = e^{-D^2(u,v)/2D_0^2}$$



Perspective plot  
of the transfer function



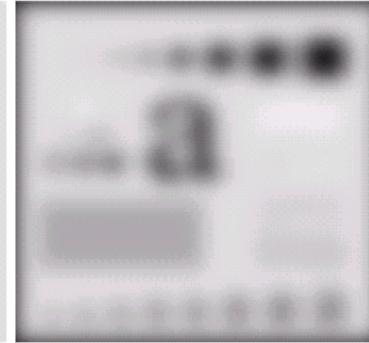
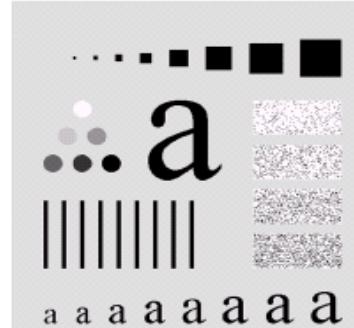
Displayed as an  
image



Radial cross section

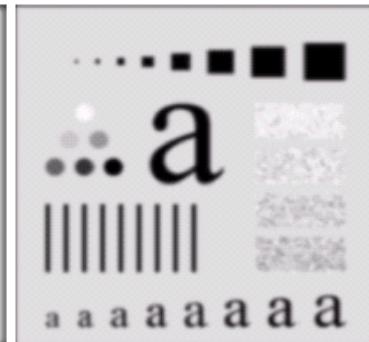
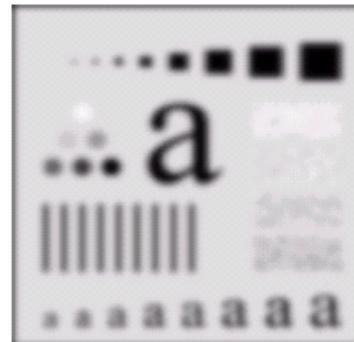
# Gaussian Low Pass Filter

Original image



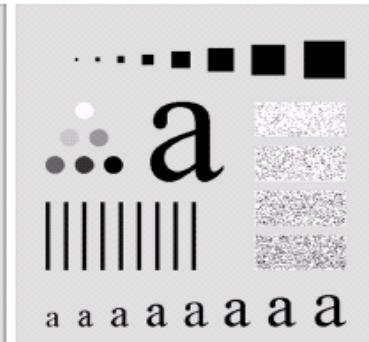
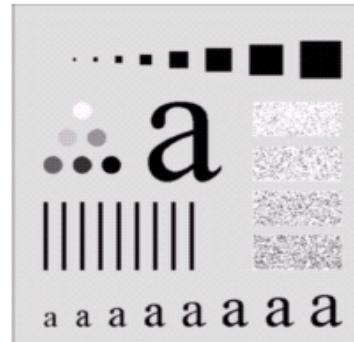
Result of filtering with Gaussian filter with cutoff radius 5

Result of filtering with Gaussian filter with cutoff radius 15



Result of filtering with Gaussian filter with cutoff radius 30

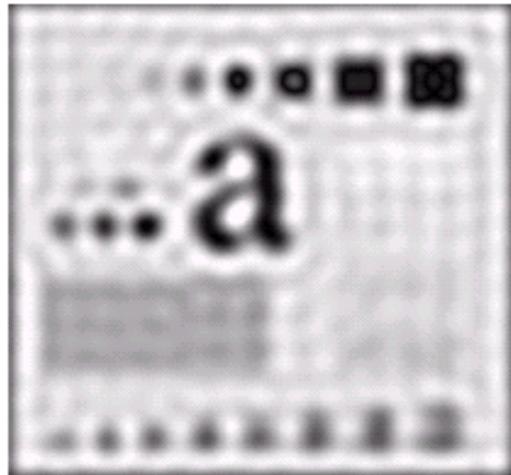
Result of filtering with Gaussian filter with cutoff radius 85



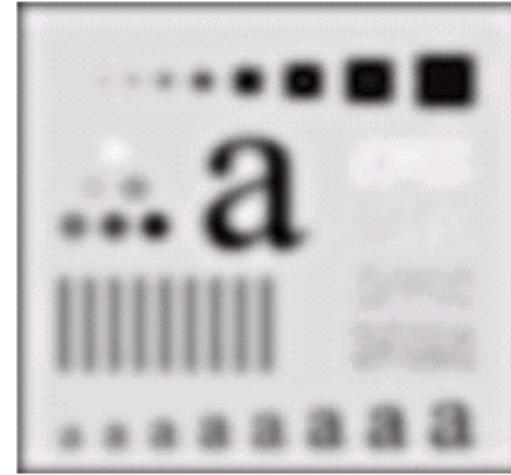
Result of filtering with Gaussian filter with cutoff radius 230

# Low Pass Filters Comparison

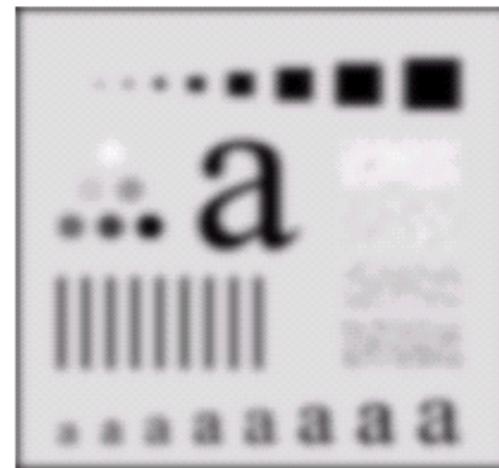
Result of filtering  
with ideal low pass  
filter of radius 15



Result of filtering  
with Butterworth  
filter of order 2  
and cutoff radius  
15



Result of filtering  
with Gaussian  
filter with cutoff  
radius 15



Gaussian filter did not  
achieve as much  
smoothing as the  
Butterworth filter.

This is happened  
because the profile of  
Gaussian is not as "tight"  
as the profile of the  
Butterworth filter.

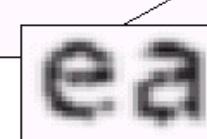
# Low Pass Filtering Example

A low pass Gaussian filter is used to connect broken text.

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.



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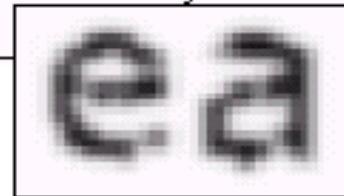


# Low Pass Filtering Example

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# Low Pass Filtering Example

Different lowpass Gaussian filters used to remove blemishes (small mark) in a photograph.



# Low Pass Filtering Example



Original image of  
size  $1028 \times 732$



Filtered with a GLPF  
with  $D_0 = 100$



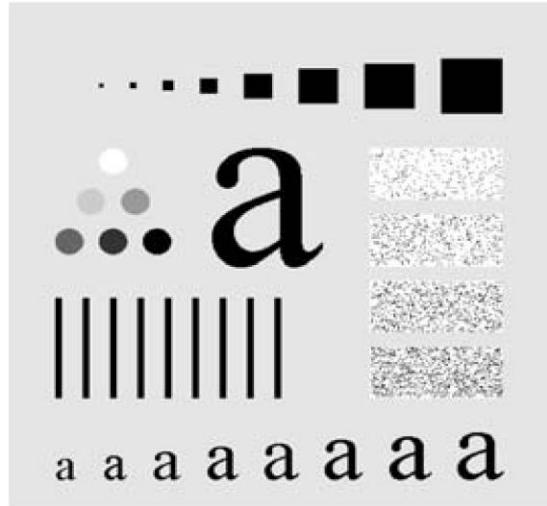
Filtered with a GLPF  
with  $D_0 = 80$

Reduction in skin  
lines in the magnified  
sections of the filtered  
images.

