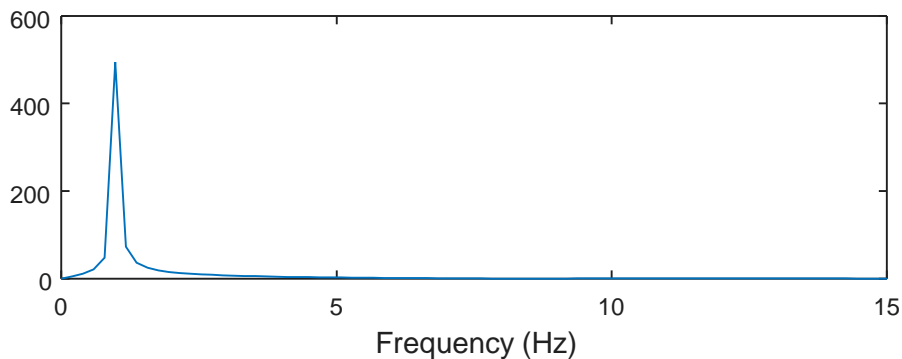
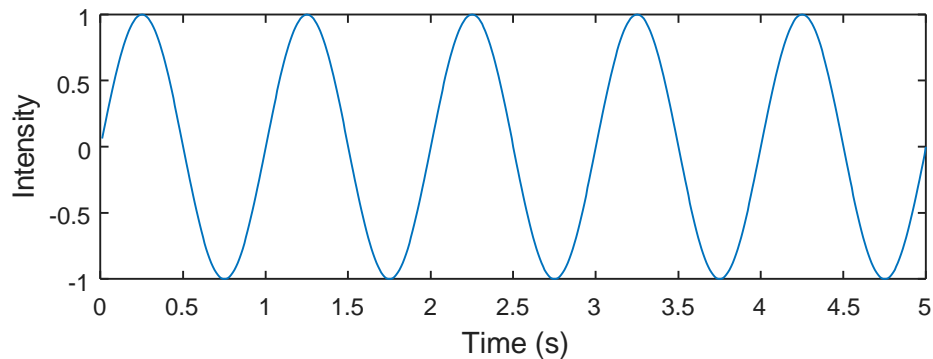


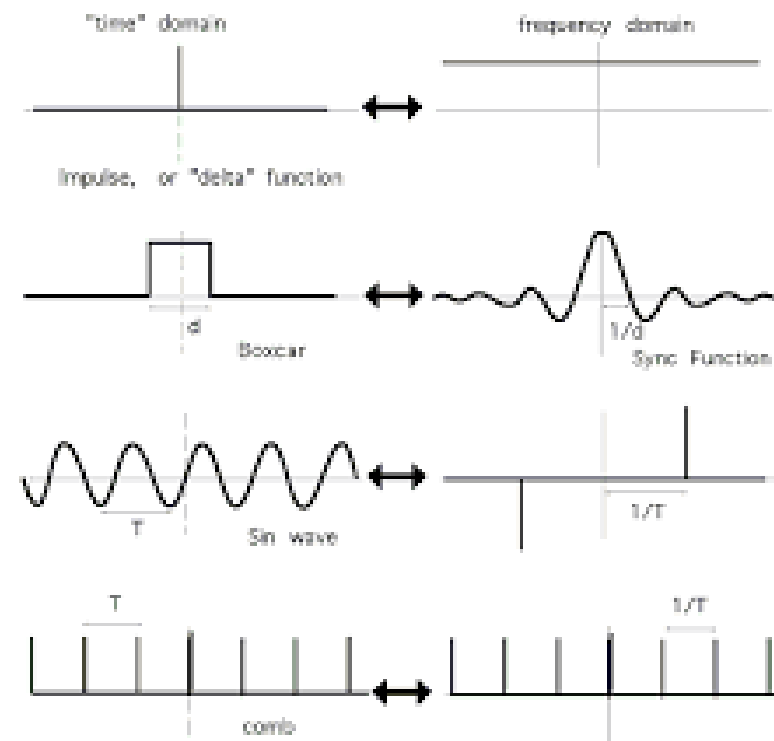
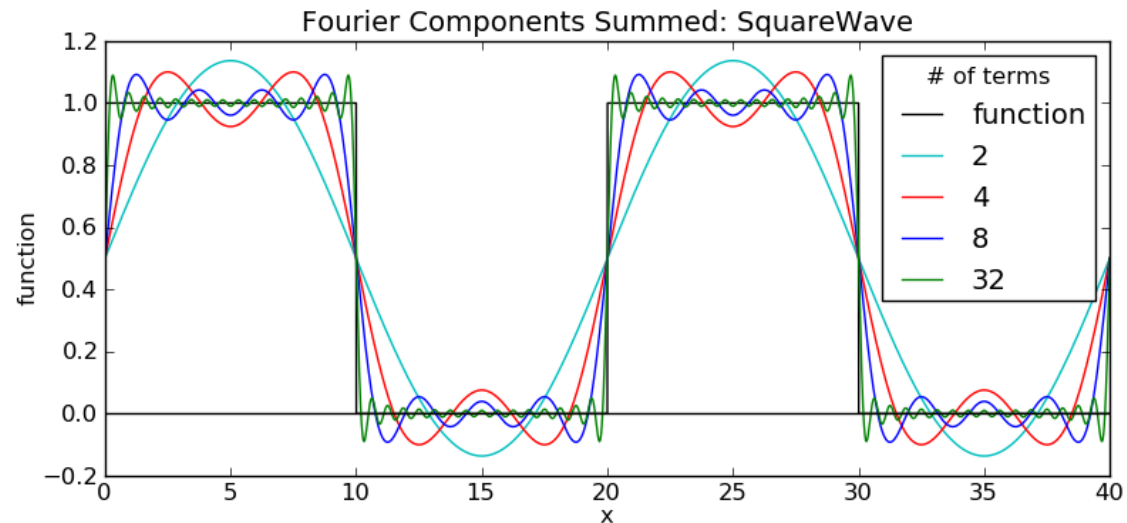
Fourier transforms, SIM

- Last class
  - More STED
  - Minflux
  - Fourier transforms
- This class
  - More FTs
  - 2D FTs
  - SIM



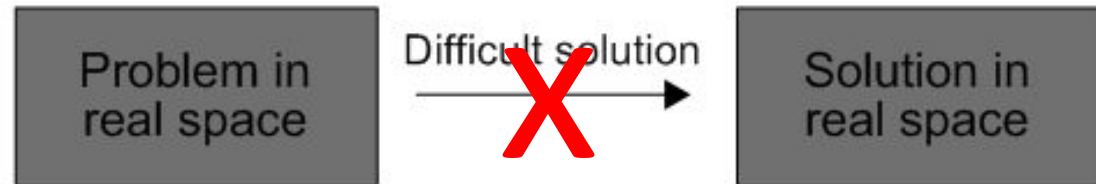
$$f(t) = \sum_n c_n \sin(2\pi n f)$$

If we can figure out  $c_n$  we can describe any function as a sum of sine waves



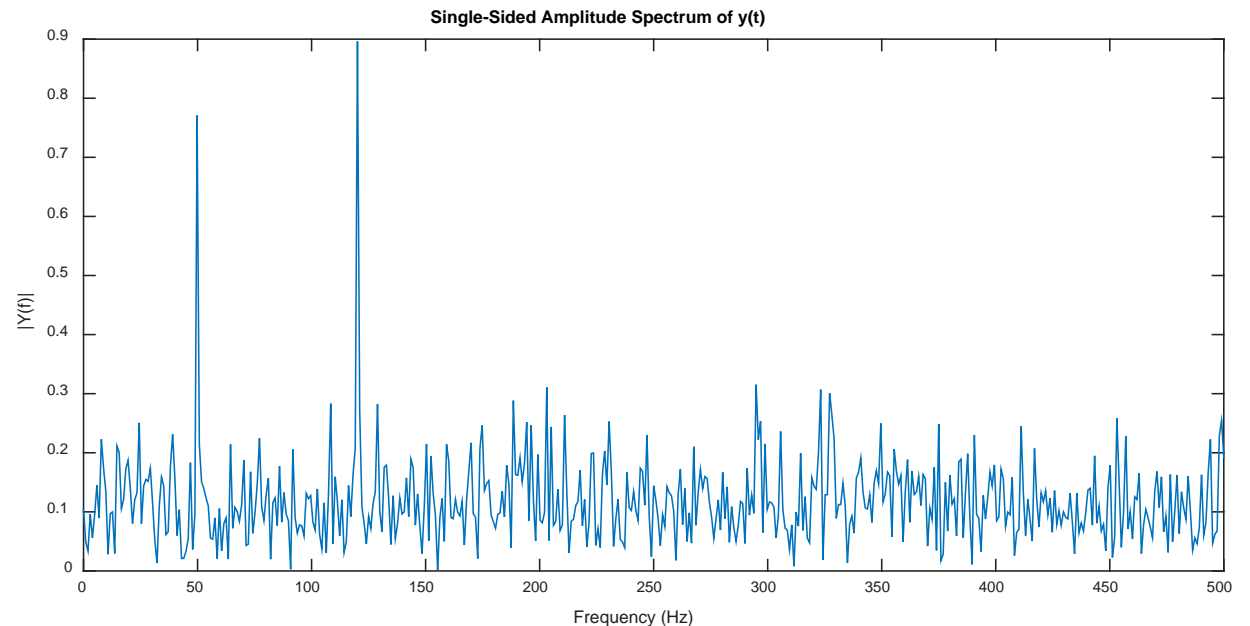
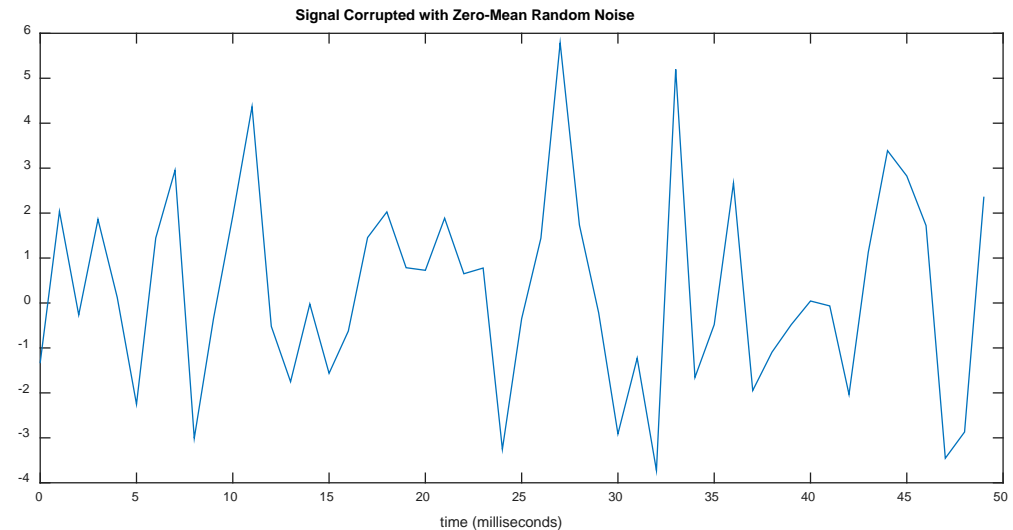
# Why do we convert to frequency domain

