

DIGITAL IMAGE PROCESSING

Image Enhancement:
Filtering in the Frequency Domain

Contents



In this lecture we will look at image enhancement in the frequency domain

- ▣ Jean Baptiste Joseph Fourier
- ▣ The Fourier series & the Fourier transform
- ▣ Image Processing in the frequency domain
 - ▣ Image smoothing
 - ▣ Image sharpening
- ▣ Fast Fourier Transform

Jean Baptiste Joseph Fourier



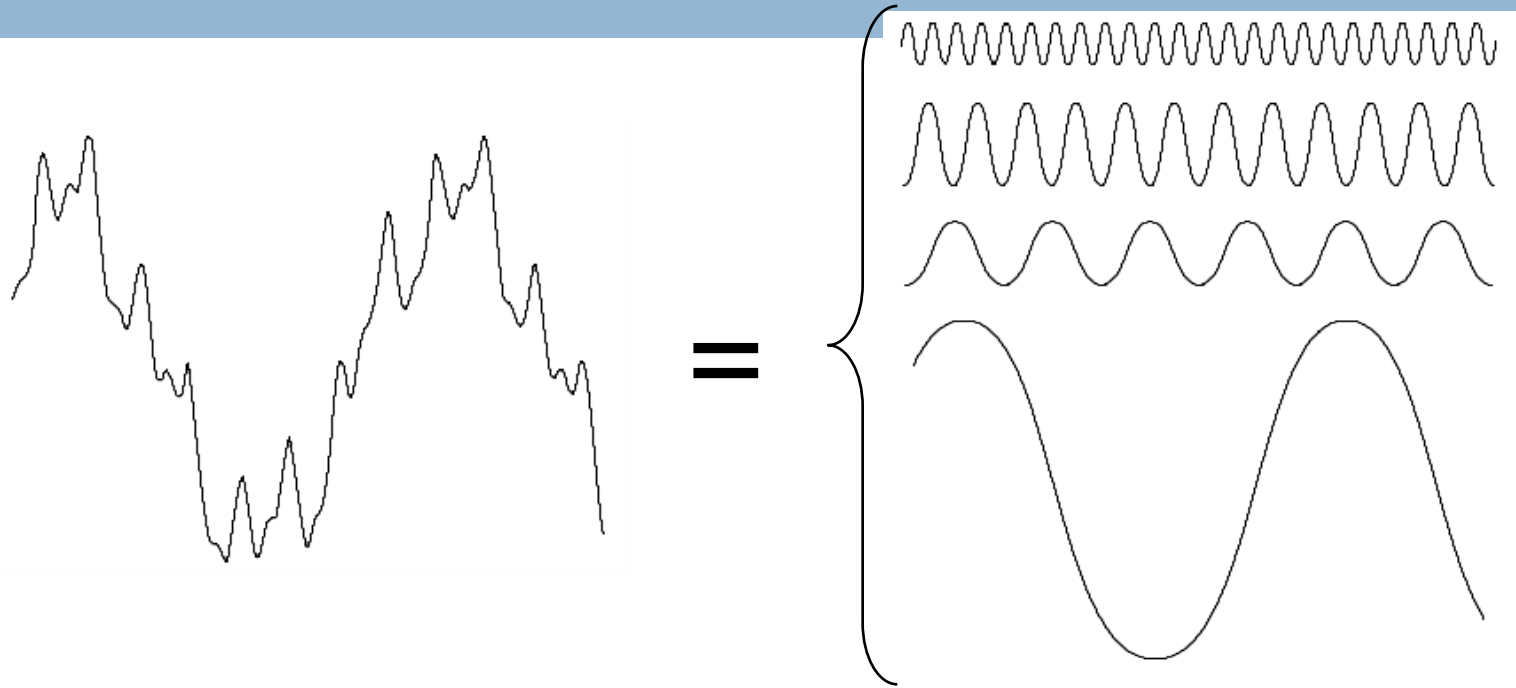
Fourier was born in Auxerre,
France in 1768

- Most famous for his work “*La Théorie Analytique de la Chaleur*” published in 1822
- Translated into English in 1878: “*The Analytic Theory of Heat*”

Nobody paid much attention when the work was first published

One of the most important mathematical theories in modern engineering

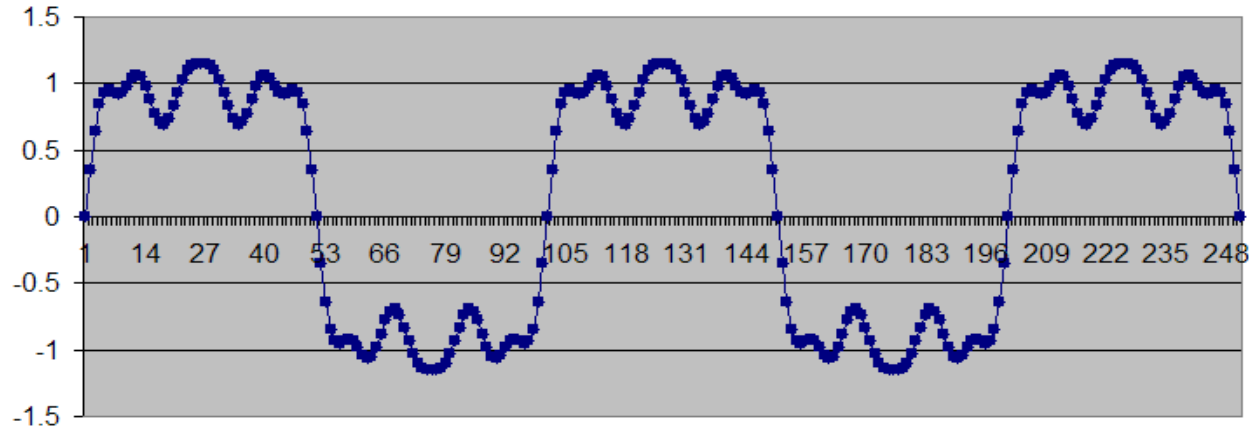
The Big Idea



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

The Big Idea (cont...)

Filtered Signal



we get closer and closer to the original function as we add more and more frequencies

Filtered Signal

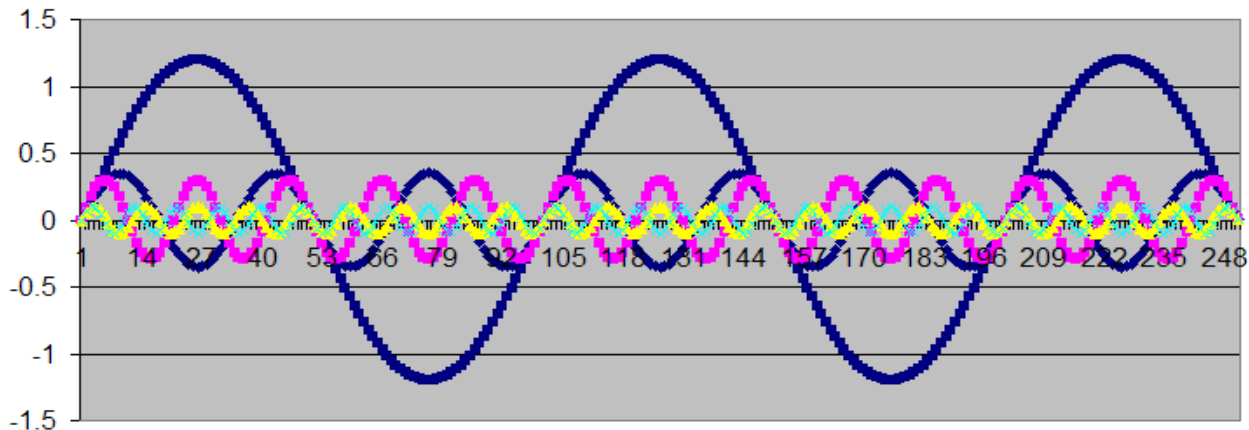


Image processing in the Frequency domain

- Why bother going into frequency domain?
- How this transformation is done?
- What does Fourier Transform tell us?
- What is the relationship between images in Spatial and Frequency domain?
- Analysis of frequency domain filters and their applications on images

Why bother going into frequency domain?

- Frequency domain representation makes it easy to visualize some characteristics of images
- It is easy to conceptualize filters in frequency domain
- Once a filter is selected in the frequency domain, it is usually implemented in the spatial domain
- Frequency domain steps:
 - ▣ Transformation from spatial to frequency domain
 - ▣ Image processing in the frequency domain
 - ▣ Inverse transformation back to the Spatial domain

What do frequencies mean in an image?

- Frequency in a digital image corresponds to the pace of variation of the pixels' gray level values
- Low frequencies indicate and correspond to slow varying pixel values
- High frequencies indicate high variation in the pixel values
- So frequency domain representation gives us a measure of pixels distribution in an image
- Fourier transform makes a connection between *spatial* and *frequency* domains

Discrete Fourier Transform of One Variable

Sampled Function

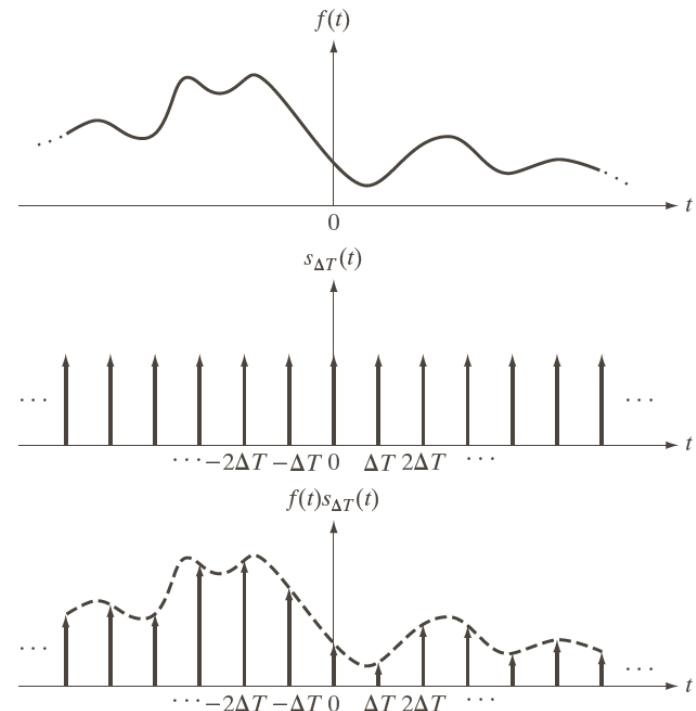
The Fourier Transform of a one variable function can be obtained using the relation:

$$F(u) = \int_{-\infty}^{\infty} \bar{f}(t) e^{-j2\pi ut} dt$$

where $\bar{f}(t) = f(t)s_{\Delta T}(t) = \sum_{n=-\infty}^{\infty} f(t)\delta(t - n\Delta T)$

Substituting...

$$\begin{aligned} F(u) &= \int_{-\infty}^{\infty} \sum_{n=-\infty}^{\infty} f(t)\delta(t - n\Delta T) e^{-j2\pi ut} dt \\ &= \sum_{n=-\infty}^{\infty} \int_{-\infty}^{\infty} f(t)\delta(t - n\Delta T) e^{-j2\pi ut} dt \\ &= \sum_{n=-\infty}^{\infty} f_n e^{-j2\pi un\Delta T} \end{aligned}$$



The function f_n is discrete, however its transform is continuous and infinitely periodic with period $1/\Delta T$

Discrete Fourier Transform of One Variable

Sampled Function

Suppose we want to have M equally spaced samples of the transform taken over the period $u = 0$ to $u = \frac{1}{\Delta T}$

This can be done by: $u = \frac{m}{M\Delta T} \quad m = 0, 1, 2, \dots, M-1$

Substituting $u \dots$

$$F_m = \sum_{n=0}^{M-1} f_n e^{-j2\pi mn / M} \quad m = 0, 1, 2, \dots, M-1$$

Discrete Fourier Transform

- From DIP's perspective, the input is a two dimensional discrete signal (image)
- For a discrete one dimensional function $f(x)$, with $x=0, 2, \dots M-1$

- The DFT is:
$$F(u) = \frac{1}{M} \sum_{x=0}^{M-1} f(x) e^{-j2\pi ux/M}, \quad u = 0, \dots, M-1$$

- Or :

$$F(u) = \frac{1}{M} \sum_{x=0}^{M-1} f(x) \left(\cos \frac{2\pi ux}{M} - j \sin \frac{2\pi ux}{M} \right)$$

- The **F(u)** is computed for **0 to M-1** frequencies
- Each **F(u)** represents a frequency component of the input image
- The inverse DFT is:

$$f(x) = \sum_{u=0}^{M-1} F(u) e^{j2\pi ux/M}, \quad x = 0, \dots, M-1$$

Discrete Fourier Transform

- The first value of $F(u)$ starts from $u=0$ to $u=M-1$
- And for every value of 'u' all values of x which also range from 0 to $M-1$ are summed up and multiplied with the exponential term

$$F(u) = \frac{1}{M} \sum_{x=0}^{M-1} f(x) e^{-j2\pi ux/M}, \quad u = 0, \dots, M-1$$

- Or :

$$F(u) = \frac{1}{M} \sum_{x=0}^{M-1} f(x) \left(\cos \frac{2\pi ux}{M} - j \sin \frac{2\pi ux}{M} \right)$$

- The $\mathbf{F(u)}$ is computed for **0 to M-1** frequencies

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

- At $u=0, v=0$, we get: $F(0,0) = \frac{1}{MN} \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x, y)$ Average value

DFT- Translation

- The duration 0 to M-1 consists of two back to back half periods
- For display and filtering purposes it is often desired to have the complete period within this interval

$$f(x)e^{j2\pi(u_0x/M)} \Leftrightarrow F(u - u_0)$$

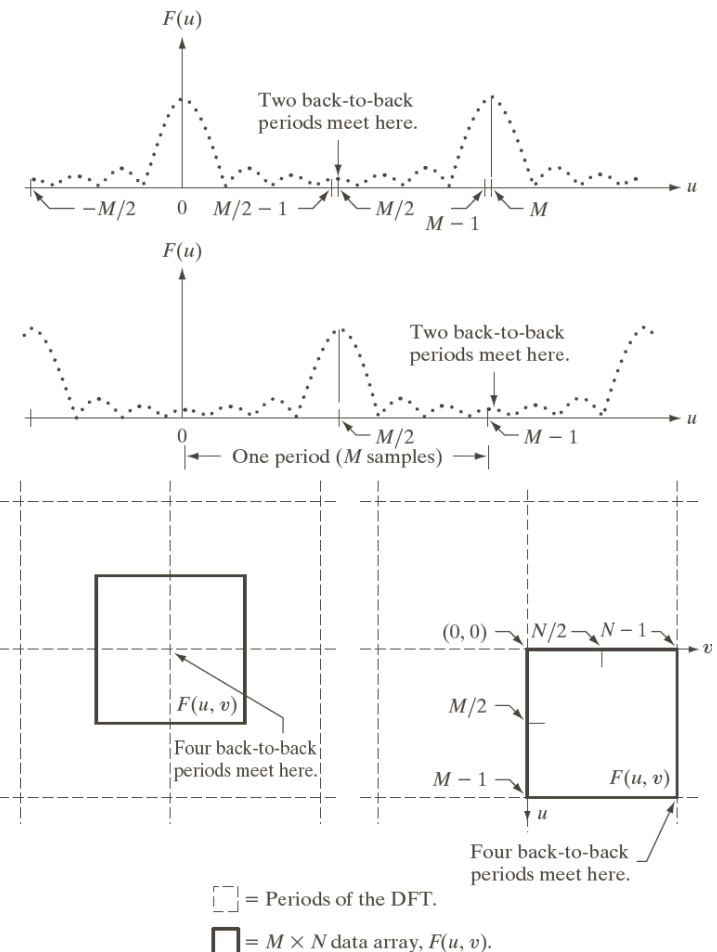
- In other words, multiplying f(x) by the exponential term shifts the transform so that the origin F(0), is located at

$$f(x, y)e^{j2\pi(u_0x/M + v_0y/N)} \Leftrightarrow F(u - u_0, v - v_0)$$

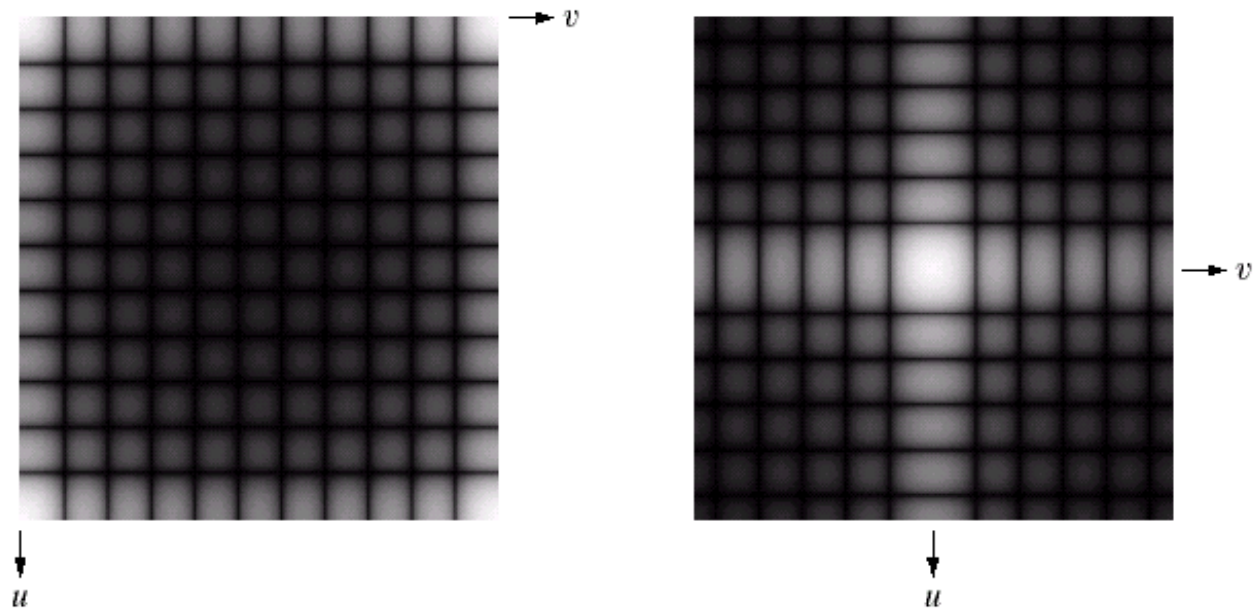
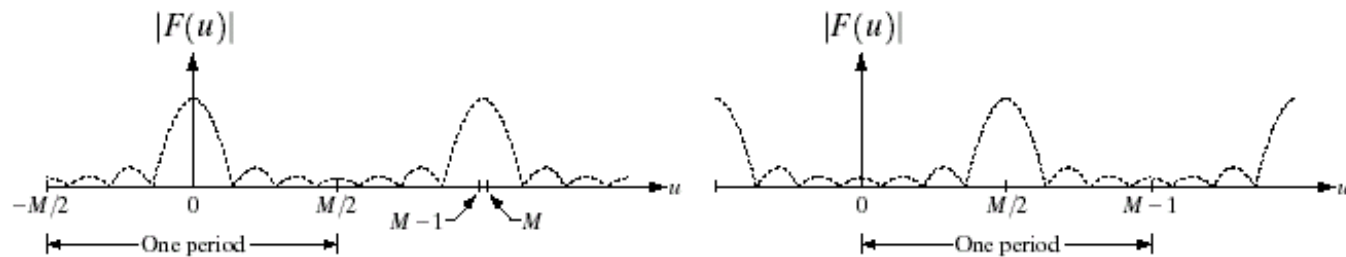
- If we let $u_0 = \frac{M}{2}$, then the exponential term becomes:

$$e^{j\pi x} \quad \text{or} \quad (-1)^x$$

$$f(x, y)(-1)^{x+y} \Leftrightarrow F(u - M/2, v - N/2)$$



PERIODICITY OF DFT



The Discrete Fourier Transform (DFT)