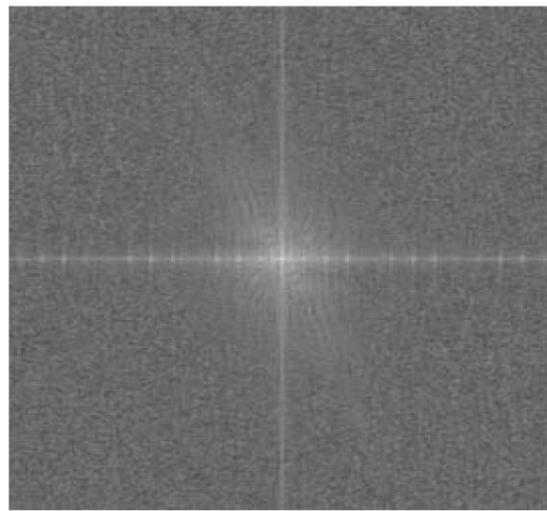


# Low Pass Filtering Example

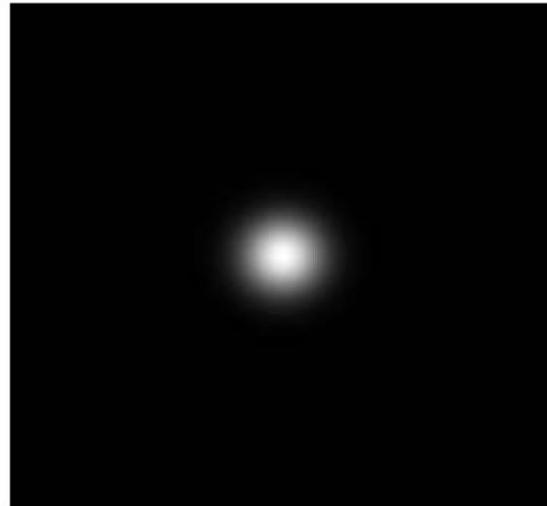
Original  
image



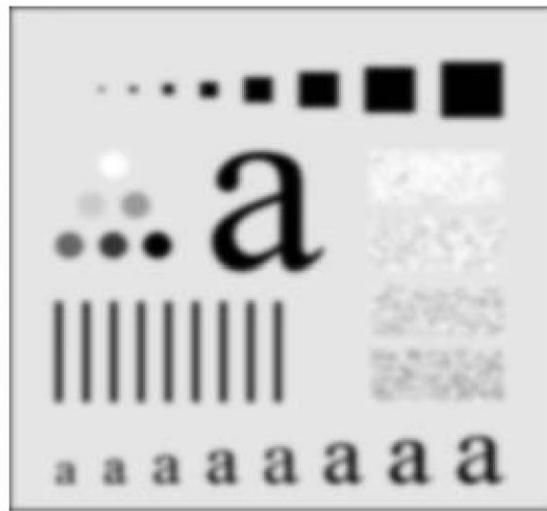
Spectrum of  
original image



Gaussian lowpass  
filter



Processed  
image



# Sharpening in the Frequency Domain

Edges and fine detail in images are associated with high frequency components.

*High pass filters – only pass the high frequencies, drop the low ones.*

High pass frequencies are precisely the reverse of low pass filters as:

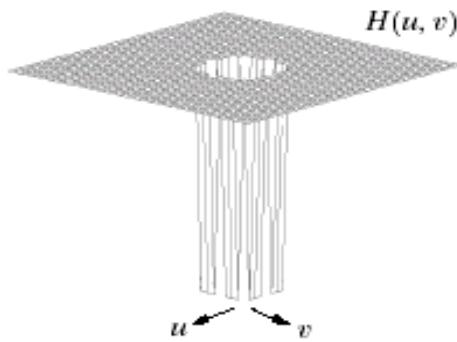
$$H_{hp}(u, v) = 1 - H_{lp}(u, v)$$

# Ideal High Pass Filter

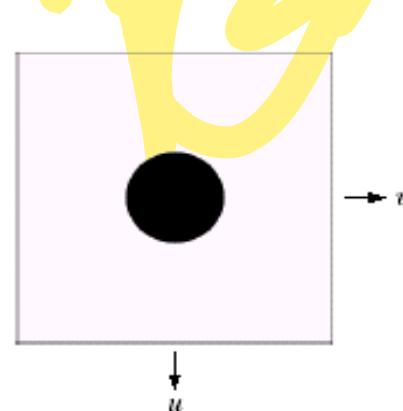
The ideal high pass filter is given as:

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

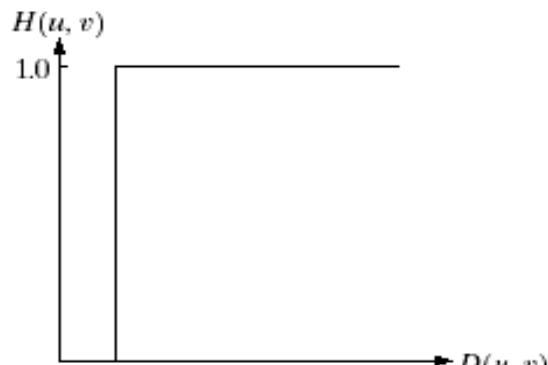
where  $D_0$  is the cut off distance as low pass filter.



Perspective plot  
of the transfer function



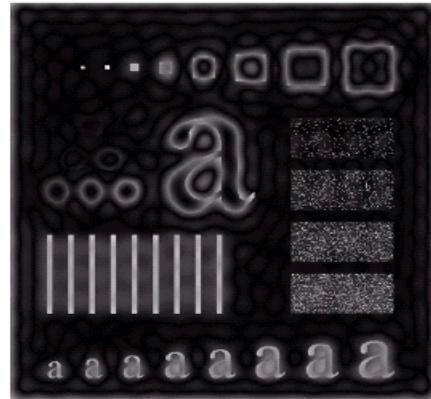
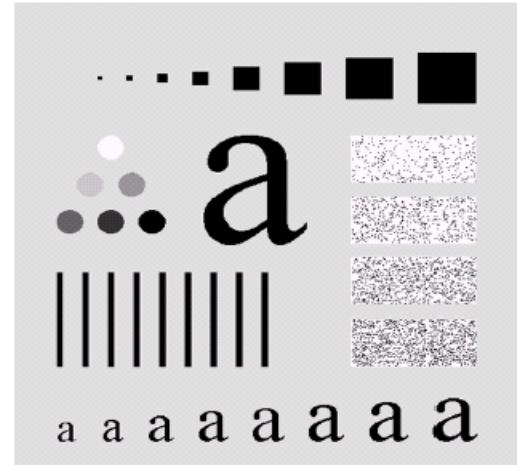
Displayed as an  
image



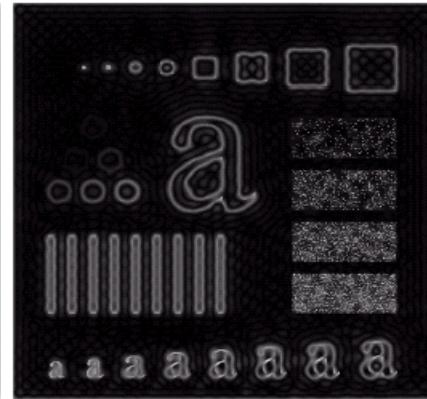
Radial cross section

# Ideal High Pass Filter

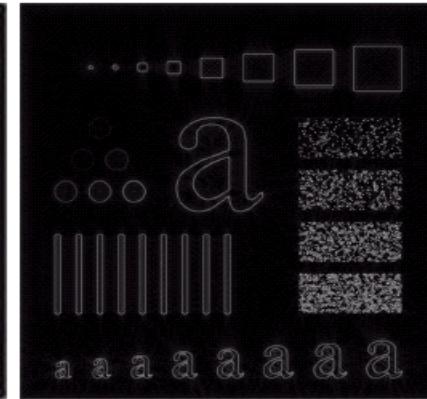
Distorted thickened object boundary.



