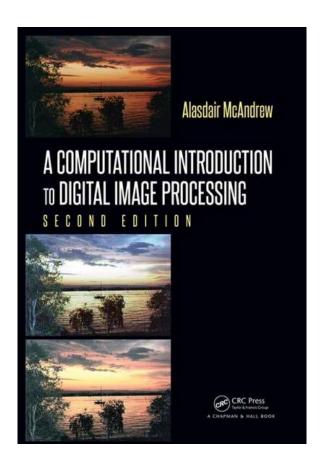
Chapter 7

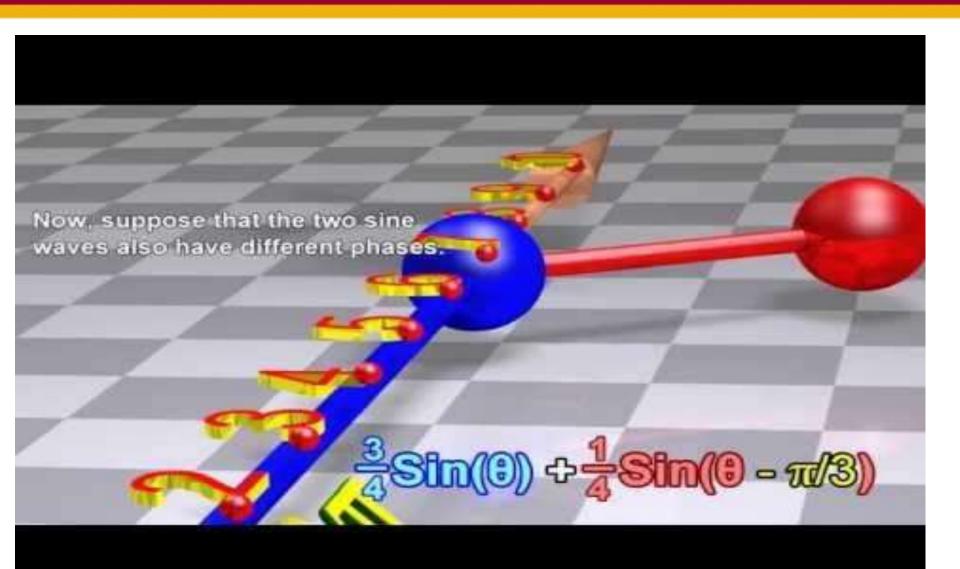


The Fourier Transform



Fourier Transform Video





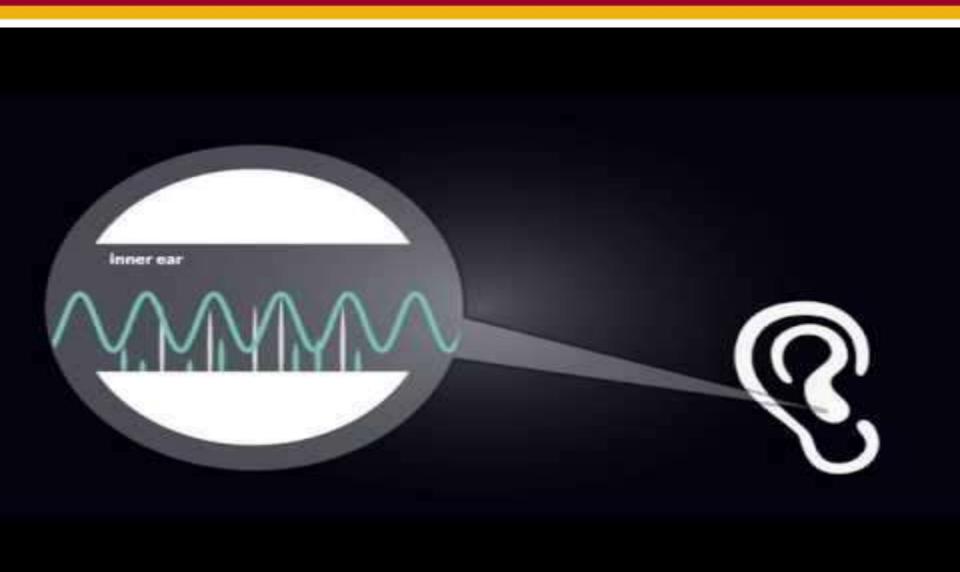
Fourier Transform Video





Fourier Transform Video





Introduction



Why Fourier Transforms?