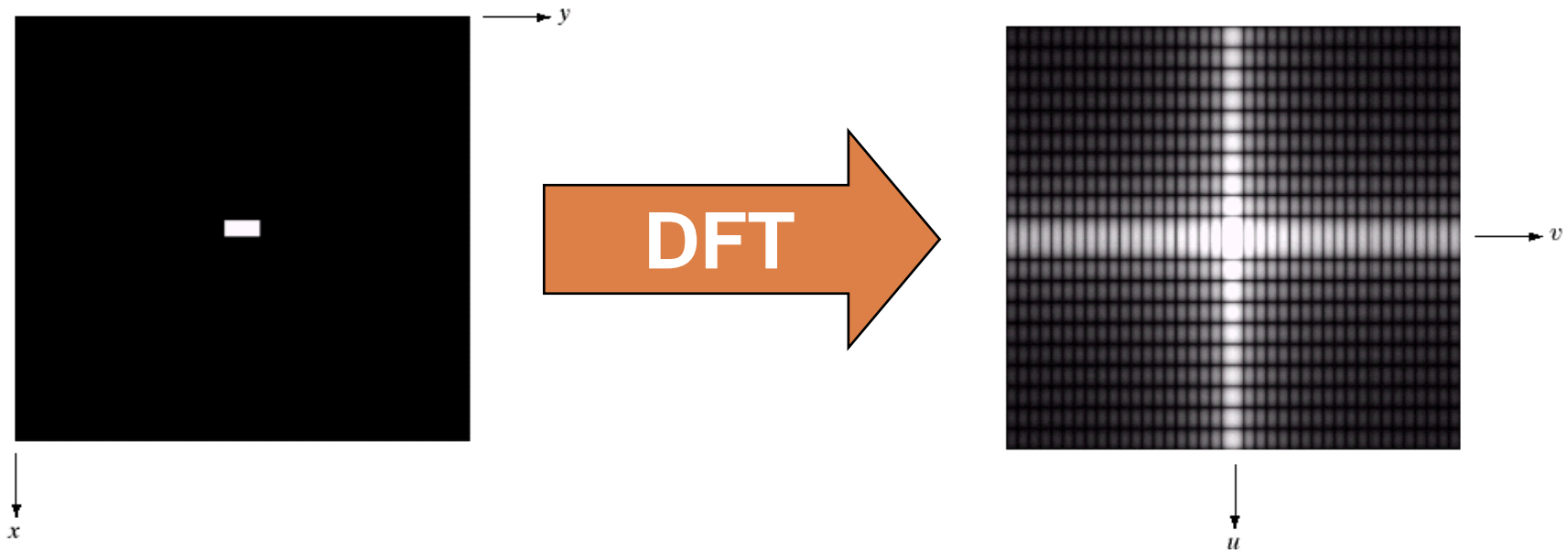
The *Discrete Fourier Transform* of $f(x, y)$, for $x = 0, 1, 2 \dots M-1$ and $y = 0, 1, 2 \dots N-1$, denoted by $F(u, v)$, is given by the equation:

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

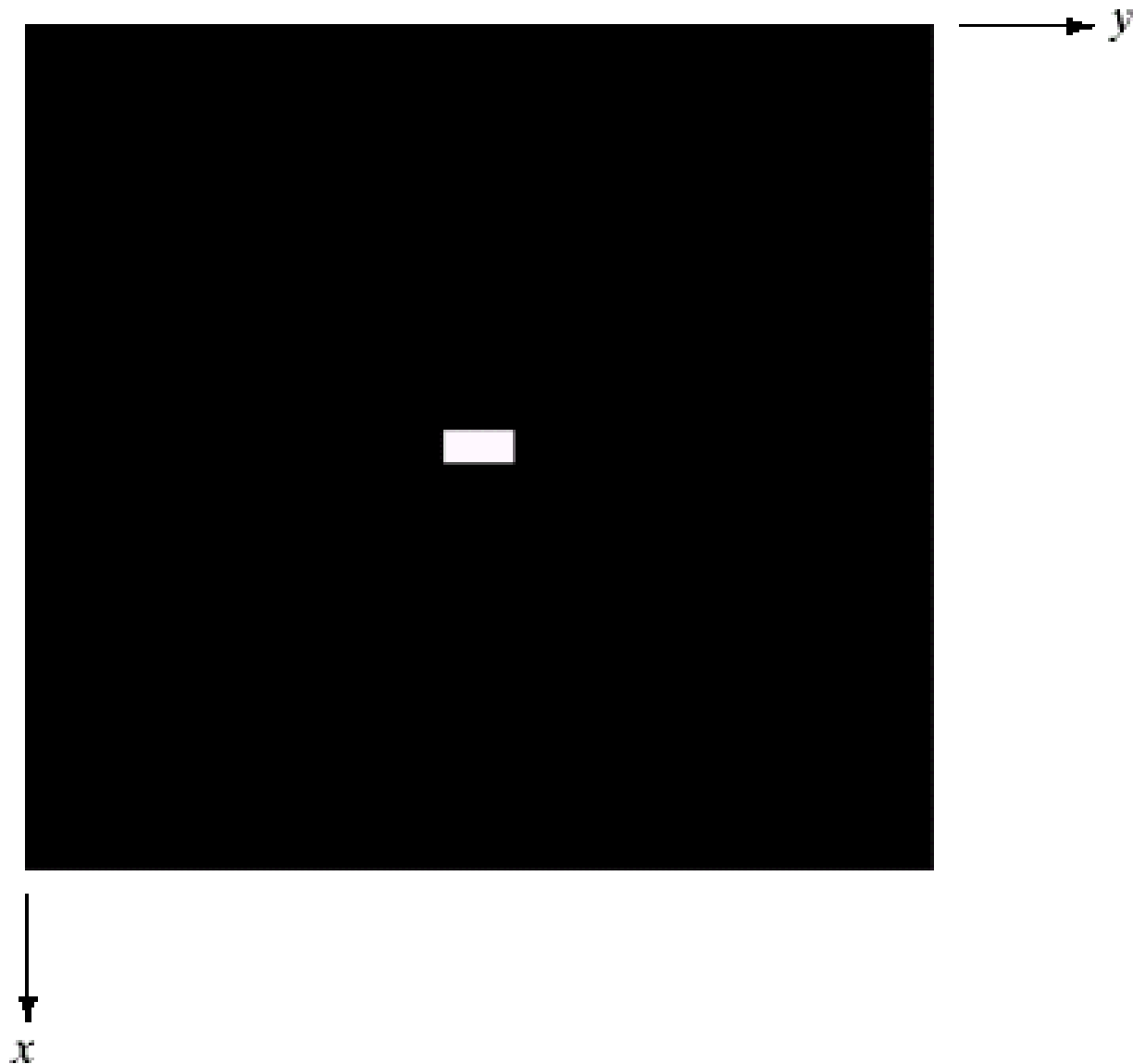
for $u = 0, 1, 2 \dots M-1$ and $v = 0, 1, 2 \dots N-1$.

DFT & Images

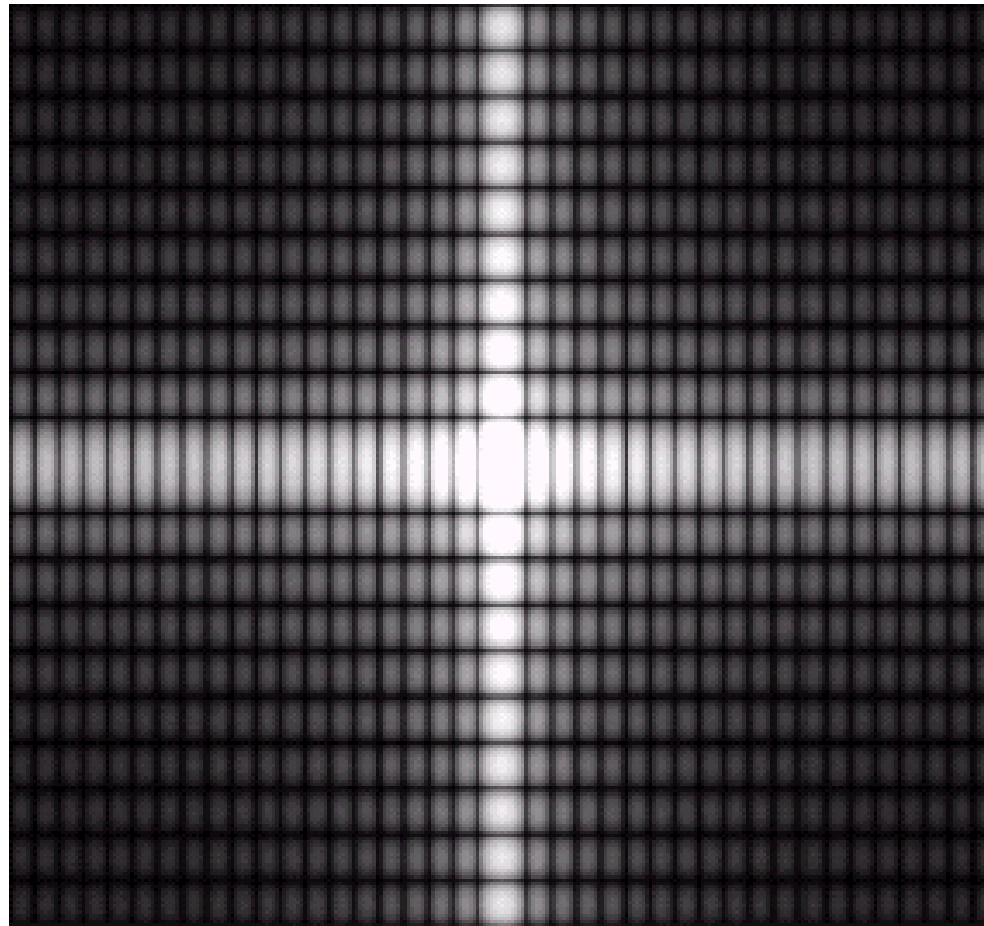
The DFT of a two dimensional image can be visualised by showing the spectrum of the images component frequencies



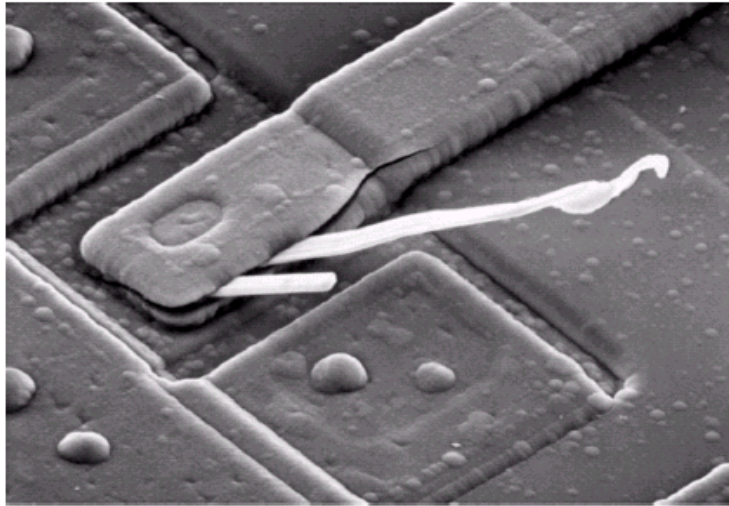
DFT & Images



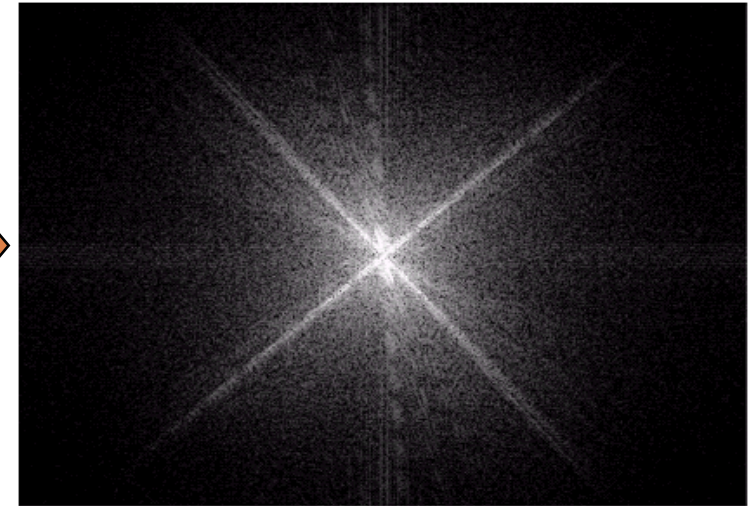
DFT & Images



DFT & Images (cont...)

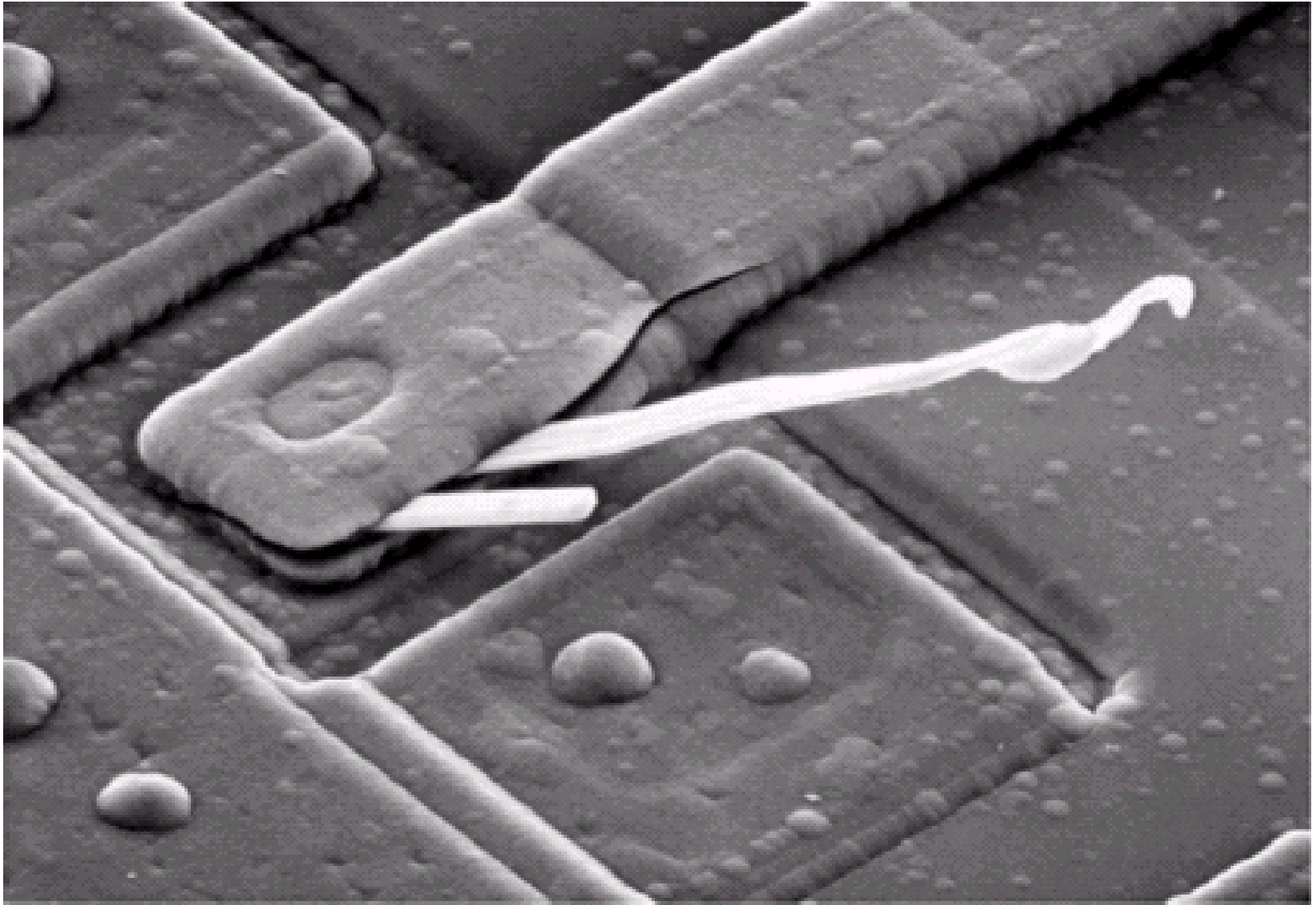


Scanning electron microscope image of an integrated circuit magnified ~2500 times

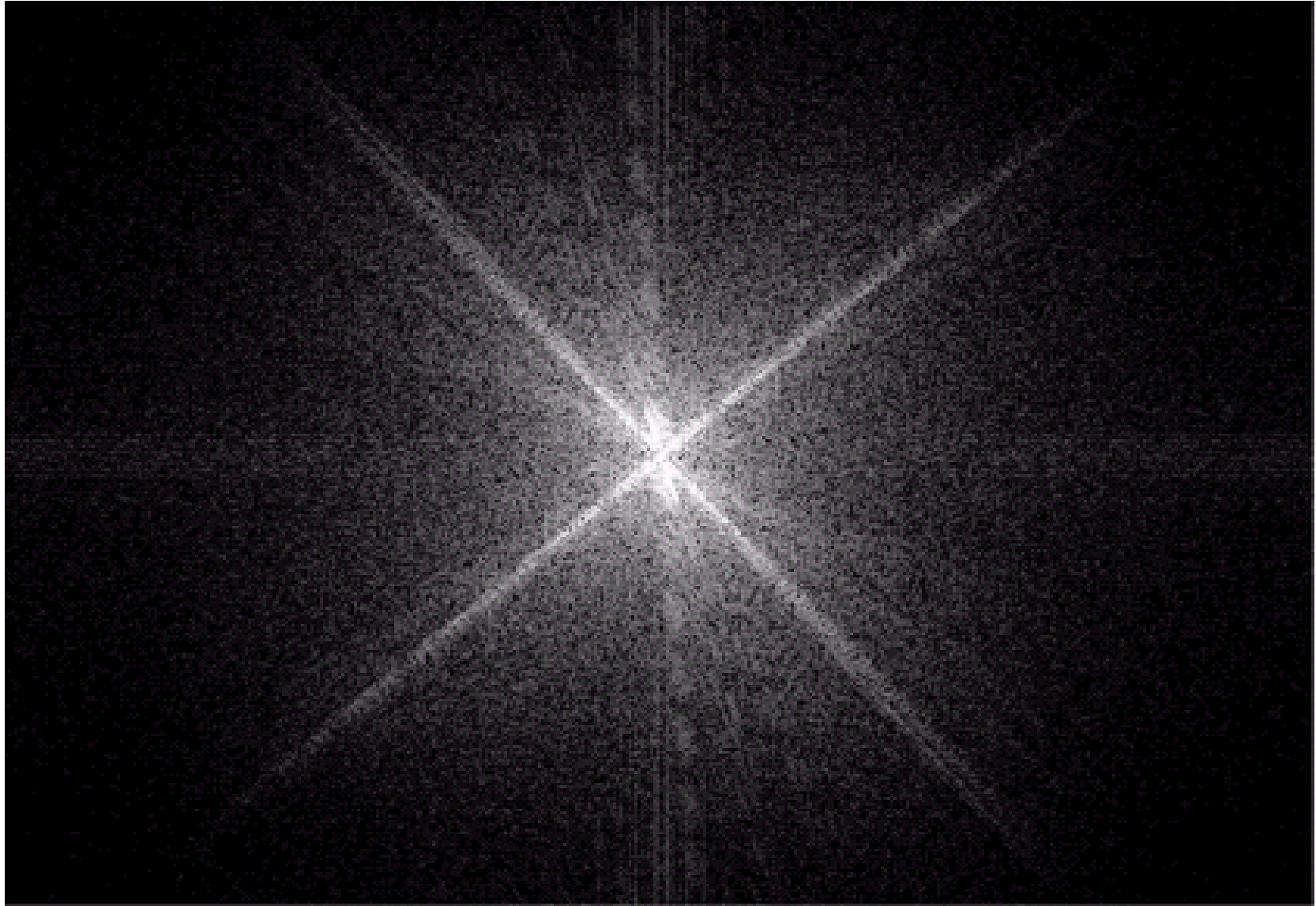


Fourier spectrum of the image

DFT & Images (cont...)



DFT & Images (cont...)



The Inverse DFT

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

The inverse DFT is given by:

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

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

Filtering in the Frequency domain

- Apart from trivial cases, establishing a relation between spatial domain and frequency domain characteristics of images is difficult
- When $u=v=0$, this corresponds to average value
- Moving away from this point, the low frequencies correspond to slowly varying components in an image
- The higher frequencies correspond to faster gray level changes
- Such relationships (although gross) can help establishing enhancement techniques in the frequency domain

Basics of Filtering in the Frequency domain

Filtering in FD consists modifying the Fourier transform of an image with a filter function and then computing the inverse transform to get the output

With an image of size **$M \times N$** :

$$g(x, y) = \mathcal{F}^{-1}[H(u, v)F(u, v)]$$

The functions F , H and g are arrays of size $M \times N$

PROPERTIES OF DFT

CHOICE OF DFT INTERVAL

- If the interval lengths of $f(x)$ and $h(x)$ are M and N respectively interval length for $f(x)*h(x)$ will be $M+N-1$

Effect on frequency domain filtering!