Results of ideal  
high pass filtering  
with  $D_0 = 15$



Results of ideal  
high pass filtering  
with  $D_0 = 30$



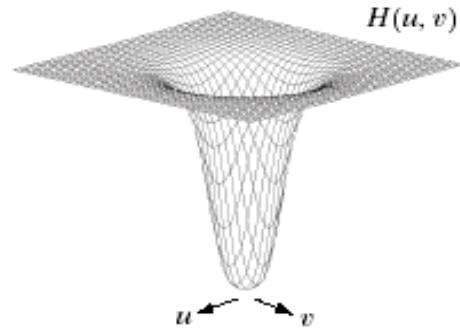
Results of ideal  
high pass filtering  
with  $D_0 = 80$

# Butterworth High Pass Filter

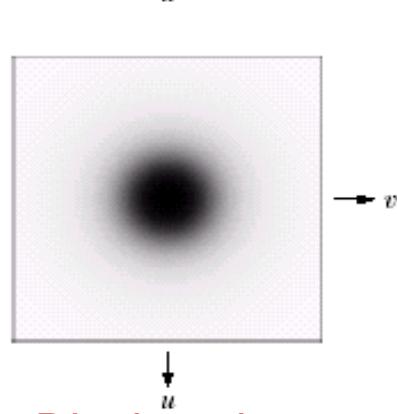
The Butterworth high pass filter is given as:

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

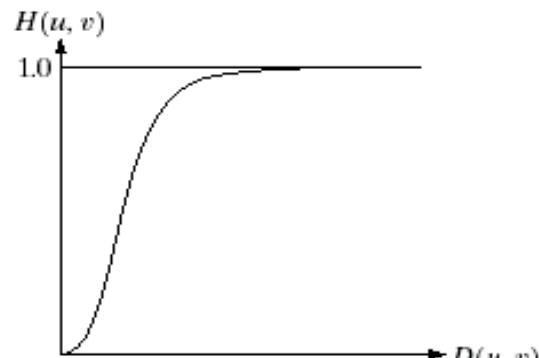
where  $n$  is the order and  $D_0$  is the cut off distance as low pass filter.



Perspective plot  
of the transfer function



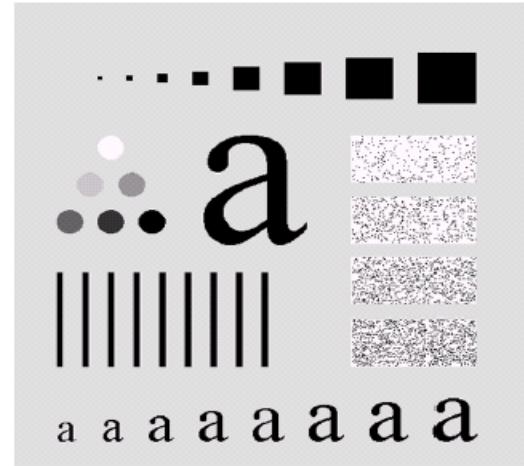
Displayed as an  
image



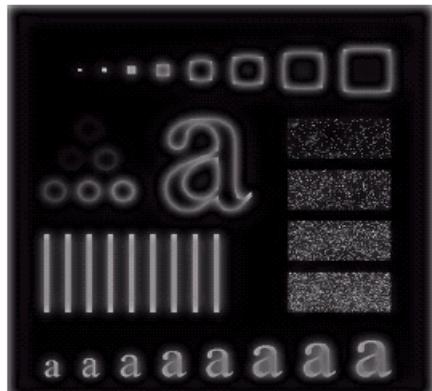
Radial cross section



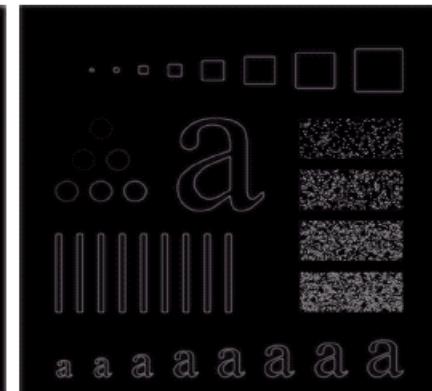
# Butterworth High Pass Filter



Results of  
Butterworth  
high pass  
filtering of  
order 2 with  
 $D_0 = 15$



Results of  
Butterworth  
high pass  
filtering of  
order 2 with  
 $D_0 = 80$



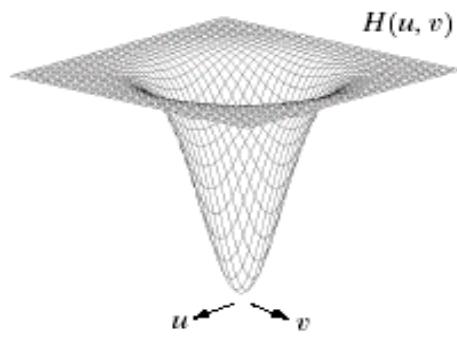
Results of  
Butterworth high  
pass filtering of  
order 2 with  $D_0 = 30$

# Gaussian High Pass Filter

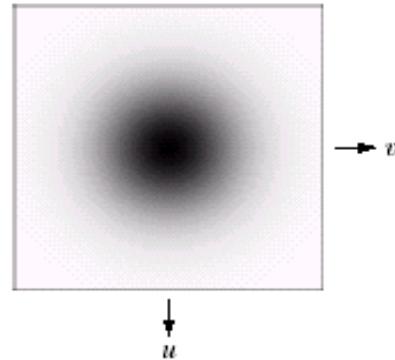
The Gaussian high pass filter is given as:

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

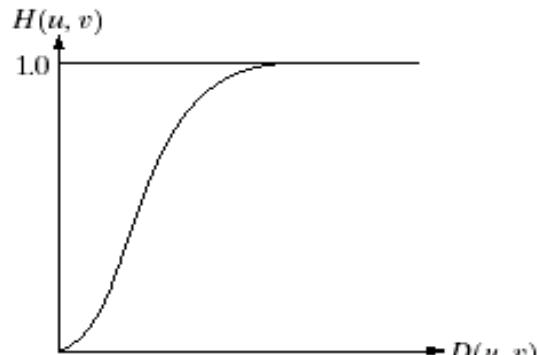
where  $D_0$  is the cut off distance as before.



