Linear Filtering – Part II

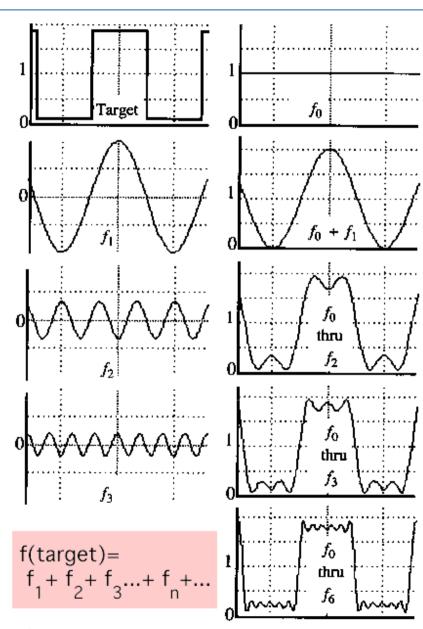
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Fourier theory

- Jean Baptiste Joseph Fourier had a crazy idea: Any periodic function can be written as a weighted sum of sines and cosines of different frequencies (1807).
 - → Fourier series
- Even functions that are not periodic (but whose area under the curve is finite) can be expressed as the integral of sines and cosines multiplied by a weighing function.
 - → Fourier transform

Fourier theory

- The Fourier theory shows how most real functions can be represented in terms of a basis of sinusoids.
- The building block:
 - A $\sin(\omega x + \Phi)$
- Add enough of them to get any signal you want.



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

$$F(u) = \int_{-\infty}^{\infty} f(x) e^{-j2\pi ux} dx.$$

• Given F(u), we can obtain f(x) using the *inverse* Fourier transform

$$f(x) = \int_{-\infty}^{\infty} F(u) \ e^{j2\pi ux} \ du.$$

• The discrete Fourier transform (DFT), F(u), of a discrete function of one variable, f(x), x = 0, 1, 2, ..., M-1, is defined by

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

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

• Given F(u), we can obtain the original function back using the inverse DFT

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

for $x = 0, 1, 2, \dots, M - 1$.

- These formulas can be extended for functions of two variables.
- Fourier transform:

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

Inverse Fourier transform:

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

Discrete Fourier transform:

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

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

Inverse discrete Fourier transform:

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

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

 \bullet F(u,v) can also be expressed in polar coordinates as

$$F(u,v) = |F(u,v)| e^{j\phi(u,v)}$$

where

$$|F(u,v)| = \left(\Re^2 \{F(u,v)\} + \Im^2 \{F(u,v)\}\right)^{1/2}$$

is called the magnitude or spectrum of the Fourier transform, and

$$\phi(u,v) = \tan^{-1}\left(\frac{\Im\{F(u,v)\}}{\Re\{F(u,v)\}}\right)$$

is called the *phase angle* or *phase spectrum*.

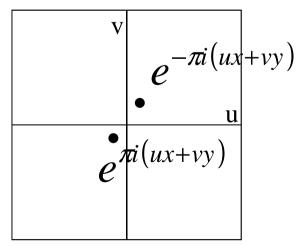
• $\Re\{F(u,v)\}$ and $\Im\{F(u,v)\}$ are the real and imaginary parts of F(u,v), respectively.

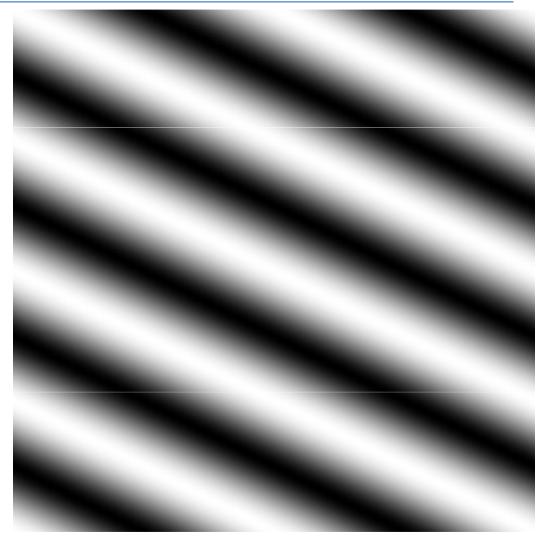
- The spectrum need not be interpreted as an image, but rather as a 2D display of the power in the original image versus the frequency components u and v.
- The value F(0,0) is the average of f(x,y).
- Fourier transform is conjugate symmetric $(F(u, v) = F^*(-u, -v))$ and its spectrum is symmetric about the origin (|F(u, v)| = |F(-u, -v)|) (when f(x, y) is real).
- Usually the input image function is multiplied by $(-1)^{x+y}$ prior to computing the Fourier transform so that

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

The origin of the Fourier transform is located at u=M/2 and v=N/2.

To get some sense of what basis elements look like, we plot a basis element --- or rather, its real part --- as a function of x,y for some fixed u, v. We get a function that is constant when (ux+vy) is constant. The magnitude of the vector (u, v) gives a frequency, and its direction gives an orientation. The function is a sinusoid with this frequency along the direction, and constant perpendicular to the direction.