- Efficiency
- Alternative to spatial filtering
- Perform low pass and high pass filtering with great degree of precision

One-Dimensional DFT



- Discrete function
 - only have to obtain finite number of values
 - only need finite number of functions

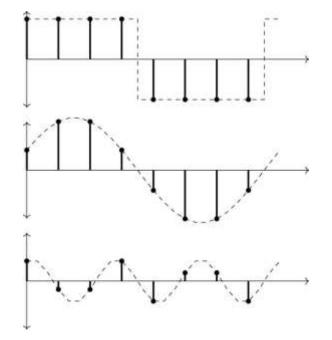


Figure 7.3: Expressing a discrete function as the sum of sines

One-Dimensional DFT



Matlab or Octave

- Functions
 - fft
 - ifft

Python

- Libraries
 - scipy
 - numpy

Two-Dimensional DFT



- Input: matrix
- Output: second matrix of same size
- F is the Fourier Transform of f and is written:

$$F = \mathcal{F}(f)$$
.

Definition

$$F(u,v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x,y) \exp\left[-2\pi i \left(\frac{xu}{M} + \frac{yv}{N}\right)\right].$$
 (7.4)

$$f(x,y) = \frac{1}{MN} \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} F(u,v) \exp\left[2\pi i \left(\frac{xu}{M} + \frac{yv}{N}\right)\right]. \tag{7.5}$$

Two-Dimensional DFT: Similarity



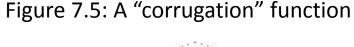
Forward and inverse transforms very similar

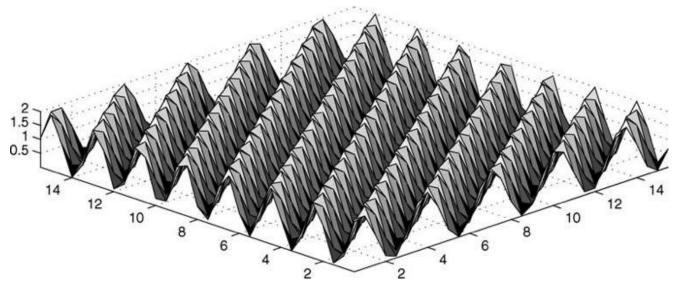
- Exceptions:
 - scale factor 1/MN in inverse transform,
 - negative sign in exponent of forward transform
- same algorithm, only very slightly adjusted, can be used for both the forward and inverse transforms.

Two-Dimensional DFT: Spatial Filter



- What is does:
 - multiplies all elements under a mask with fixed values, and adds them all up.
- Can consider DFT as linear spatial filter







The convolution theorem: One of the most powerful advantages of using the DFT.

Suppose we wish to convolve an image M with a spatial filter S. Our method has been place S over each pixel of M in turn, calculate the product of all corresponding gray values of M and elements of S, and add the results. The result is called the *digital* convolution of M and S, and is denoted:

M * S

This method of convolution can be very slow, especially if *S* is large.



Convolution Theorem states that M * S can be obtained by:

- 1. Pad S with zeros so that it is the same size as M; denote this padded result by S'.
- 2. Form the DFTs of both M and S', to obtain F(M) and F(S').
- 3. Form the element-by-element product of these two transforms: $\mathcal{F}(M) \cdot \mathcal{F}(S')$.
- 4. Take the inverse transform of the result:

$$\mathcal{F}^{-1}(\mathcal{F}(M) \cdot \mathcal{F}(S')).$$



Put simply, the convolution theorem states:

$$M * S = \mathcal{F}^{-1}(\mathcal{F}(M) \cdot \mathcal{F}(S'))$$



To convolve a 512x512 image with a 32x32 filter:

Directly: $32^2 = 1024$ multiplications for each pixel Total $1024 \times 512 \times 512 = 268,435,456$

DFT/FFT: 4608 multiplications per row & per col = $4608 \times 512 \times 2 = 4,718,592$ Same operations for the DFT of filter and inverse DFT, & 512×512 for product of two transforms:

 $4,718,592 \times 3 + 262,144 = 14,417,920$

Experimenting with Fourier Transforms



Relevant functions

- fft, takes the DFT of a vector
- ifft, takes the inverse DFT of a vector
- fft2, takes the DFT of a matrix
- ifft2, takes the inverse DFT of a matrix
- fftshift, shifts a transform

Filtering in the Frequency Domain



Why Use FFT in image processing?