Perspective plot  
of the transfer function

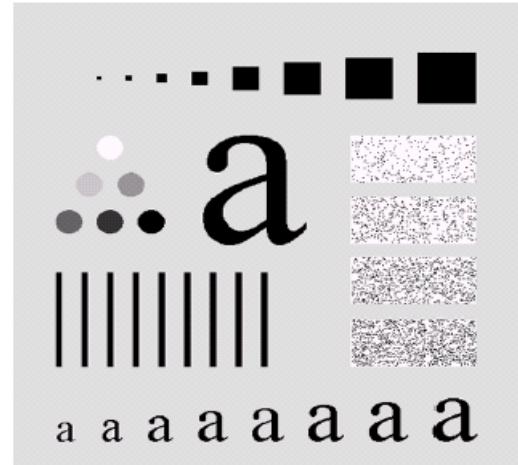


Displayed as an  
image

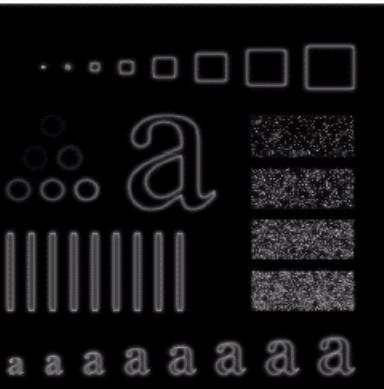
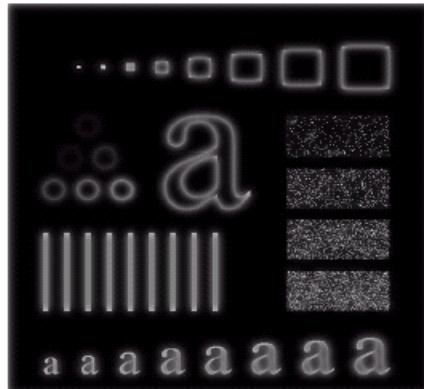


Radial cross section

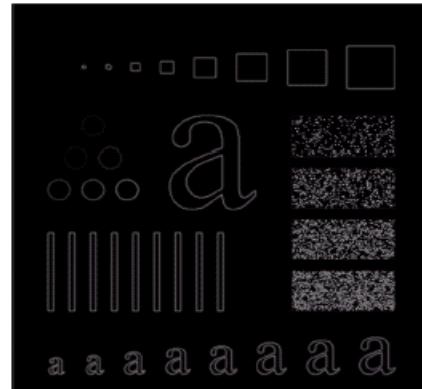
# Gaussian High Pass Filter



Results of  
Gaussian  
high pass  
filtering with  
 $D_0 = 15$



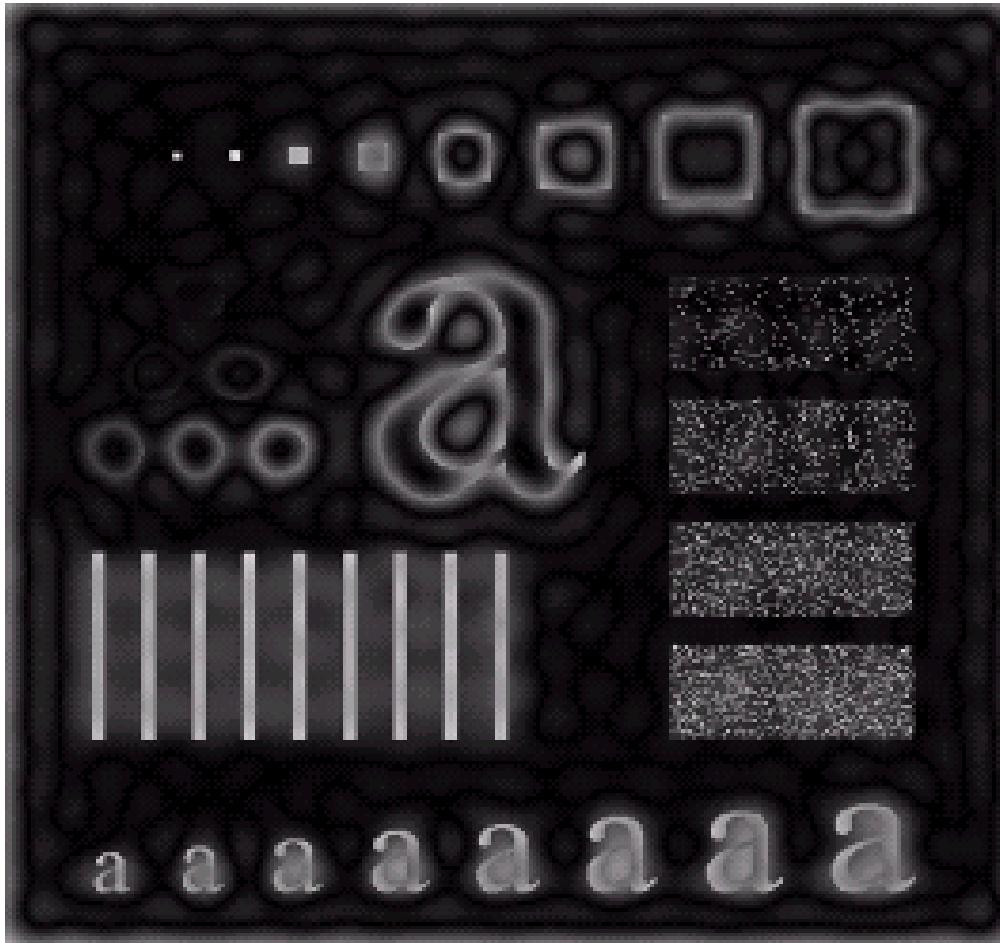
Results of  
Gaussian high  
pass filtering with  
 $D_0 = 30$