- Some problems are very hard to solve in the time domain
- Imagine a repeating source of noise
- There may be a very easy solution in the frequency domain
- It's easy (in Matlab) to convert to frequency, apply the correction, and invert back to time domain



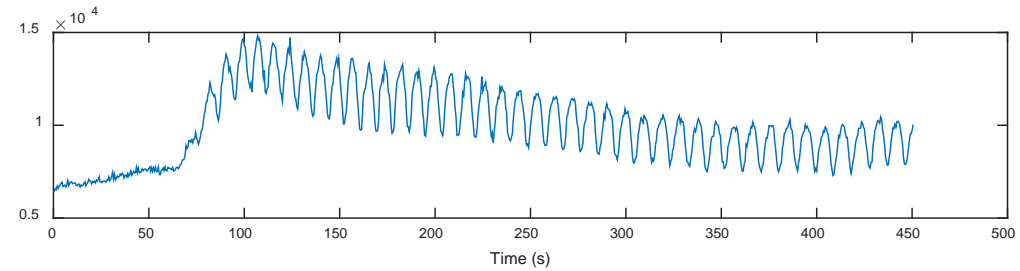
# Fourier transforms useful to pick out signal in noise

- If you think you have an oscillating signal, but it is too noisy, you can often pick it out in frequency space
- Noise is often distributed randomly in frequency space, but your signal remains at one peak
- Conversion from time to frequency domain can clear things up
- Frequency space is heavily used in audio processing

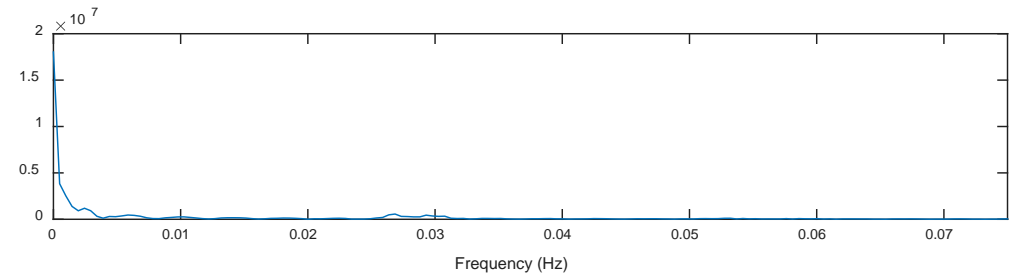


# DC components of signals

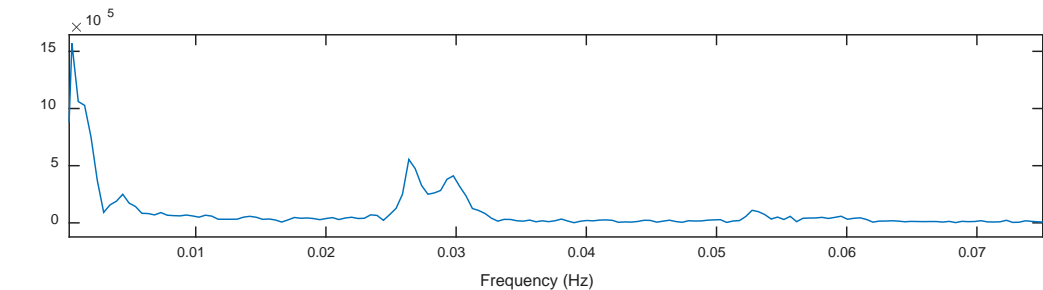
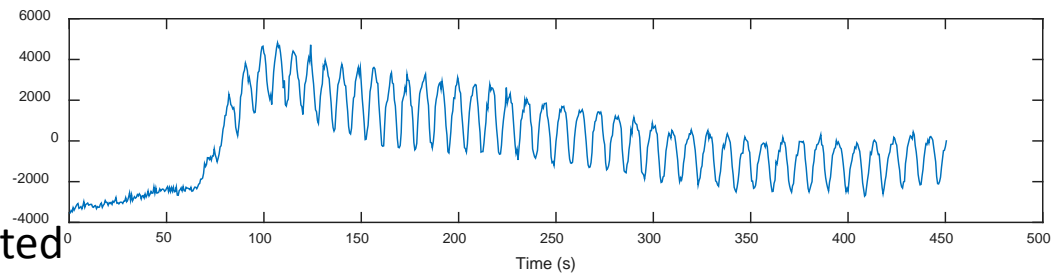
- If the mean of your signal is  $\neq 0$ , then you will have a DC component (freq = 0)
- This can overwhelm your lower frequencies
- Common to subtract the mean off your signal first
- Can also take the log of the frequencies to display nicely



Original

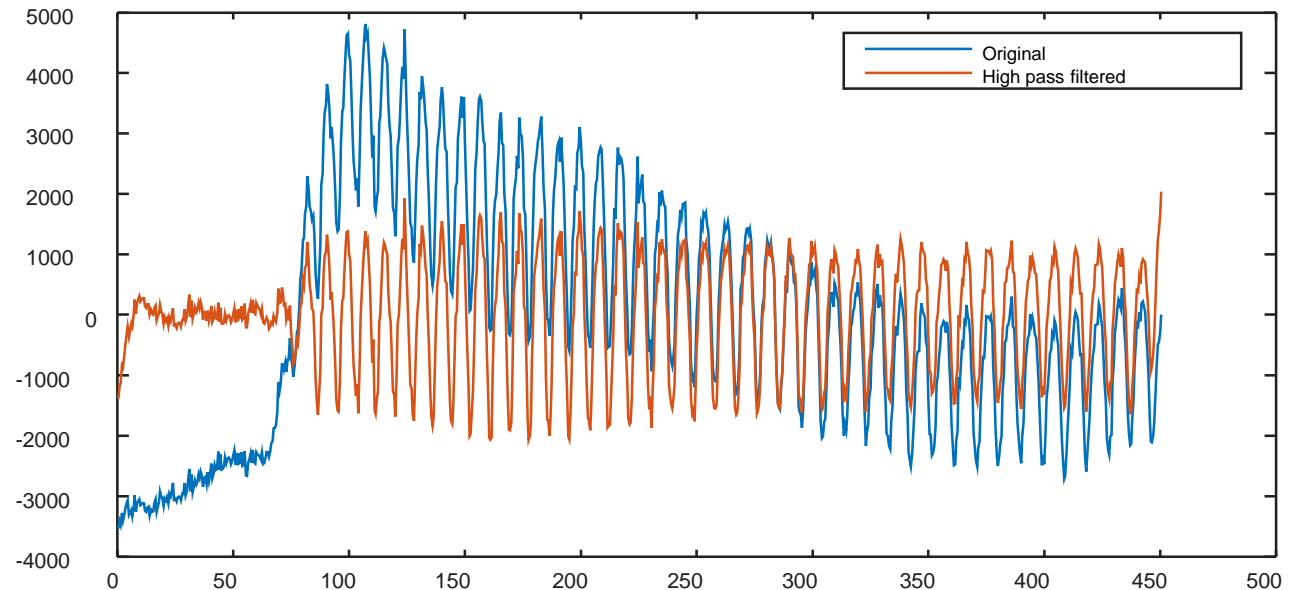
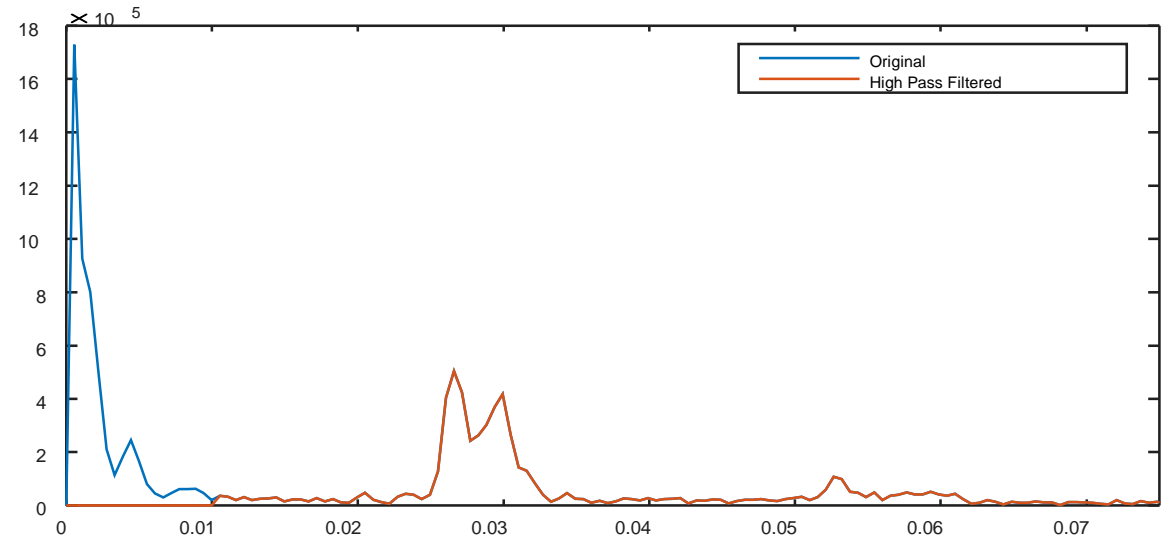