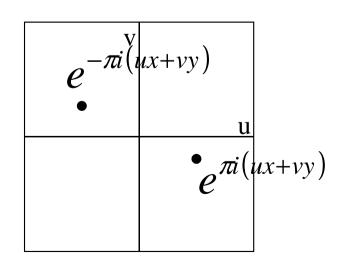




Adapted from Antonio Torralba

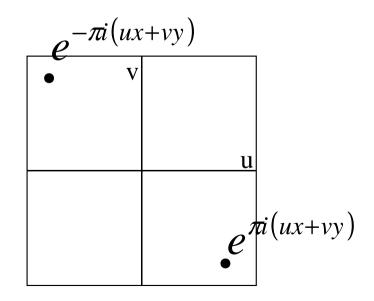
10

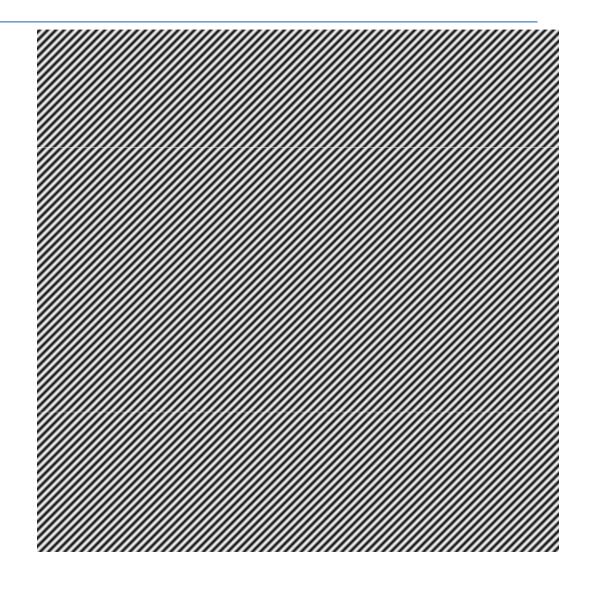
Here u and v are larger than in the previous slide.



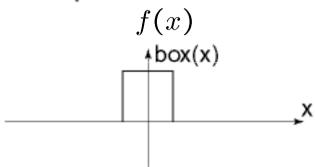


And larger still...

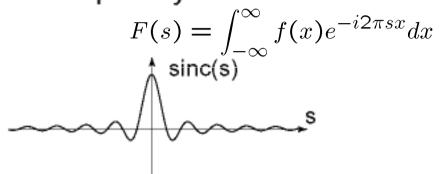


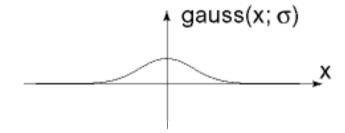


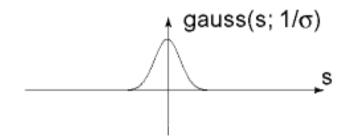
Spatial domain

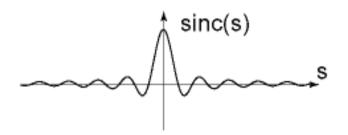


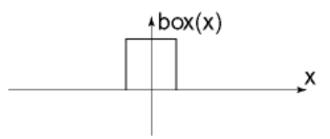
Frequency domain



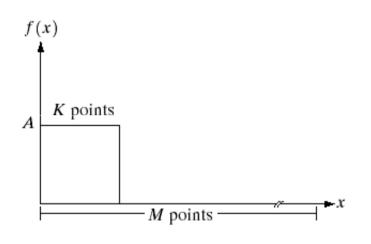


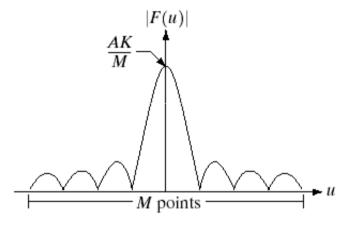






Adapted from Alexei Efros, CMU





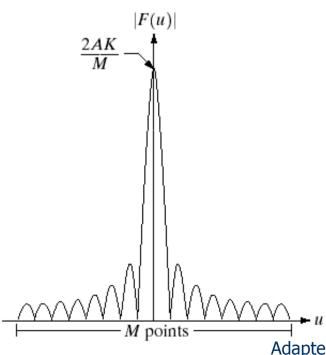
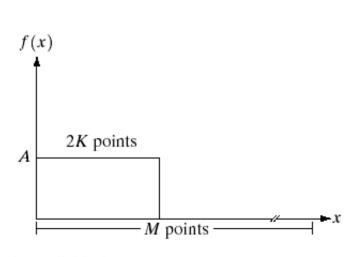
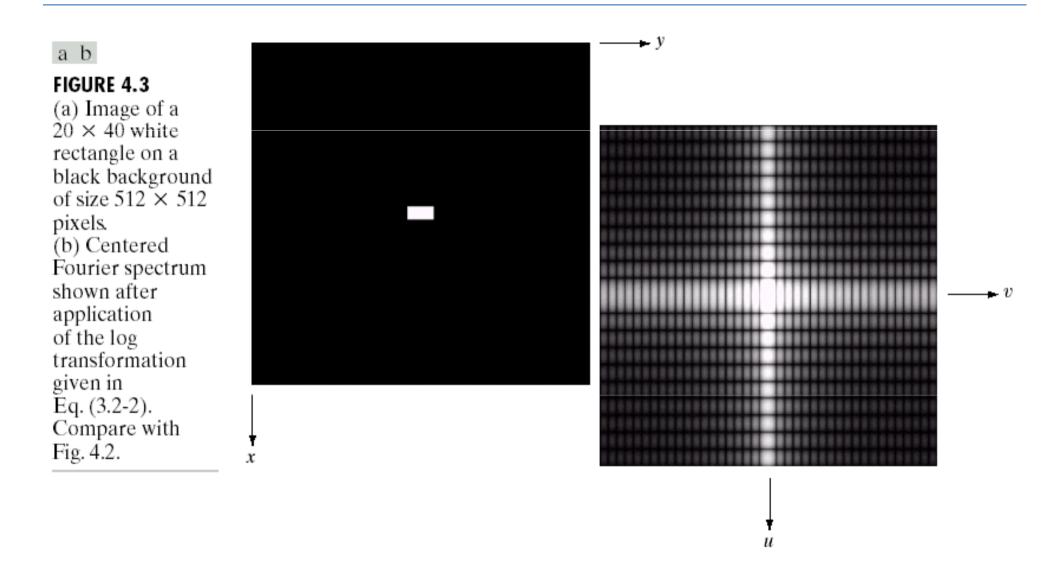




figure 4.2 (a) A discrete function of *M* points, and (b) its Fourier spectrum. (c) A discrete function with twice the number of nonzero points, and (d) its Fourier spectrum.