Results of  
Gaussian  
high pass  
filtering with  
 $D_0 = 80$



# Highpass Filter Comparison

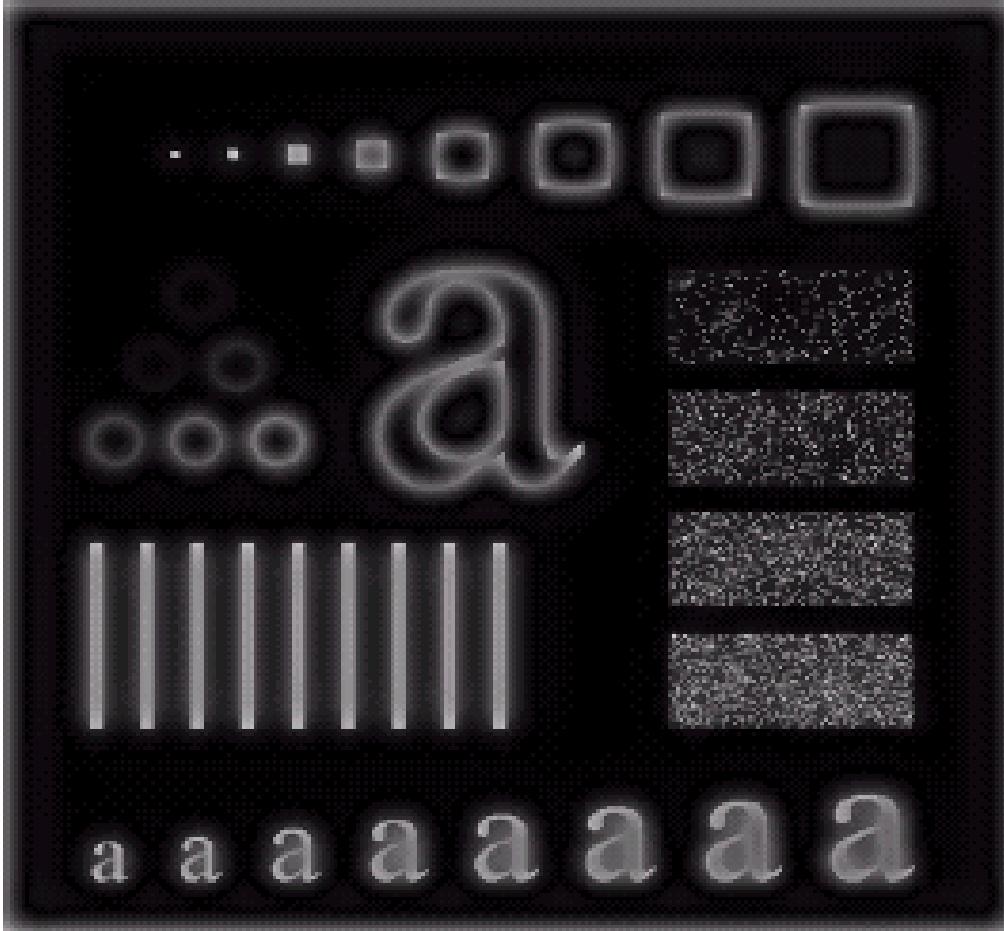


Results of ideal  
high pass filtering  
with  $D_0 = 15$

Distorted thickened  
object boundary.



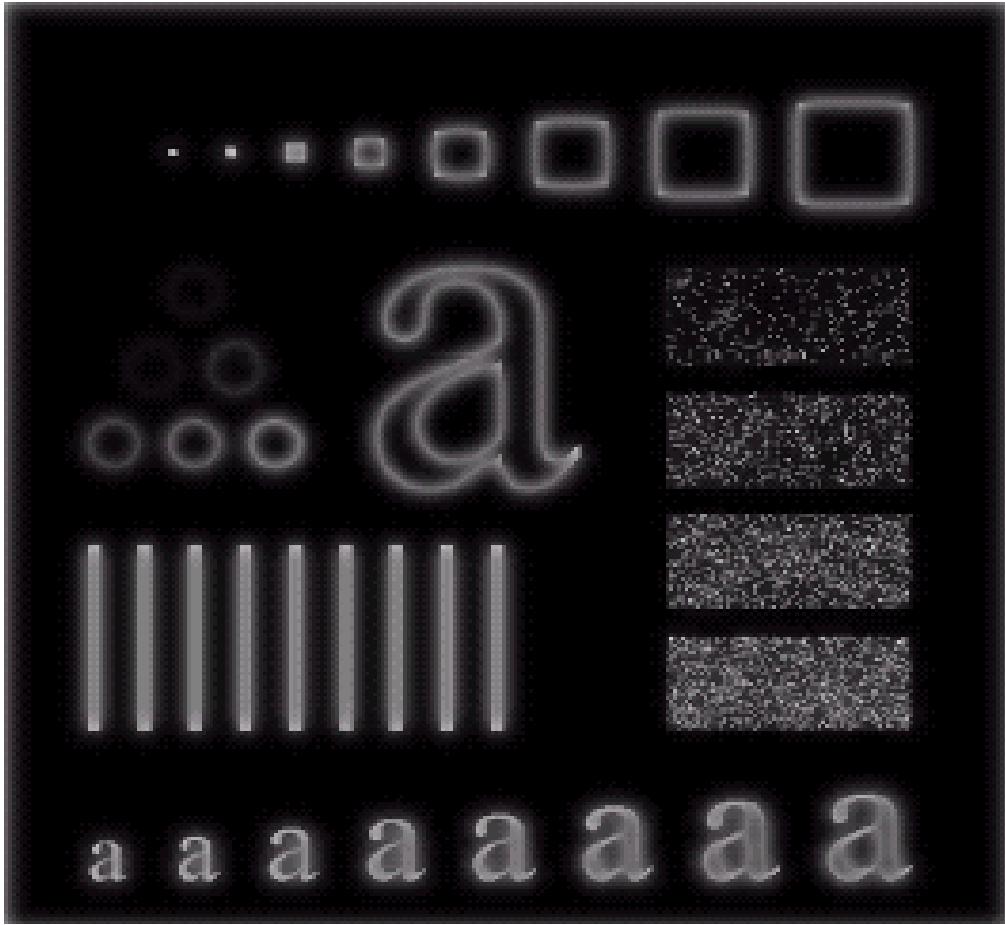
# Highpass Filter Comparison



Results of Butterworth  
high pass filtering of order  
2 with  $D_0 = 15$

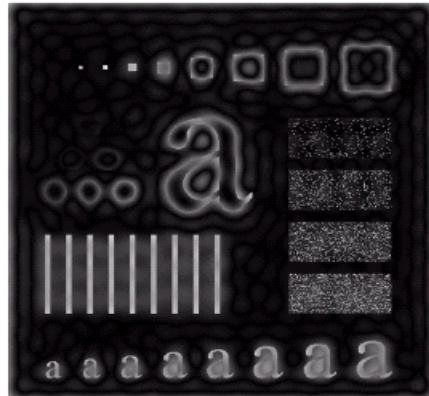


# Highpass Filter Comparison

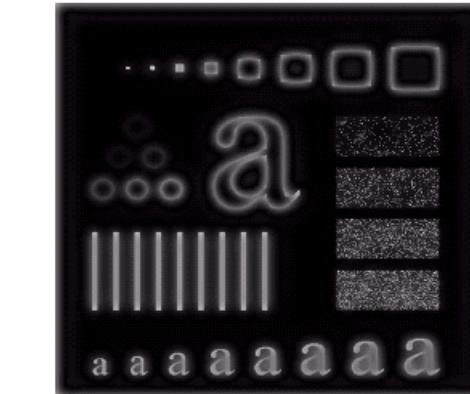
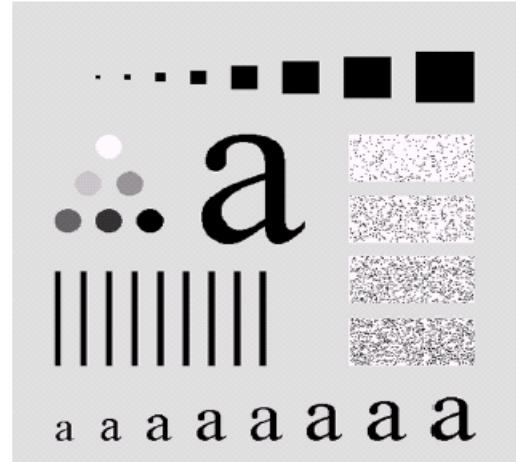


Results of Gaussian  
high pass filtering with  
 $D_0 = 15$

# Highpass Filter Comparison



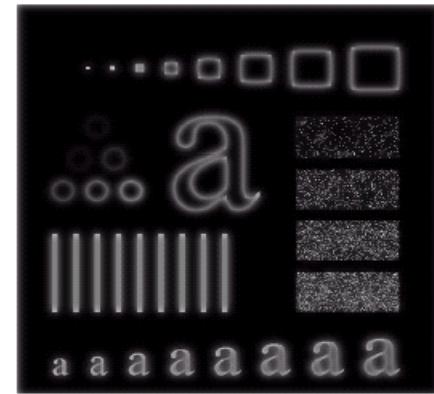
Results of ideal  
high pass filtering  
with  $D_0 = 15$