Mean subtracted

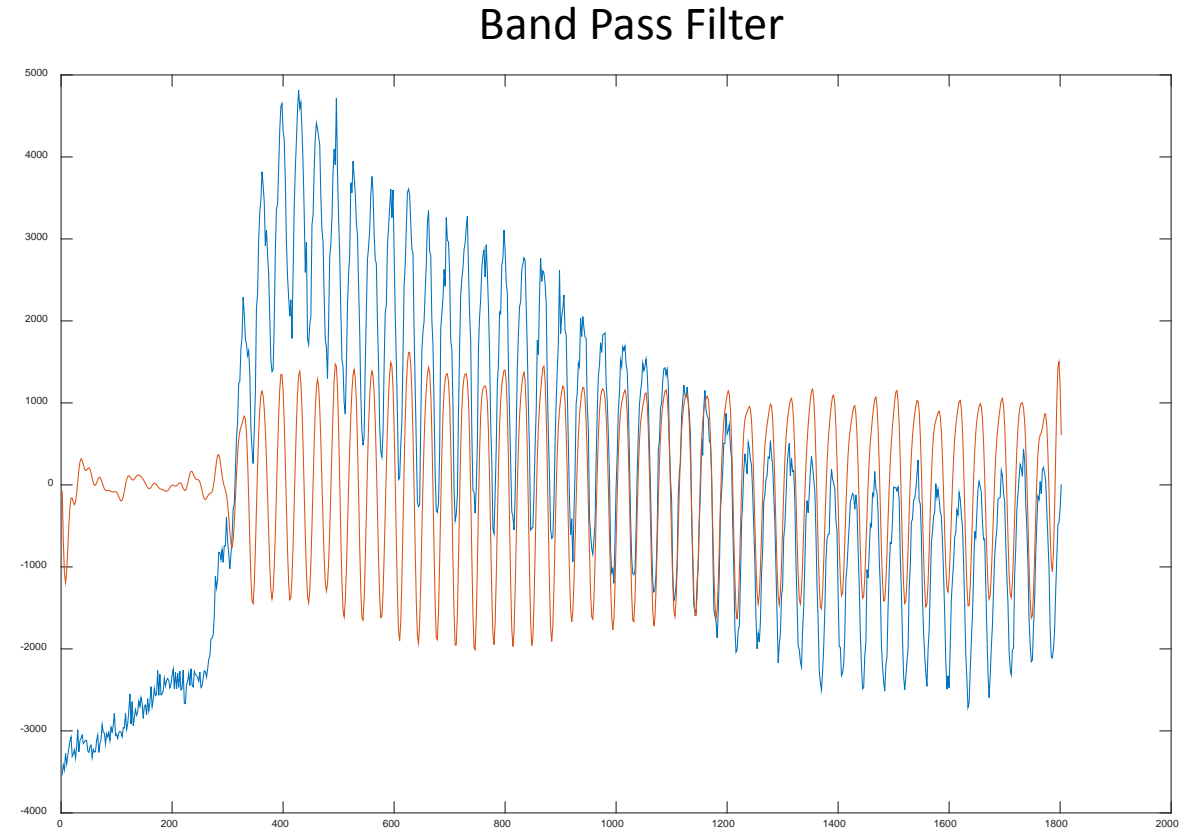
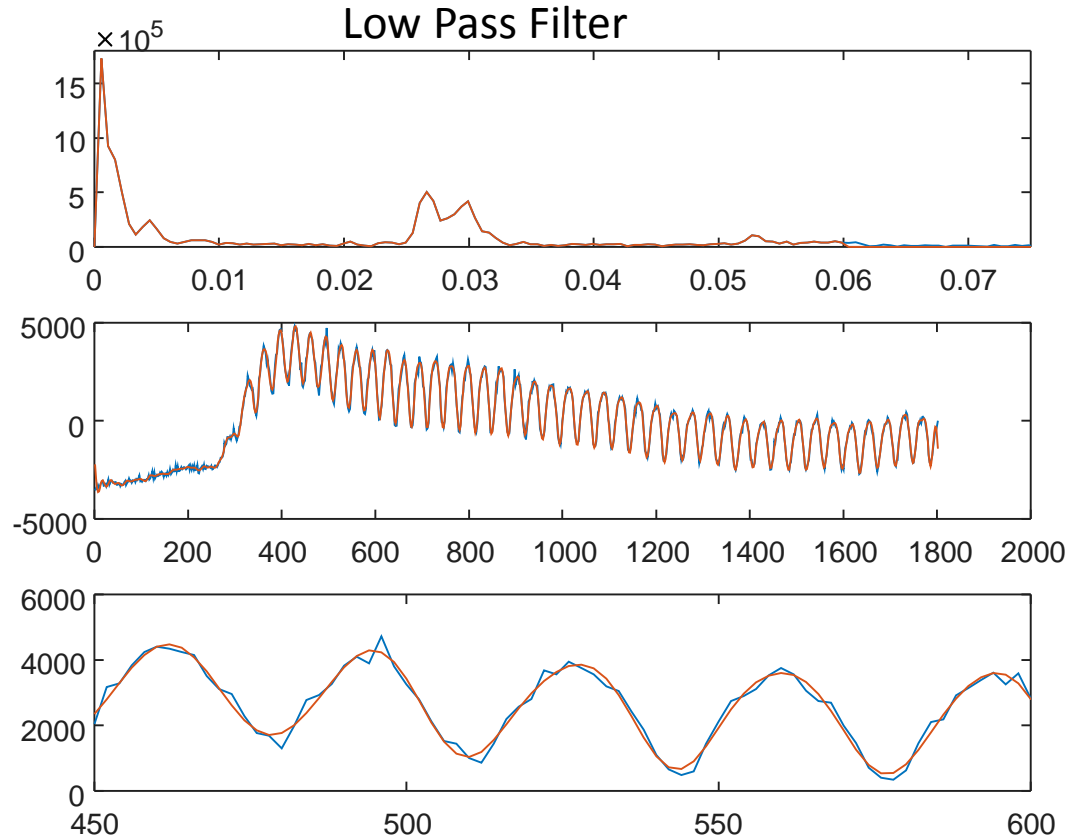


# Fourier filtering

- If we know that the signal we care about is going to fall within some range
- It is easy to:
  - convert to frequency space
  - Set unwanted frequencies to 0
  - Inverse Fourier transform
- Very easy to set low pass, high pass, or band pass filters



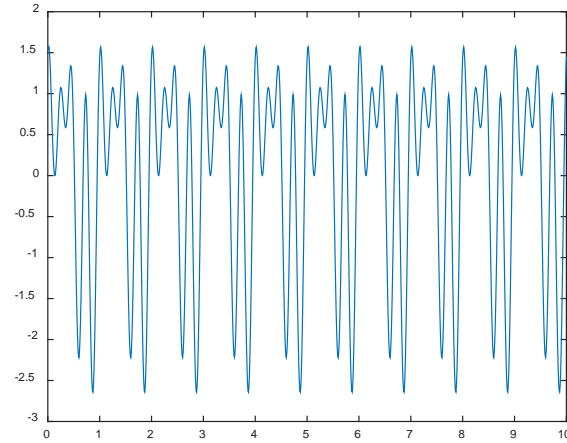
# More Fourier filtering



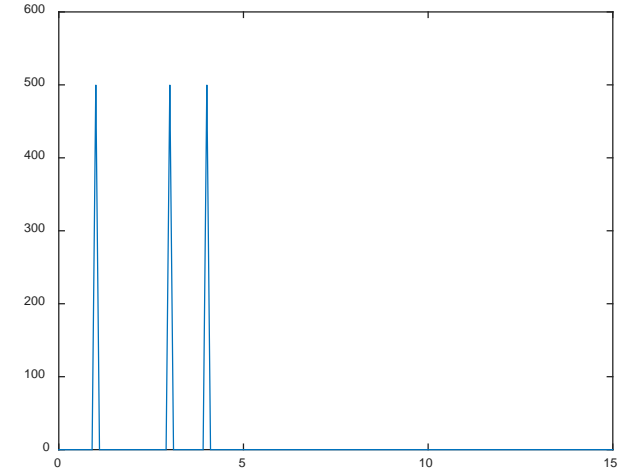


# Fourier transforms encode phase as complex number

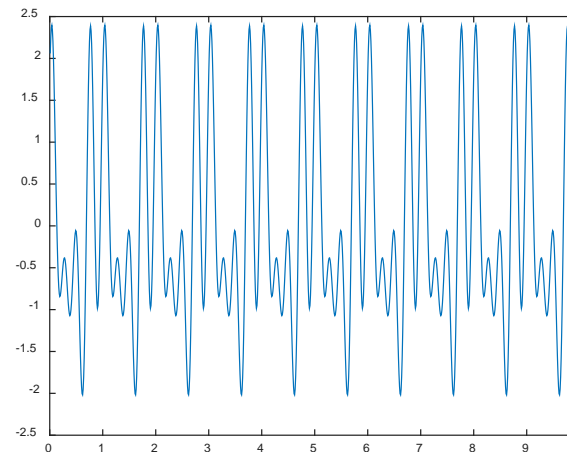
- Consider 3 sine waves with different frequencies and phases
- The FFT will look exactly the same independent of the phases
- In order to reconstruct the time varying signal, we need the phases



```
y1 = sin(2*pi*t + deg2rad(0));  
y2 = sin(6*pi*t + deg2rad(30));  
y3 = sin(8*pi*t + deg2rad(110));
```



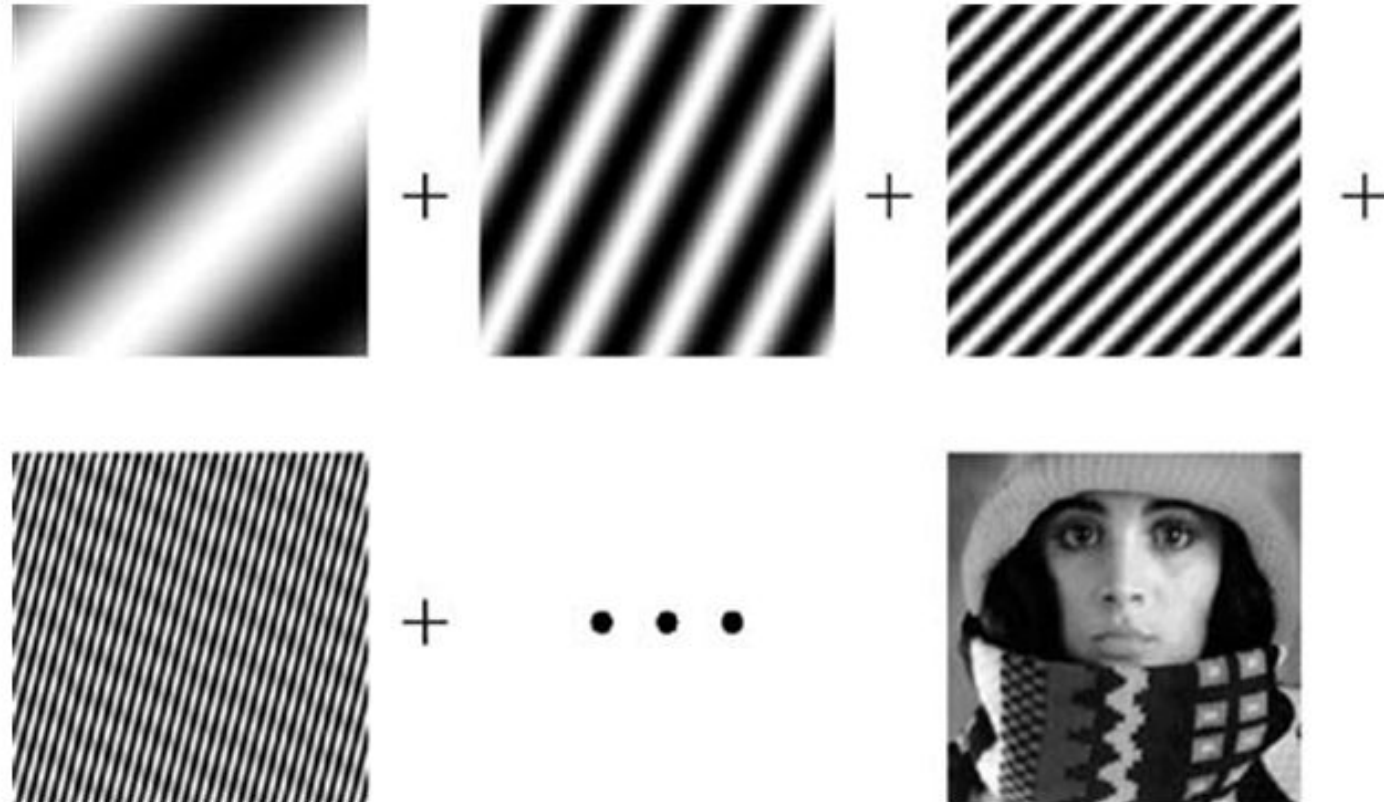
Note, I plotted the absolute value of the fourier transform