Adapted from Gonzales and Woods



• The *power spectrum* is defined as the square of the Fourier spectrum:

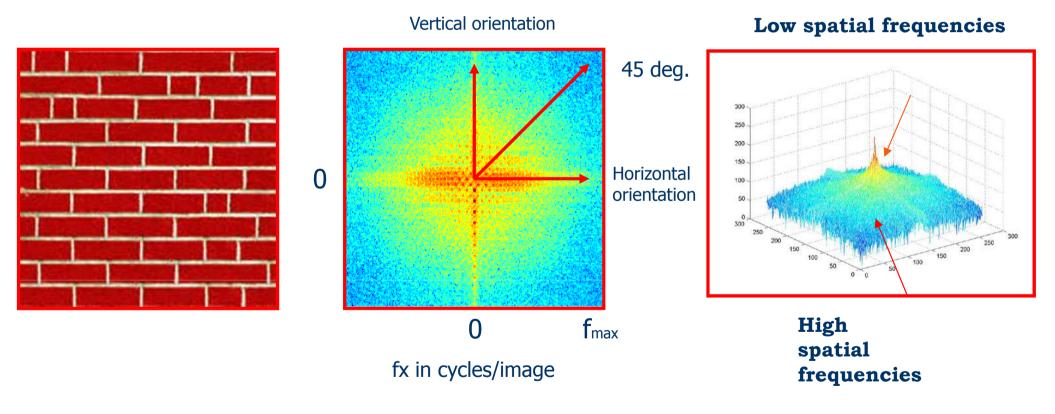
$$P(u,v) = |F(u,v)|^2$$

= $\Re^2 \{ F(u,v) \} + \Im^2 \{ F(u,v) \}.$

- The radial distribution of values in the Fourier spectrum of an image is sensitive to texture coarseness in that image.
 - A coarse texture will have high values concentrated near the origin of the spectrum.
 - A fine texture will cause the values to be spread out.

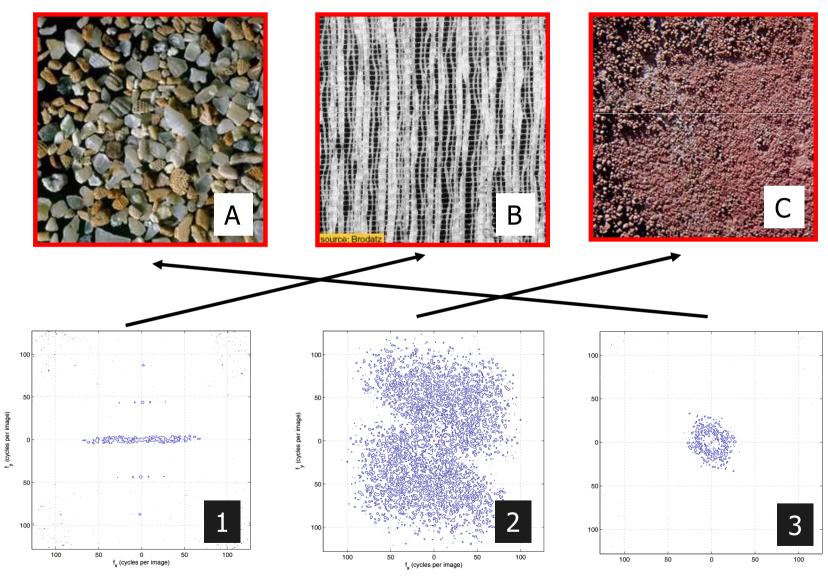
- The angular distribution of values in the spectrum is sensitive to the directionality of the texture in the image.
 - A texture with many edges in a given direction θ will have high values of the spectrum concentrated around the perpendicular direction $\theta + \pi/2$.
 - For a non-directional texture, the spectrum is also non-directional.
- We will come back to this when we talk about texture.

How to interpret a Fourier spectrum:



Log power spectrum

Adapted from Antonio Torralba



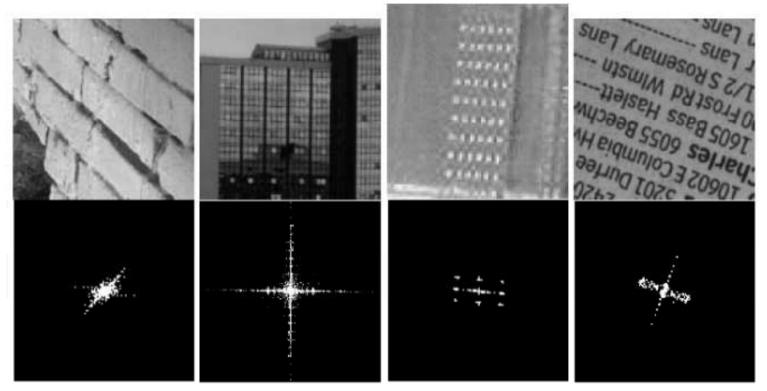
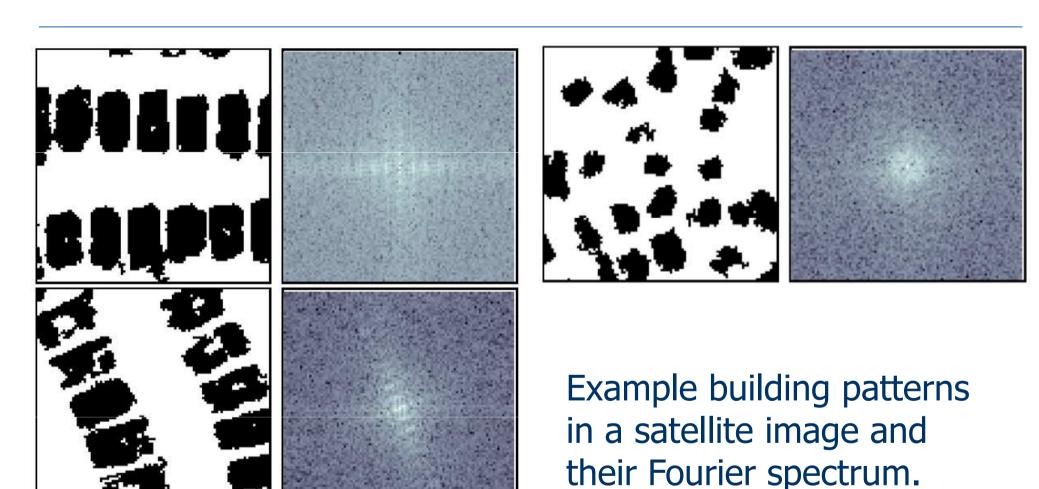