Results of Butterworth  
high pass filtering of order  
2 with  $D_0 = 15$

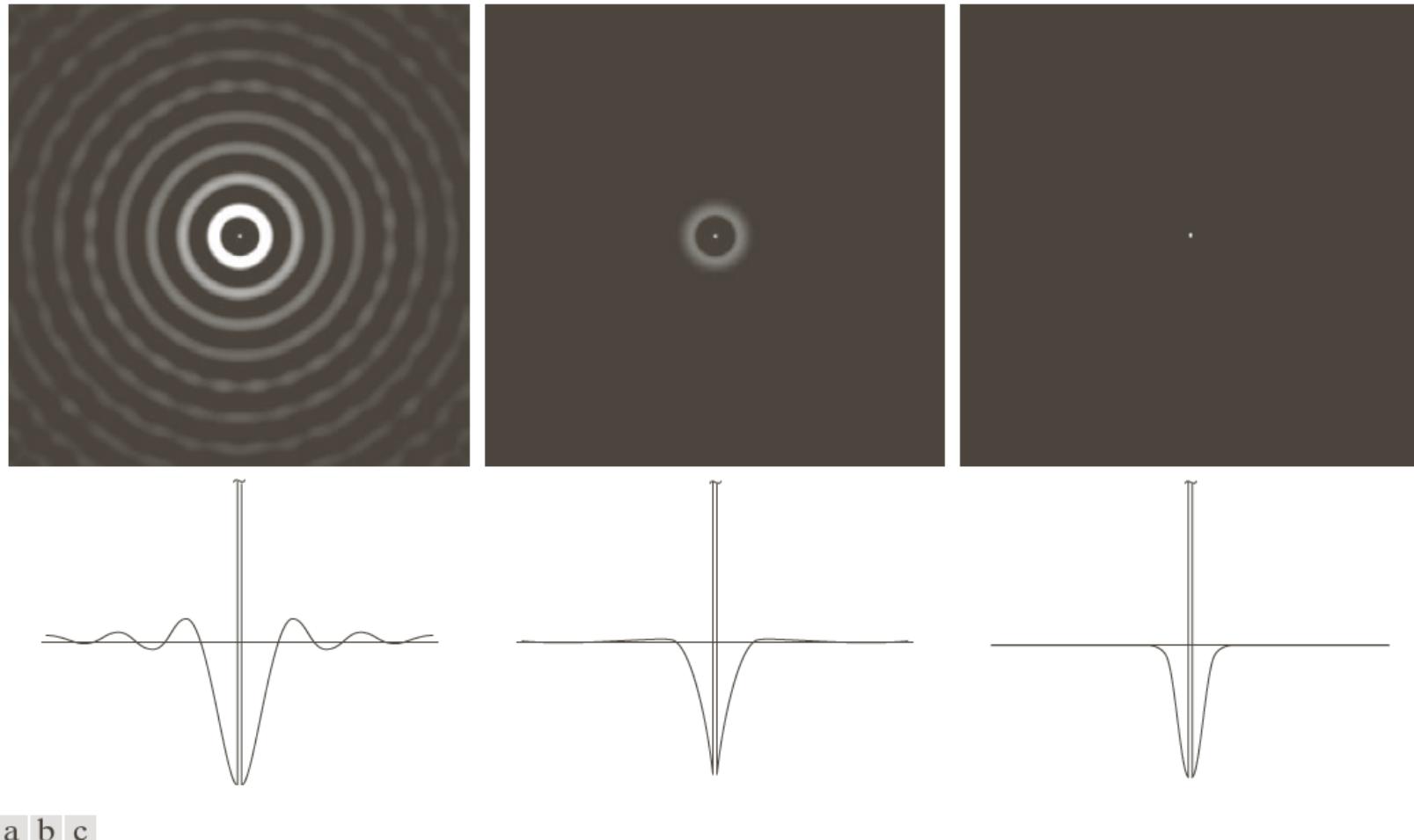
The results obtained by Gaussian filter are smoother than with the other two filters.

Even the filtering of the smaller objects and thin bars is clear with this filter.



Results of Gaussian  
high pass filtering with  
 $D_0 = 15$

# Highpass Filter Comparison



a b c

**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.

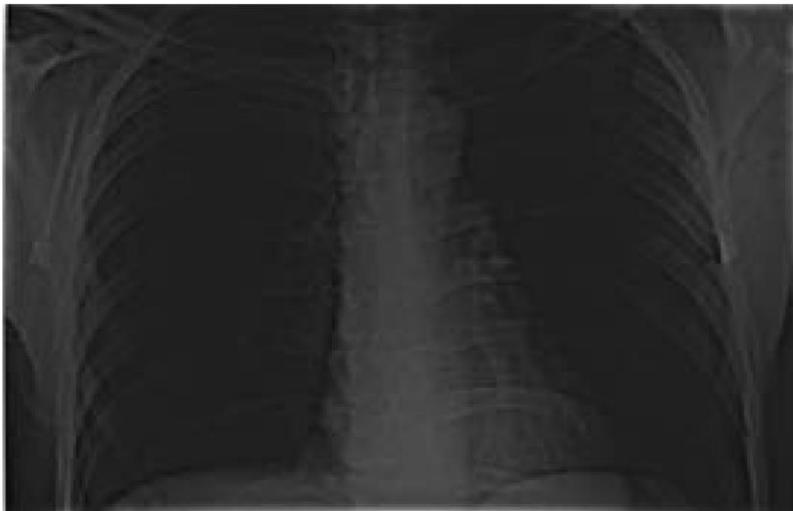
# Highpass Filtering Example

Images taken from Gonzalez & Woods, Digital Image Processing (2002)

Original image



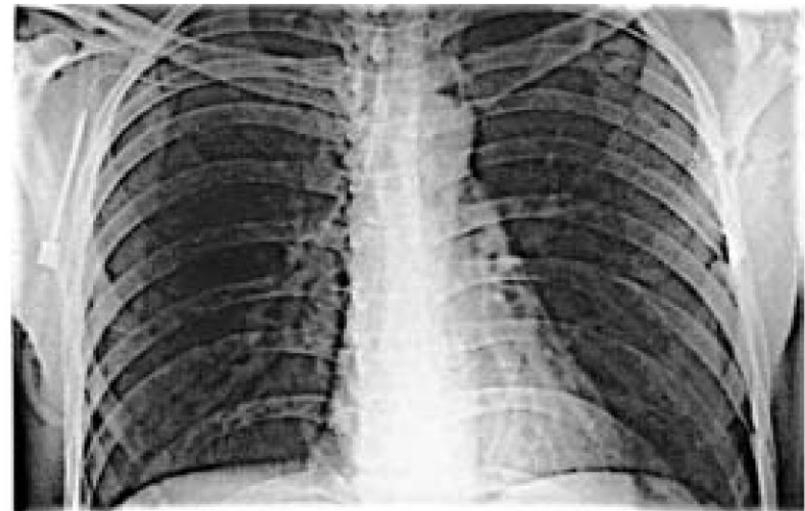
High frequency  
emphasis result



Highpass filtering result



After histogram  
equalisation



# Convolution

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

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

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

Convolution theorem defines the correspondence between filtering in the spatial domain and frequency domain.