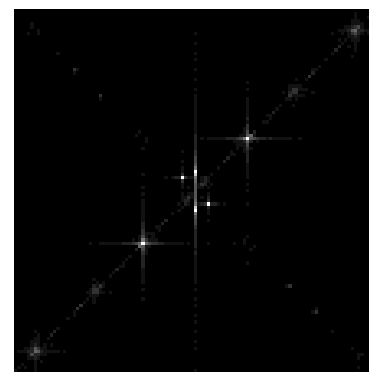
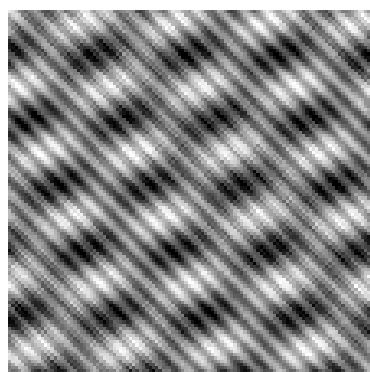
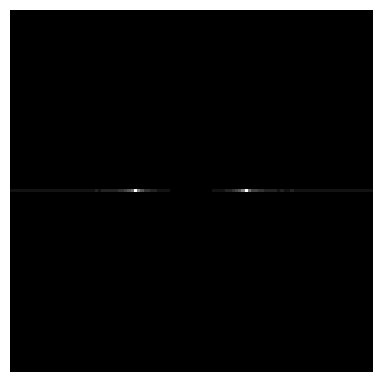
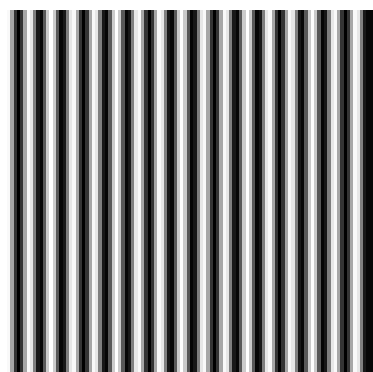
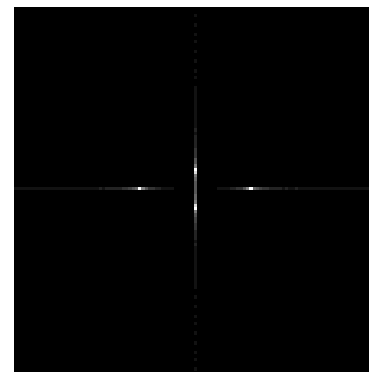
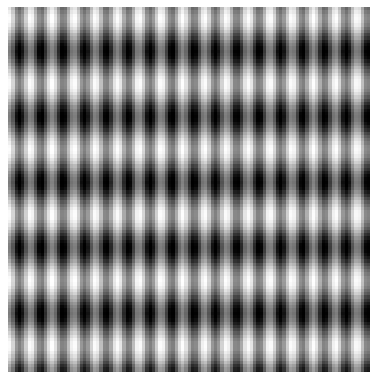
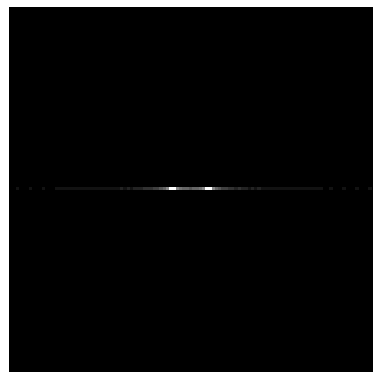
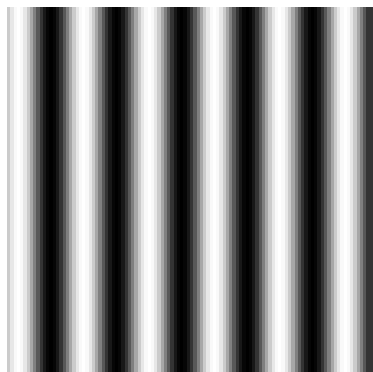
```
y4 = sin(2*pi*t + deg2rad(110));  
y5 = sin(6*pi*t + deg2rad(0));  
y6 = sin(8*pi*t + deg2rad(60));
```

# 2D Fourier transform

- In exactly an analogous way, linear combinations of 2d sine functions can be combined to form any image
- If you can calculate the amplitude and phase of each set, you can reconstruct any arbitrary image
- Matlab has a 2D FFT function that allows you to calculate amplitudes very quickly
- Allows image processing in the frequency domain

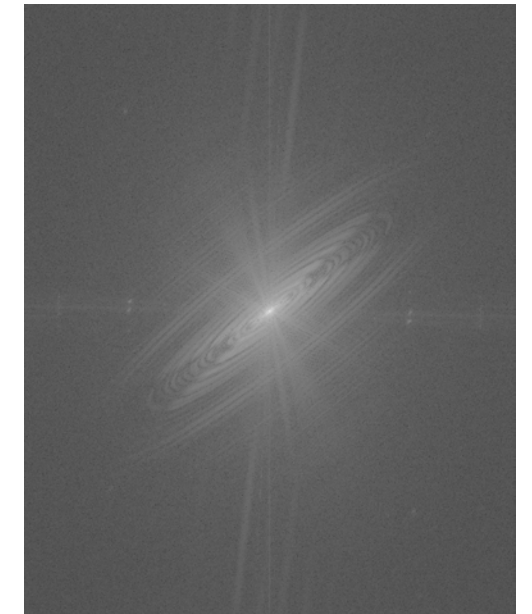
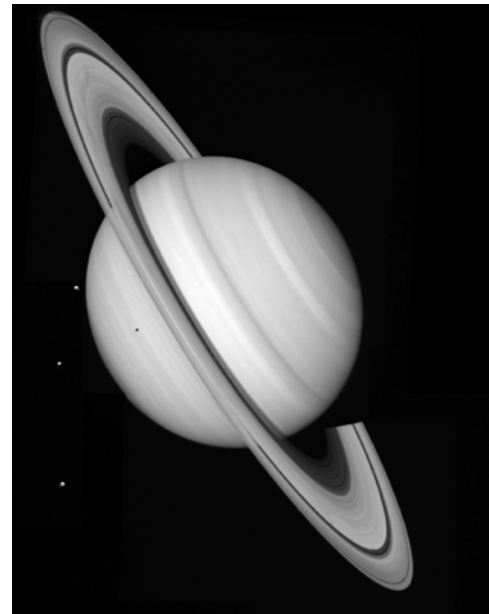
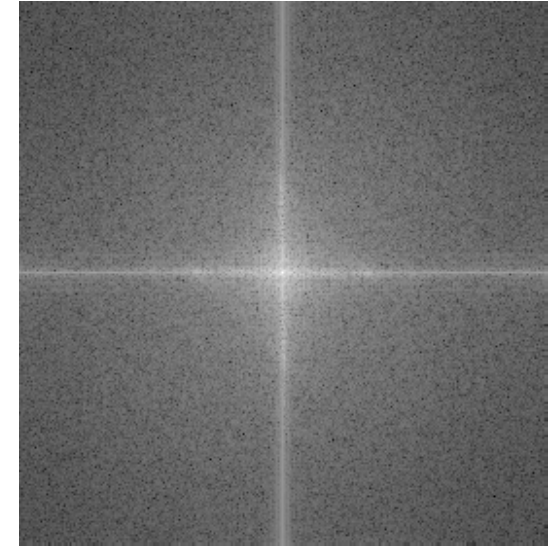
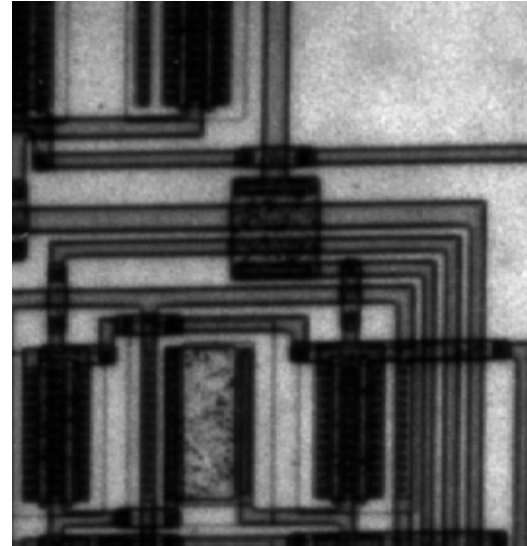


# 2D Fourier transforms



# 2D Fourier Transform of Images

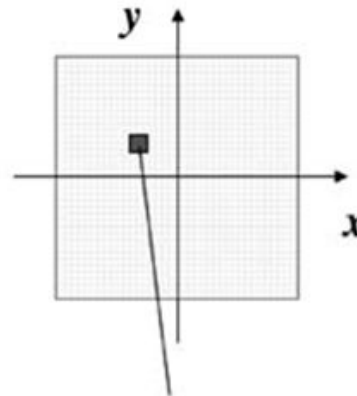
- Now frequencies are represented by amplitude in space
- X axis (frequency) is periodic signal in y dimension, and vice versa
- The center of the image is the DC signal
- Diagonal points represent diagonal periodicity
- Edges represent the highest frequencies encoded



# 2D Fourier transform meaning

- In space, each pixel represents an intensity
- In frequency, each pixel represents the amount of that spatial frequency
- In the camera man, there are vertical lines (buildings), horizontal lines (skyline), and diagonal lines (tripod)