Figure 5.42: Four images (above) and their power spectrums (below). The power spectrum of the brick texture shows energy in many sinusoids of many frequencies, but the dominant direction is perpendicular to the 6 dark seams running about 45 degrees with the X-axis. There is noticable energy at 0 degrees with the X axis, due to the several short vertical seams. The power spectrum of the building shows high frequency energy in waves along the X-direction and the Y-direction. The third image is an aerial image of an orchard: the power spectrum shows the rows and columns of the orchard and also the "diagnonal rows". The far right image, taken from a phone book, shows high frequency power at about 60° with the X-axis, which represents the texture in the lines of text. Energy is spread more broadly in the perpendicular direction also in order to model the characters and their spacing.



Convolution theorem

• The discrete *convolution* of two functions f(x,y) and h(x,y) of size $M\times N$ is defined as

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

- This is equivalent to the *correlation* of f(x,y) with h(x,y) flipped about the origin.
- Convolution theorem:

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

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

where "⇔" indicates a Fourier transform pair.

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Frequency domain filtering

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(1) Fourier transform the image f(x, y) to obtain its frequency rep. F(u, v).
(2) Fourier transform the mask h(x, y) to obtain its frequency rep. H(u, v)
(3) multiply F(u, v) and H(u, v) pointwise to obtain F'(u, v)
(4) apply the inverse Fourier transform to F'(u, v) to obtain the filtered image f'(x, y).
```

Algorithm 3: Filtering image f(x,y) with mask h(x,y) using the Fourier transform

Frequency domain filtering operation

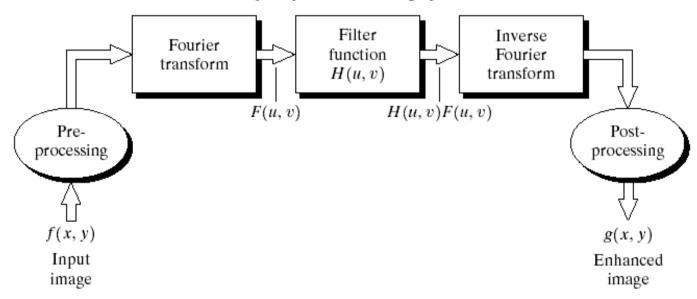
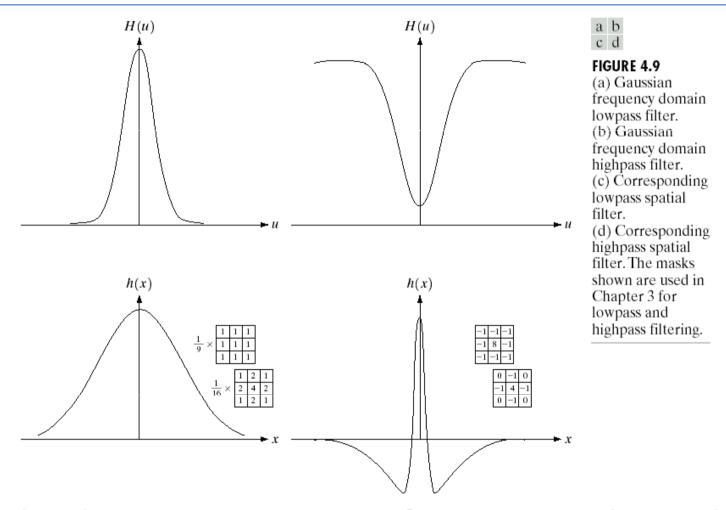


FIGURE 4.5 Basic steps for filtering in the frequency domain.

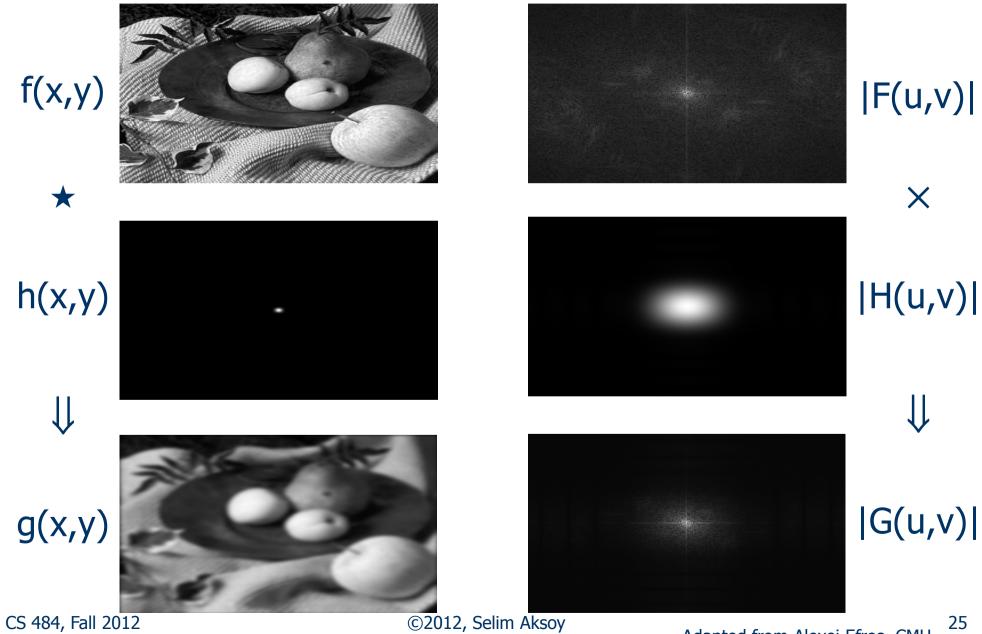
Adapted from Shapiro and Stockman, and Gonzales and Woods

Frequency domain filtering



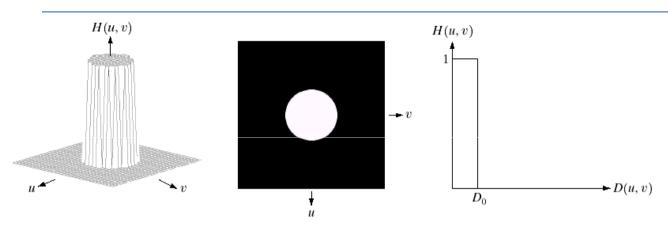
 Since the discrete Fourier transform is periodic, padding is needed in the implementation to avoid aliasing (see section 4.6 in the Gonzales-Woods book for implementation details).

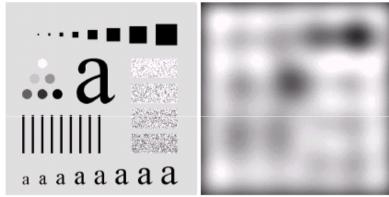
Frequency domain filtering



Adapted from Alexei Efros, CMU

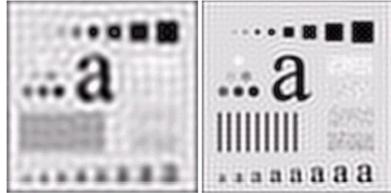
Smoothing frequency domain filters

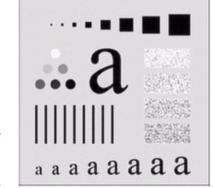


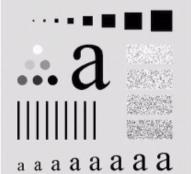