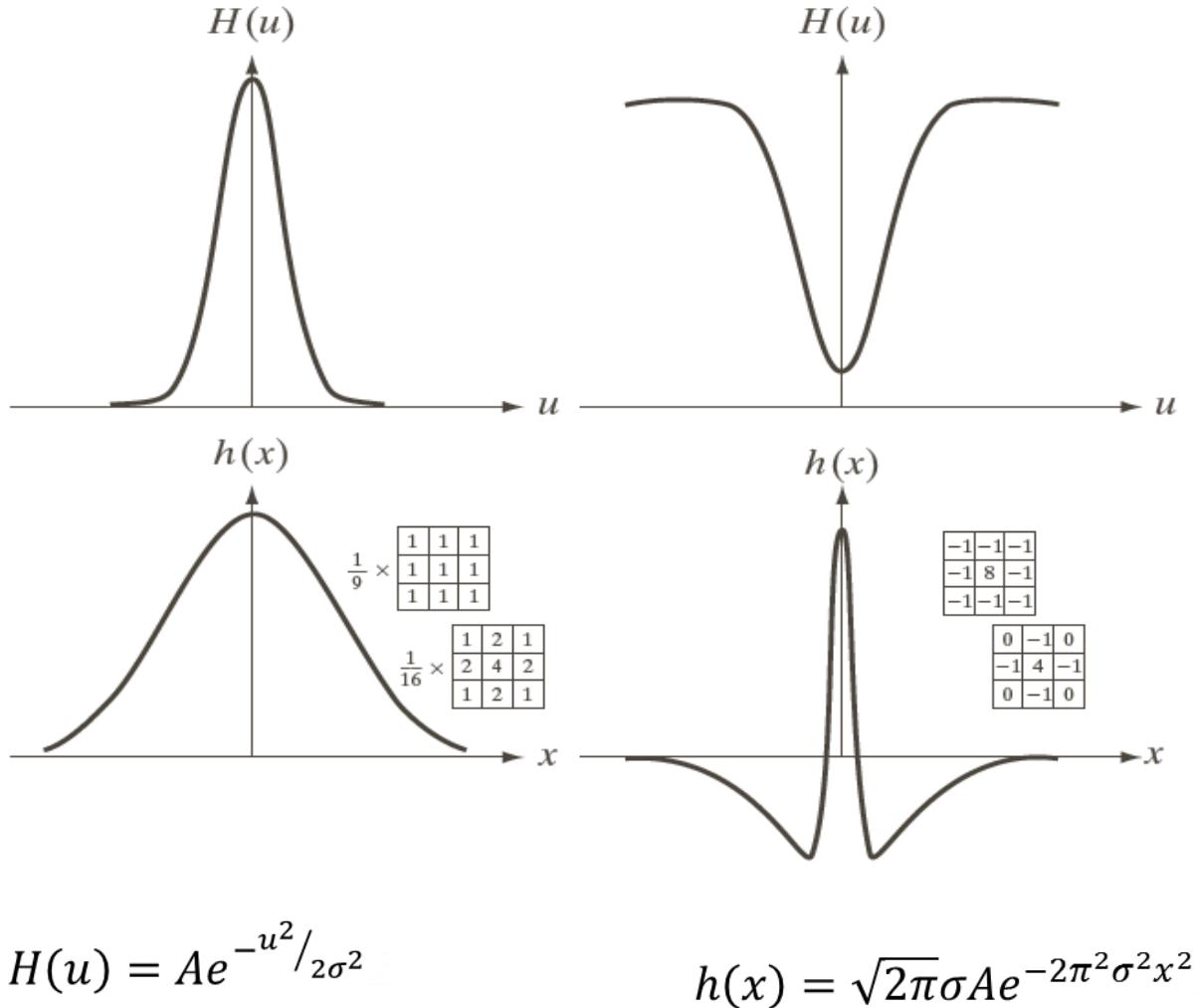
# Convolution

Convolution equation is nothing more than an implementation for

- (1) flipping one function about origin;
- (2) shifting that function with respect to the other by changing the values of  $(x, y)$ ; and
- (3) Computing a sum of products over all values of  $m$  and  $n$ , for each displacement  $(x, y)$ .

The displacements  $(x, y)$  are integer increments that stop when the function no longer overlap.

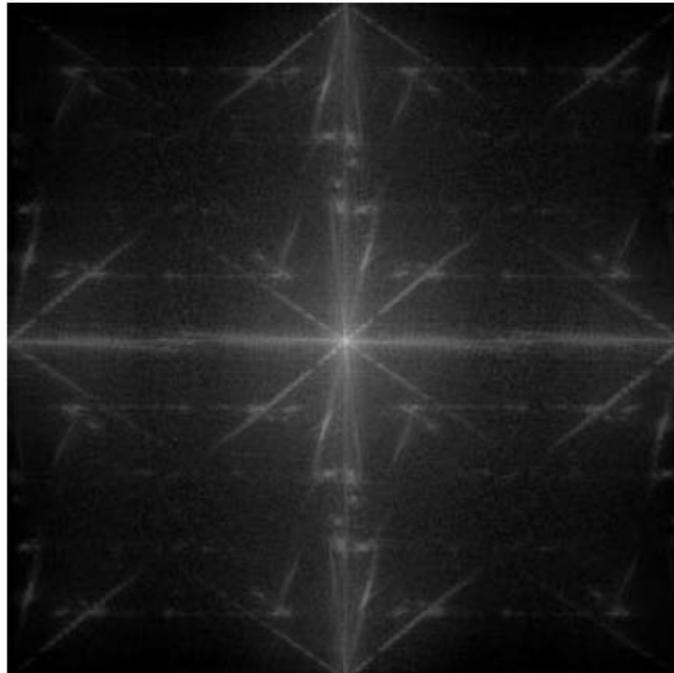
# Correspondence between Filtering in the Spatial domain and Frequency Domain



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.

# Correspondence between Filtering in the Spatial domain and Frequency Domain

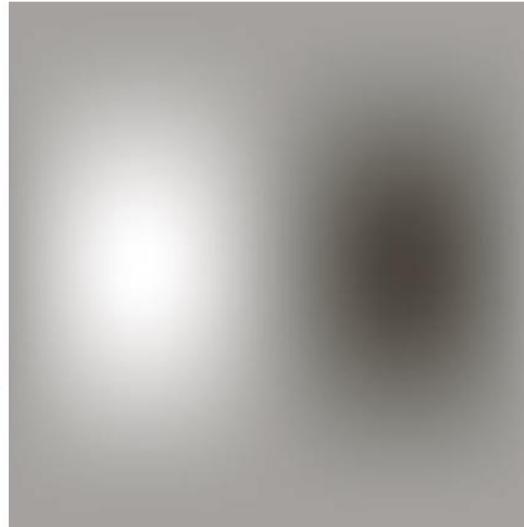
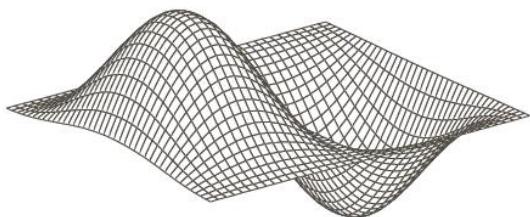