Space domain



Gives the intensity of the image at that point

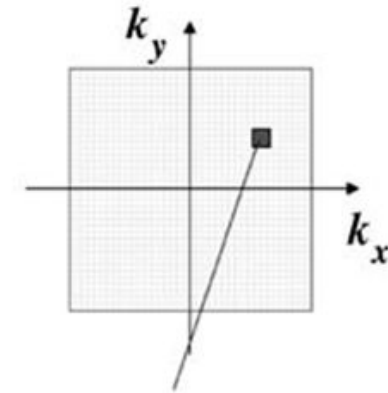
Fourier transform



Inverse Fourier transform

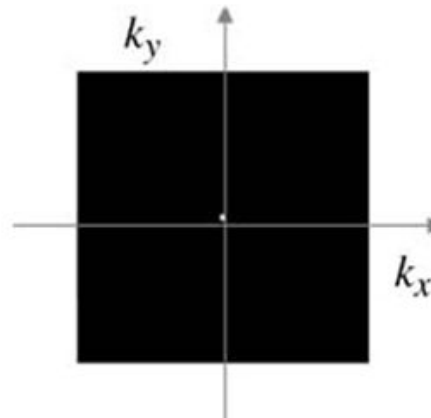


Frequency domain



Gives the contribution of  $e^{i(k_x x + k_y y)}$  to the image

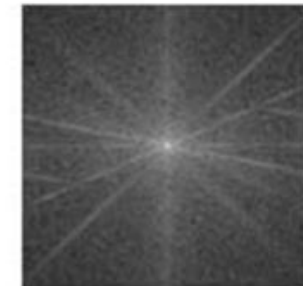
Frequency domain



Original image

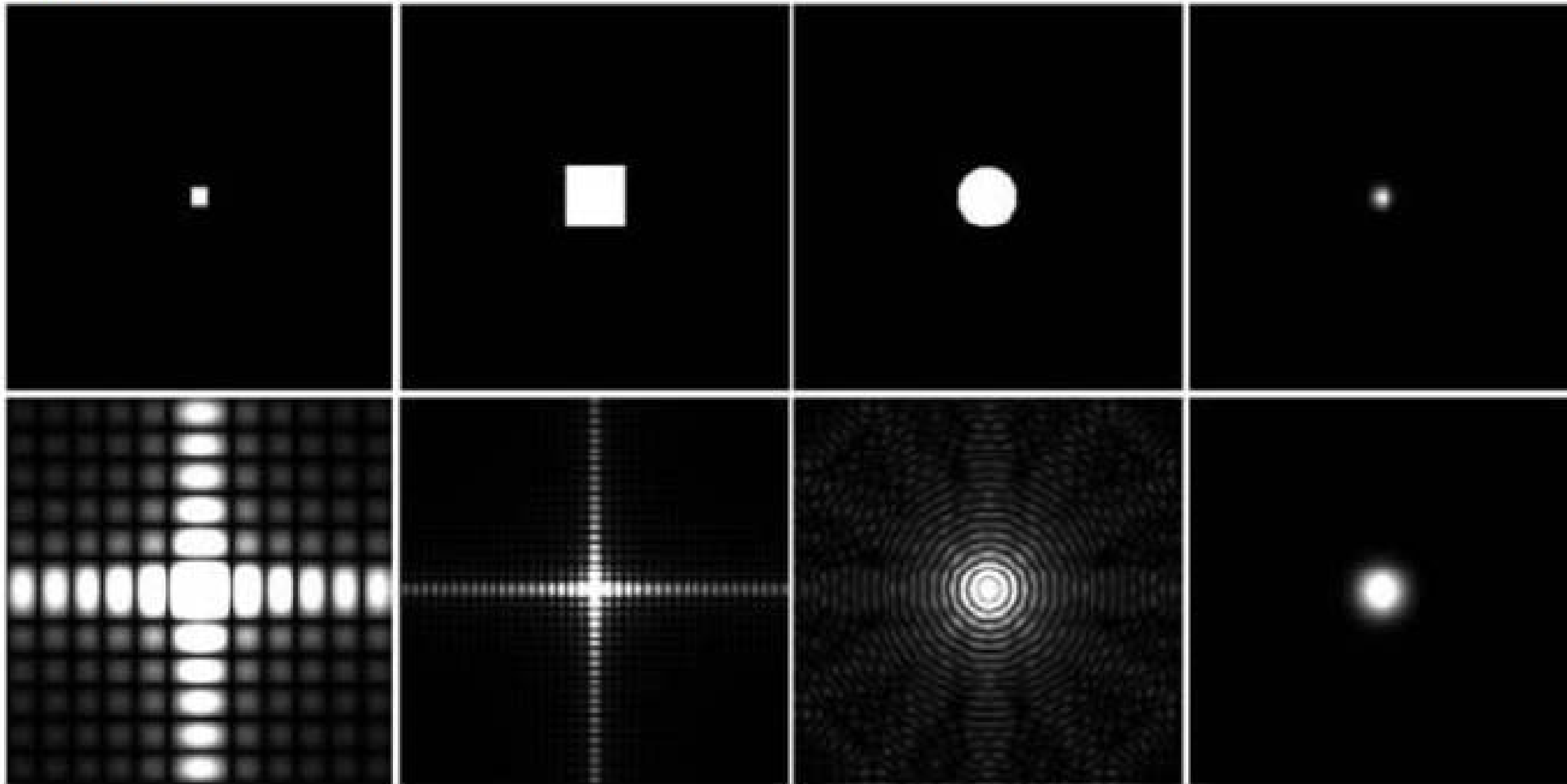


FT of image (modulus)



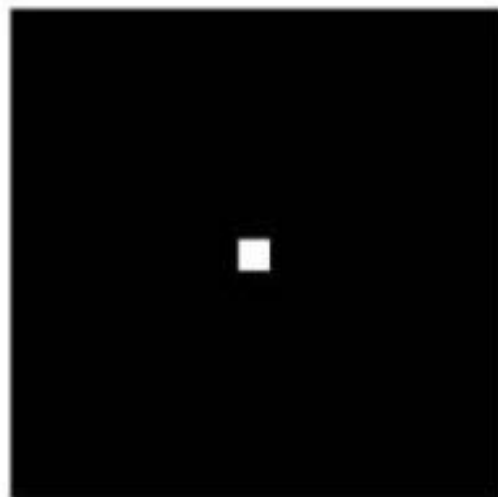
# Amplitude spectra

Images

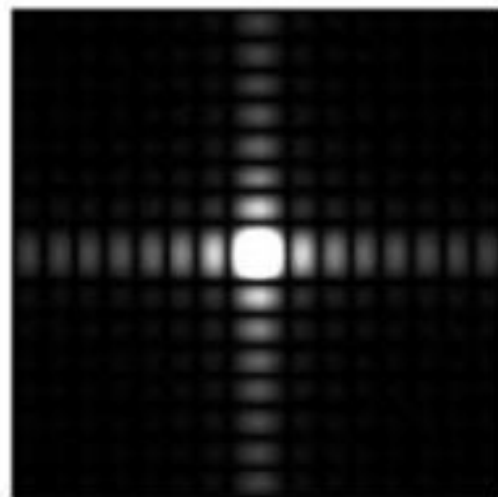


Fourier Transforms

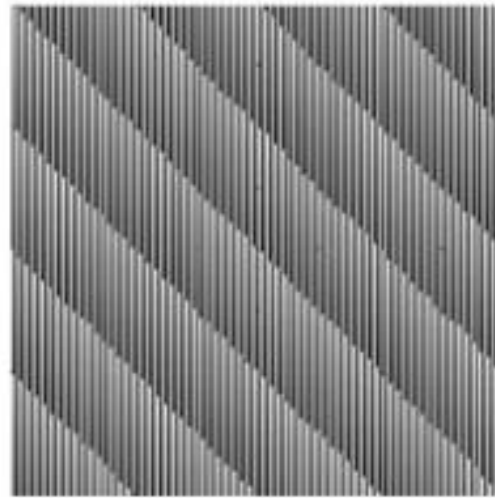
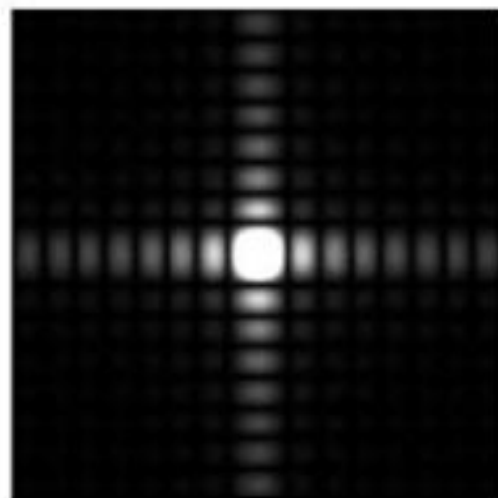
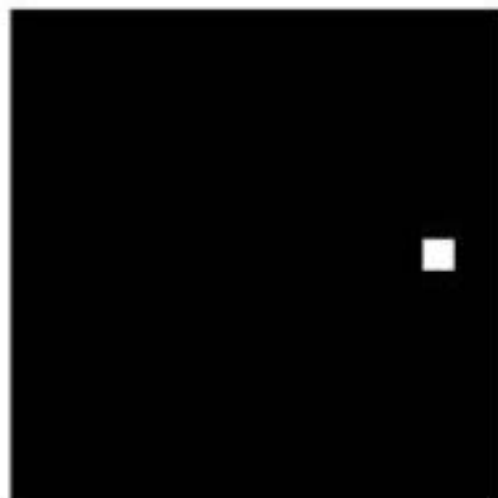
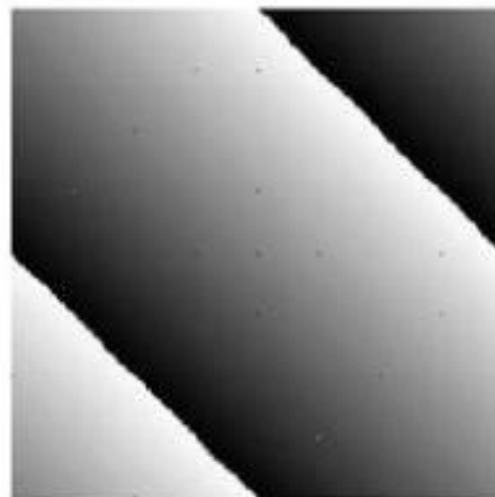
Original