a b c

FIGURE 4.10 (a) Perspective plot of an ideal lowpass filter transfer function. (b) Filter displayed as an image. (c) Filter radial cross section.



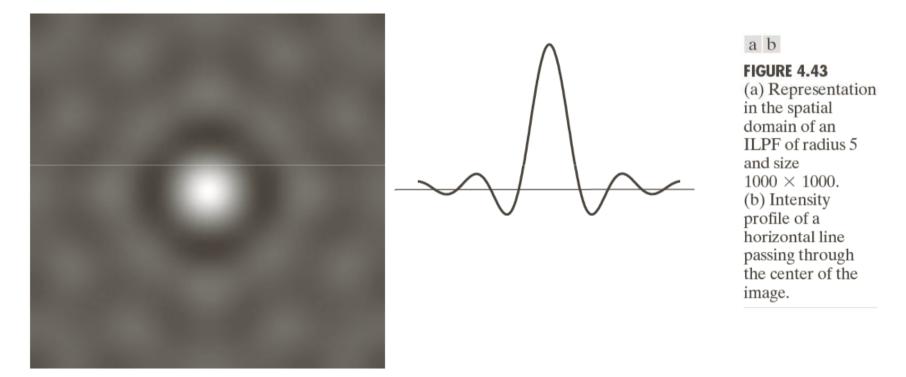




a b **FIGURE 4.12** (a) Original image. (b)–(f) Results of ideal lowpass filtering with cutoff frequencies set at radii values of 5, 15, 30, 80, and 230, as shown in Fig. 4.11(b). The power removed by these filters was 8, 5.4, 3.6, 2, and 0.5% of the total, respectively.

Smoothing frequency domain filters

The blurring and ringing caused by the ideal lowpass filter can be explained using the convolution theorem where the spatial representation of a filter is given below.



Sharpening frequency domain filters

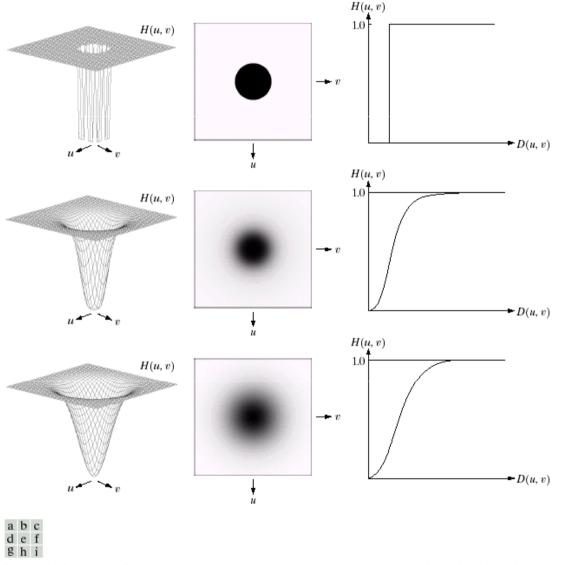
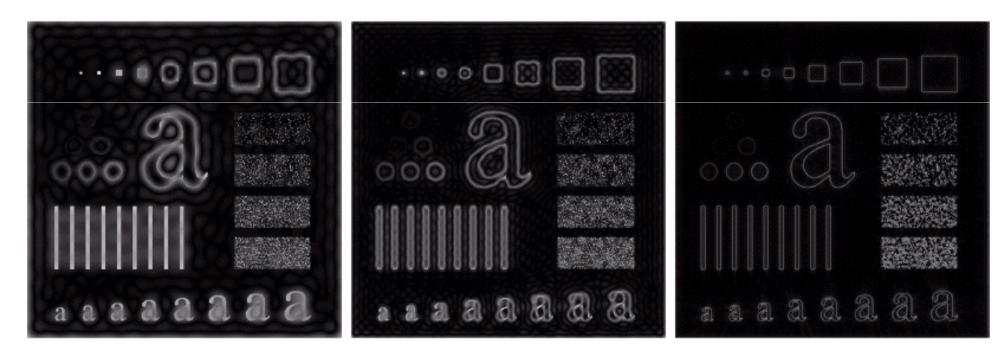


FIGURE 4.22 Top row: Perspective plot, image representation, and cross section of a typical ideal highpass filter. Middle and bottom rows: The same sequence for typical Butterworth and Gaussian highpass filters.

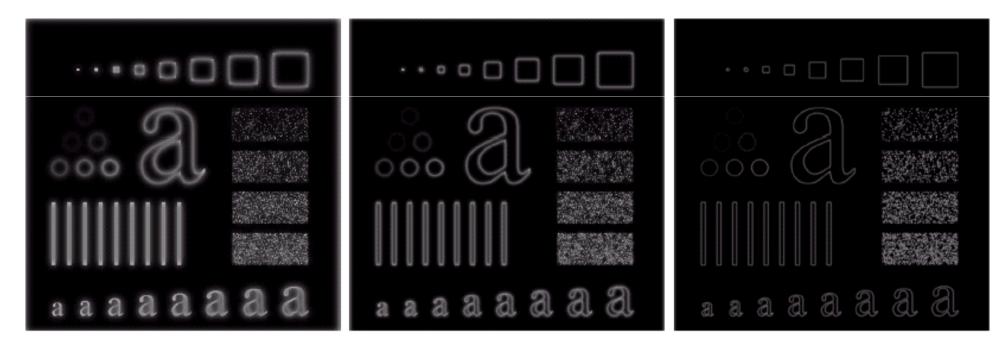
Sharpening frequency domain filters



a b c

FIGURE 4.24 Results of ideal highpass filtering the image in Fig. 4.11(a) with $D_0 = 15$, 30, and 80, respectively. Problems with ringing are quite evident in (a) and (b).

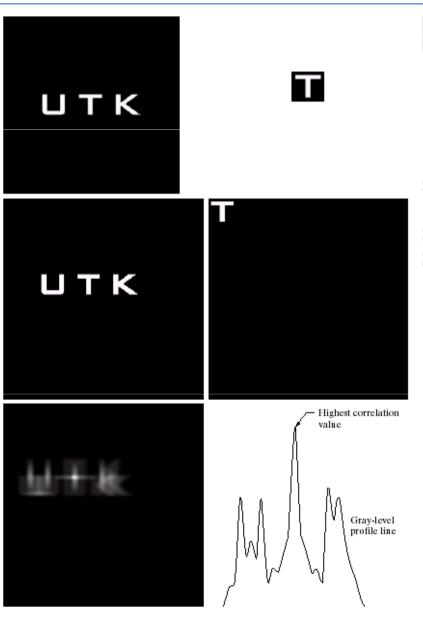
