Signal and Image processing A short intro

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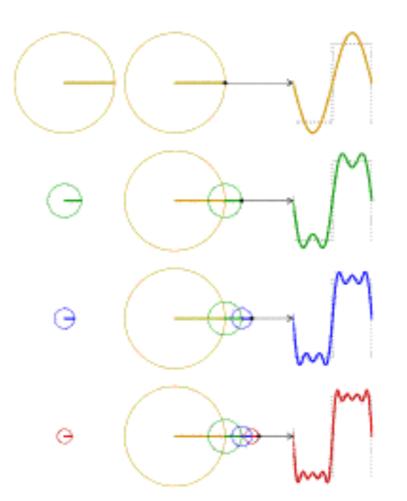
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When I say signal processing...

- Fourier series, Fourier transform, FFT
- A bit of image processing (but related to Fourier)
- A bit on image reconstruction (also related to Fourier)

Fourier Analysis what's it all about?

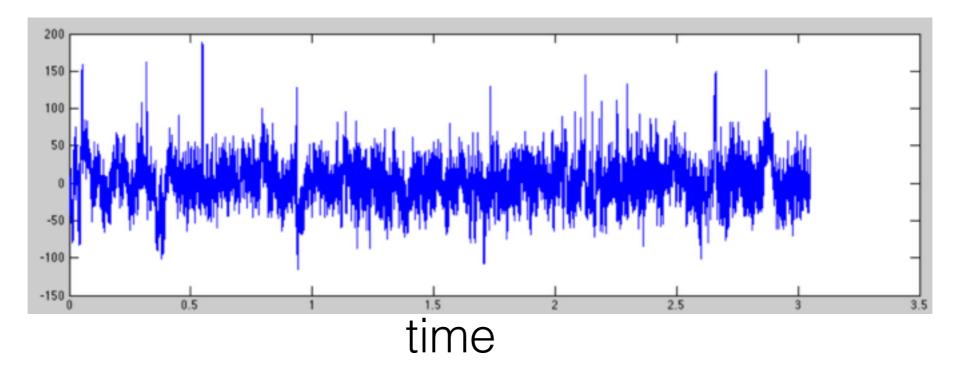
• Sines and cosines are building blocks

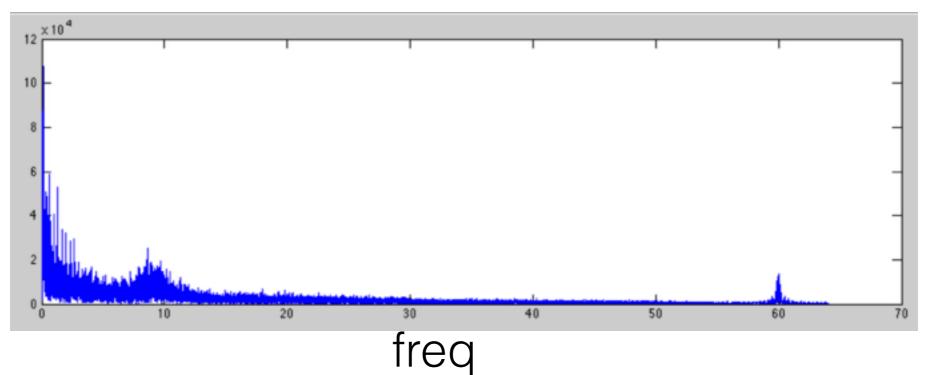


source: wikipedia

Why?

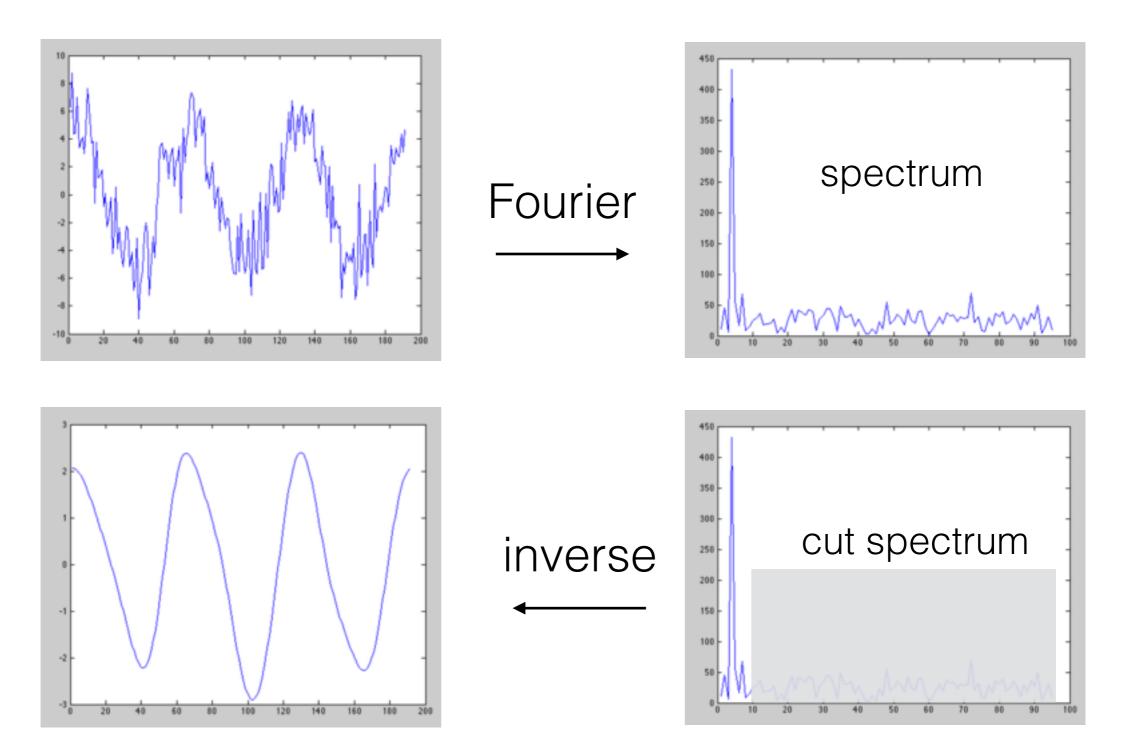
Frequency content





Why?

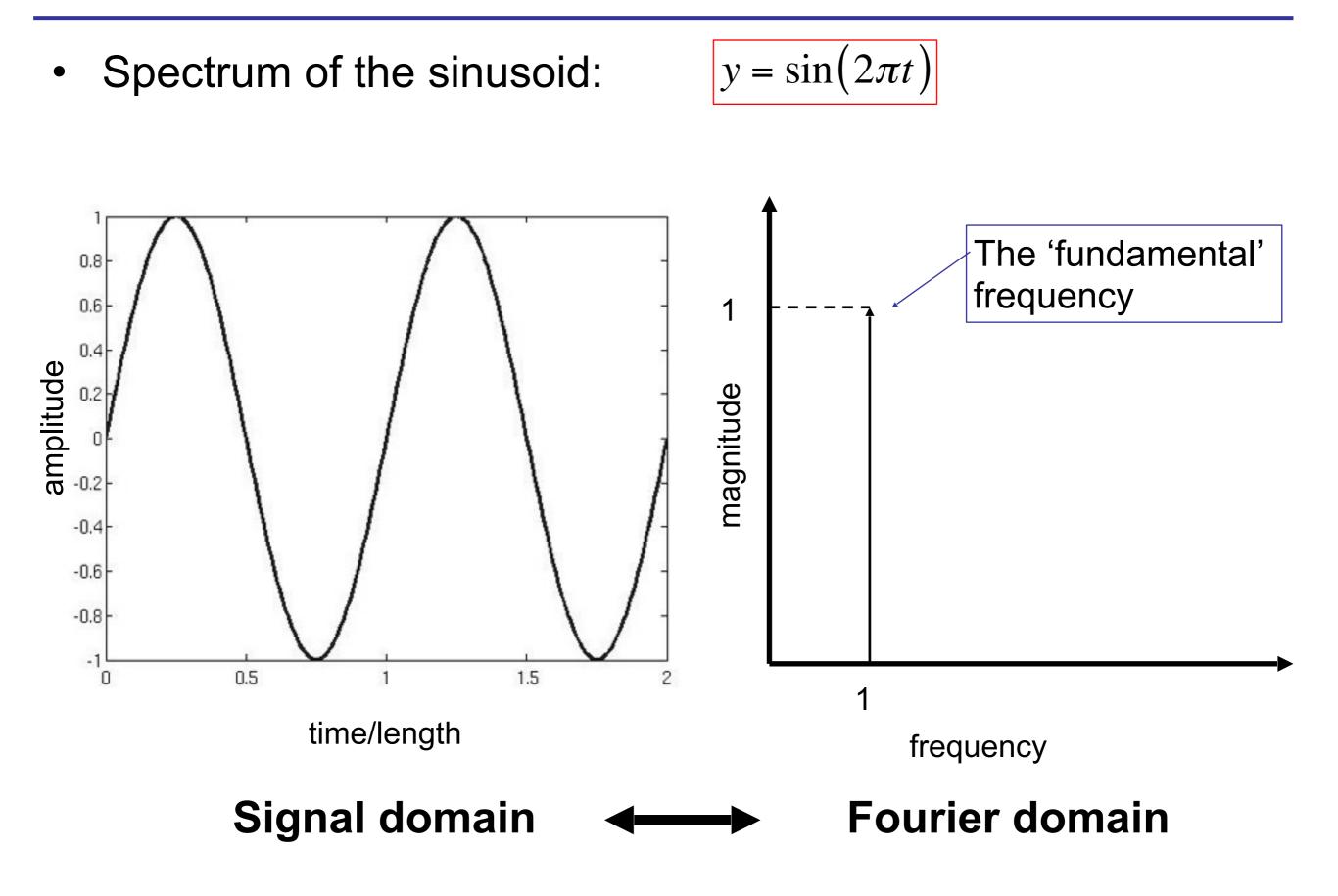
· Cleaning



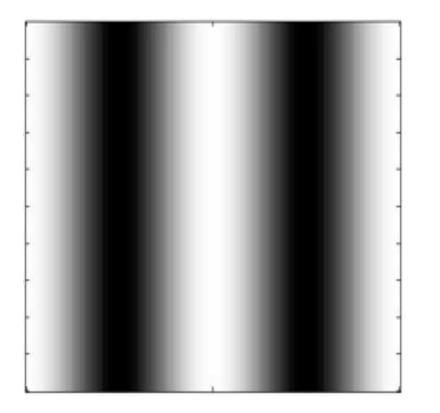
Why?

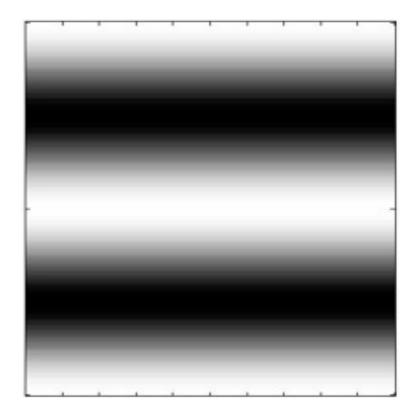
- Frequency content
- Cleaning (filtering)
- Computational efficiency
- Also very useful in abstract maths (beyond signal processing)

Fourier Series



• Define sinusoid going in x and y directions.