Amplitude

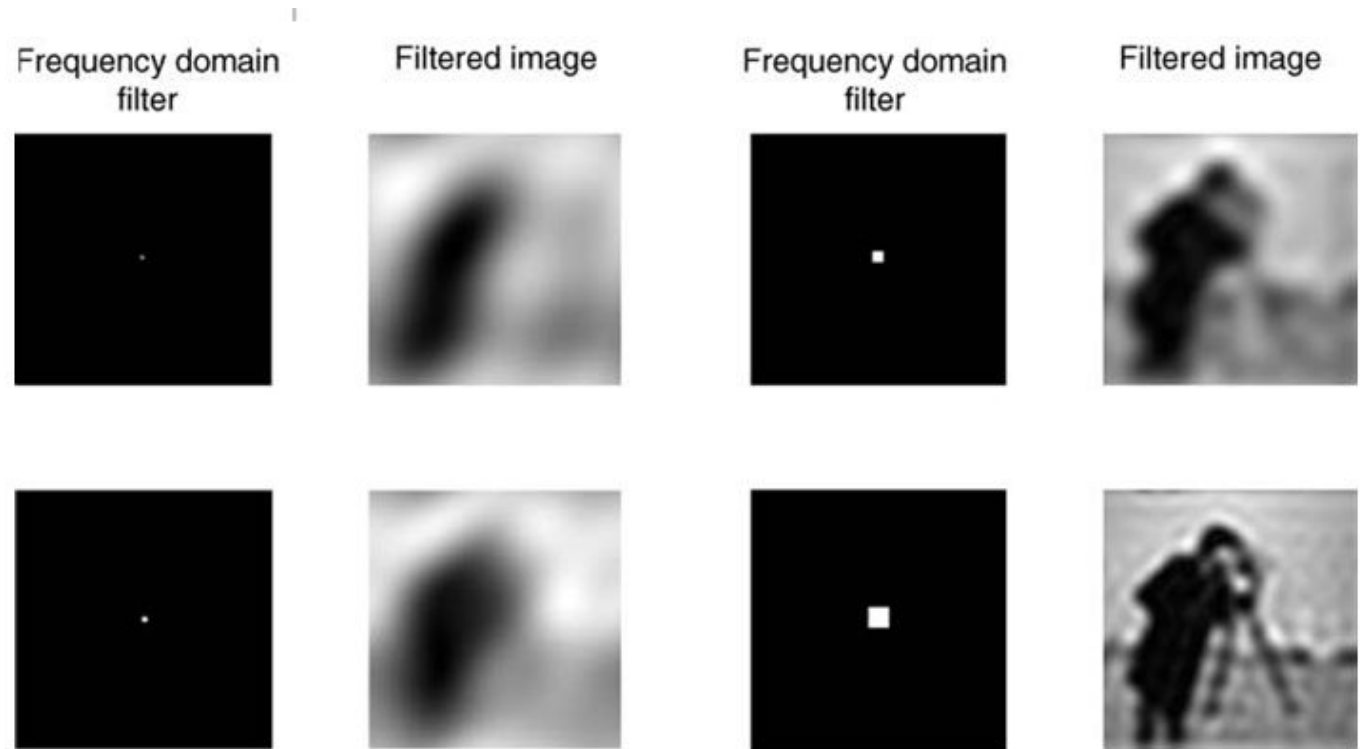


Phase



# FFT and filtering

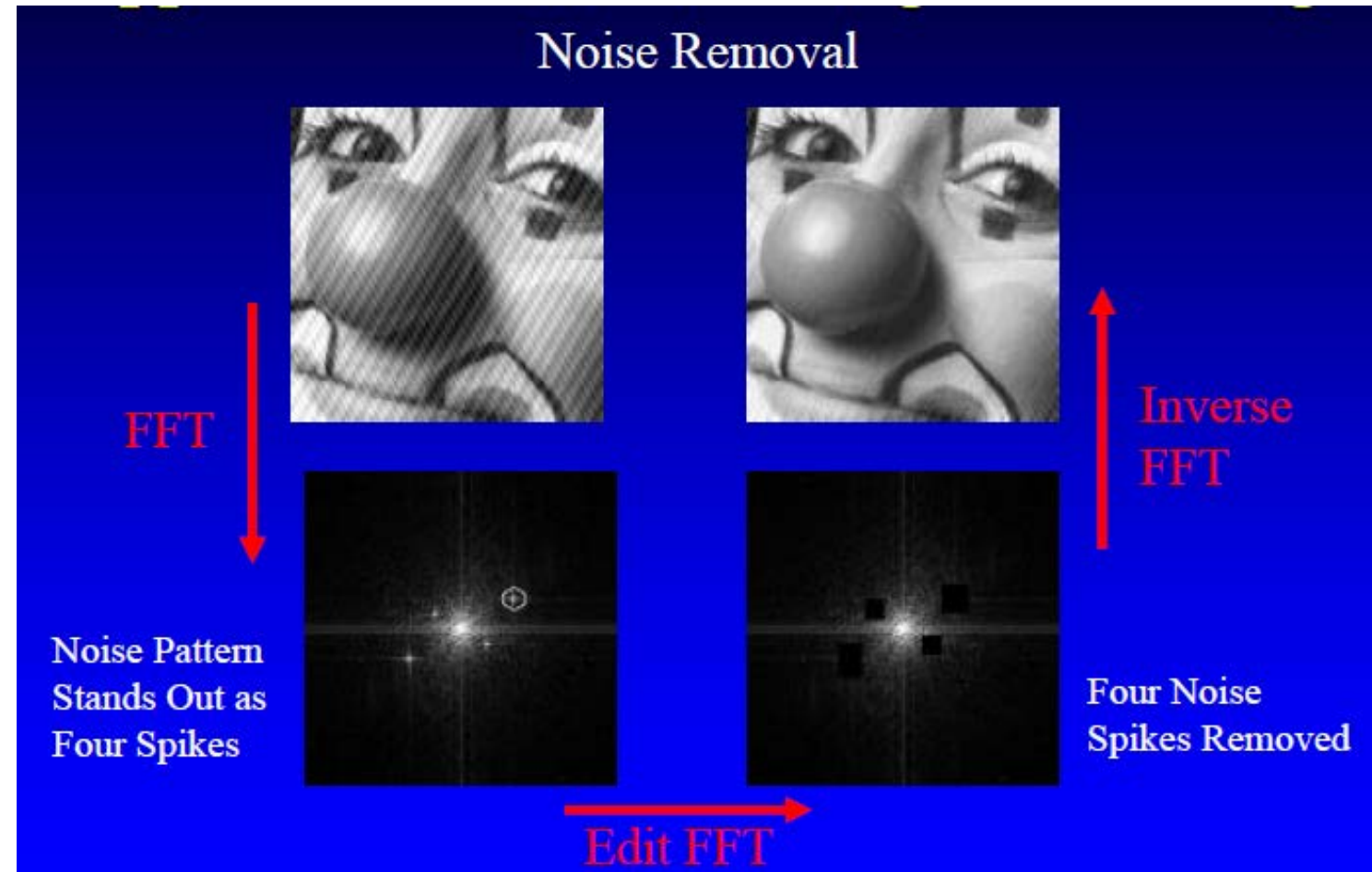
- The more frequencies encoded in an image, the sharper the detail you can resolve
- By restricting higher frequencies you can filter the image





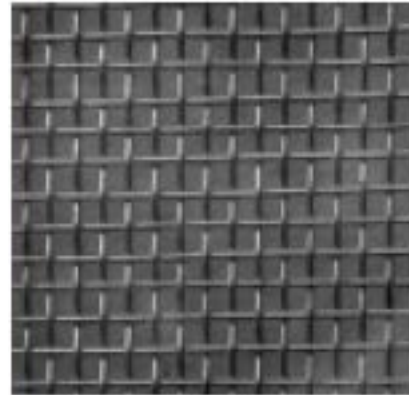
# Applications: Noise removal in images

- Similar to the 1D case, periodic noise can be easily suppressed
- Regular noise appears as points
- Set those points in frequency space = 0
- IFFT to convert back to spatial coordinates

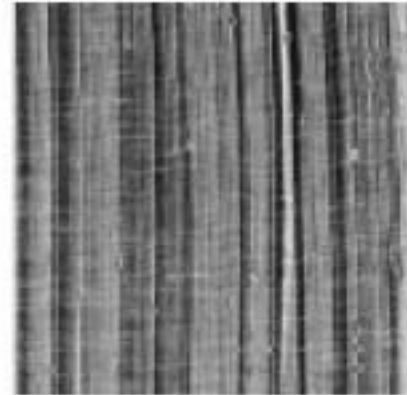


# Applications: recognition of textures

- Repeating textures will have distinct Fourier components
- Easy to pick them out in frequency space



(a) A periodic texture (D1).



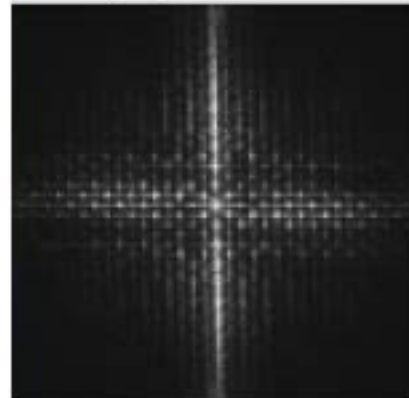
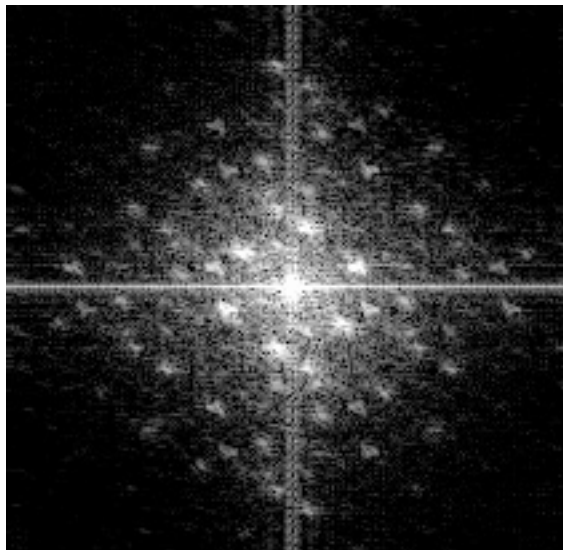
(b) A directional texture (D106).



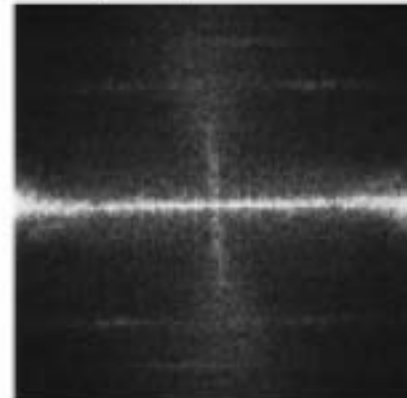
(c) A random texture (D106).



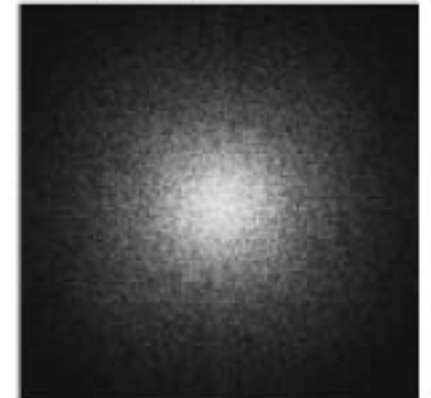
Drosophila eye  
and its 2D FFT



(d) The Fourier spectrum of (a).