a | b

**FIGURE 4.38**  
(a) Image of a building, and  
(b) its spectrum.

# Correspondence between Filtering in the Spatial domain and Frequency Domain

-1	0	1
-2	0	2
-1	0	1

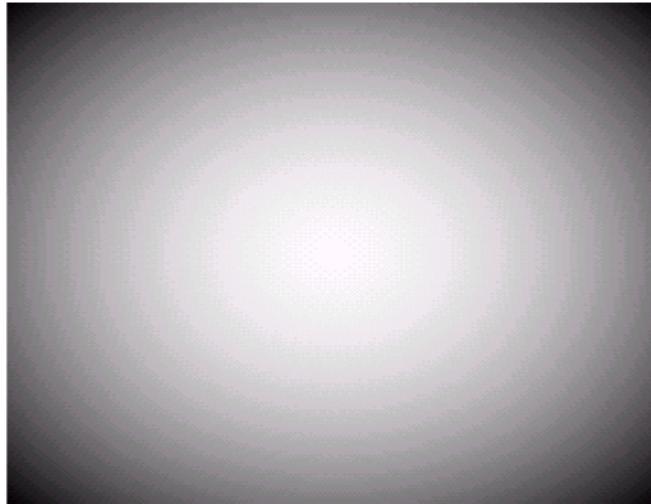
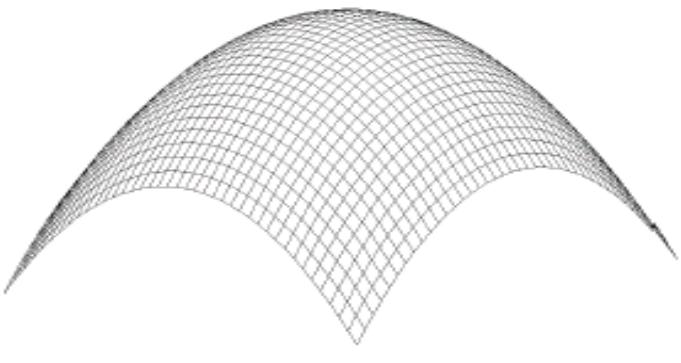
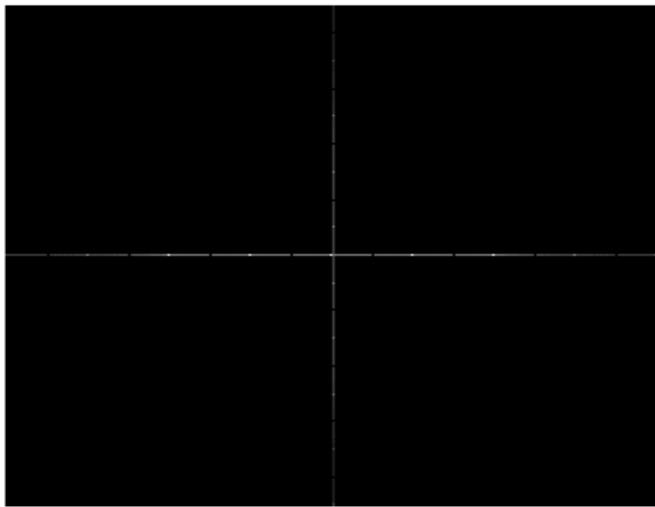


a b  
c d

**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.

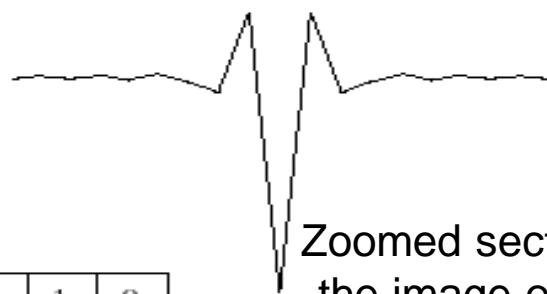
# Laplacian In The Frequency Domain

Inverse DFT of Laplacian in the frequency domain



2-D image of Laplacian in the frequency domain

0	1	0
1	-4	1
0	1	0



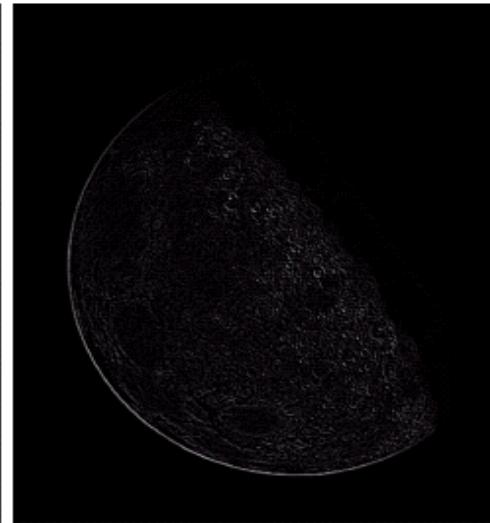
Zoomed section of the image on the left compared to spatial filter

# Frequency Domain Laplacian Example

Original  
image



Laplacian  
filtered  
image



Laplacian  
image  
scaled



Enhanced  
image



# Correlation

$$f(x, y) \circ h(x, y) = \frac{1}{MN} \sum_{m=0}^{M-1} \sum_{n=0}^{N-1} f^*(m, n) h(x + m, y + n)$$

$$f(x, y) \circ h(x, y) \Leftrightarrow F^*(u, v) H(u, v)$$

$$f^*(x, y) h(x, y) \Leftrightarrow F(u, v) \circ H(u, v)$$

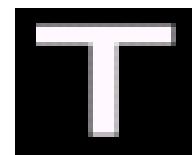
If  $f = x + iy$  then  $f^* = x - iy$

$f^*$  denotes the complex conjugate of  $f$ . We normally deal with real functions (images), in which case  $f^* = f$ .

# Application of Correlation

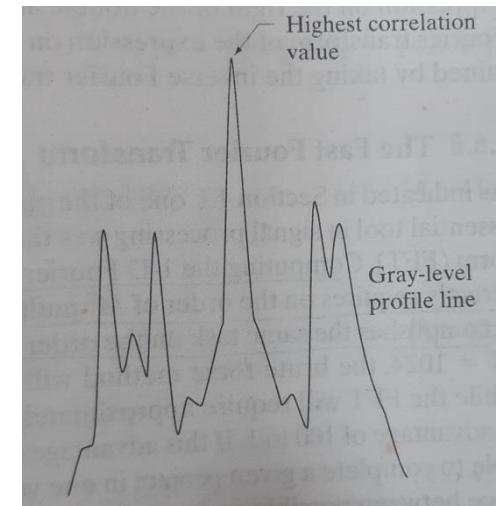


Input  
Image



Template

# Application of Correlation



Correlation  
function displayed  
as an image

# Fast Fourier Transform (FFT)

The reason that Fourier based techniques have become so popular is the development of the *Fast Fourier Transform (FFT)* algorithm.

Allows the Fourier transform to be carried out in a reasonable amount of time.

Reduces the amount of time required to perform a Fourier transform by a factor of 100 – 600 times!

# Frequency Domain Filtering & Spatial Domain Filtering

- Similar jobs can be done in the spatial and frequency domains.
- Filtering in the spatial domain can be easier to understand.
- Filtering in the frequency domain can be much faster – particularly for large images.