The 2-D Convolution Theorem

The 2-D convolution theorem is given by the expressions:

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

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

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

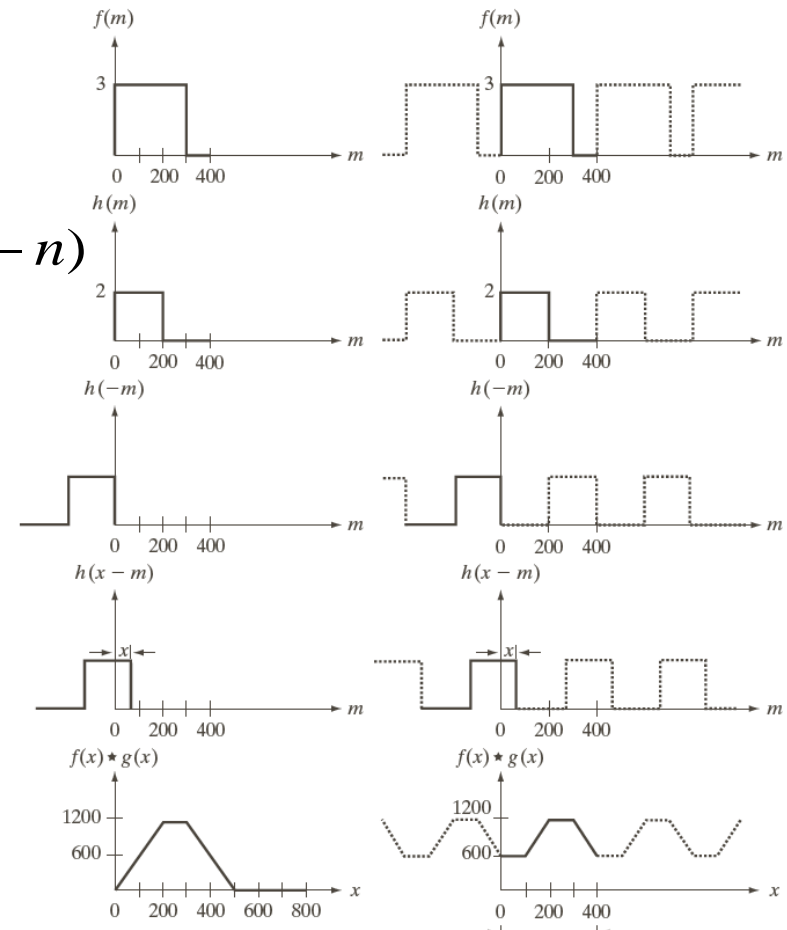
$$f(x) * h(x) = \sum_{m=0}^{399} f(x)h(x-m)$$

Taking into account the periodicity implied by the DFT yields incorrect convolution result

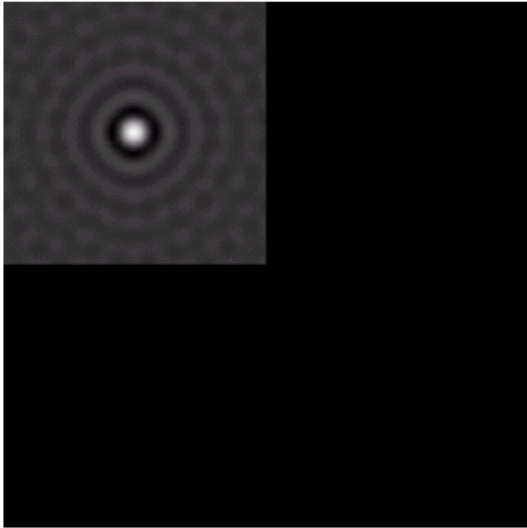
The solution is to use zero padding

If $f(x)$ and $h(x)$ consists of A and B samples, the padding should be:

$$P \geq A+B-1$$



CHOICE OF INTERVAL SIZE FOR CONVOLUTION: PADDING EXAMPLE



Basics of Filtering in the Frequency domain

Filtering in FD consists of the following steps:

1. For an input image $f(x,y)$ of size $M \times N$, determine the zero padding parameters
2. Typically, the $P=2M$ and $Q=2N$
3. Form a padded image, $f_p(x,y)$, of size $P \times Q$ by appending the necessary number of zeros to $f(x,y)$
4. Multiply the $f_p(x,y)$ by $(-1)^{x+y}$ to centre the transform at $P/2, Q/2$
5. Compute the DFT, $F(u,v)$ of the image in step 4
6. Generate a filter function, $H(u,v)$ of size $P \times Q$ with centre at coordinates $(P/2, Q/2)$
7. Form the product $G(u,v) = H(u,v)F(u,v)$ using array multiplication
8. Obtain the output: $g_p(x,y) = \{real[\mathcal{F}^{-1}[G(u,v)]]\}(-1)^{x+y}$
9. Obtain the final processed image, $g(x,y)$ by extracting the $M \times N$ region from the top, left quadrant of $g_p(x,y)$

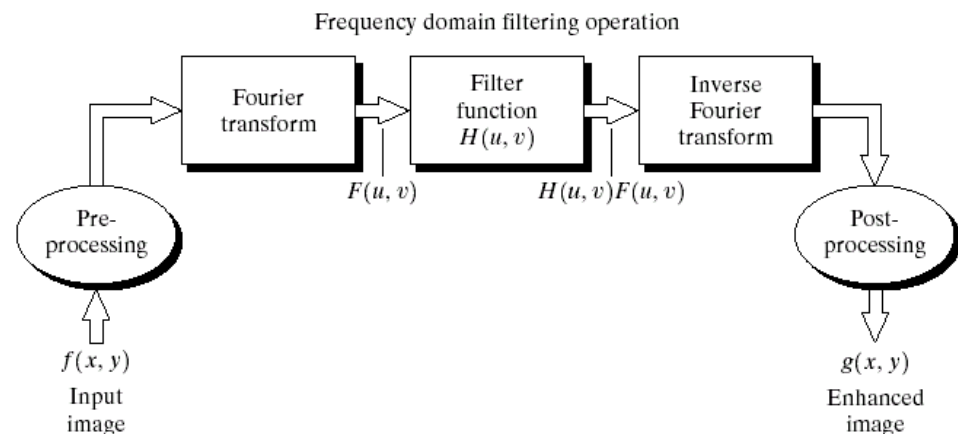
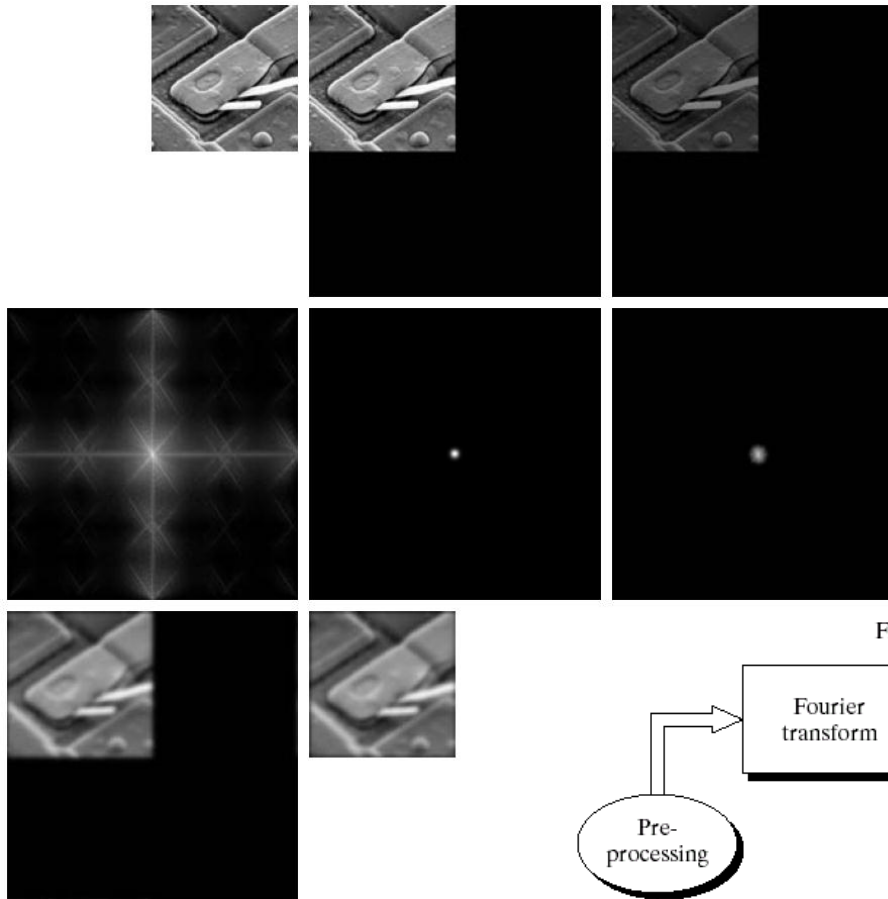
Basics of Filtering in the Frequency domain

a	b	c
d	e	f
g	h	

Steps of filtering in FD

FIGURE 4.36

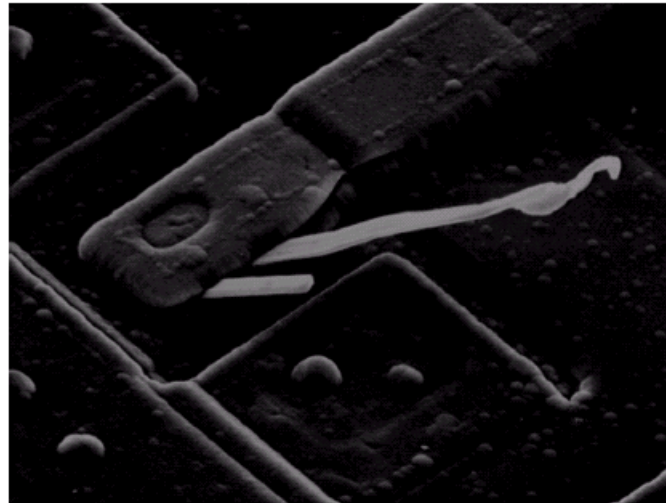
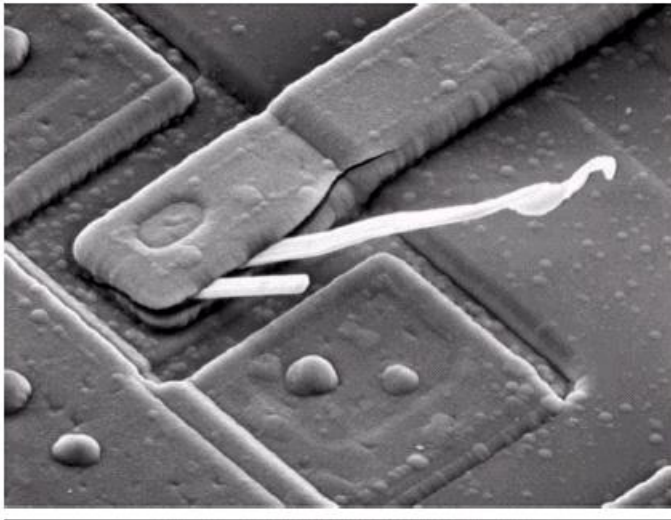
- (a) An $M \times N$ image, f .
 (b) Padded image, f_p of size $P \times Q$.
 (c) Result of multiplying f_p by $(-1)^{x+y}$.
 (d) Spectrum of F_p . (e) Centered Gaussian lowpass filter, H , of size $P \times Q$.
 (f) Spectrum of the product HF_p .
 (g) g_p , the product of $(-1)^{x+y}$ and the real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



Some basic filters and their properties

- A filter function that forces the average value of the Transform to zero:

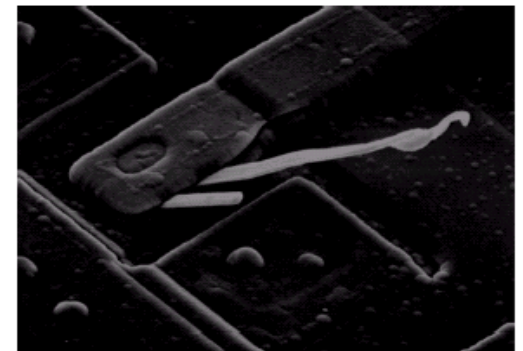
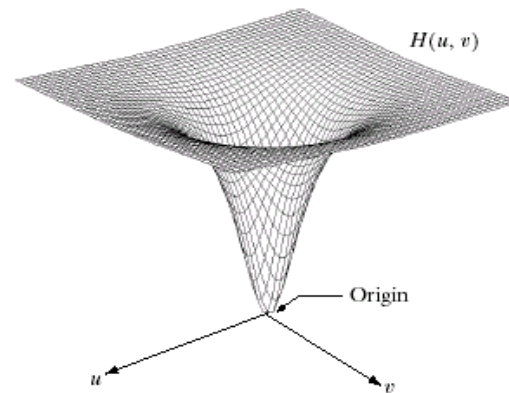
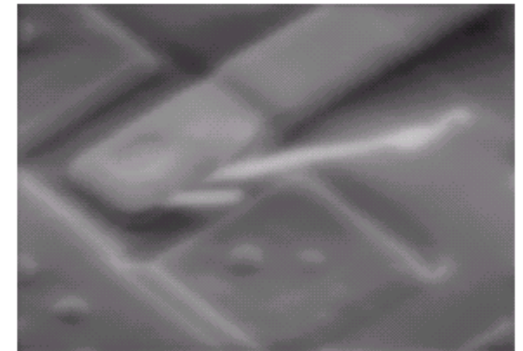
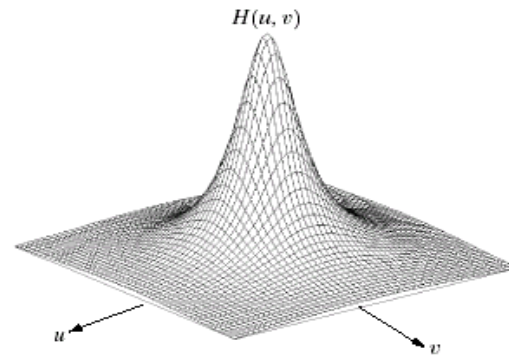
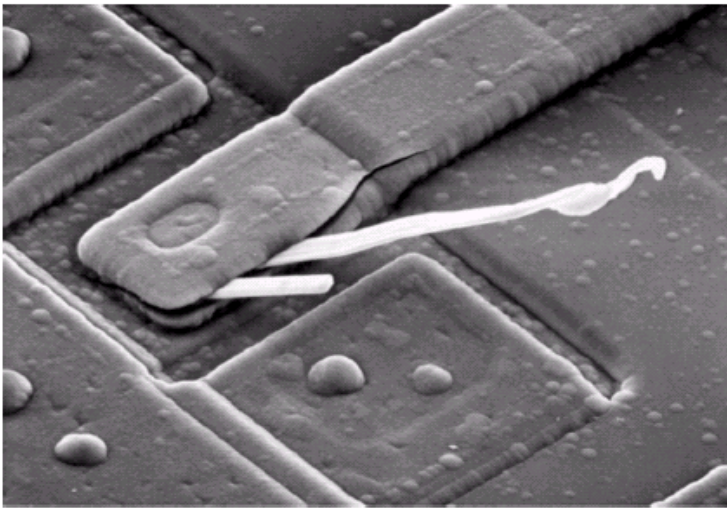
$$H(u, v) = \begin{cases} 0 & \text{if } (u, v) = (M/2, N/2) \\ 1 & \text{otherwise} \end{cases}$$



A zero average value means negative and positive values – the displayed image is scaled

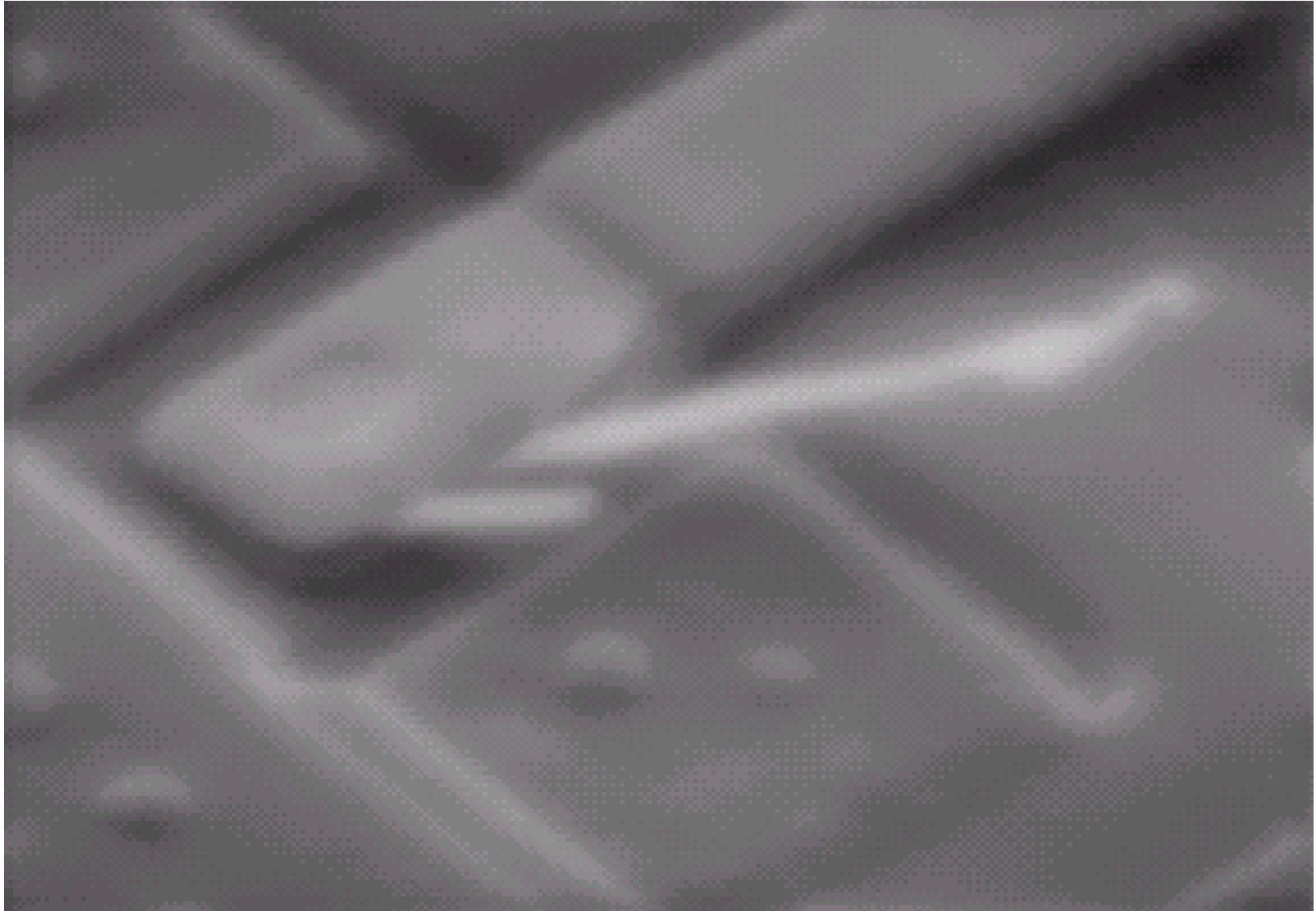
Some Basic Frequency Domain Filters

Low Pass Filter

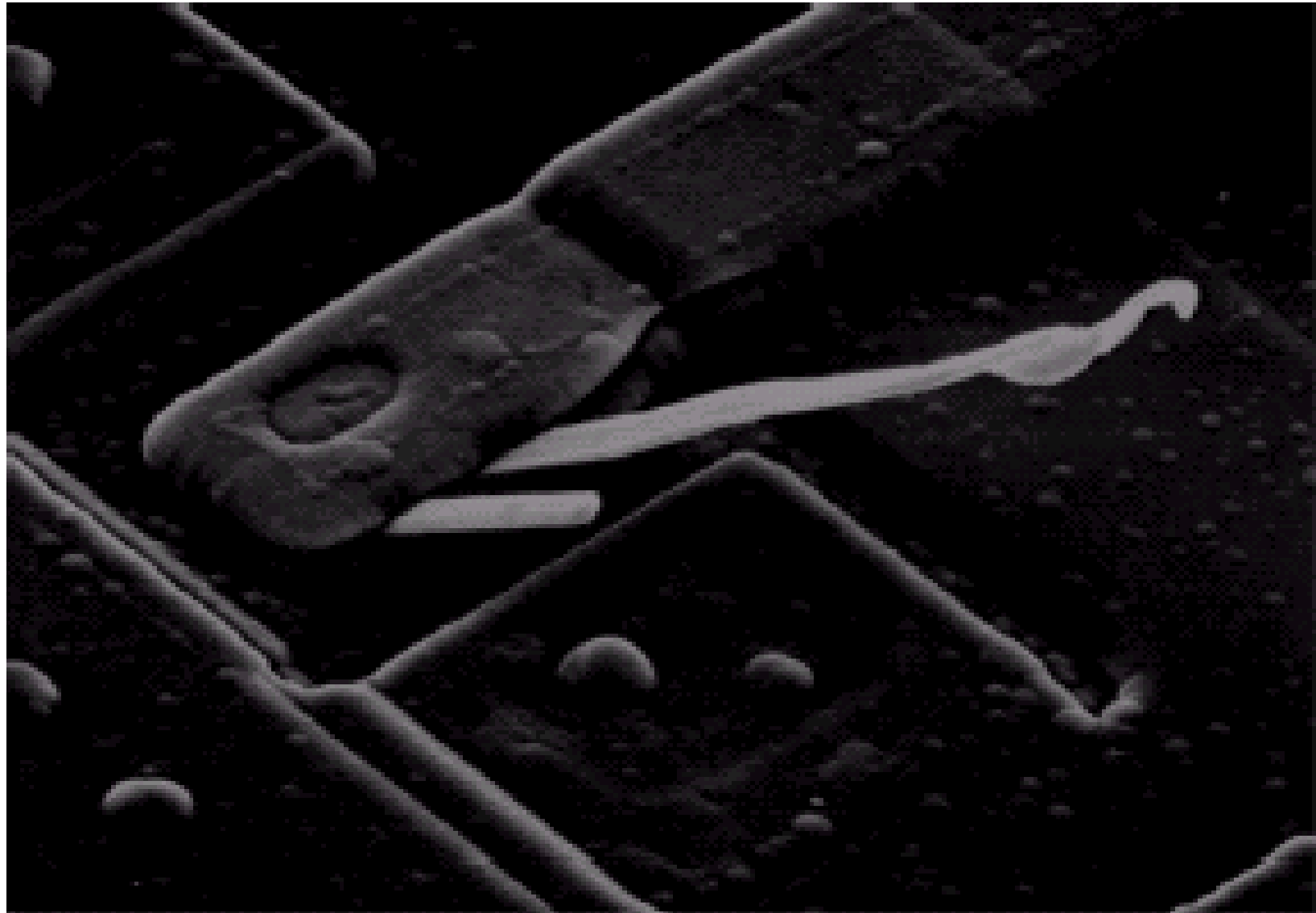


High Pass Filter

Some Basic Frequency Domain Filters



Some Basic Frequency Domain Filters



Smoothing Frequency Domain Filters

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

The basic model for filtering is:

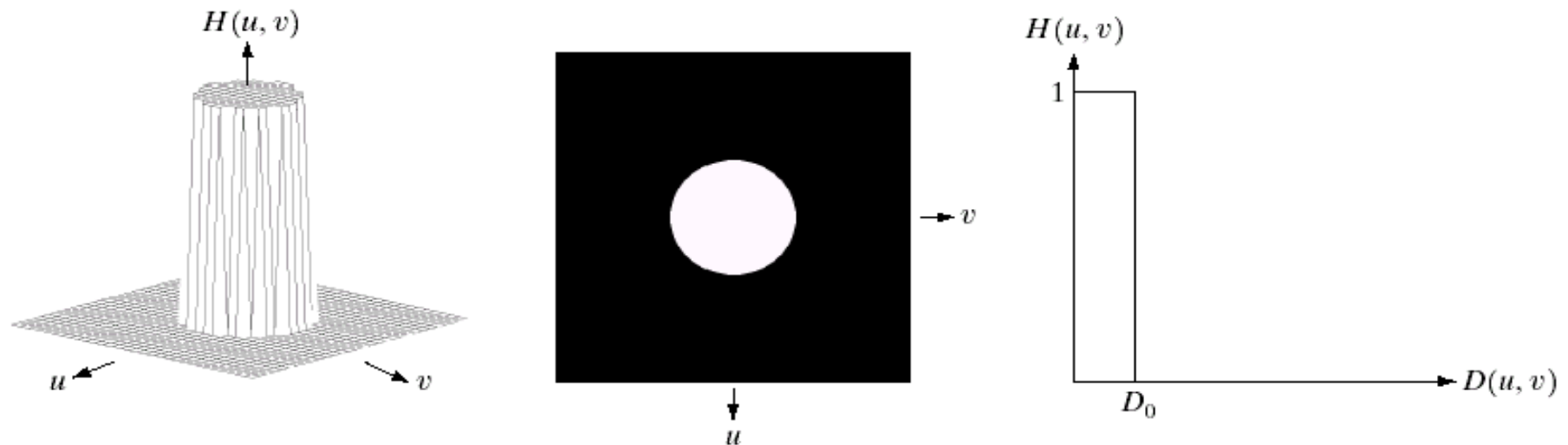
$$G(u,v) = H(u,v)F(u,v)$$

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



Ideal Low Pass Filter (cont...)

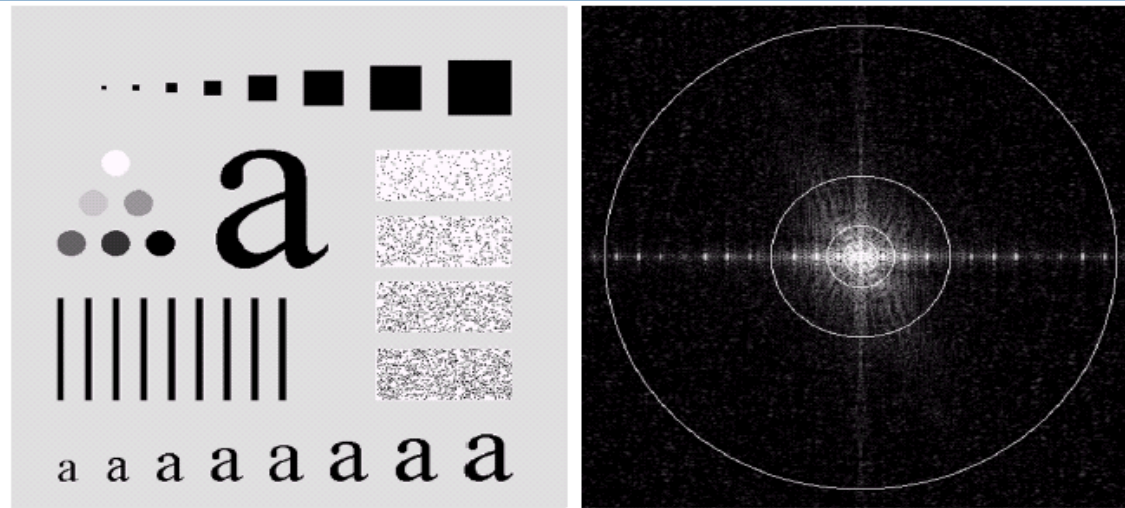
The transfer function for the ideal low pass filter can be given as:

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

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

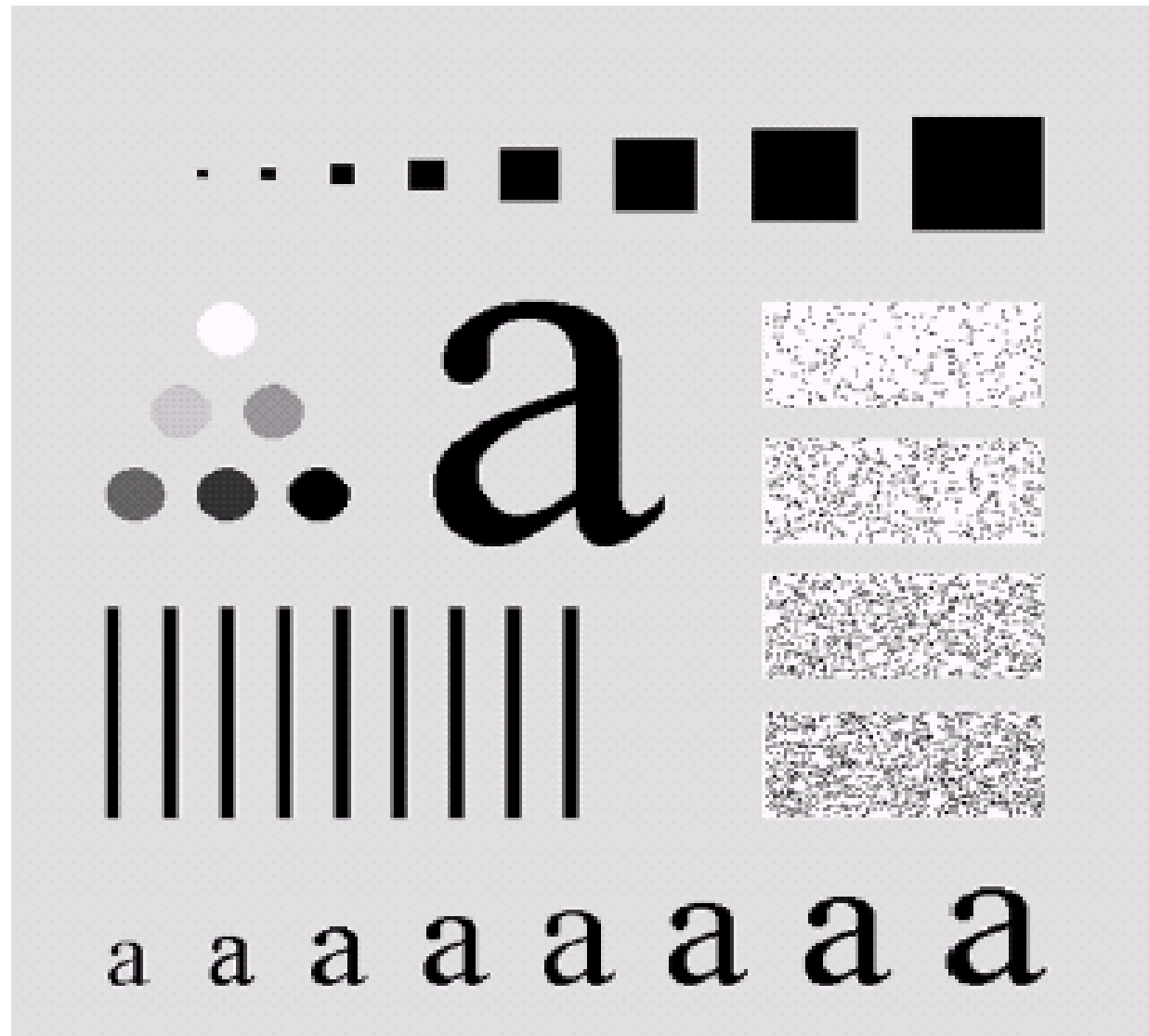
$$D(u, v) = [(u - M / 2)^2 + (v - N / 2)^2]^{1/2}$$

Ideal Low Pass Filter (cont...)

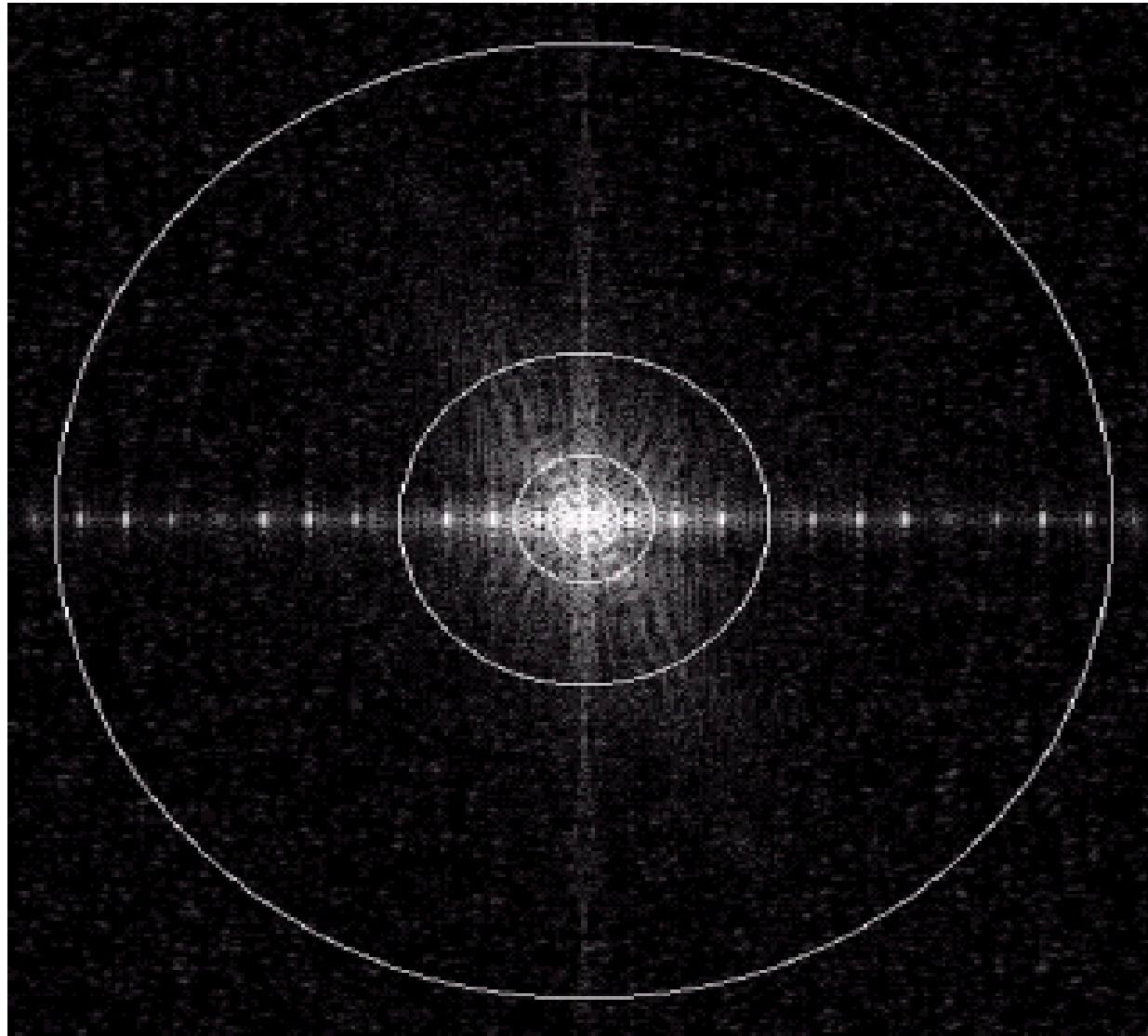


Above we show an image, it's Fourier spectrum and a series of ideal low pass filters of radius 5, 15, 30, 80 and 230 superimposed on top of it

Ideal Low Pass Filter (cont...)

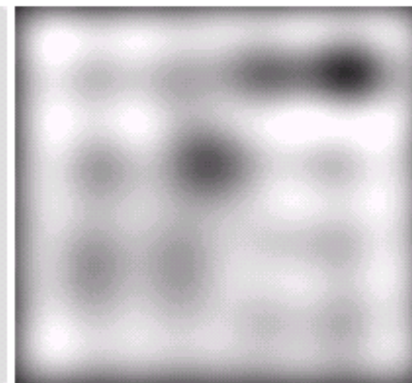


Ideal Low Pass Filter (cont...)



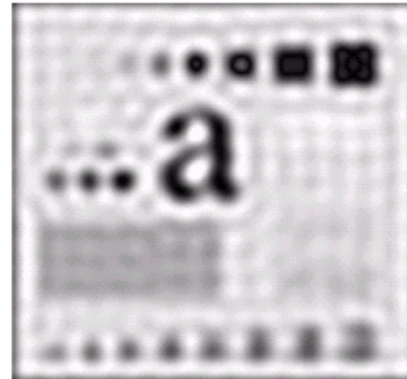
Ideal Low Pass Filter (cont...)

Original
image



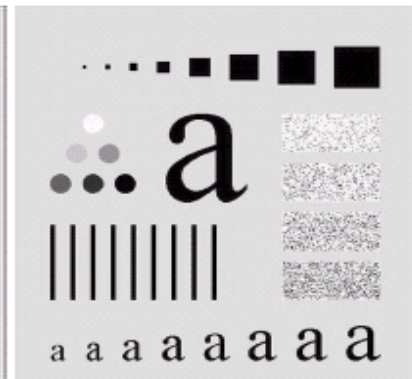
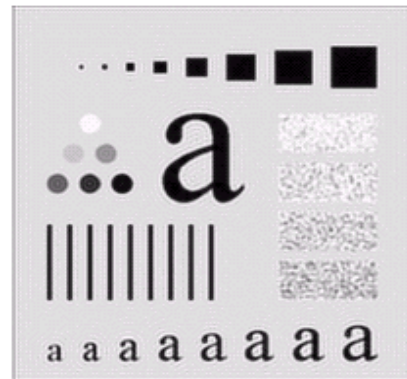
Result of filtering
with ideal low pass
filter of radius 5

Result of filtering
with ideal low pass
filter of radius 15



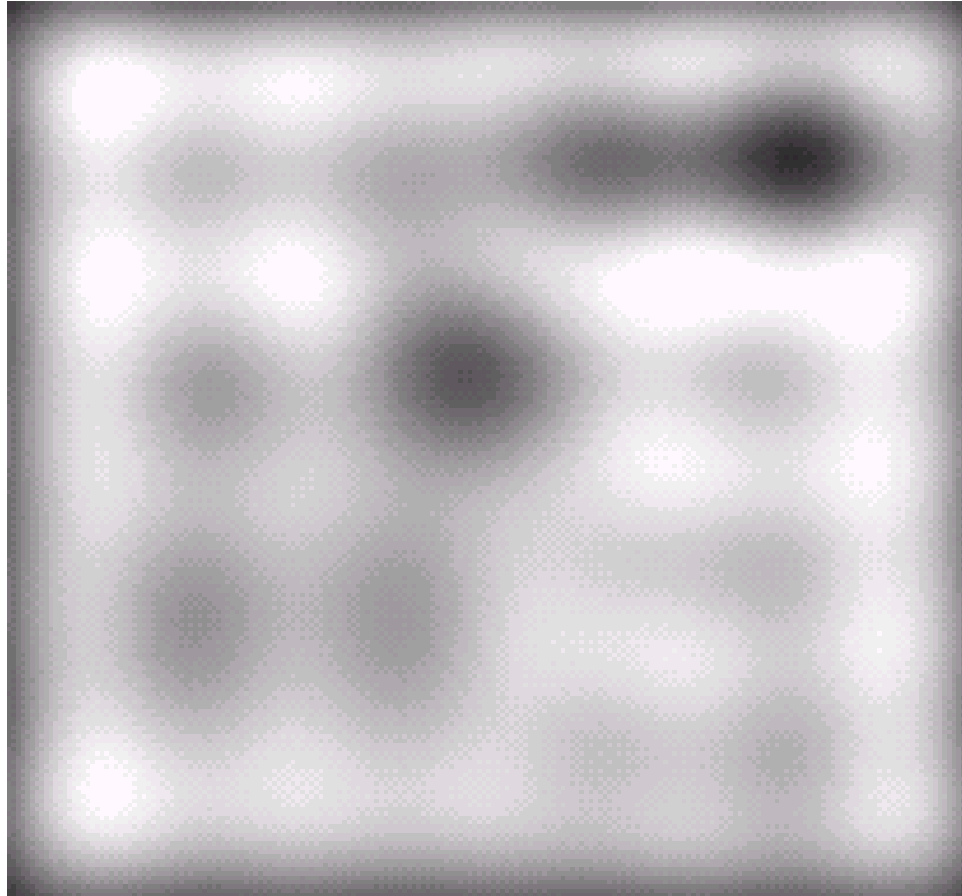
Result of filtering
with ideal low pass
filter of radius 30

Result of filtering
with ideal low pass
filter of radius 80



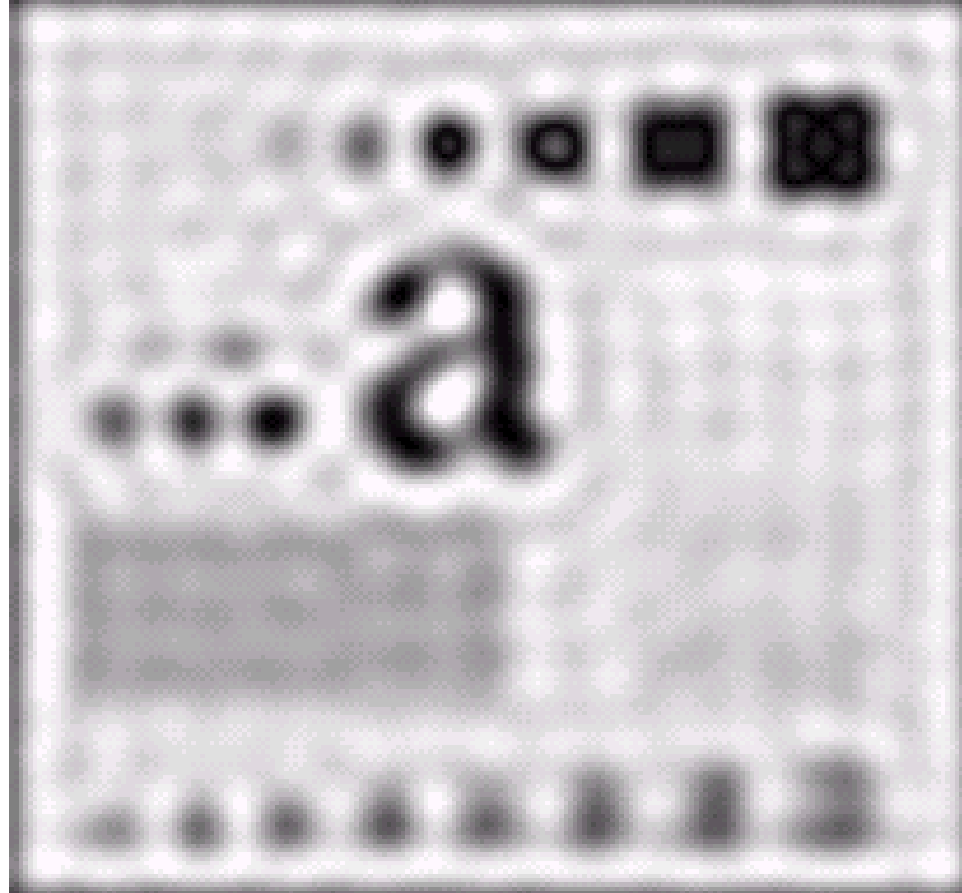
Result of filtering
with ideal low pass
filter of radius 230

Ideal Low Pass Filter (cont...)