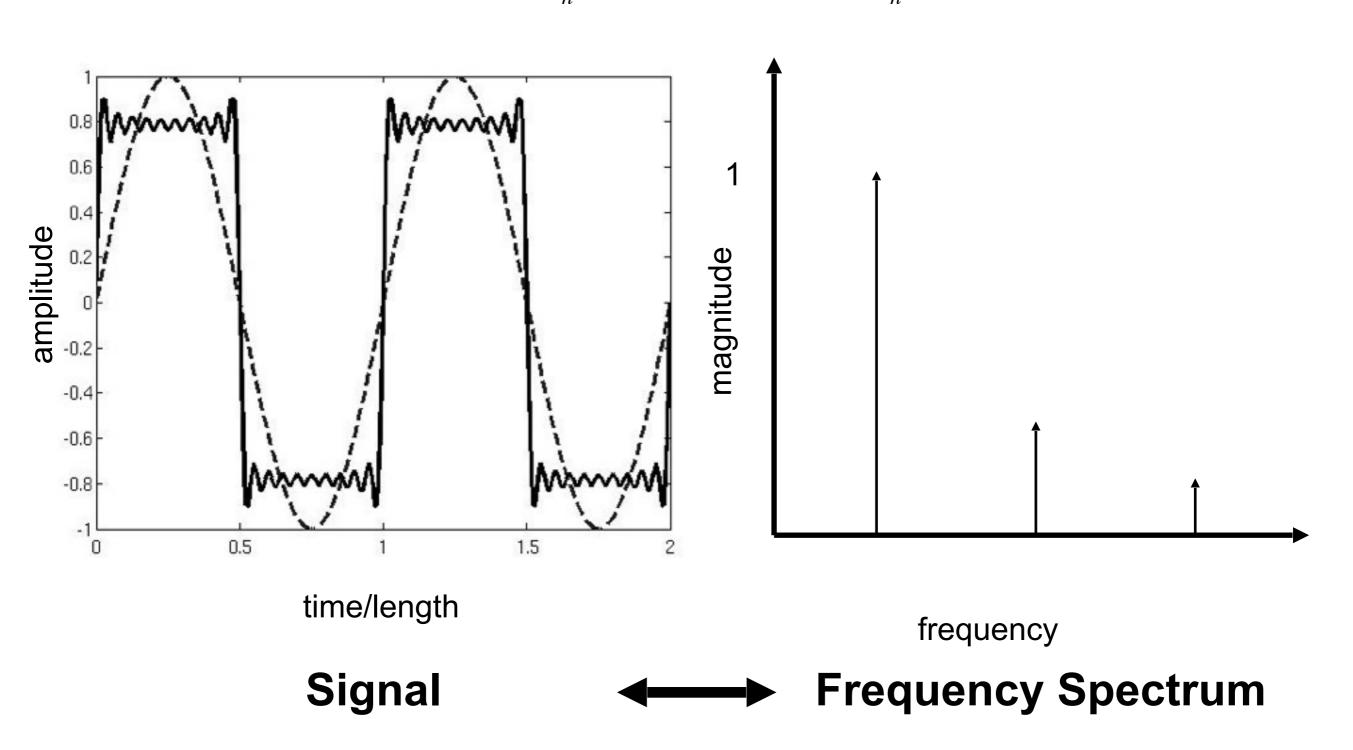


x frequency = 2 *y* frequency = 0 *x* frequency = 0 *y* frequency = 2

Fourier Series

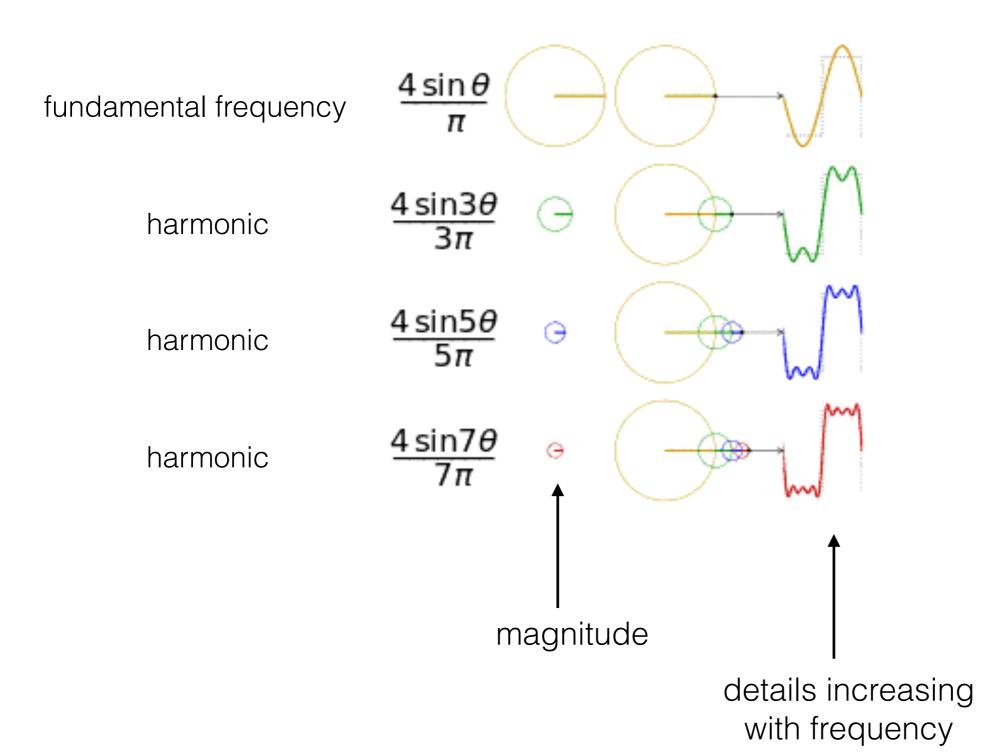
Approximate periodic signals with sines and cosines

• Keep going: $y(t) = \sum_{n} a_n \sin(nf \times 2\pi t) + \sum_{n} b_n \cos(nf \times 2\pi t)$



Fourier Analysis what's it all about?