Sharpening frequency domain filters



a b c

FIGURE 4.26 Results of highpass filtering the image of Fig. 4.11(a) using a GHPF of order 2 with $D_0 = 15$, 30, and 80, respectively. Compare with Figs. 4.24 and 4.25.

- Correlation can also be used for matching.
- If we want to determine whether an image f contains a particular object, we let h be that object (also called a template) and compute the correlation between f and h.
- If there is a match, the correlation will be maximum at the location where h finds a correspondence in f.
- Preprocessing such as scaling and alignment is necessary in most practical applications.



a b c d e f

FIGURE 4.41

- (a) Image.
- (b) Template.
- (c) and
- (d) Padded
- images.
- (e) Correlation function displayed
- as an image.
- (f) Horizontal profile line
- through the
- highest value in
- (e), showing the point at which the
- best match took

place.

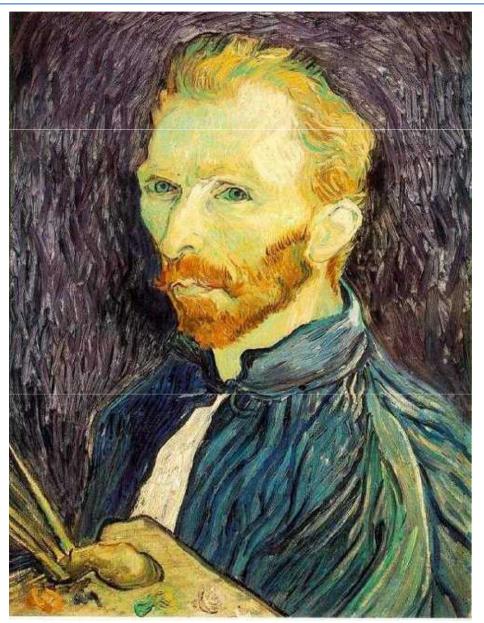
Adapted from Gonzales and Woods



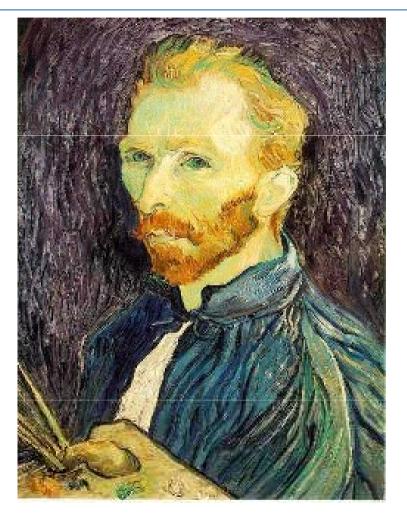
Face detection using template matching: face templates.

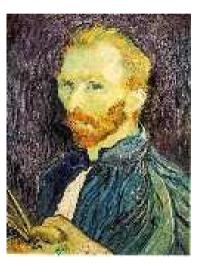


Face detection using template matching: detected faces.



How can we generate a half-sized version of a large image?



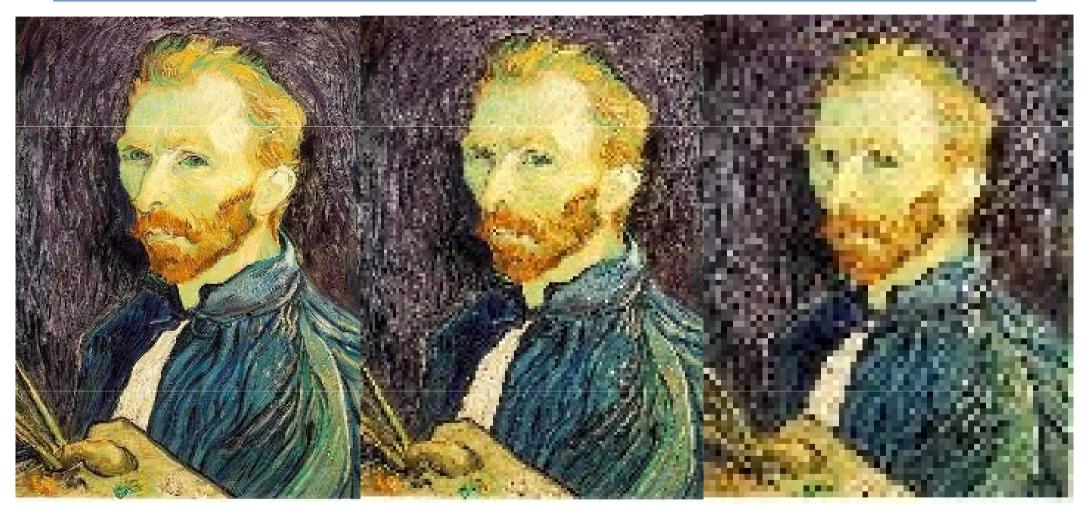




1/4

1/8

Throw away every other row and column to create a 1/2 size image (also called sub-sampling).