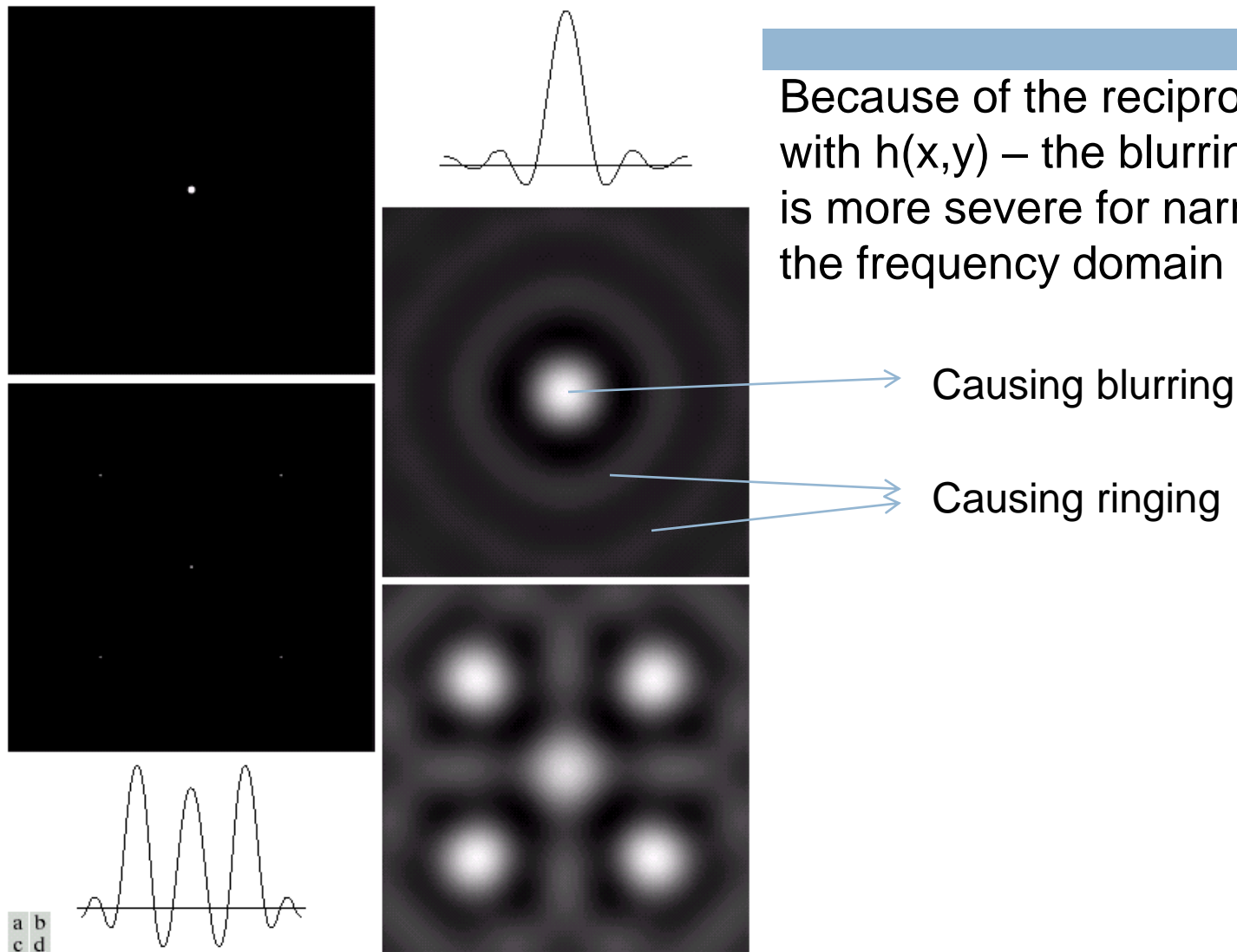
Result of filtering
with ideal low pass
filter of radius 5

Ideal Low Pass Filter (cont...)



Result of filtering
with ideal low pass
filter of radius 15

Ideal Low pass filters – blurring & ringing



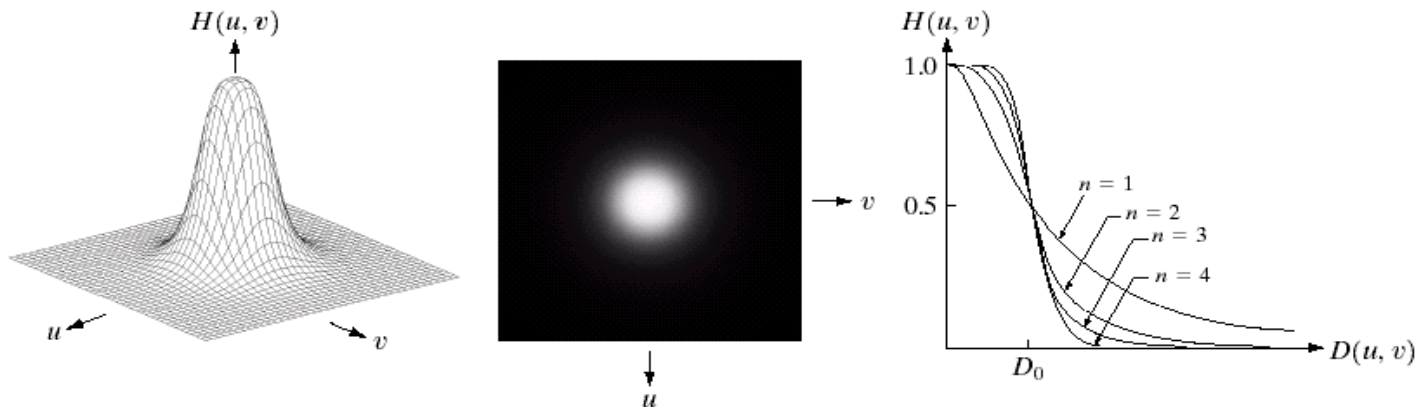
Because of the reciprocity of $H(u,v)$ with $h(x,y)$ – the blurring and ringing is more severe for narrower filters in the frequency domain

FIGURE 4.13 (a) A frequency-domain ILPF of radius 5. (b) Corresponding spatial filter (note the ringing). (c) Five impulses in the spatial domain, simulating the values of five pixels. (d) Convolution of (b) and (c) in the spatial domain.

Butterworth Lowpass Filters

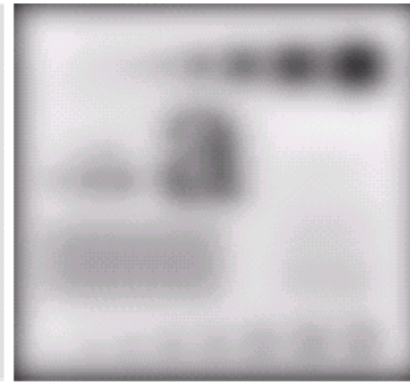
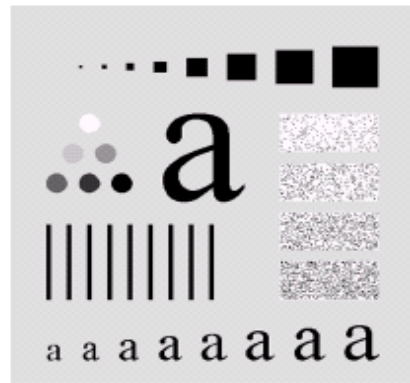
The transfer function of a Butterworth lowpass filter of order n with cutoff frequency at distance D_0 from the origin is defined as:

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



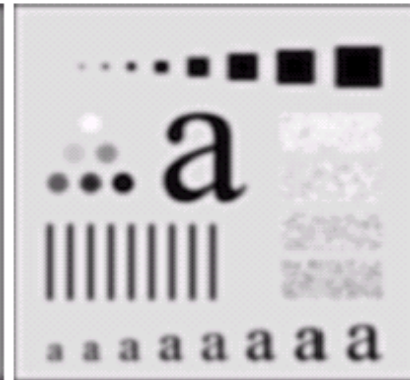
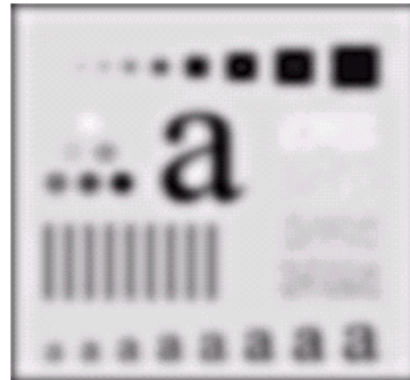
Butterworth Lowpass Filter (cont...)

Original image



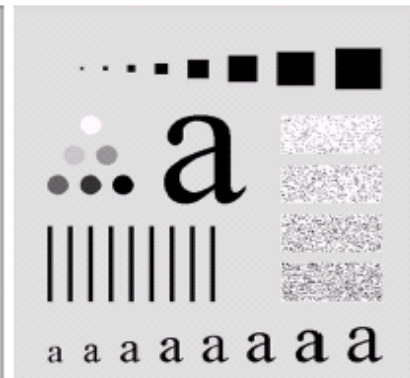
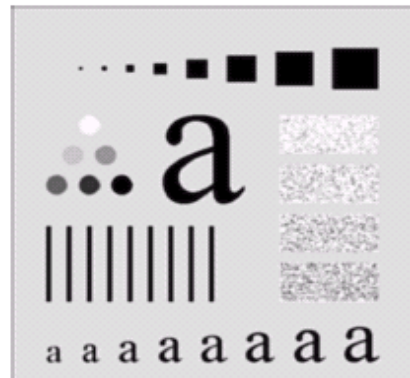
Result of filtering with Butterworth filter of order 2 and cutoff radius 5

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



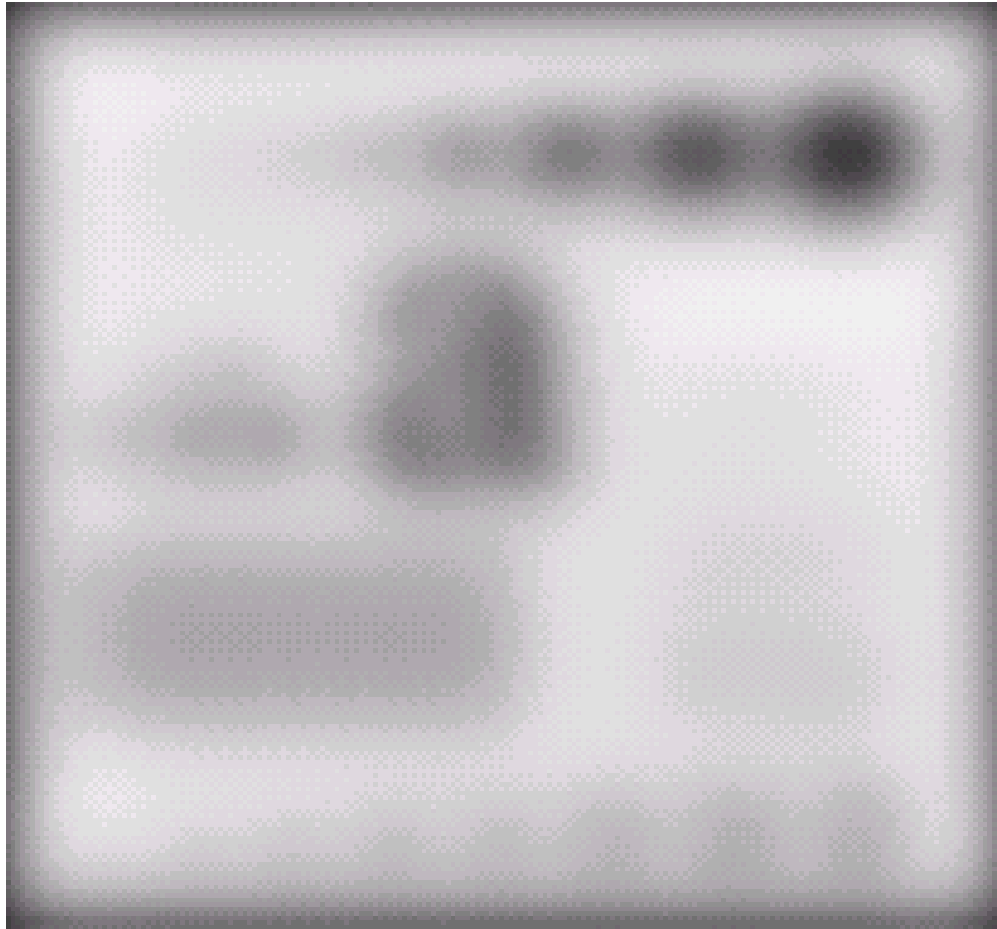
Result of filtering with Butterworth filter of order 2 and cutoff radius 30

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



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

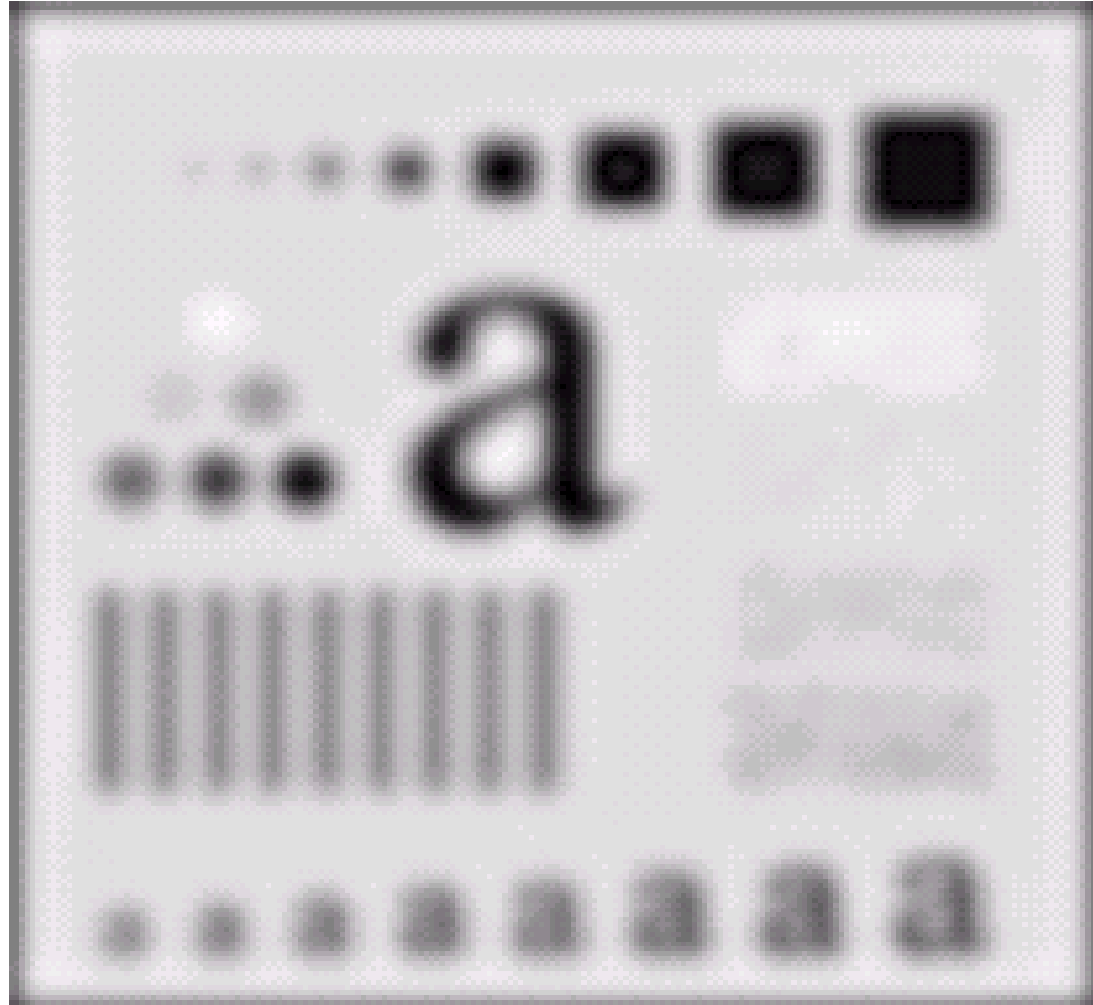
Butterworth Lowpass Filter (cont...)



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

Butterworth Lowpass Filter (cont...)

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



Comparison of Ideal Lowpass and Butterworth Lowpass filters



ILPF

Butterworth filter
of $n=2$ and radii of
5, 15, 30, 80 and
230

No ringing in this
case (with $n=2$)
ringing can be seen
in filters of higher
order

a b
c d
e f

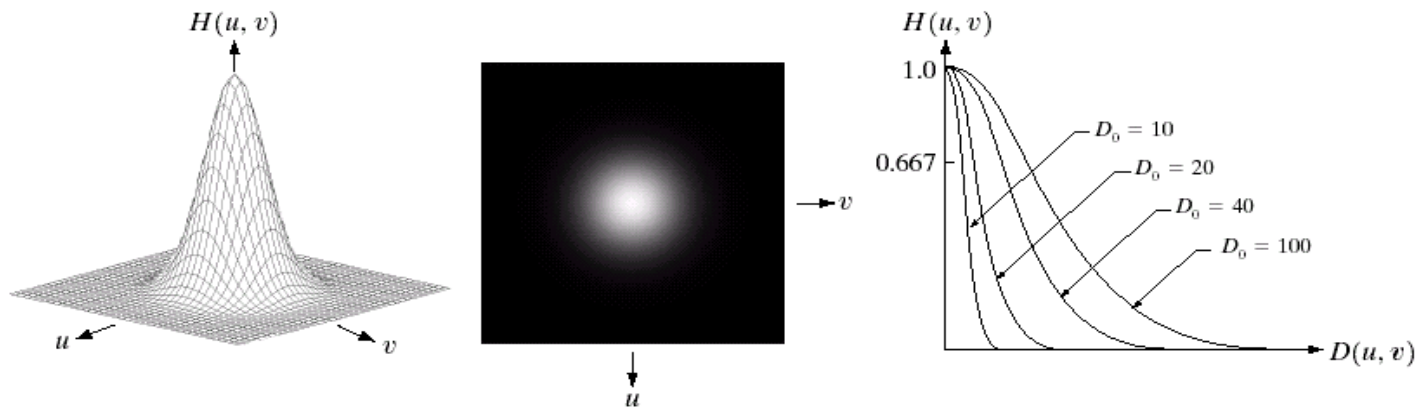
FIGURE 4.15 (a) Original image. (b)–(f) Results of filtering with BLPFs of order 2, with cutoff frequencies at radii of 5, 15, 30, 80, and 230, as shown in Fig. 4.11(b). Compare with Fig. 4.12.