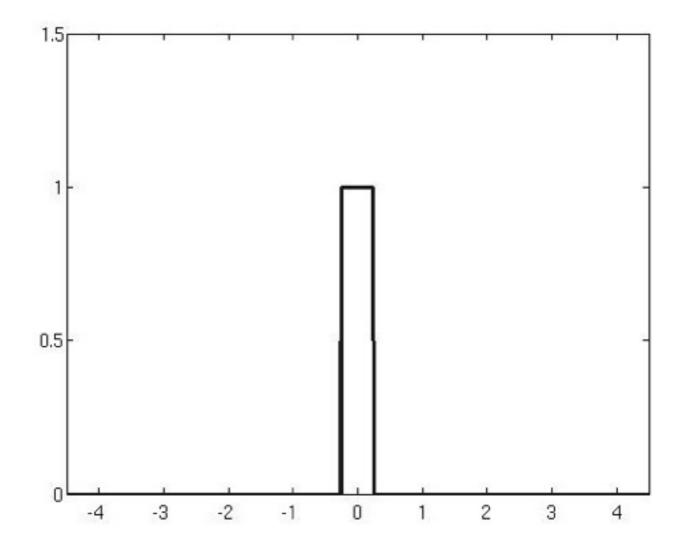
• Sines and cosines are building blocks



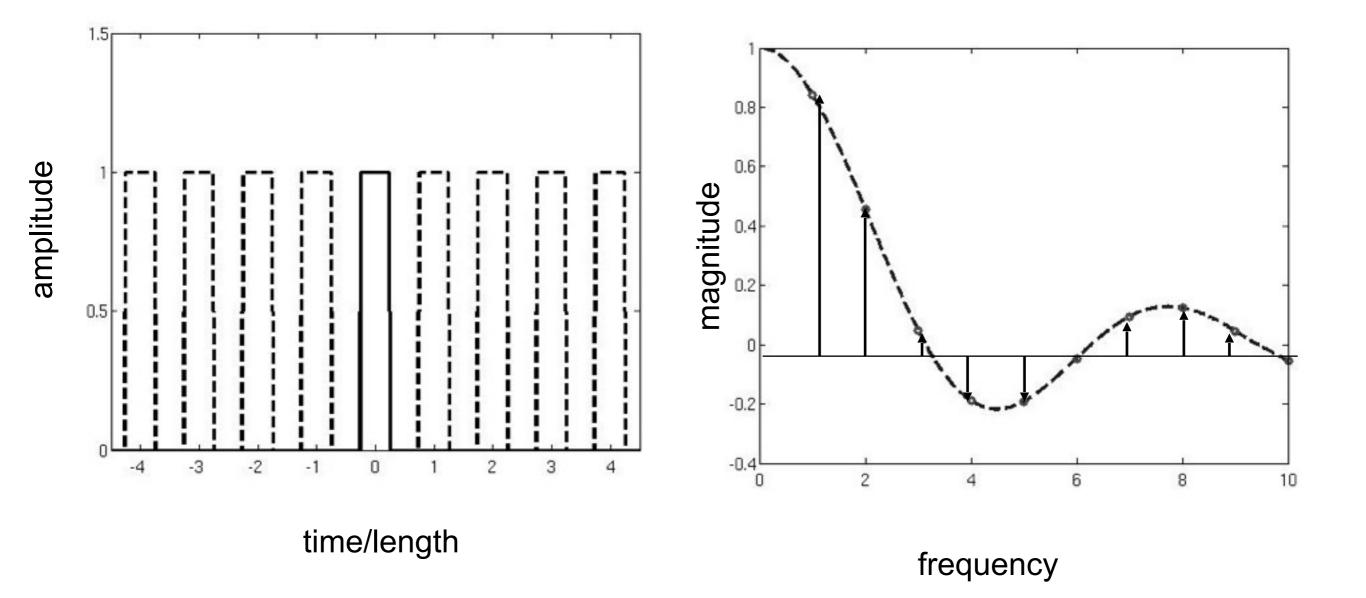
Fourier Transform

Approximate **non-periodic signals** with sines and cosines

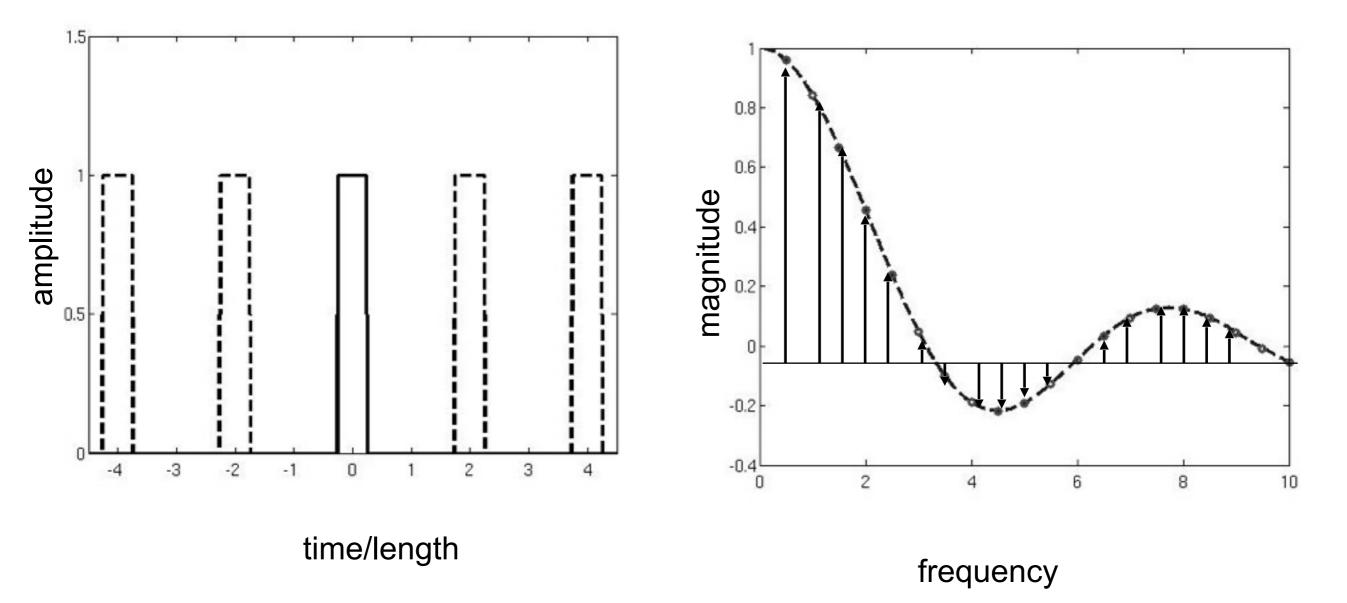
- Fourier Series good for periodic signals, what do we do if they are not?
 - Classic example: the top hat function



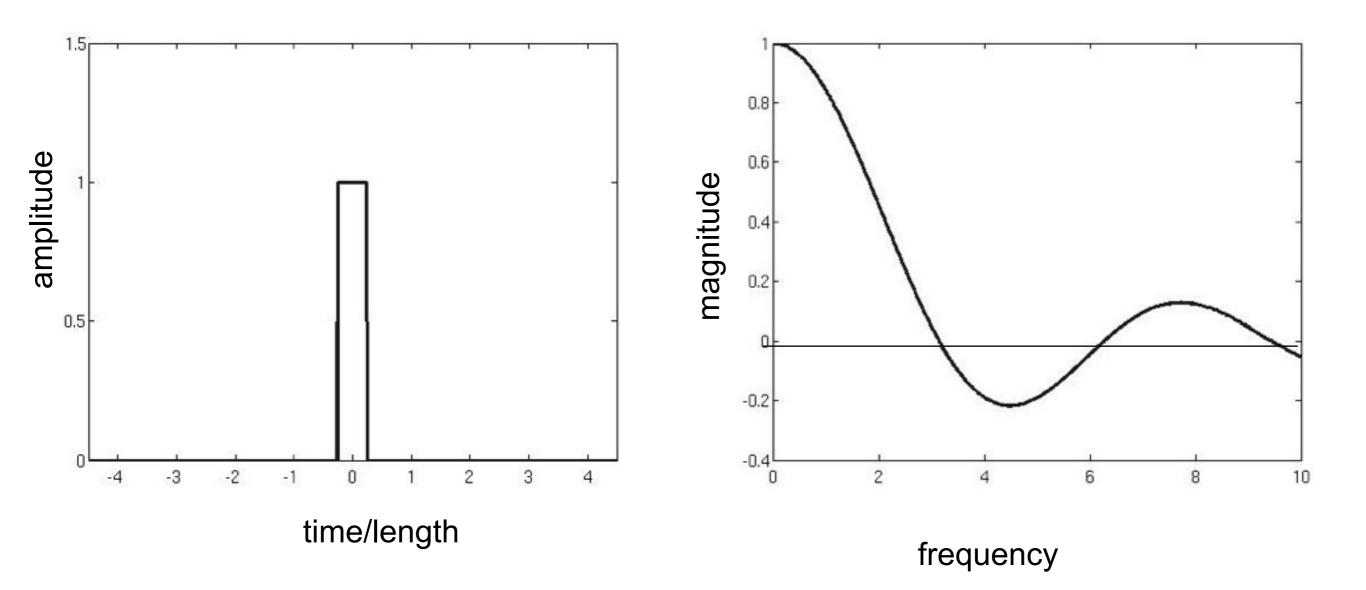
• Try making it periodic over a certain period:



 Make the spacing larger, so that there is a larger space in which only the top hat appears:



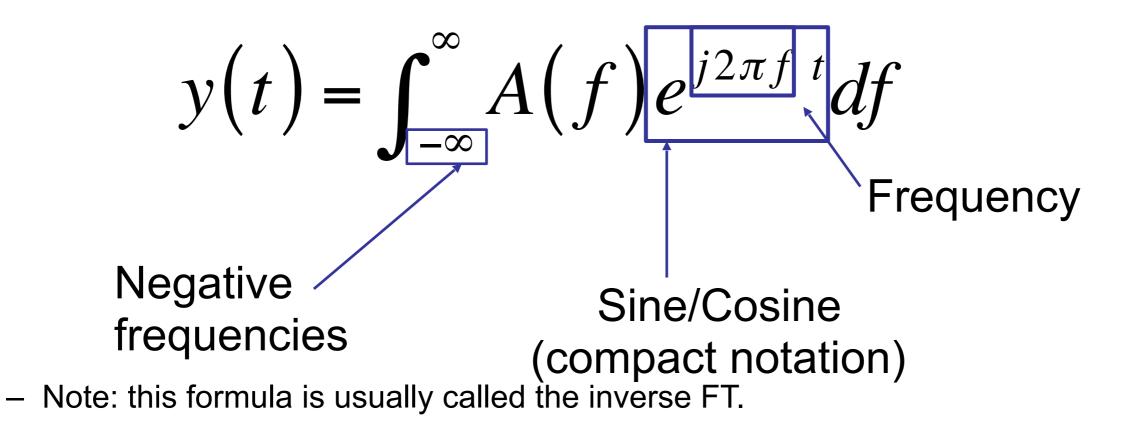
• In the limit (T = infinity) we get a smooth spectrum:



• Fourier series was a sum at specific frequencies:

$$y(t) = \sum_{n} a_{n} \sin(nf \times 2\pi t) + \sum_{n} b_{n} \cos(nf \times 2\pi t)$$

• Fourier transform is a sum over all frequencies:



• To find the frequency spectrum we need to do an integral:

$$A(f) = \int_{-\infty}^{\infty} y(t) e^{-j2\pi f t} dt$$

Basically: at frequency *f*, how much of $sin(f \ge 2\pi t)$ do I need?

 In practice we get computers do deal with this using the Fast Fourier Transform (FFT).

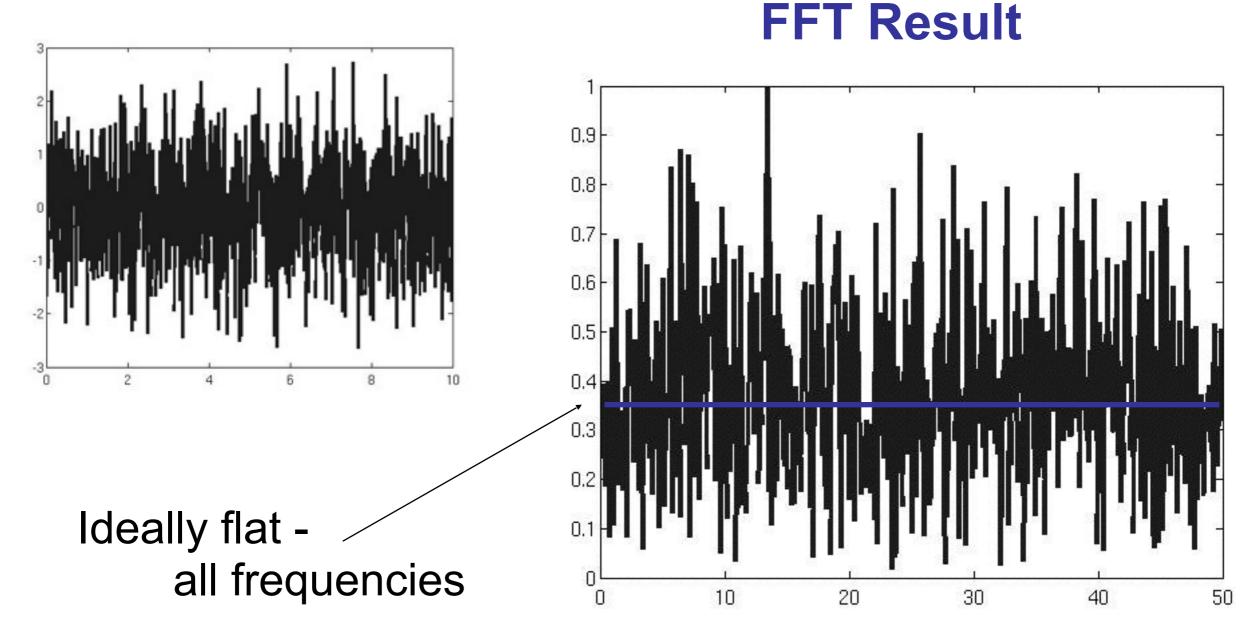
Fourier transform Mathematical operation Continuous

Fast Fourier Transform Algorithm Discrete

Fourier Transform

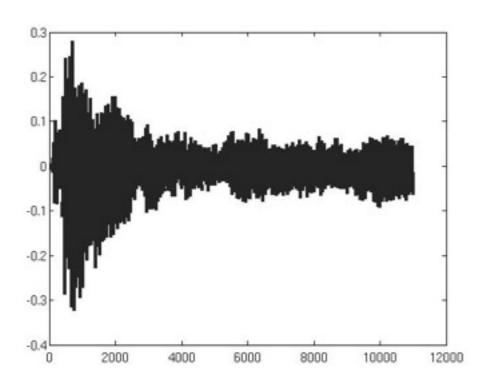
• Examples:

'White' noise

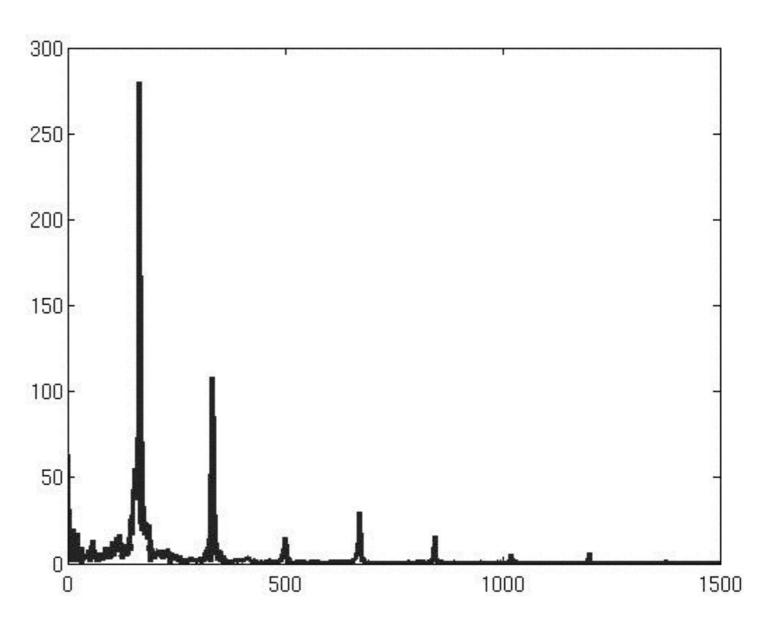


• Examples:

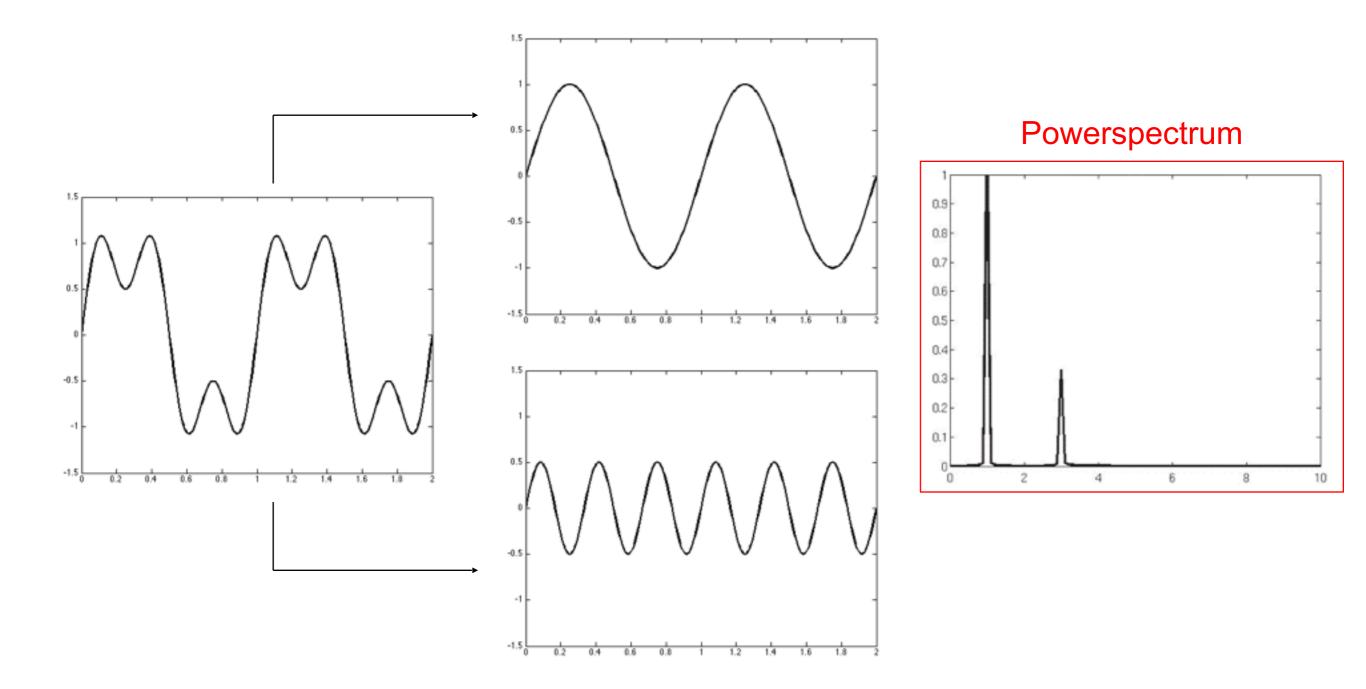
Piano note



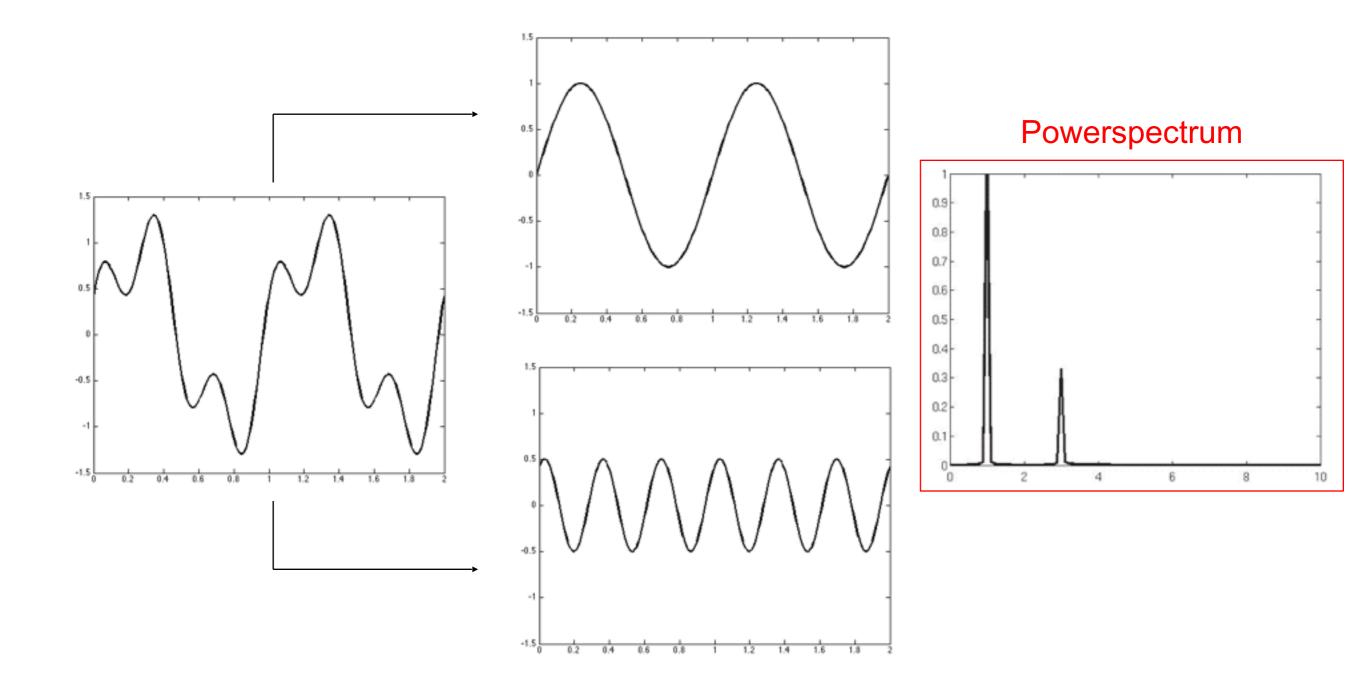
FFT Result

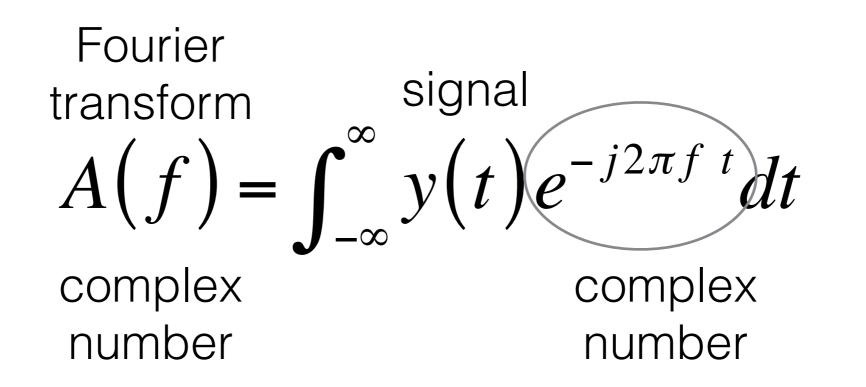


Fourier transform



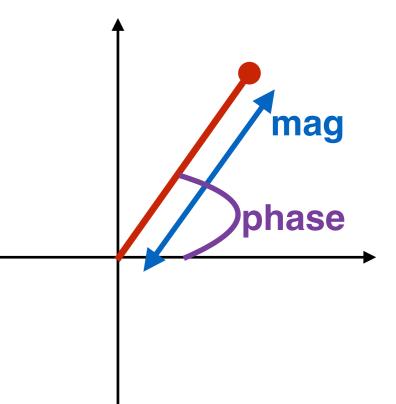
Power spectrum lacks phase information





Complex numbers easier to manipulate than sine and cosine functions

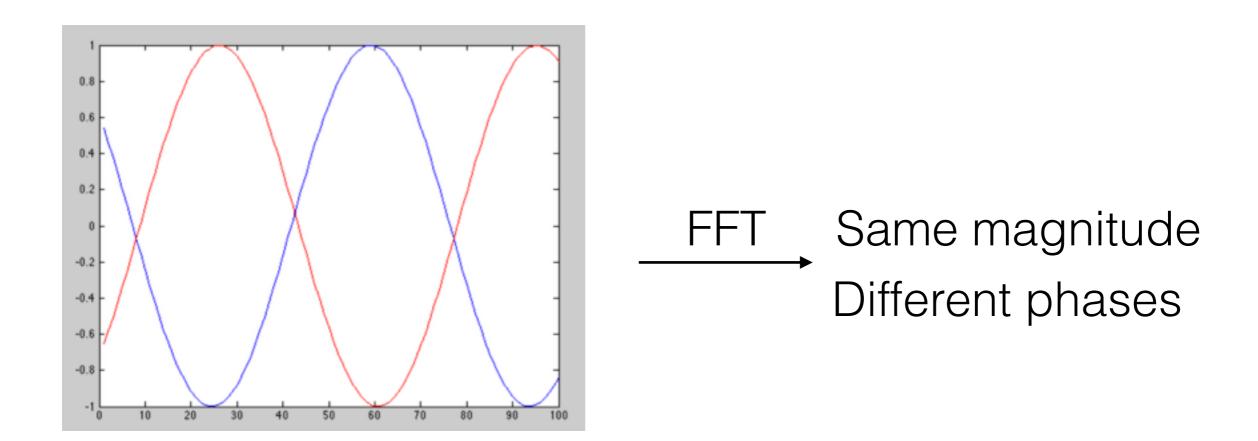
Information on phase and magnitude



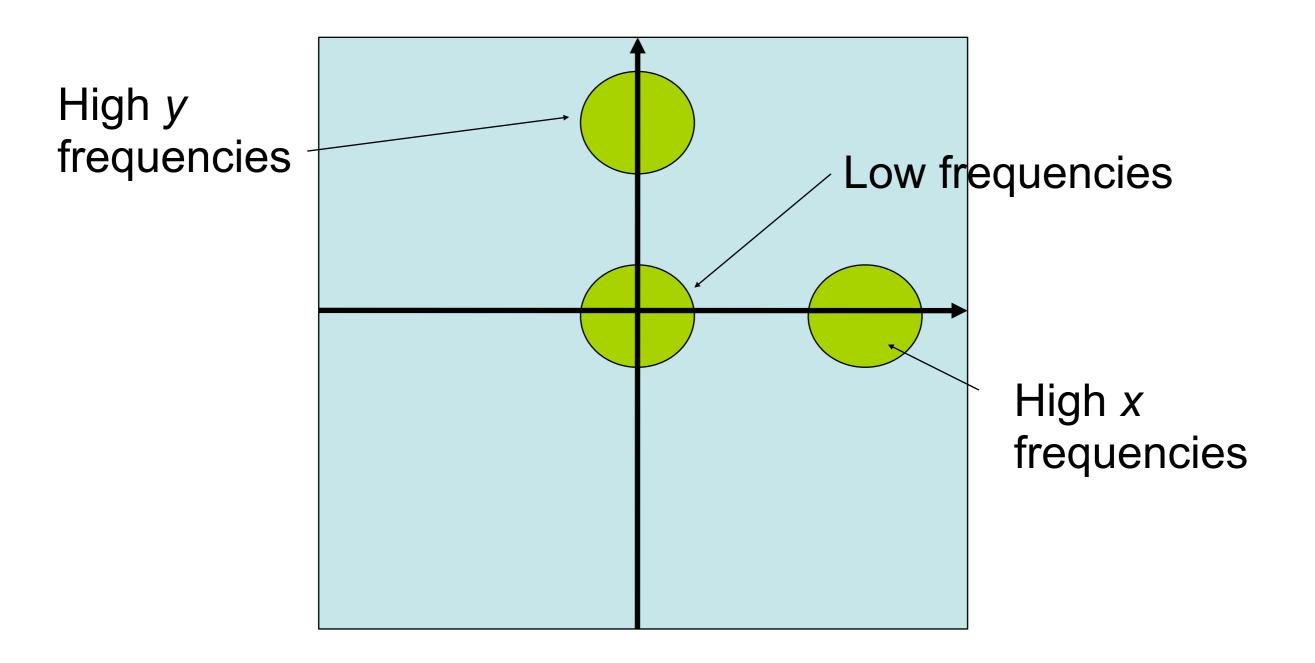
Magnitude, Phase

What?

Where?



- 2D signals, i.e. images have 2D Fourier transforms
- We now have x and y frequencies:



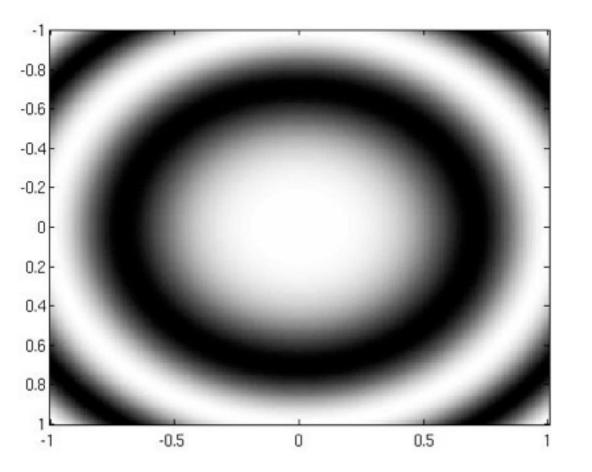
• Sinusoid in x direction

2D spectrum Image -1 -0.8 -0.6 -0.4 -0.2 0 0.2 0.4 0.6 0.8 1 -0.5 0.5 -1 0 1

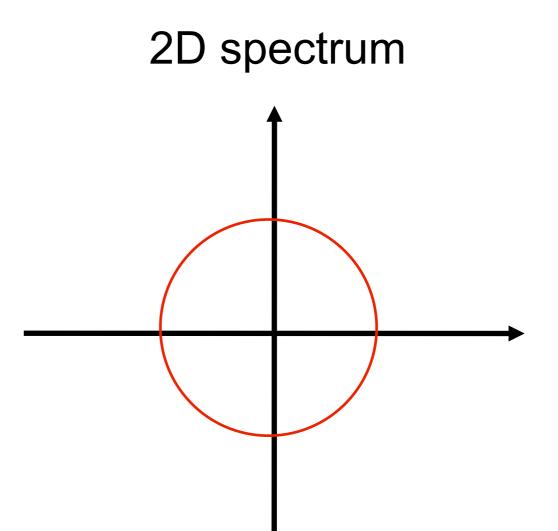
Sinusoid in x plus sinusoid in y direction

2D spectrum Image -1 -0.8 -0.6 -0.4 -0.2 0 0.2 0.4 0.6 0.8 14 -0.5 0.5 0 1

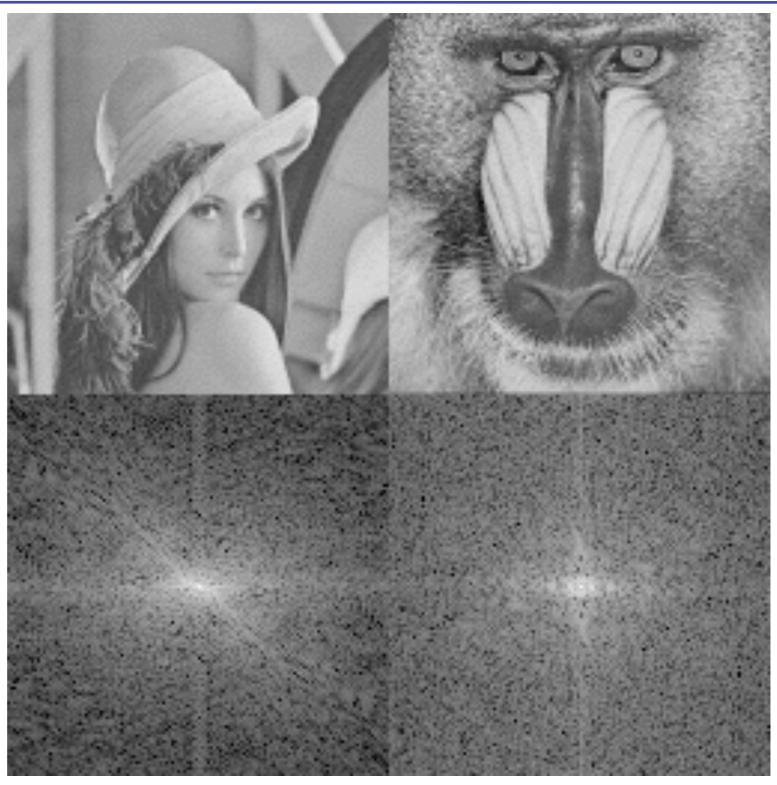
• Single frequency in all directions



Image



- Examples:
- Spectrum is 'bright' in the centre.
- Detail involves high frequency.
- Spectrum is symmetric.



http://www.cs.unm.edu/~brayer/vision/fourier.html

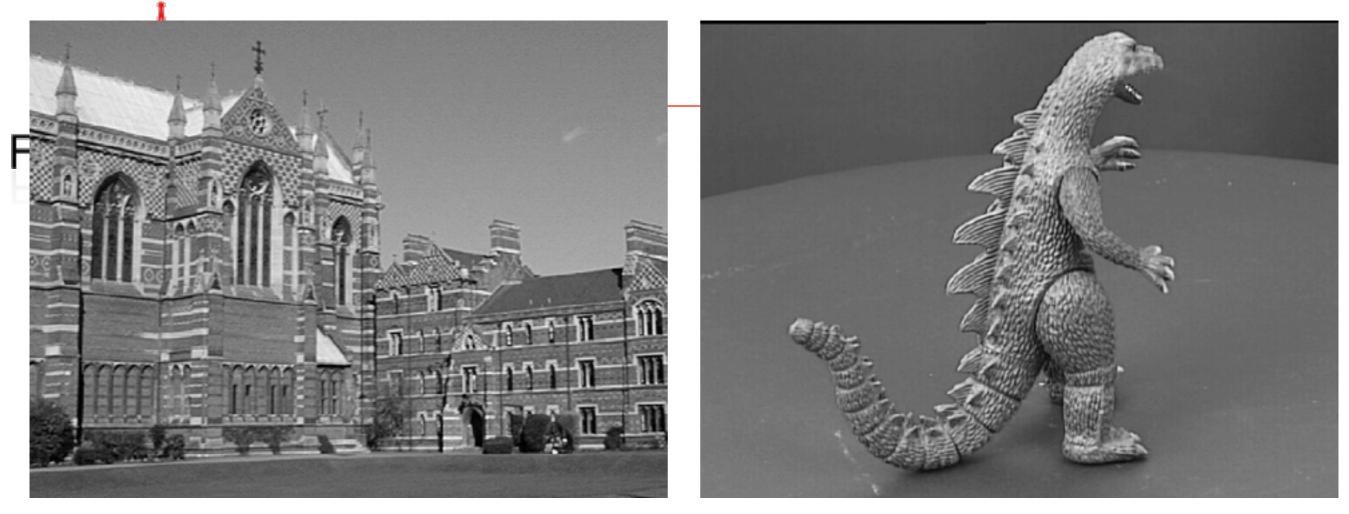


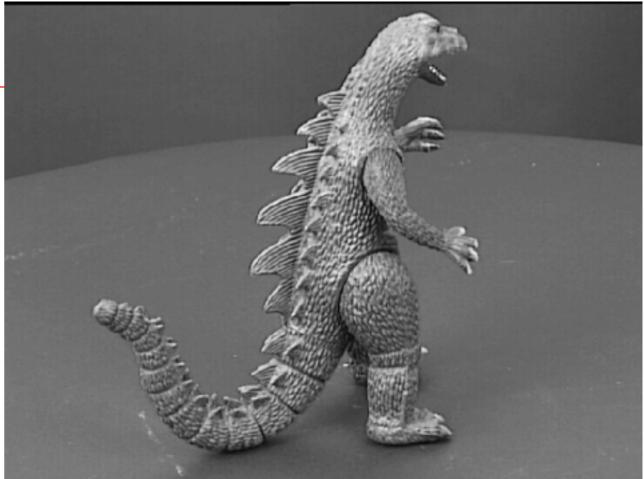
Image 1

Image 2

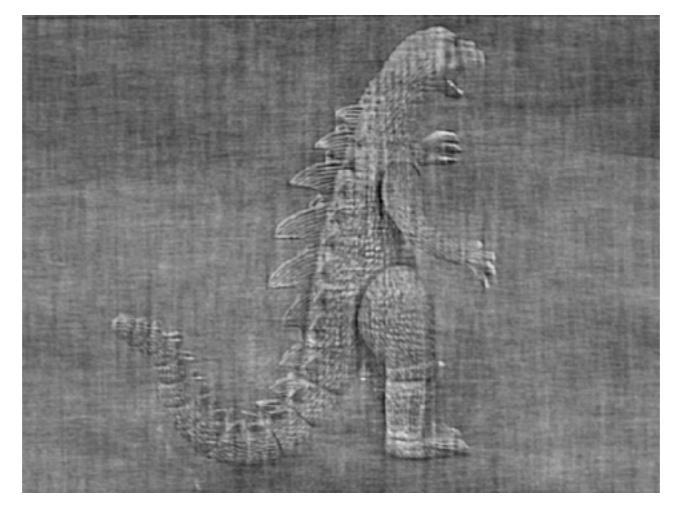
Courtesy of JM Brady

What kind of image would we get if we mix the phase and magnitude from these?