Convolution theorem: Spatial convolution can be performed by element-wise multiplication of the Fourier Transform by a suitable "filter matrix"





Definition:

- Eliminate the outer values and keep the inner ones

Ideal low pass matrix

Binary matrix m defined by

$$m(x,y) = \begin{cases} 1 & \text{if } (x,y) \text{ is closer to the center than some value } D, \\ 0 & \text{if } (x,y) \text{ is further from the center than } D. \end{cases}$$



Figure 7.14: The "cameraman" image and its DFT



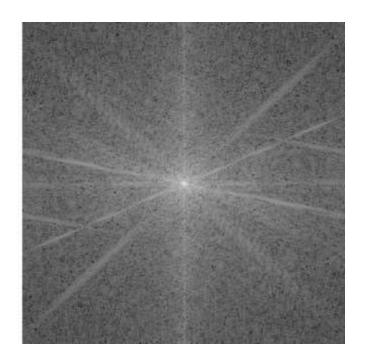
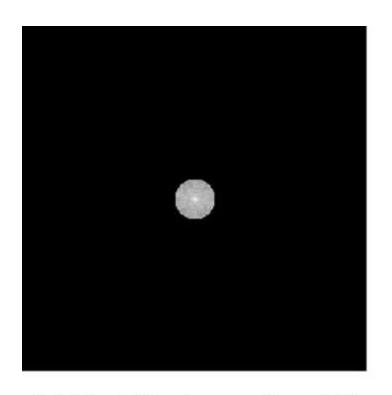




Figure 7.15: Applying ideal low pass filtering



(a) Ideal filtering on the DFT

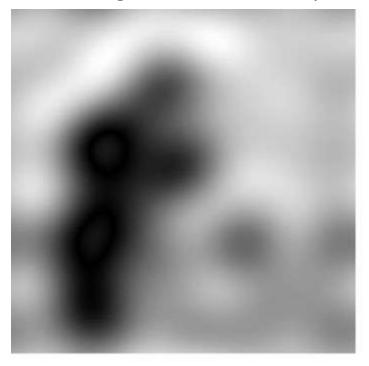


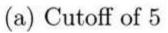
(b) After inversion



Expect:

- Smaller the circle, the more blurred the image
- Larger the circle, the less blurred
 Figure 7.16: Ideal low pass filtering with different cutoffs







(b) Cutoff of 30

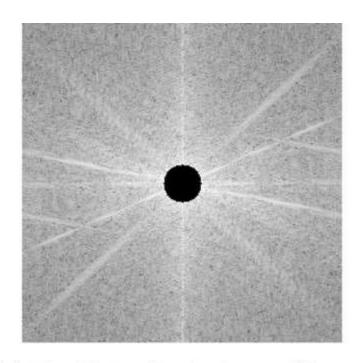
High Pass Filtering



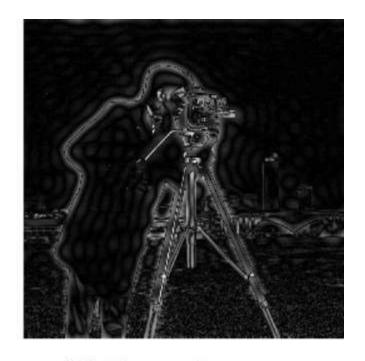
Definition:

Eliminating center values and keeping the others

Figure 7.17: Applying an ideal high pass filter to an image



(a) The DFT after high pass filtering

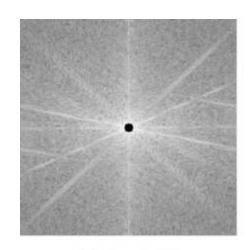


(b) The resulting image

High Pass Filtering



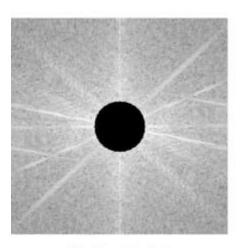
Figure 7.18: Ideal high pass filtering with different cutoffs



(a) Cutoff of 5



(b) The resulting image



(a) Cutoff of 30



(b) The resulting image



Butterworth filter functions are based on the following functions:

– Low pass filters:
$$f(x) = \frac{1}{1 + (x/D)^{2n}}$$

- High pass filters:
$$f(x) = \frac{1}{1 + (D/x)^{2n}}$$

Where in each case the parameter *n* is called the *order* of the filter



Figure 7.19: Ideal filter functions

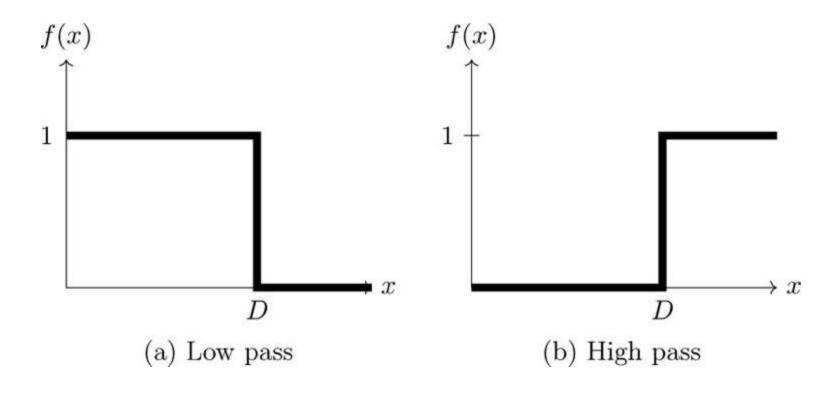
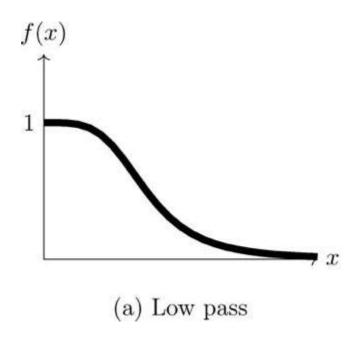




Figure 7.20: Butterworth filter functions with n = 2



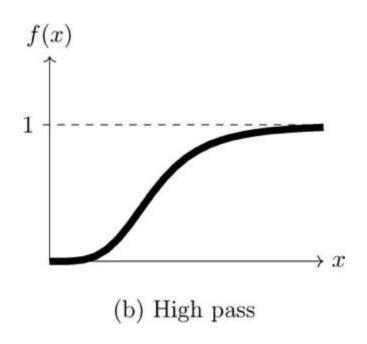




Figure 7.21: Butterworth filter functions with n = 4

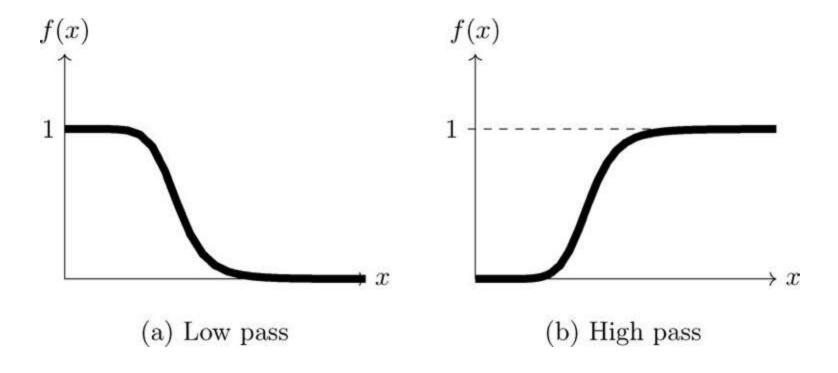




Figure 7.22: Butterworth low pass filtering



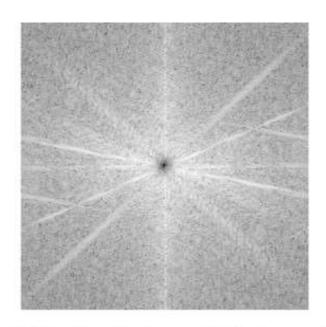
(a) The DFT after Butterworth low pass filtering



(b) The resulting image



Figure 7.23: Butterworth high pass filtering



(a) The DFT after Butterworth high pass filtering



(b) The resulting image

Gaussian Filtering



Implementation

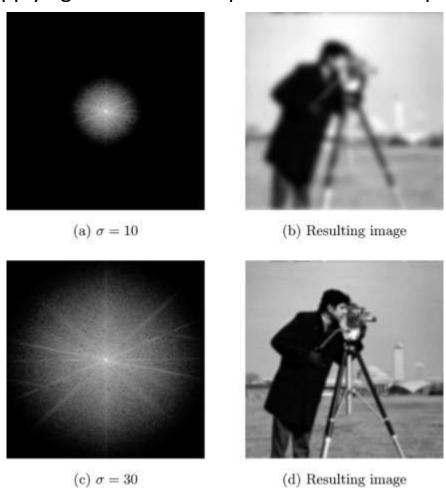
- Create Guassian filter
- Multiply it by the image transform
- Invert the result

Considered the most "smooth"

Gaussian Filtering



Figure 7.24: Applying a Gaussian low pass filter in the frequency domain



Gaussian Filtering



Figure 7.25: Applying a Gaussian high pass filter in the frequency domain



(a) Using $\sigma = 10$



(b) Using $\sigma = 30$