Gaussian Lowpass Filters

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

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

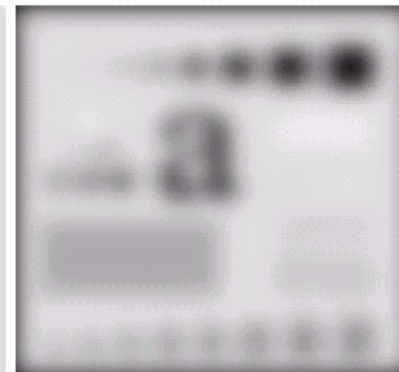
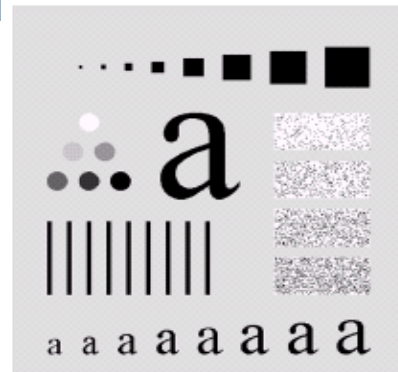
D_0 is the cutoff frequency



Gaussian Lowpass Filters (cont...)

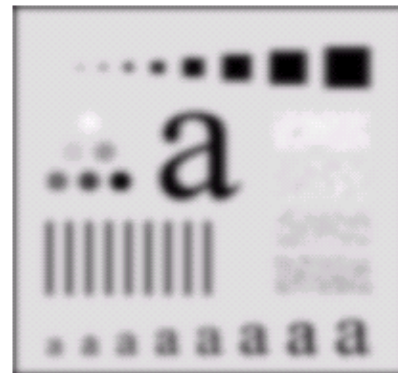
- D_0 is set to 5 different radii of
- 5, 15, 30, 80 and 230
- Smooth transition in
- blurring as a function of
- increasing cutoff frequency

Original
image



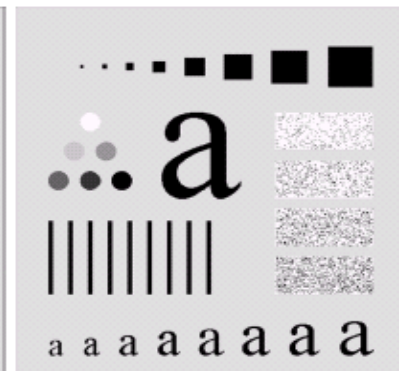
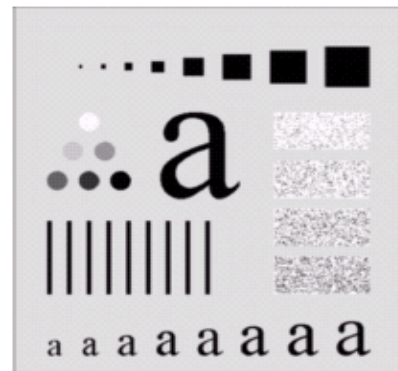
Result of filtering
with Gaussian filter
with cutoff radius 5

Result of filtering
with Gaussian
filter with cutoff
radius 15



Result of filtering
with Gaussian filter
with cutoff radius 30

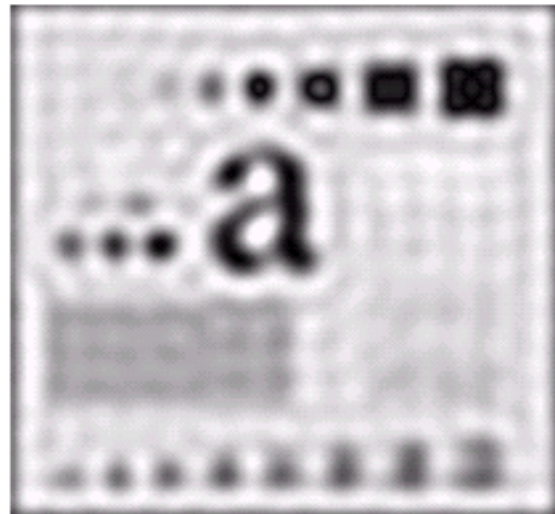
Result of filtering
with Gaussian
filter with cutoff
radius 85



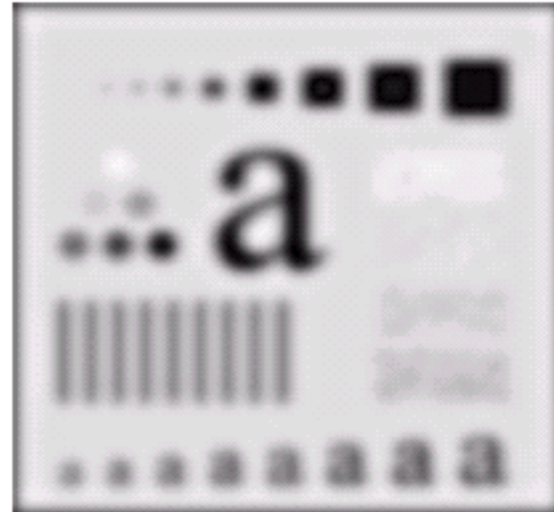
Result of filtering
with Gaussian filter
with cutoff radius
230

Lowpass Filters Compared

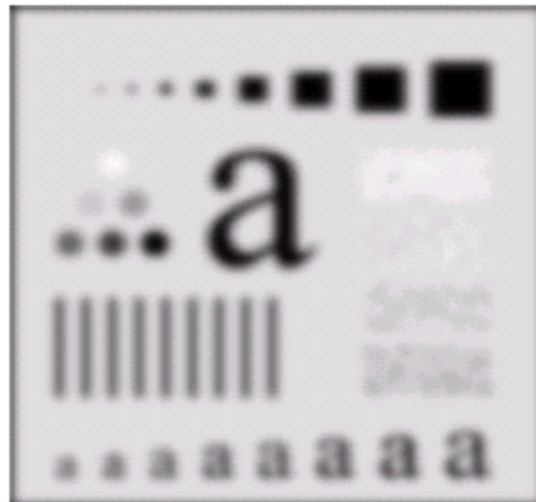
Result of filtering
with ideal low pass
filter of radius 15



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



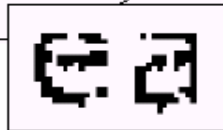
Result of filtering
with Gaussian
filter with cutoff
radius 15



Lowpass Filtering Examples

- A low pass Gaussian filter is used to connect broken text
 - To improve readability (for instance in a character recognition system), typically the solution is to blur the broken corners

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.

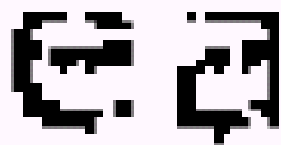


Historically, certain computer programs were written using only two digits rather than four to define the applicable year. Accordingly, the company's software may recognize a date using "00" as 1900 rather than the year 2000.

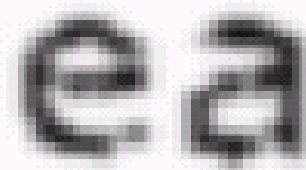


Lowpass Filtering Examples

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.



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.



Lowpass Filtering Examples (cont...)

Different lowpass Gaussian filters used to remove blemishes in a photograph

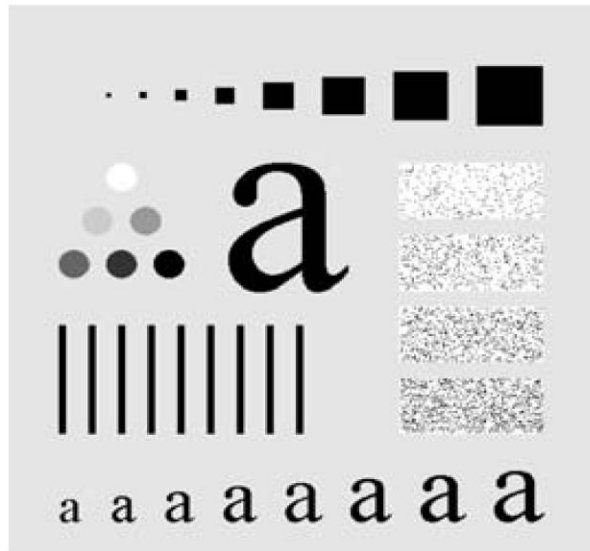


Lowpass Filtering Examples (cont...)

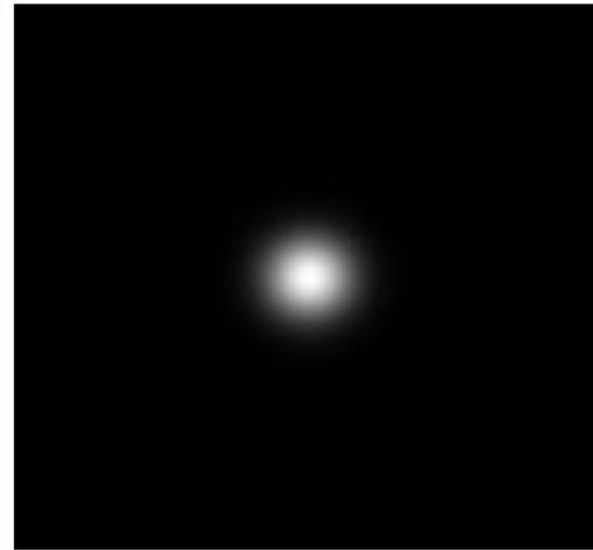


Lowpass Filtering Examples (cont...)

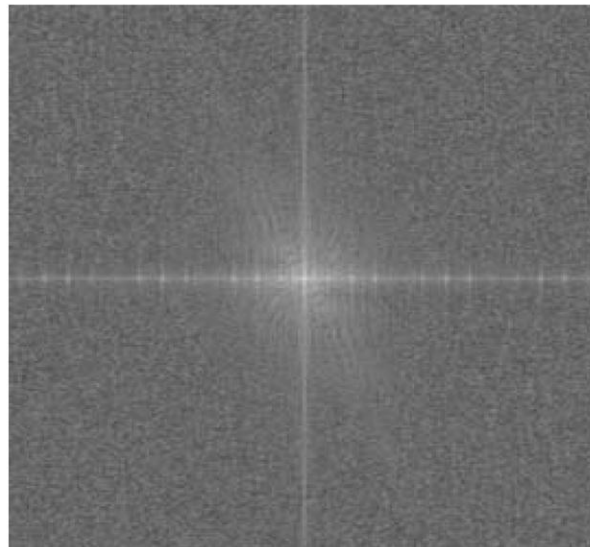
Original
image



Gaussian lowpass
filter



Spectrum of
original image



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, so:

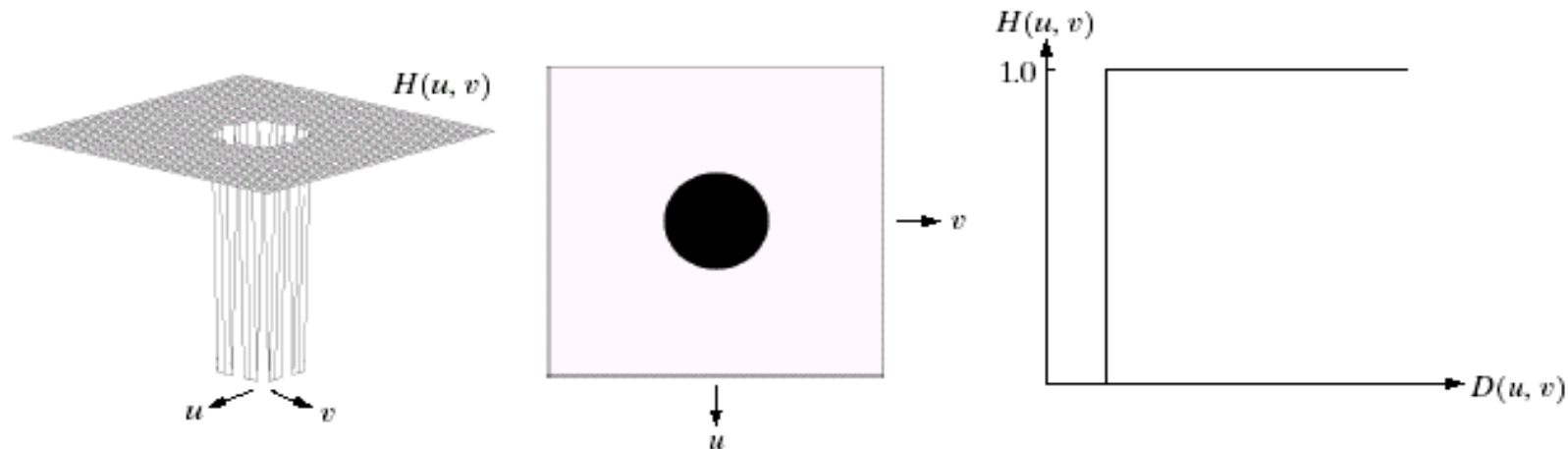
$$H_{hp}(u, v) = 1 - H_{lp}(u, v)$$

Ideal High Pass Filters

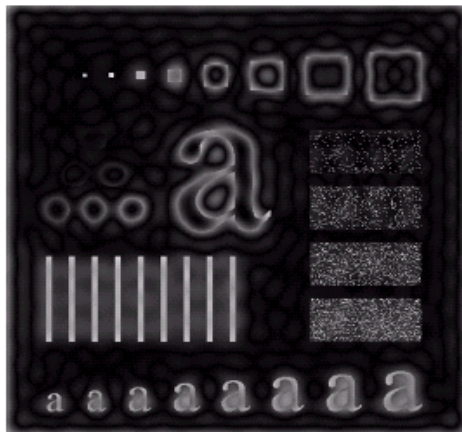
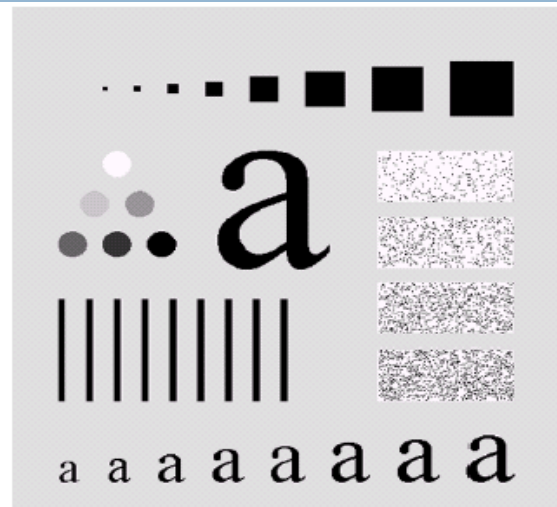
The ideal high pass filter is given as:

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

where D_0 is the cut off distance as before



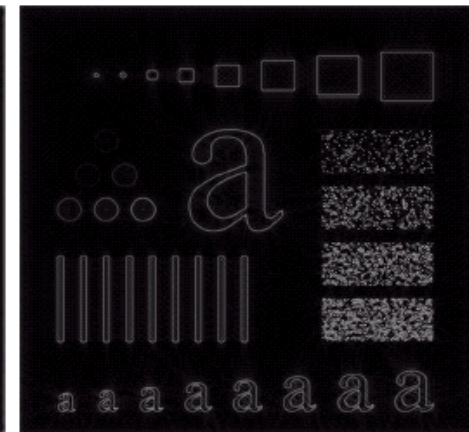
Ideal High Pass Filters (cont...)



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



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



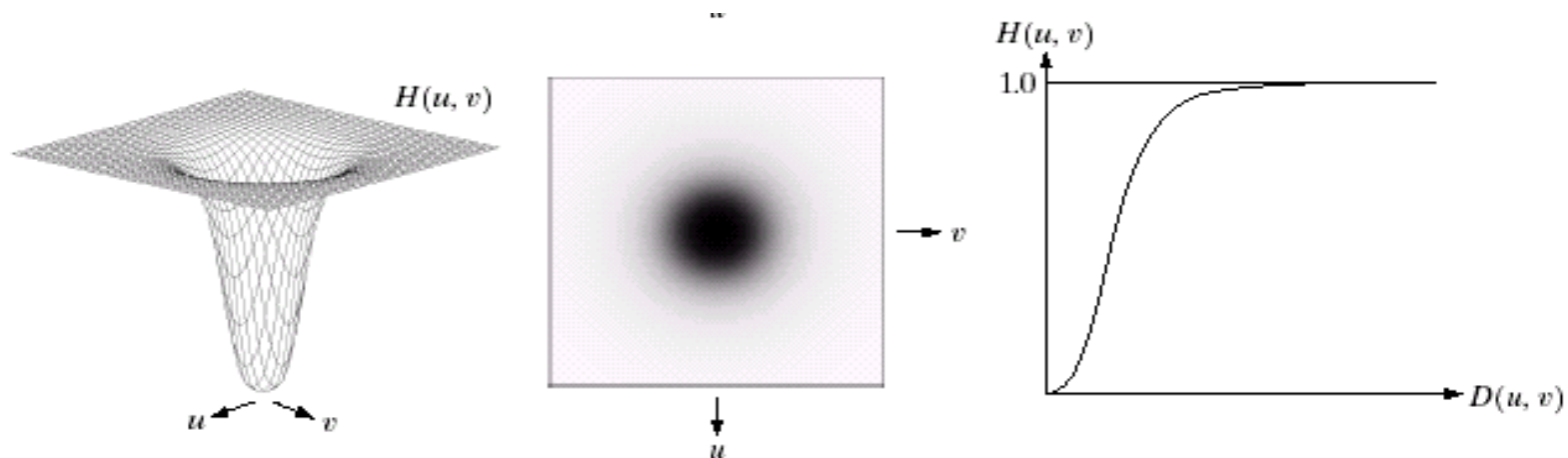
Results of ideal
high pass filtering
with $D_0 = 80$

Butterworth High Pass Filters

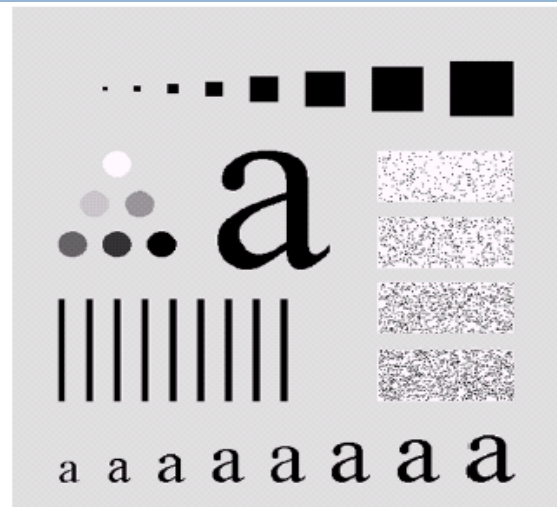
The Butterworth high pass filter is given as:

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

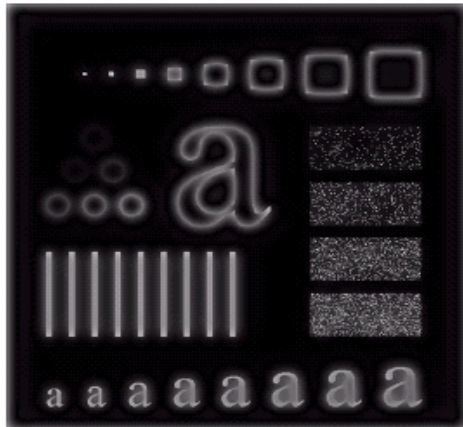
where n is the order and D_0 is the cut off distance as before



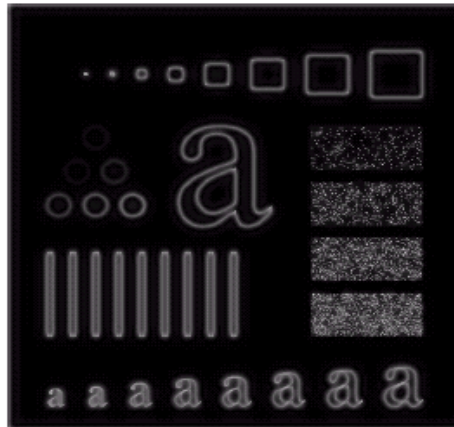
Butterworth High Pass Filters (cont...)