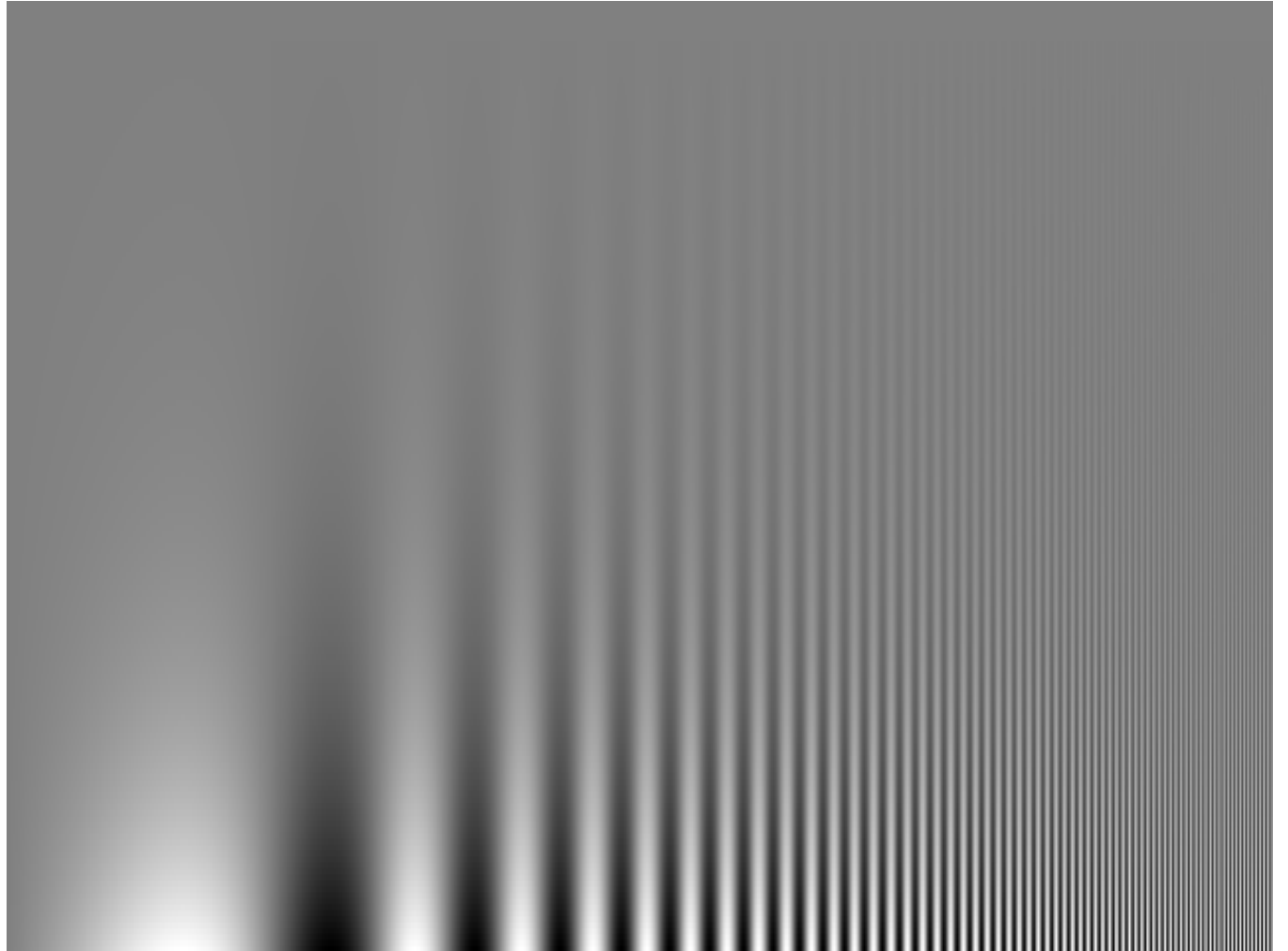
(e) The Fourier spectrum of (b).



(f) The Fourier spectrum of (c).

# Frequency sensitivity of visual system

- Campbell Robson sensitivity curve
- The U shape is a pattern of your visual system, not the image
- The computer doesn't care, but you (and readers) can misinterpret images that contain too high or low of frequencies



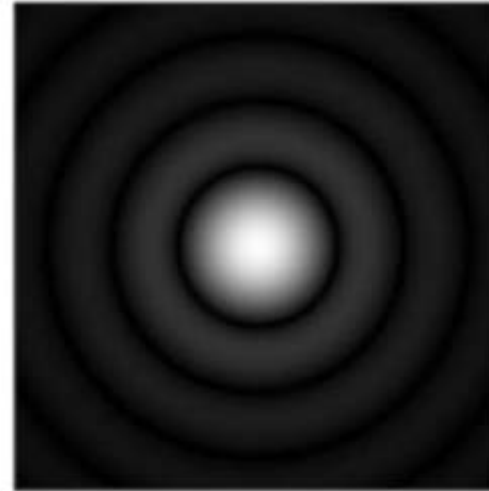
# SIM

- 3<sup>rd</sup> class of superresolution

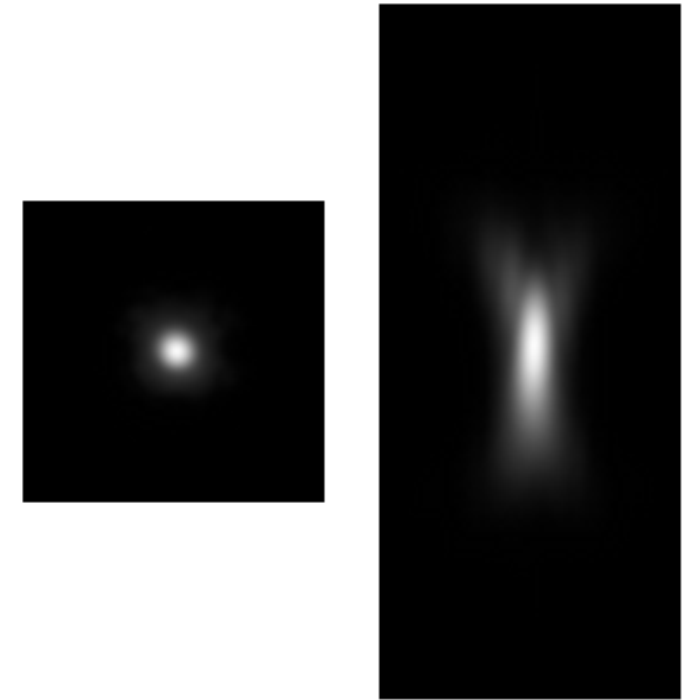
# Origins of diffraction limit

- The airy disk size is determined by the wavelength and NA
- The resolution between two objects is set by Rayleigh criterion
- It is the maximum angle that sets our size

Airy Disk

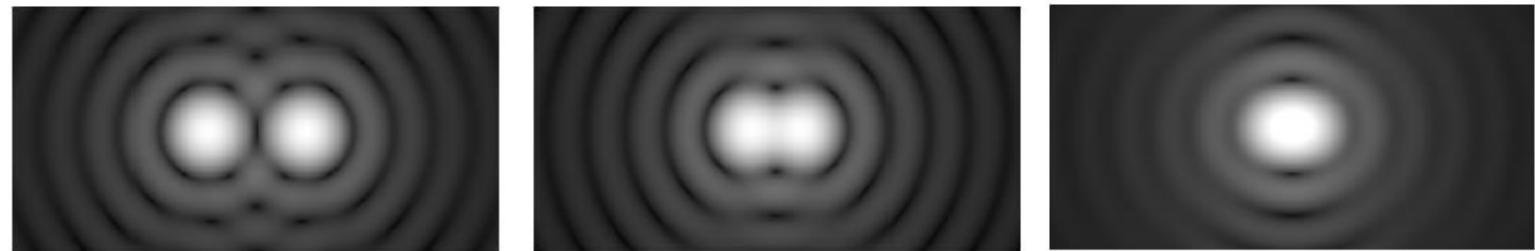


Point Spread Function



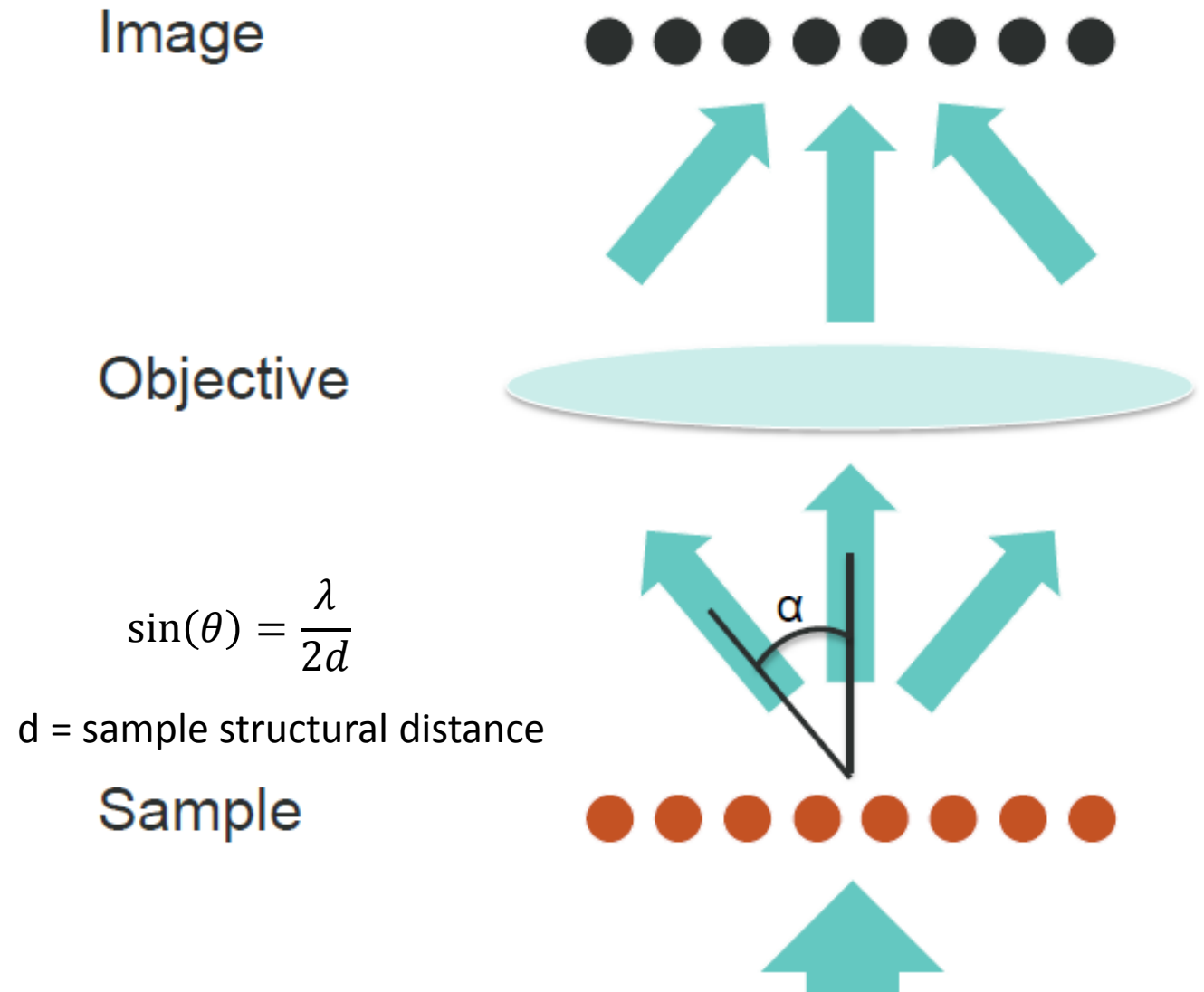
$$R = 1.22 \frac{\lambda}{2n \sin(\theta)} = .61 \frac{\lambda}{NA}$$