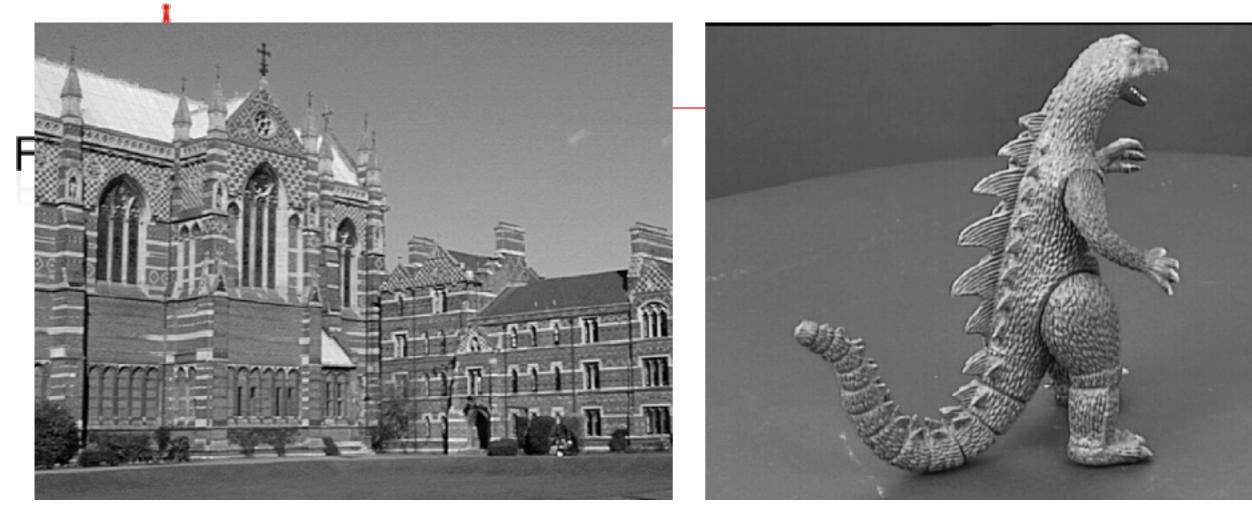


A: Phase or B: Magnitude

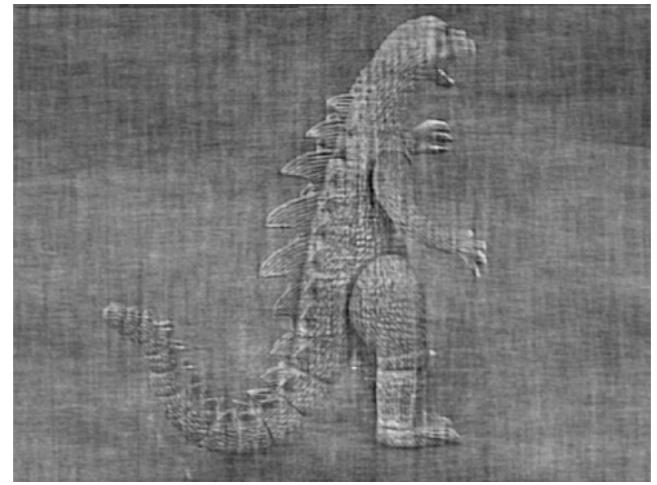


A: Magnitude or B: Phase

Courtesy of JM Brady



B: Magnitude

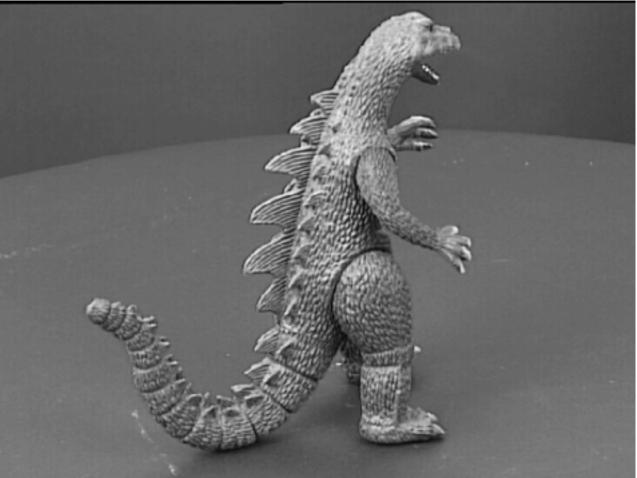




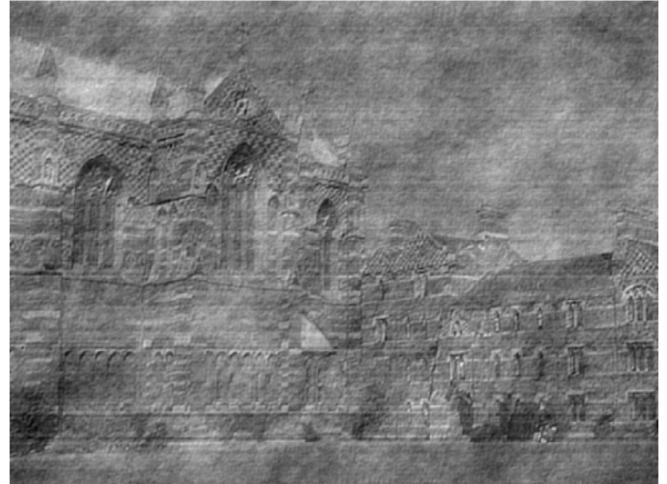
B: Phase

Courtesy of JM Brady





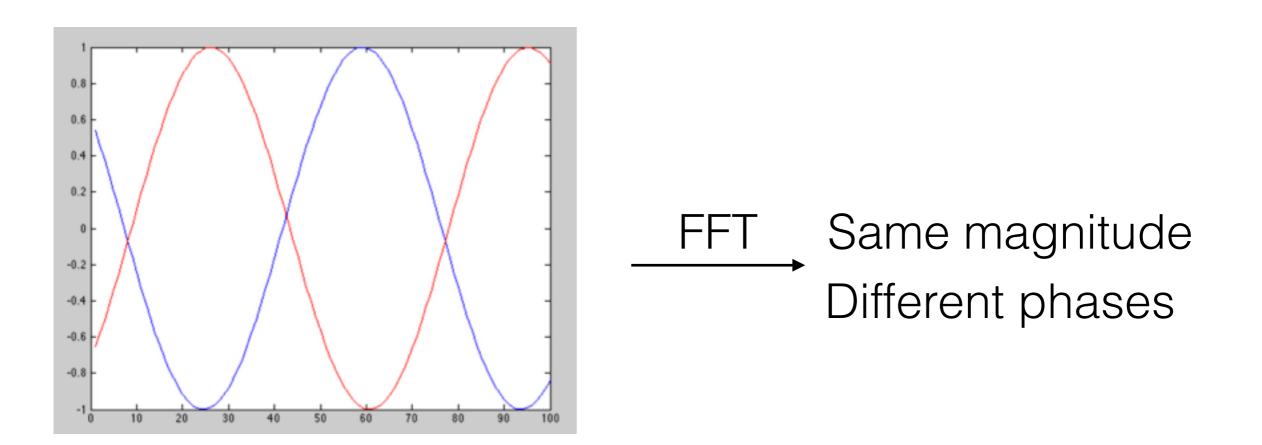






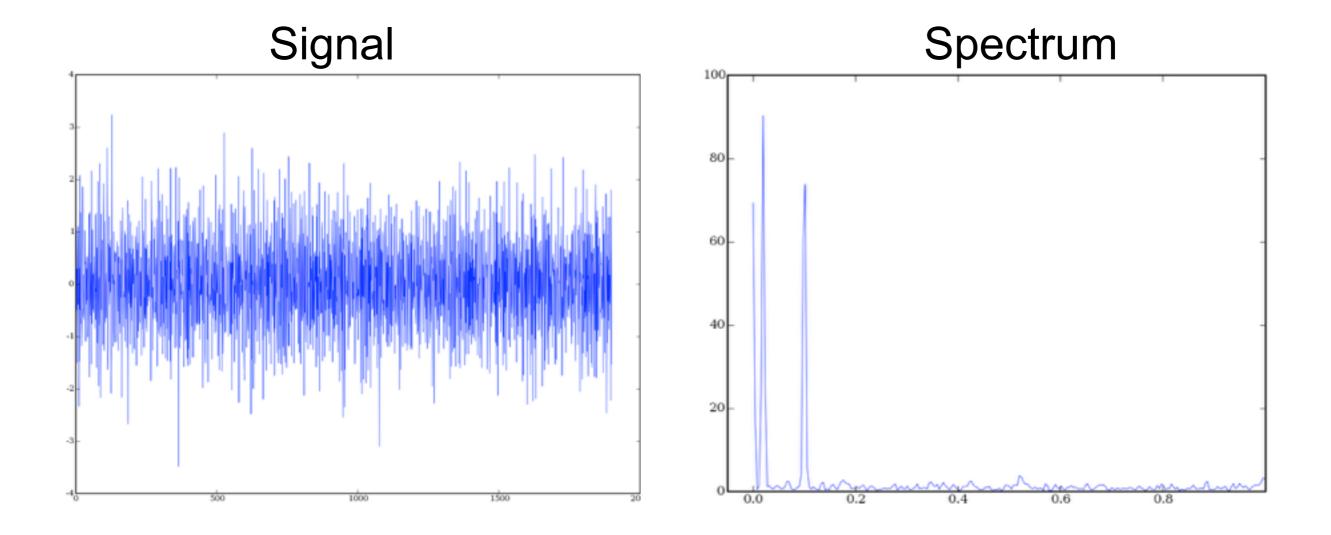
Courtesy of JM Brady

Magnitude What? Phase Where?