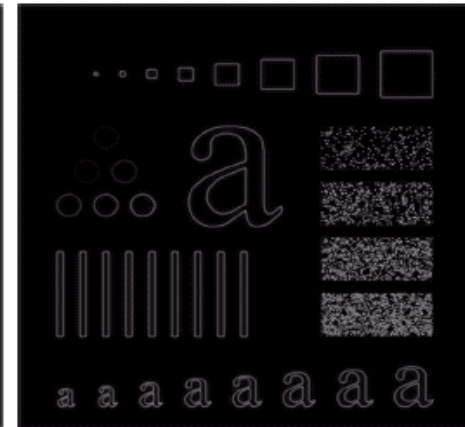
Results of
Butterworth
high pass
filtering of
order 2 with
 $D_0 = 15$



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



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

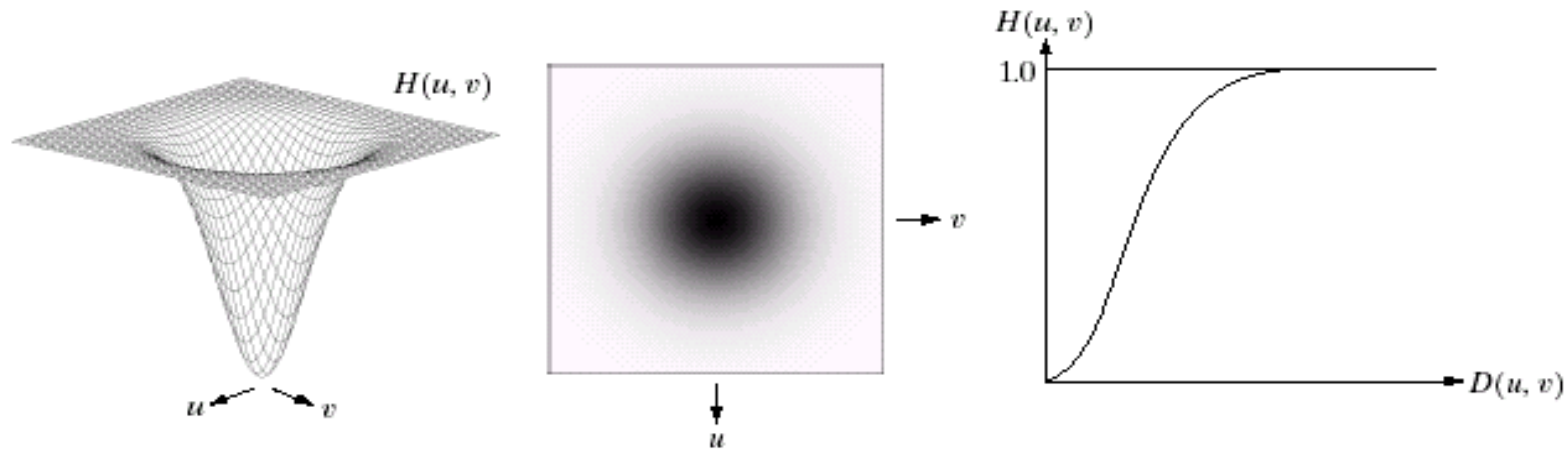


Gaussian High Pass Filters

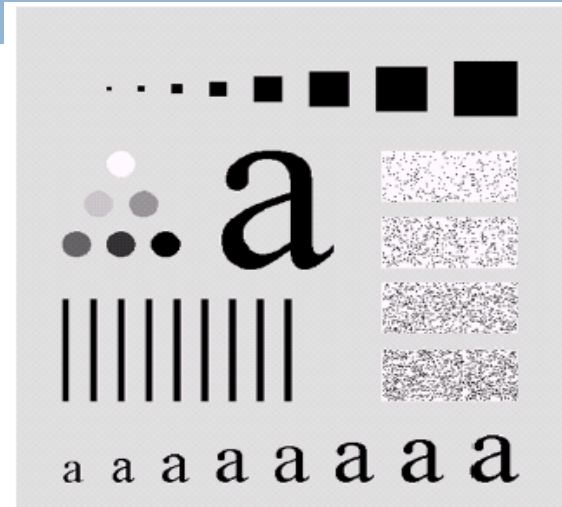
The Gaussian high pass filter is given as:

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

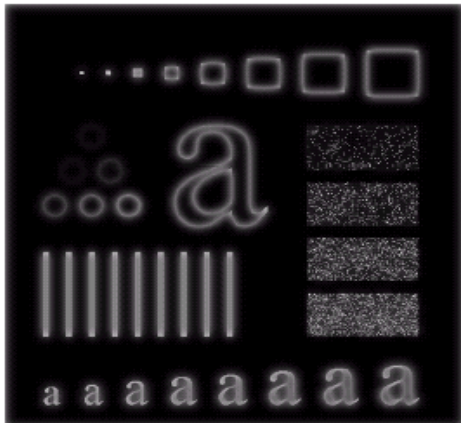
where D_0 is the cut off distance as before



Gaussian High Pass Filters (cont...)



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



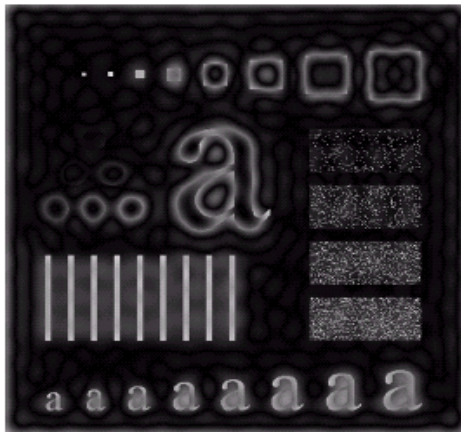
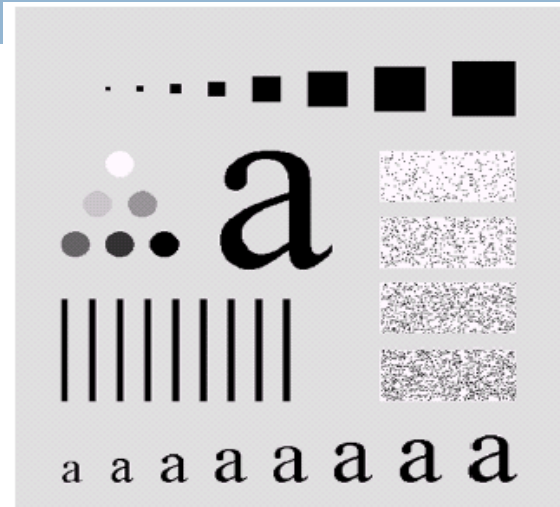
Results of Gaussian high pass
filtering with $D_0 = 30$



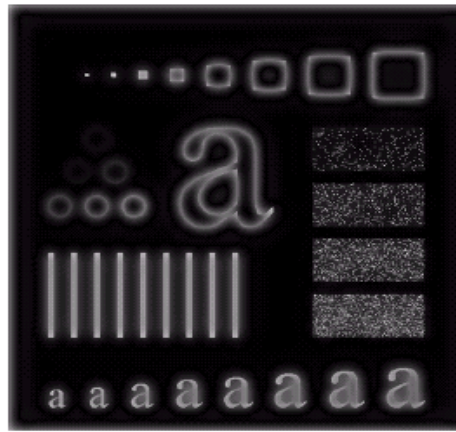
Results of
Gaussian
high pass
filtering with
 $D_0 = 80$



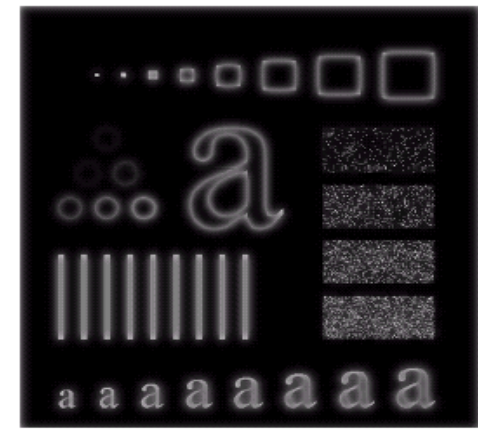
Highpass Filter Comparison



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

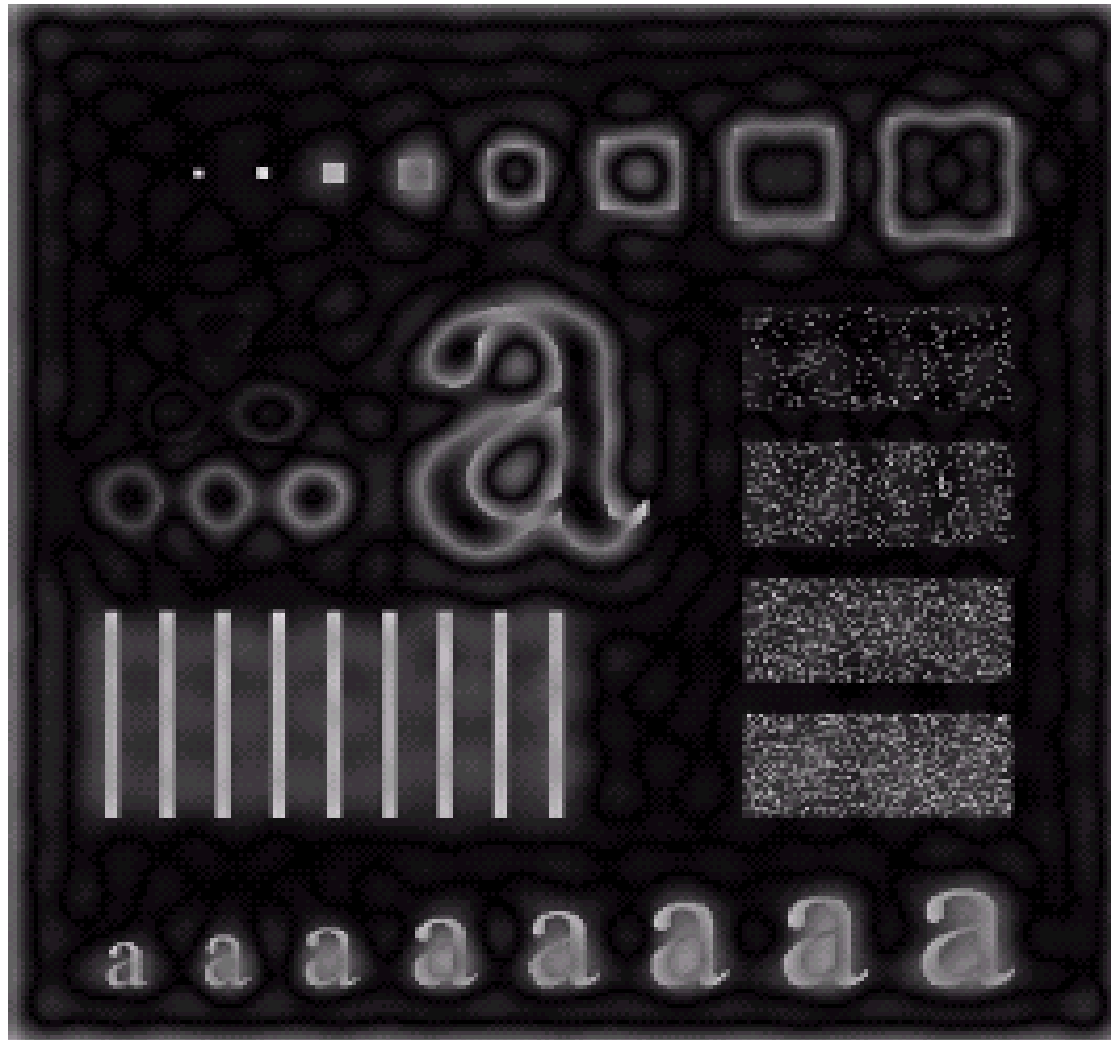


Results of Butterworth
high pass filtering of order
2 with $D_0 = 15$



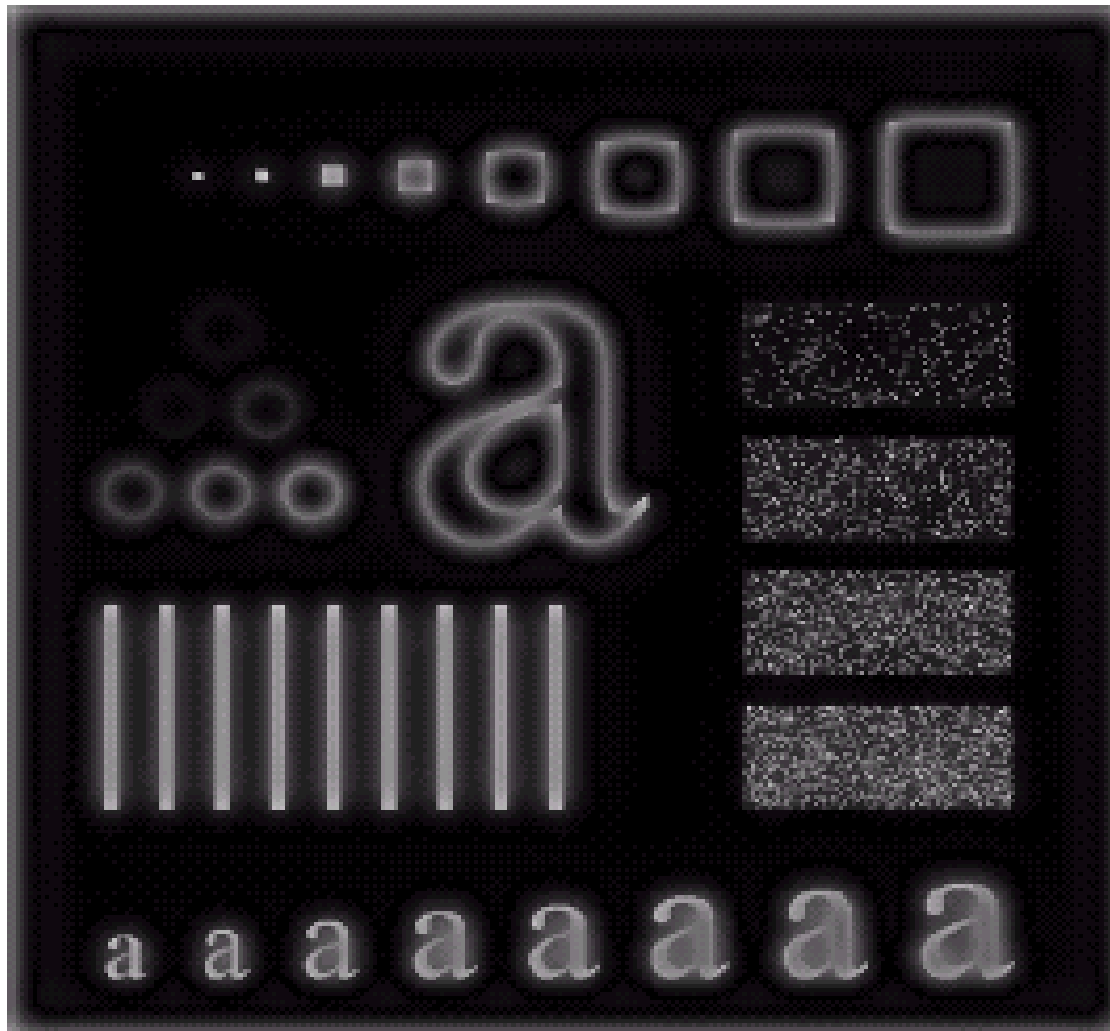
Results of Gaussian
high pass filtering with
 $D_0 = 15$

Highpass Filter Comparison



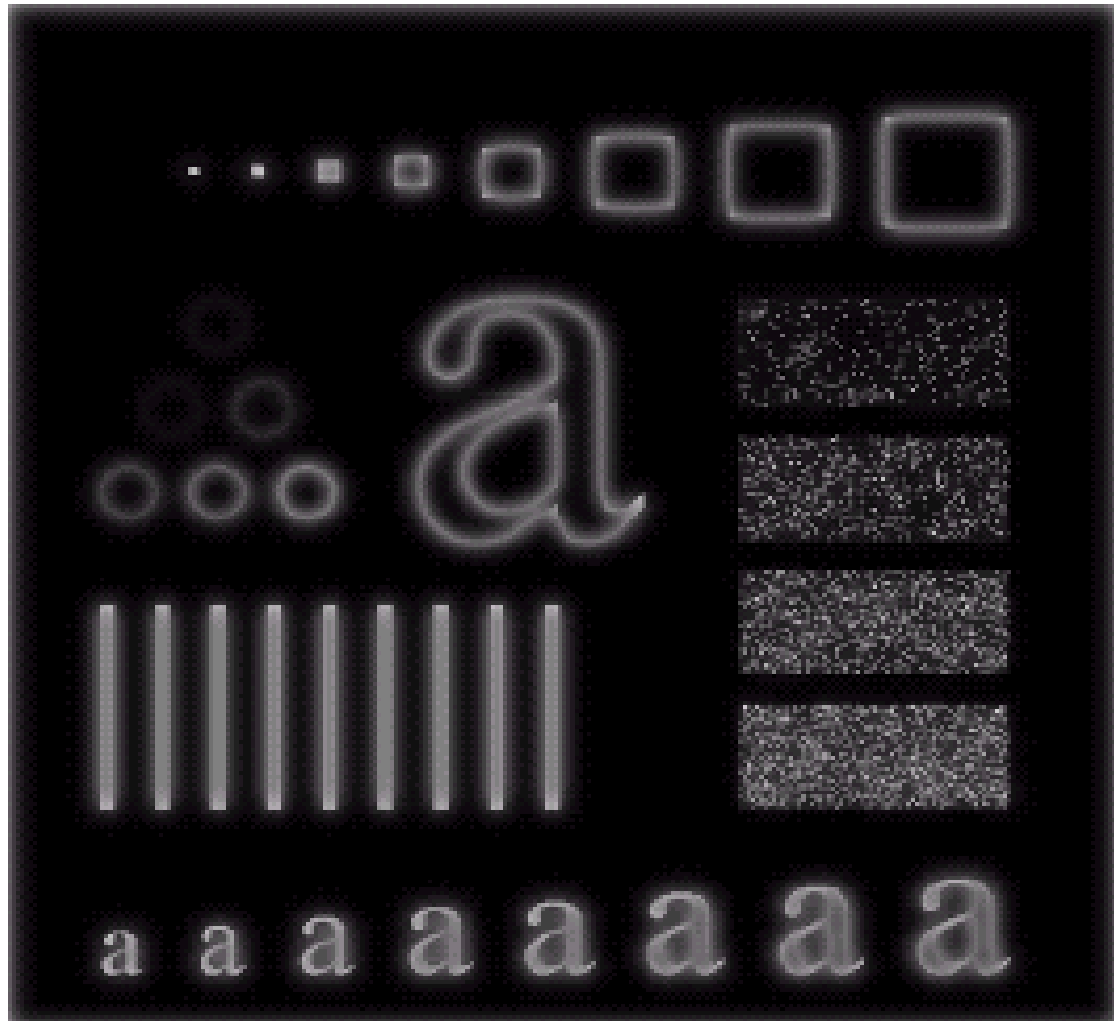
Results of ideal
high pass filtering
with $D_0 = 15$

Highpass Filter Comparison



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

Highpass Filter Comparison



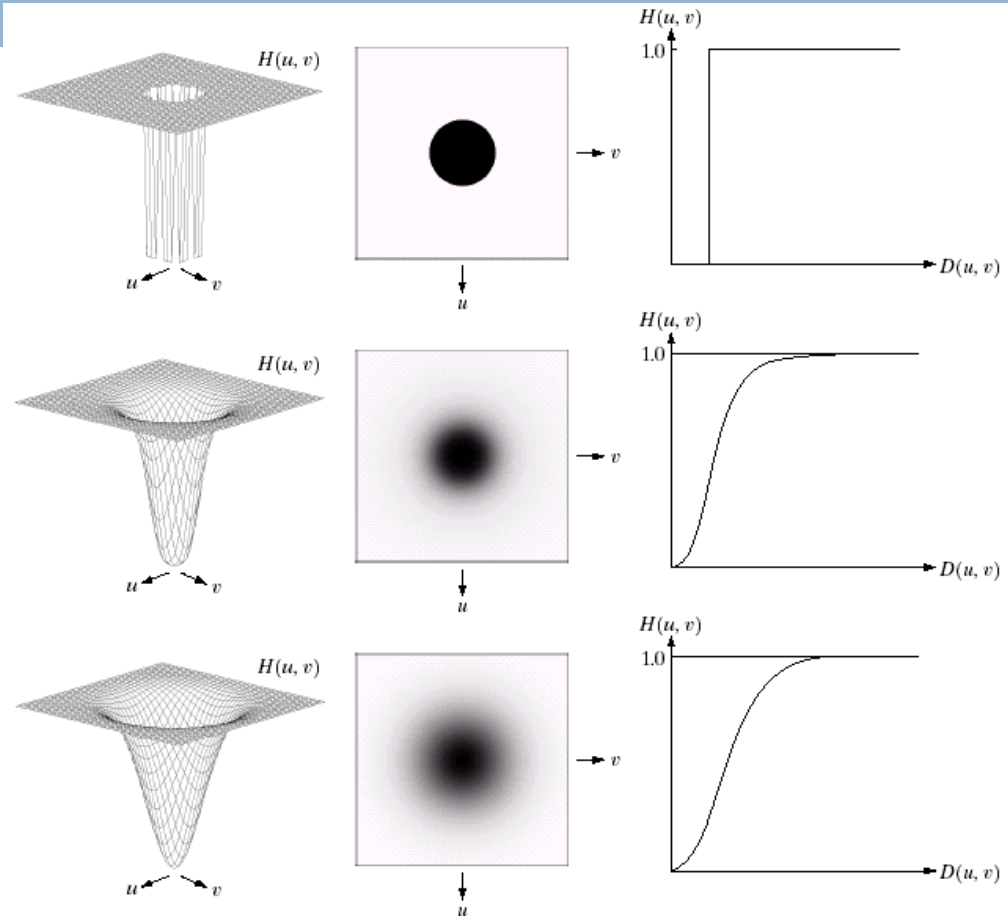
Results of Gaussian
high pass filtering with
 $D_0 = 15$

HIGHPASS FILTERS: ILLUSTRATION

Ideal

Butterworth

Gaussian



a	b	c
d	e	f
g	h	i

FIGURE 4.22 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.