Smallest distance at which we can resolve 2 points



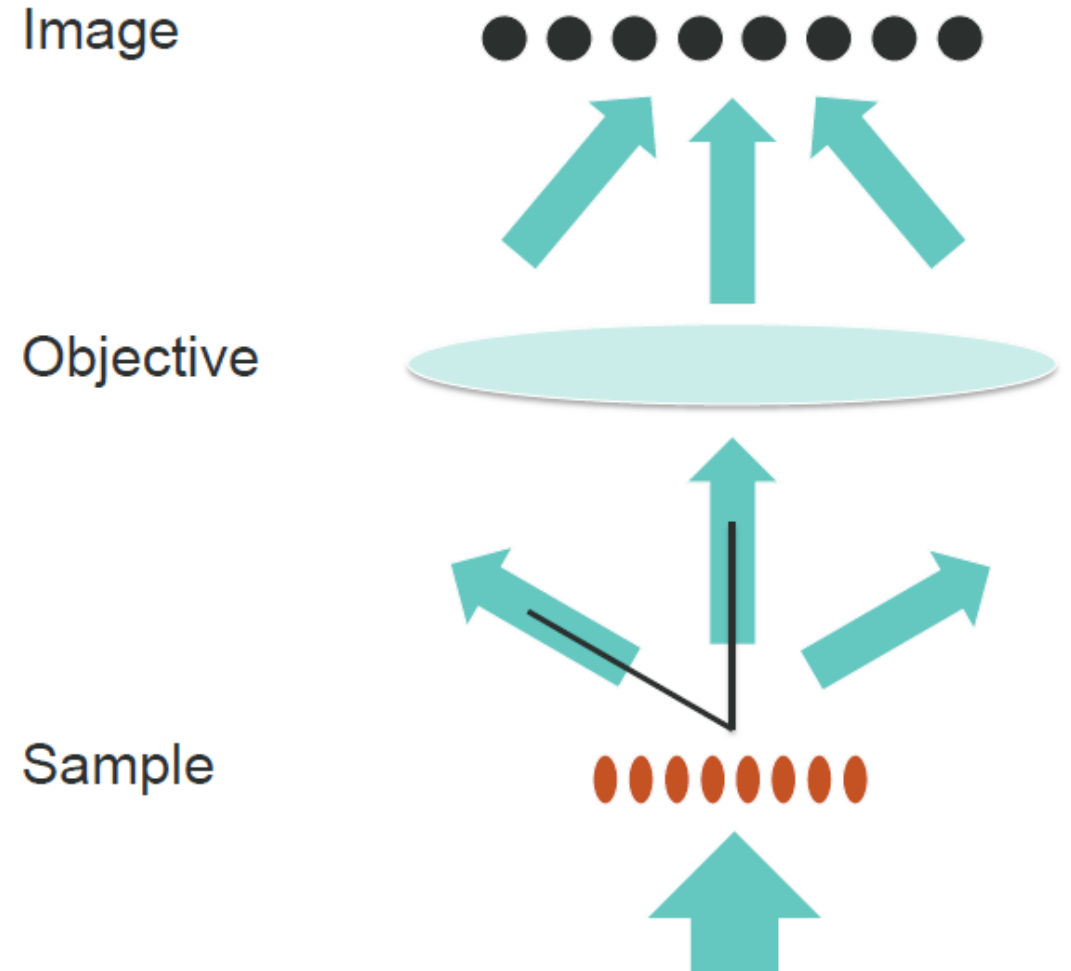
# Limit is result of sample diffracting light

- Abbe realized we can think of the sample diffracting the light
- The lens re-images those diffraction patterns back onto the sample plane
- The angle of diffraction is proportional to the spacing



# Effects of finite objective size

- The fact that we can't capture all the diffracted rays means that we lose the largest angles
- These angles correspond to the smallest features in the sample
- To resolve smaller features, we need to capture these higher frequencies



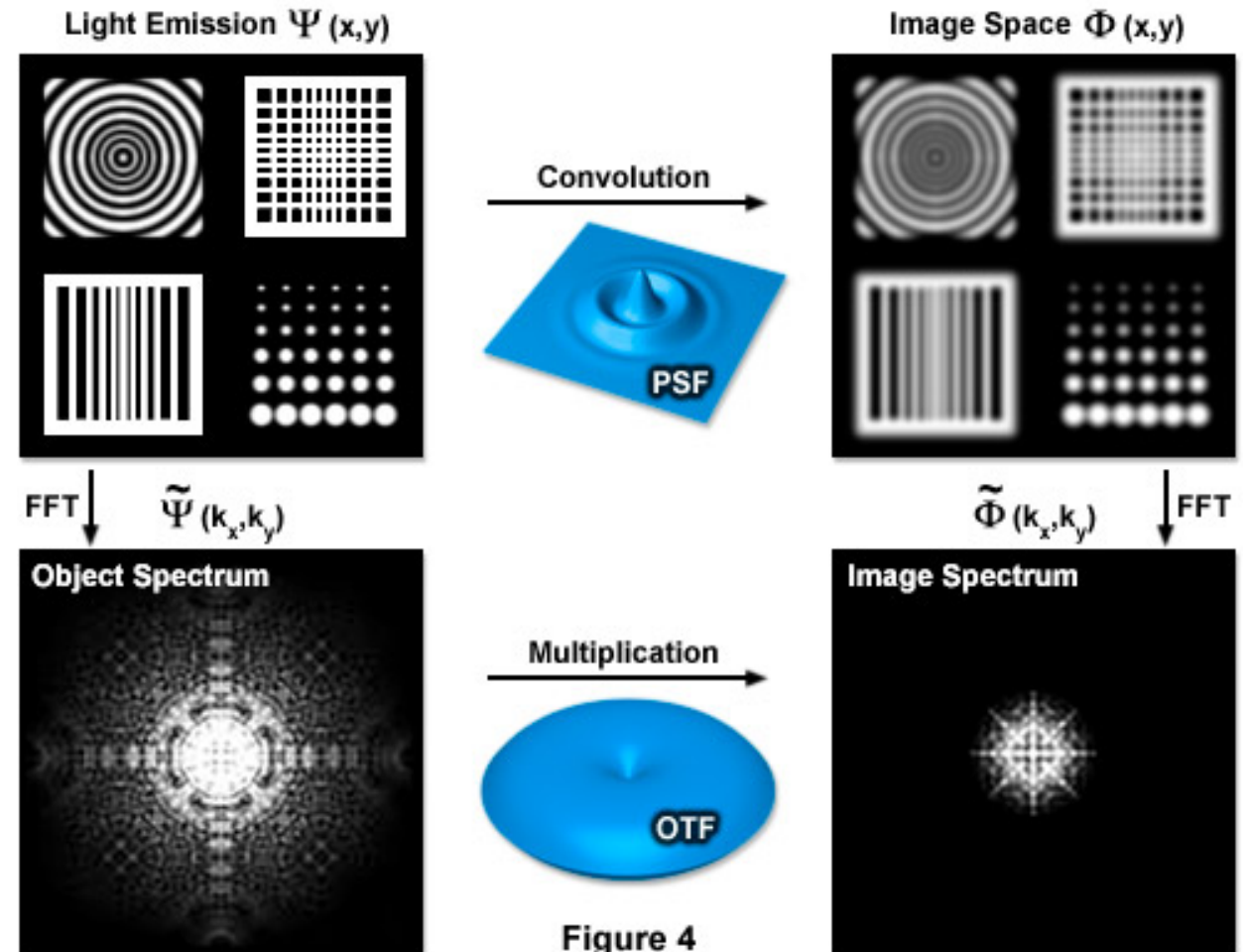
# Optical transfer function

$k$  = spatial frequency

$k_0 = 2NA/\lambda_{em}$  = maximum observable spatial frequency

- Optical transfer function is the Fourier transform of the PSF
- Frequency space representation of how the image is formed on the camera
- Convolution in image space = multiplication in frequency space

The Process of Image Formation in Fluorescence Microscopy

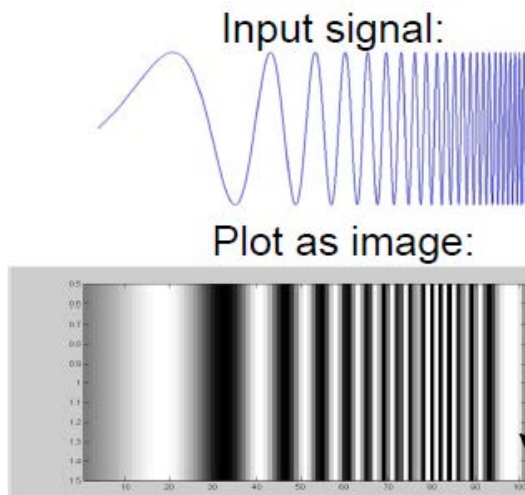




On to Matlab...

# Applications: Help with anti-aliasing

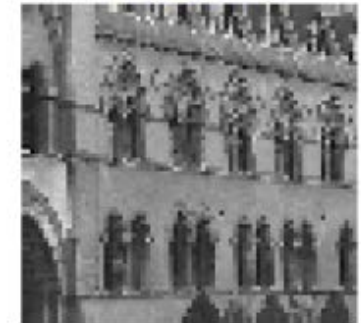
- Aliasing is a feature of sampling too slowly
- Adds frequencies not present in original signal
- Shows up in images as blotchy regions



Example



down sample by  
factor of 4

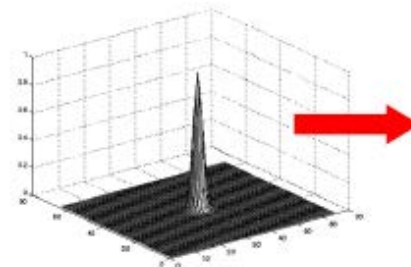


4 x zoom



convolve with  
Gaussian

\*



down sample by  
factor of 4