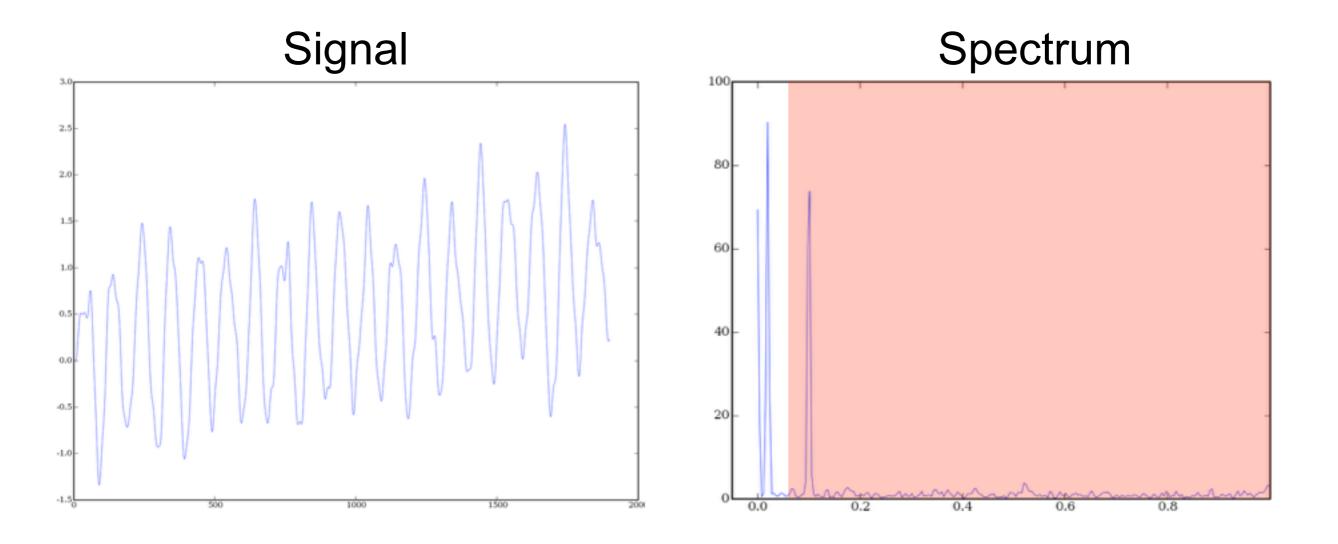


Filtering

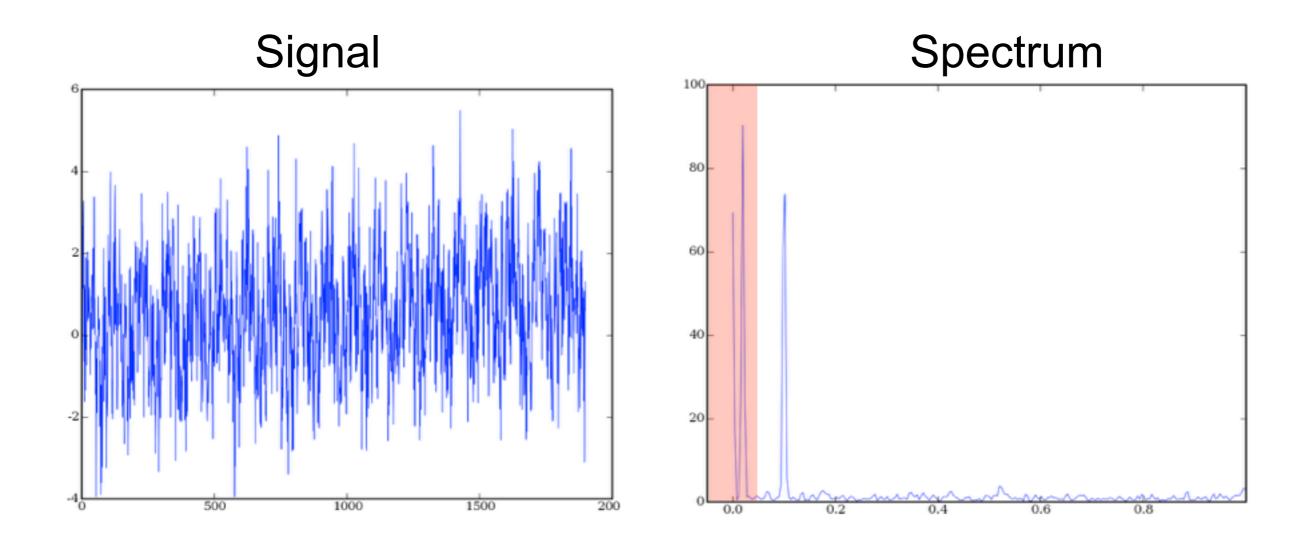
- In Fourier signals are a mixture of different frequency components.
- Often we want some components and not others.



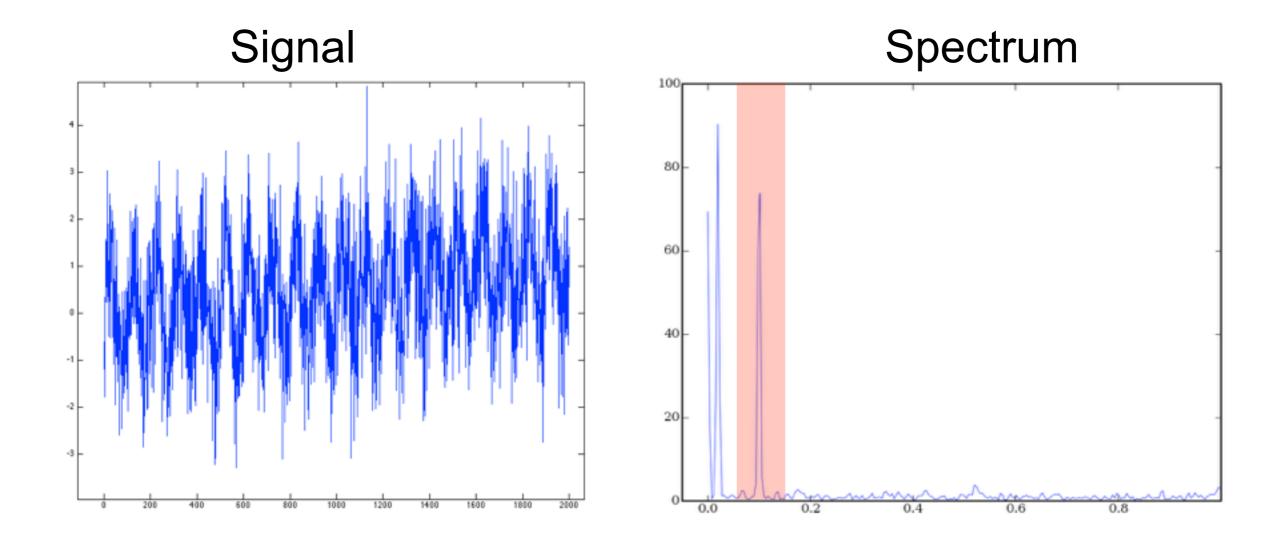
- I want to get rid of high frequency noise component.
- Low Pass filter throw away high frequencies



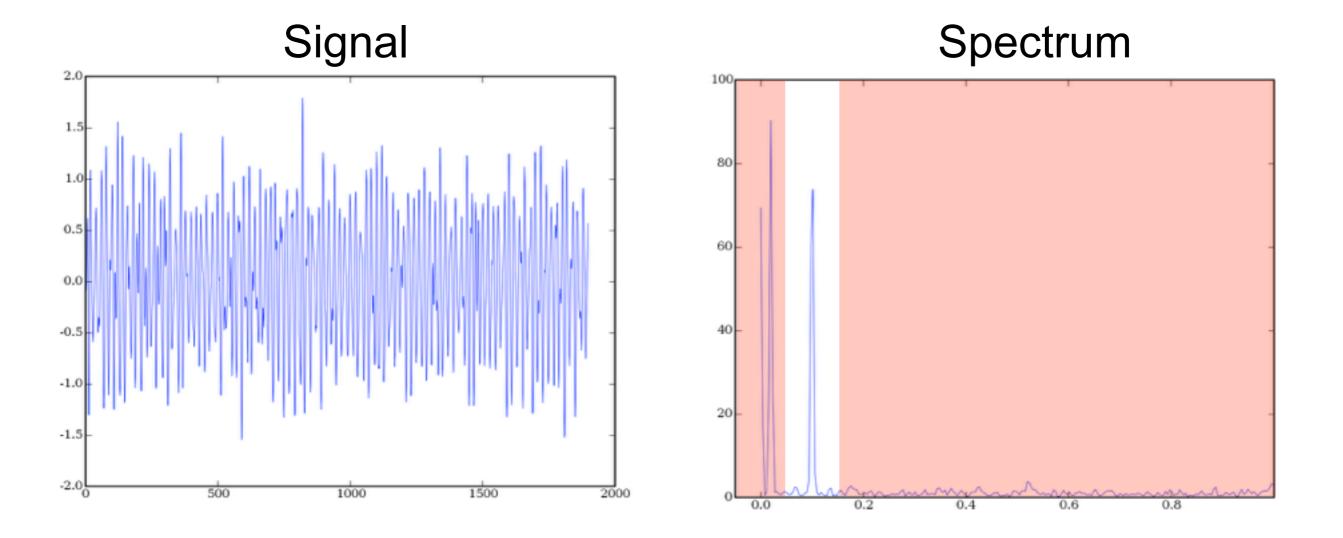
- I want to get rid of low frequency nosie/drift.
- High Pass filter throw away low frequencies.



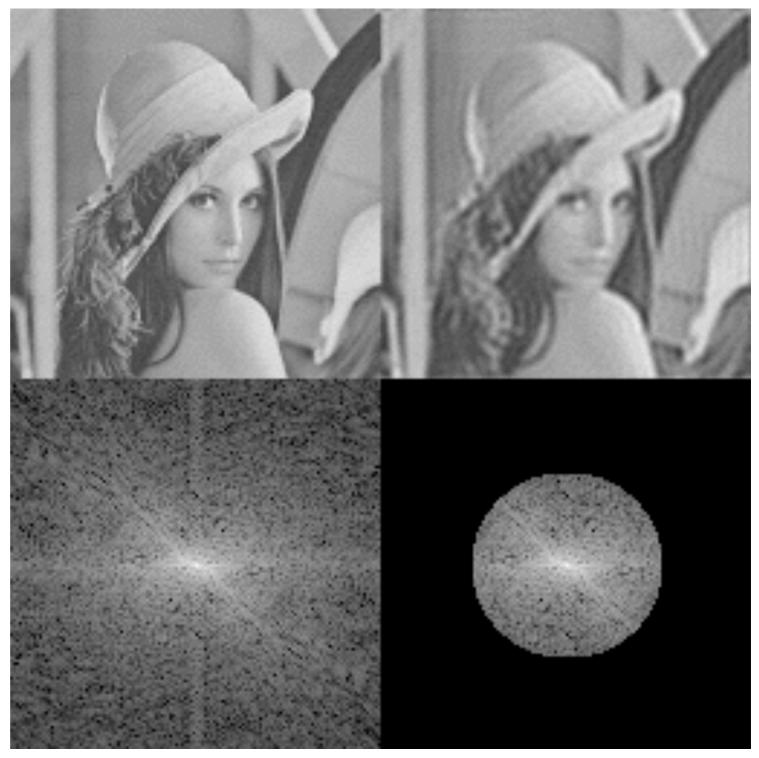
- I want to get rid of that annoying component at [pick a frequency] (e.g. mains noise).
- Band Stop filter let everything through except...



- I want to get rid of all this other stuff that is not my signal (not always this simple!)
- Band Pass filter get rid of everything but...