HIGHPASS FILTERS: ILLUSTRATION

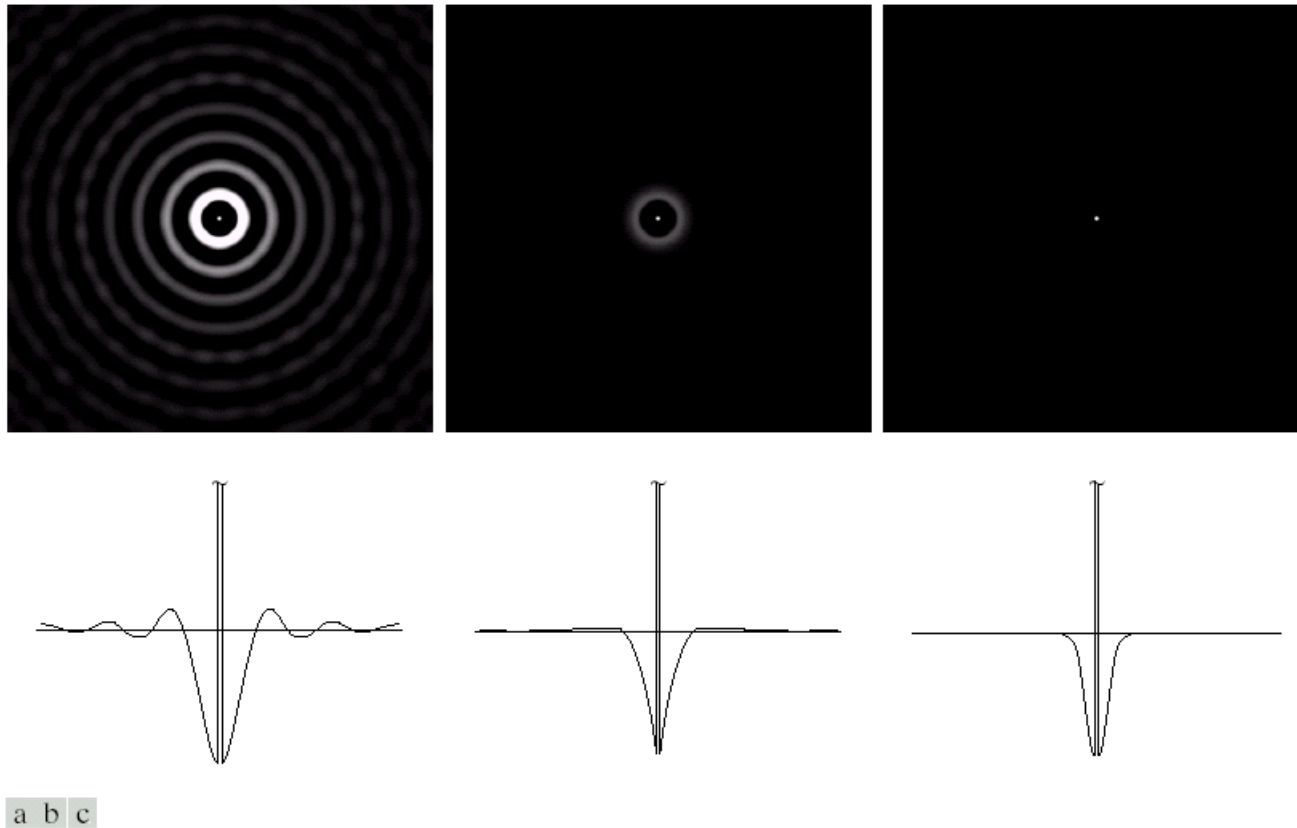


FIGURE 4.23 Spatial representations of typical (a) ideal, (b) Butterworth, and (c) Gaussian frequency domain highpass filters, and corresponding gray-level profiles.

HIGH PASS FILTER

- Ideal High Pass

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

where $D(u,v)$ is the distance from $(M/2, N/2)$

- Butterworth Highpass

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

- Gaussian Highpass

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

Highpass Filtering Example

Original image



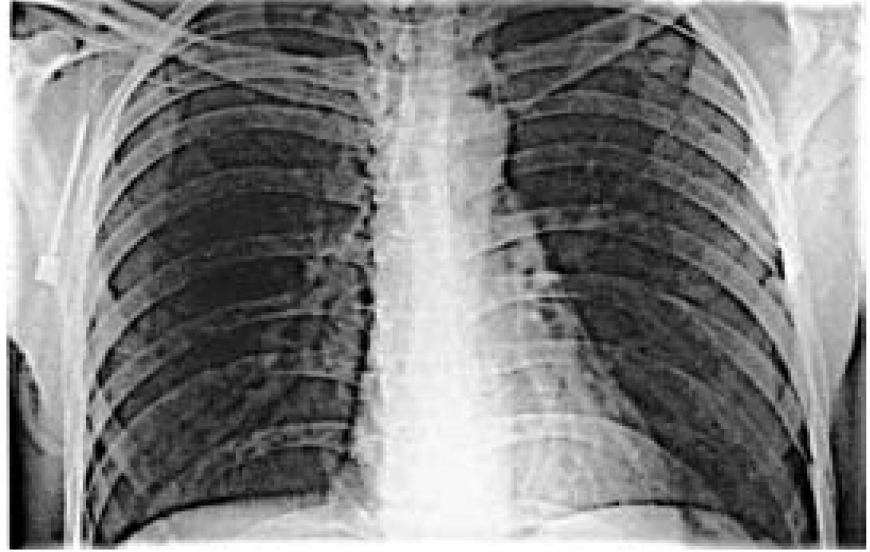
High frequency
emphasis result



Highpass filtering result

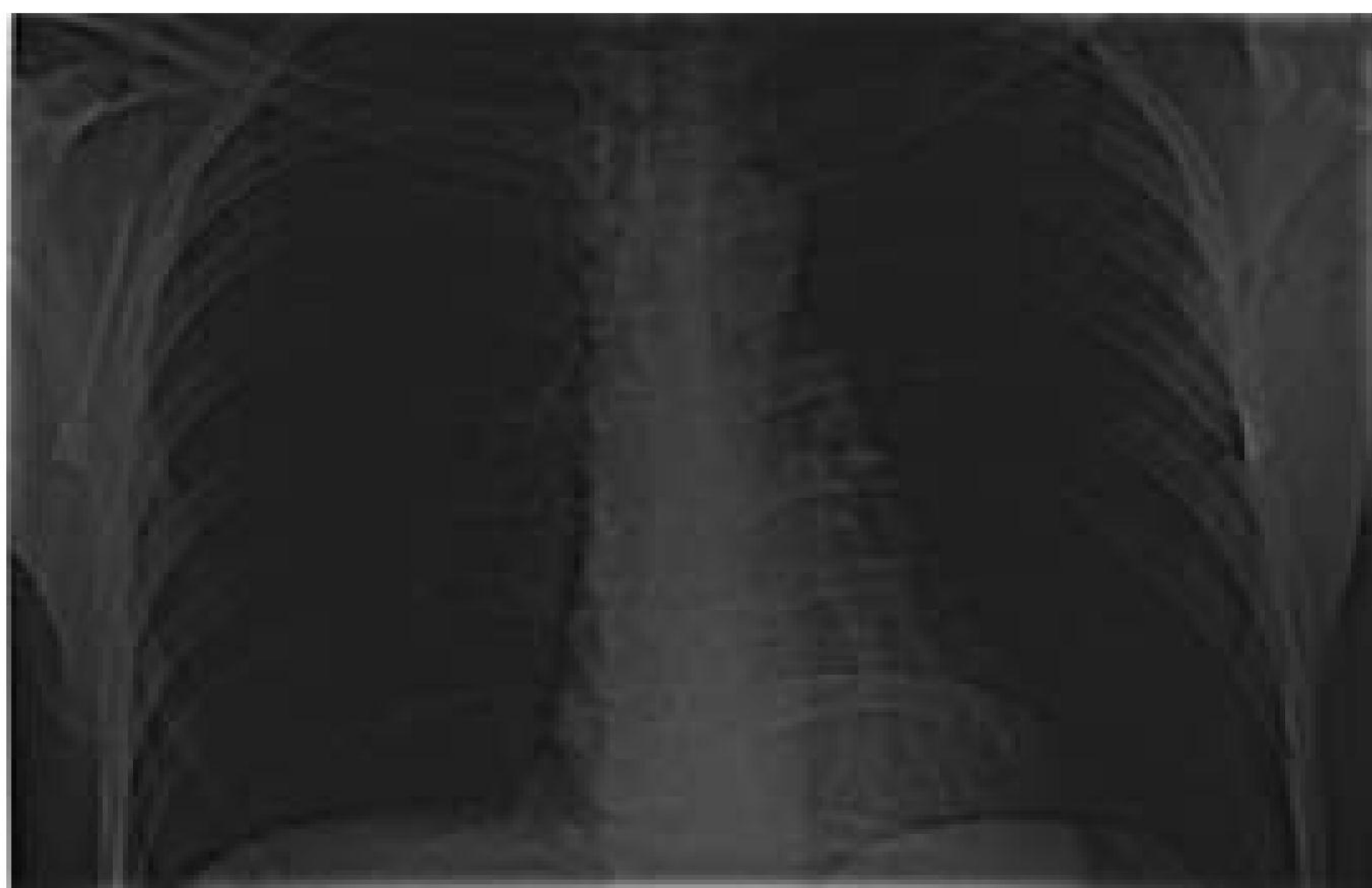


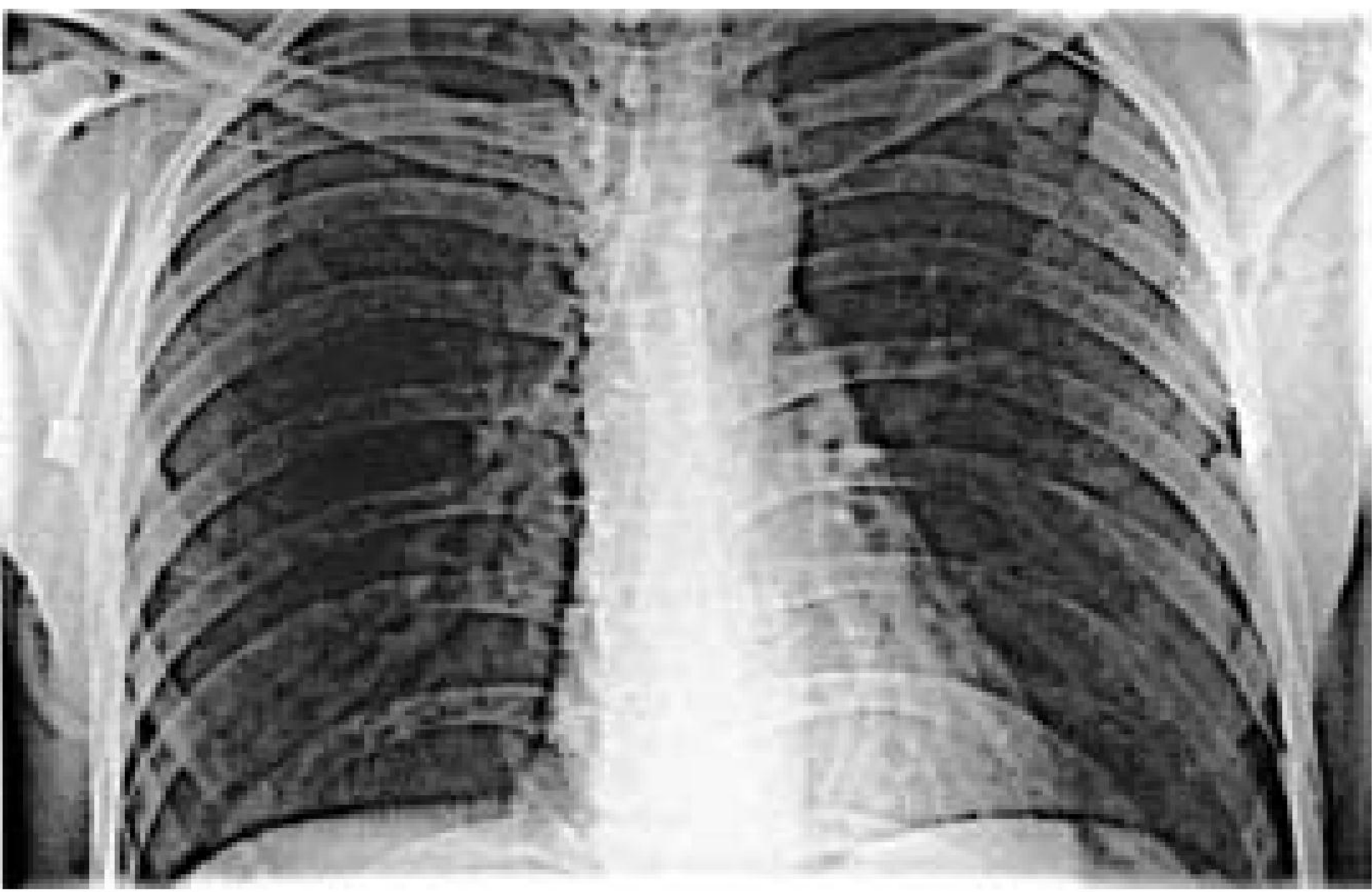
After histogram
equalisation





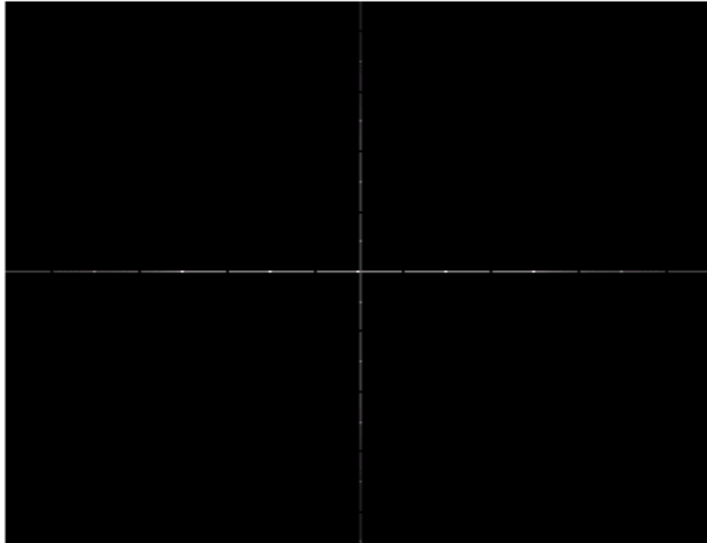




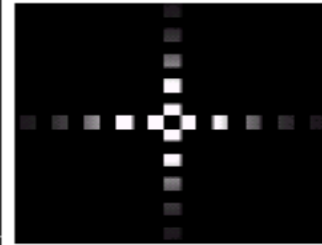
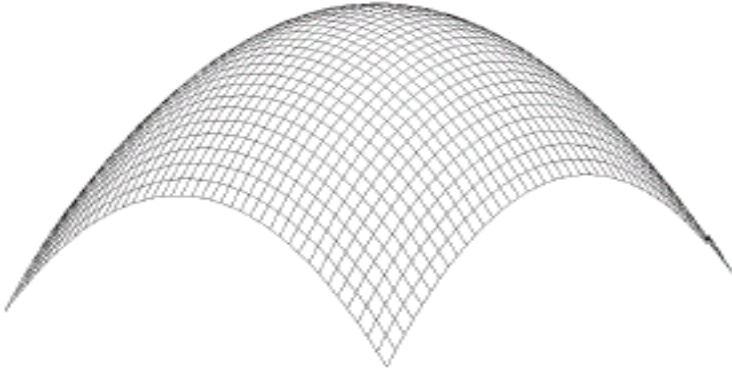


Laplacian In The Frequency Domain

Inverse DFT of
Laplacian in the
frequency domain



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

2-D image of Laplacian
in the frequency
domain

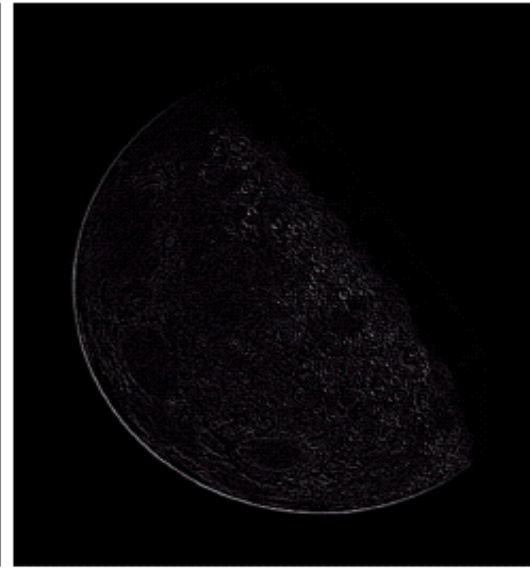


Frequency Domain Laplacian Example

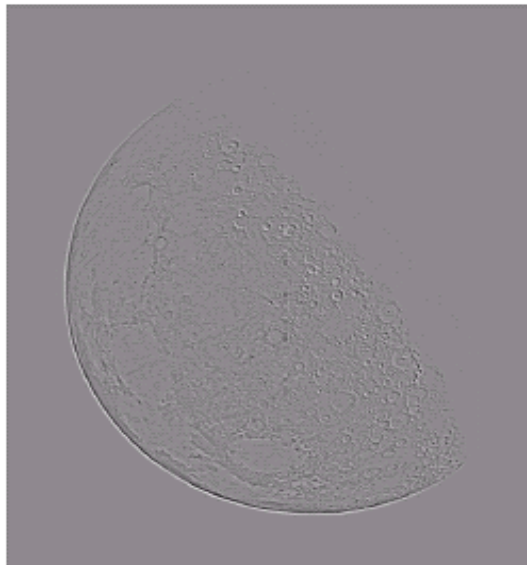
Original
image



Laplacian
filtered
image



Laplacian
image scaled



Enhanced
image



Fast Fourier Transform

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 – especially for large images


Summary

In this lecture we examined image enhancement in the frequency domain

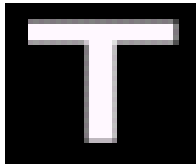
- ▣ The Fourier series & the Fourier transform
- ▣ Image Processing in the frequency domain
 - Image smoothing
 - Image sharpening
- ▣ Fast Fourier Transform

Next time we will begin to examine image restoration using the spatial and frequency based techniques we have been looking at

Interesting Application Of Frequency Domain Filtering



UTK



T

Interesting Application Of Frequency Domain Filtering



Questions?