1/2

1/4 (2x zoom)

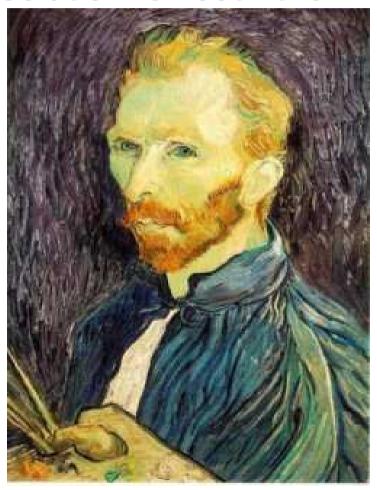
1/8 (4x zoom)

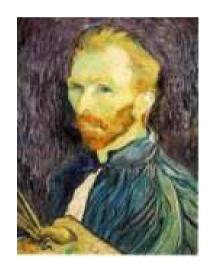
Does this look nice?

Adapted from Steve Seitz, U of Washington

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- We cannot shrink an image by simply taking every k'th pixel.
- Solution: smooth the image, then sub-sample.

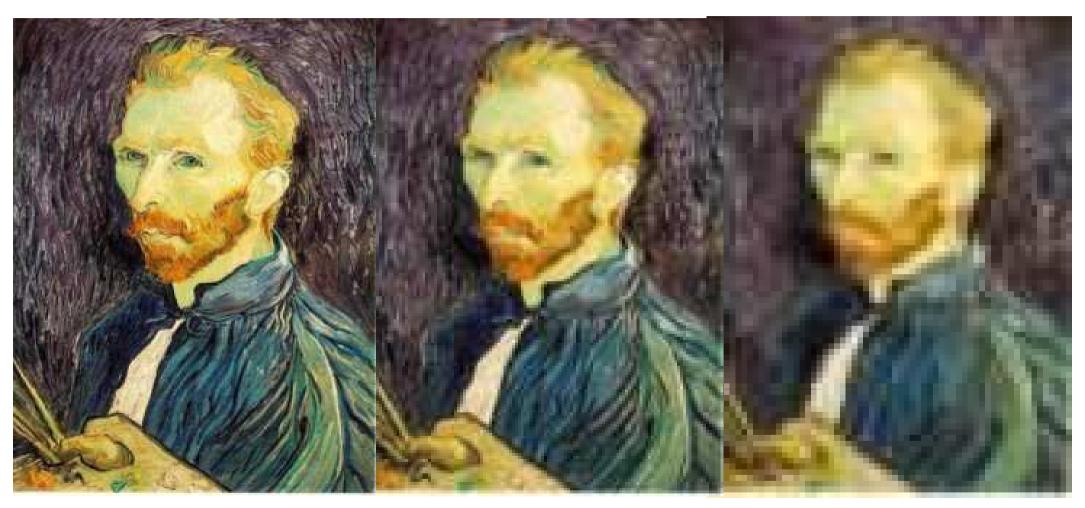




Gaussian 1/4



Gaussian 1/8



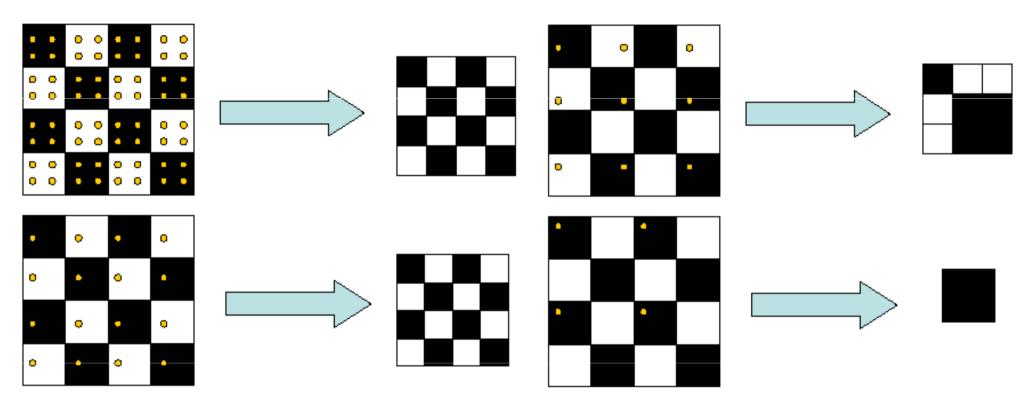
Gaussian 1/2

Gaussian 1/4 (2x zoom)

Gaussian 1/8 (4x zoom)

Adapted from Steve Seitz, U of Washington

Sampling and aliasing



Examples of GOOD sampling

Examples of BAD sampling -> Aliasing

Sampling and aliasing

- Errors appear if we do not sample properly.
- Common phenomenon:
 - High spatial frequency components of the image appear as low spatial frequency components.
- Examples:
 - Wagon wheels rolling the wrong way in movies.
 - Checkerboards misrepresented in ray tracing.
 - Striped shirts look funny on color television.

Gaussian pyramids

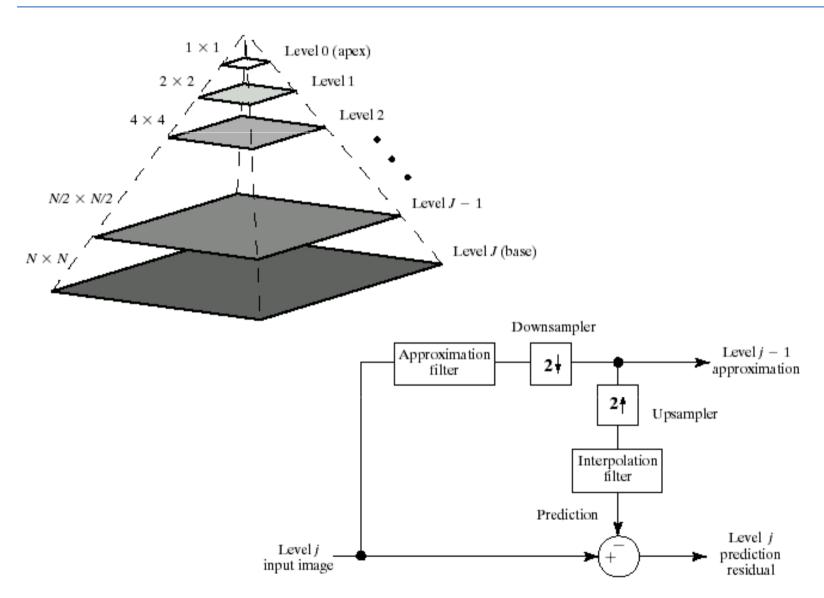
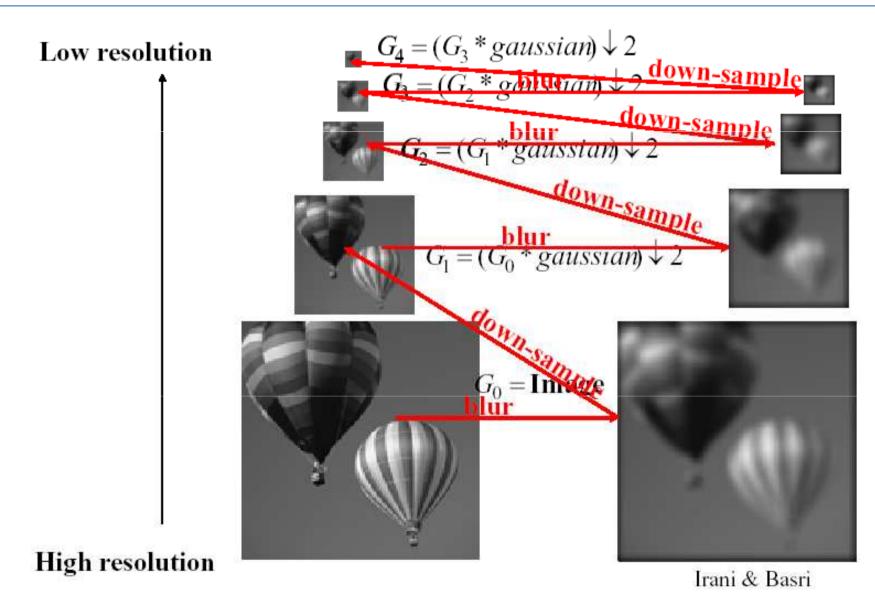


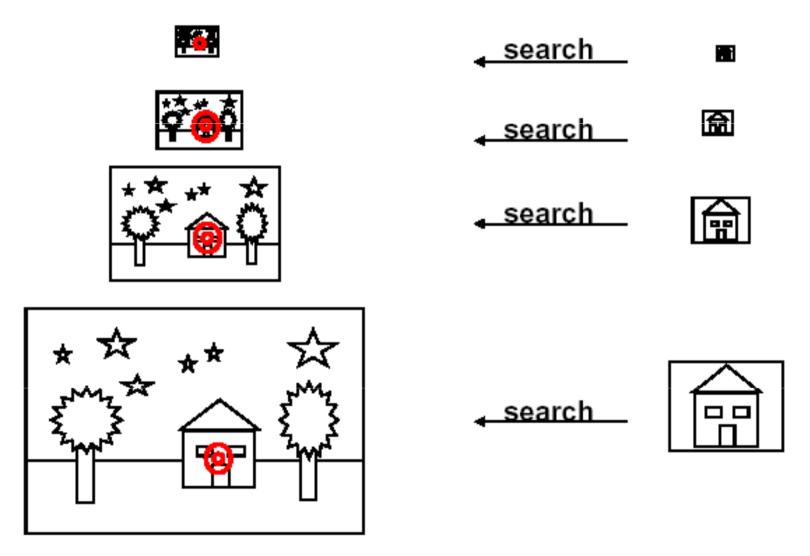


FIGURE 7.2 (a) A pyramidal image structure and (b) system block diagram for creating it.

Gaussian pyramids



Gaussian pyramids



Irani & Basri