- Same principles as in 1D.
- Low Pass
- Remove high frequencies.
- Loose detail.

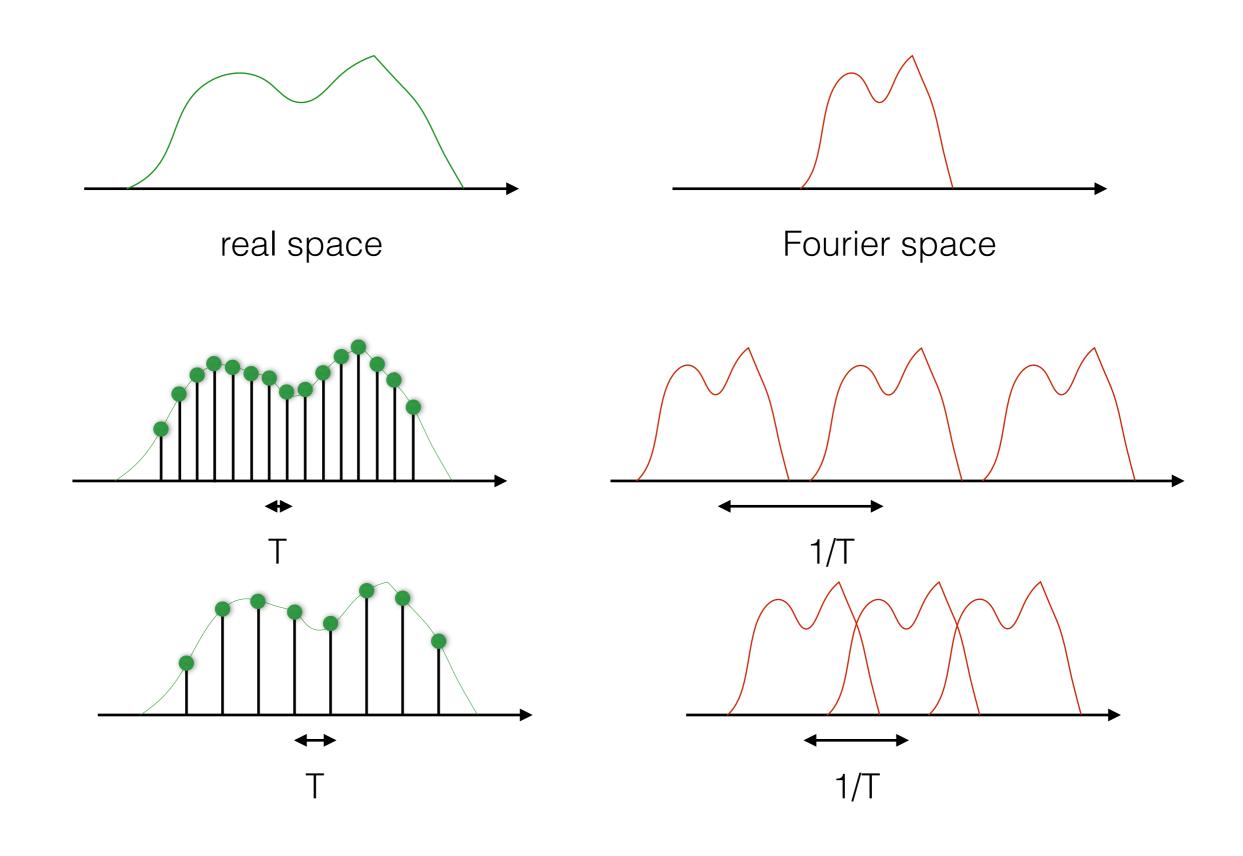


http://www.cs.unm.edu/~brayer/vision/fourier.html

Sampling, aliasing

--> details in the practical

Sampling in real space is like repeating in Fourier space and vice versa

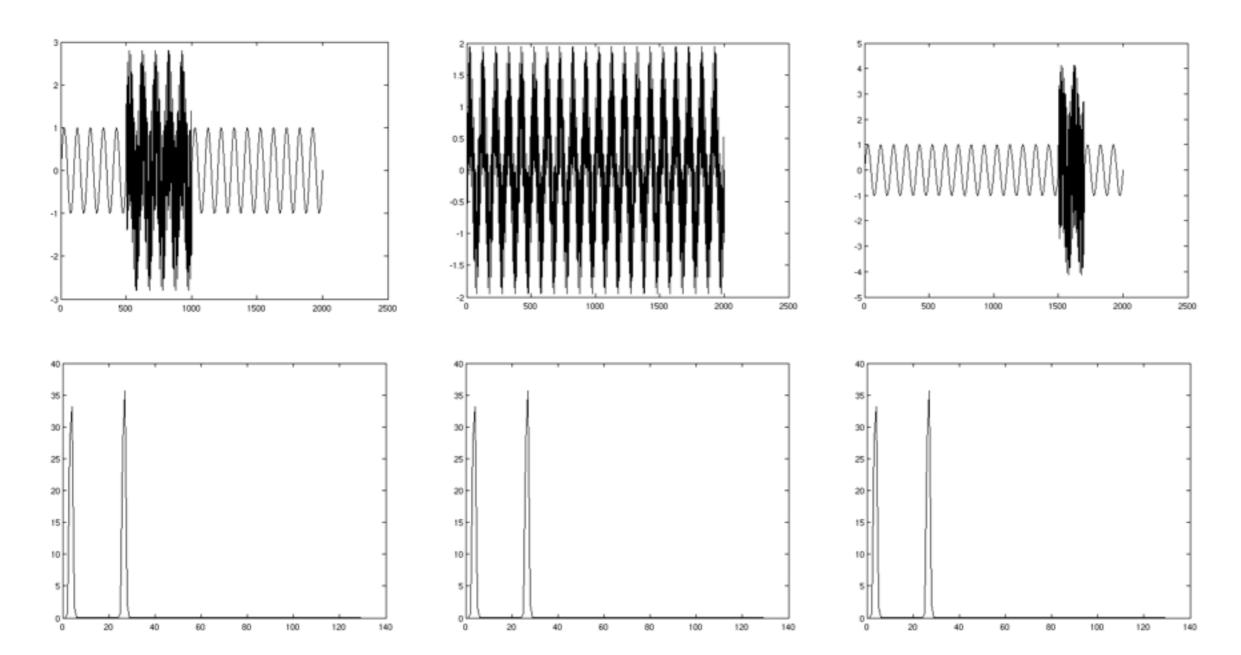


Time-frequency analysis

--> details in the practical

Time-frequency

• Three different signals - same frequency spectrum.



Time-frequency

• Need a 'dynamic' time-frequency representation.

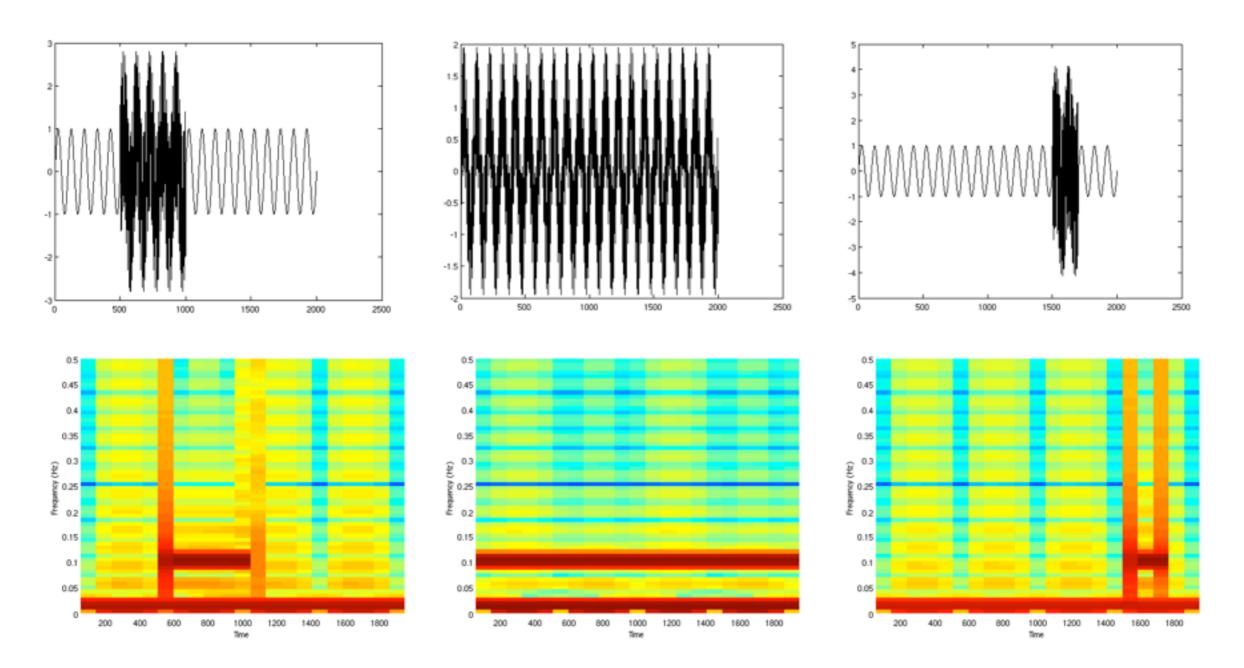
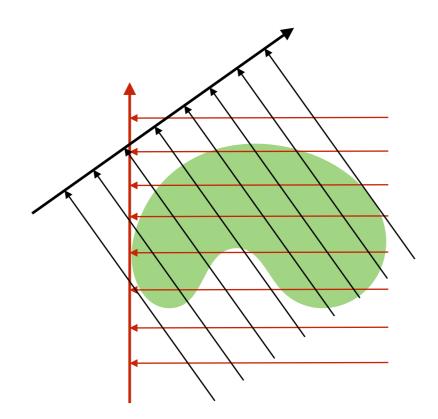


Image reconstruction

--> details in the practical

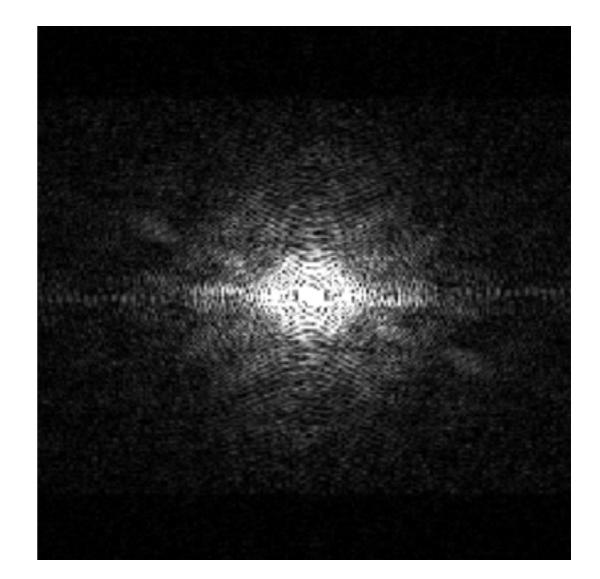
Radon transform (X-ray CT)



Magnetic Resonance Imaging



Object of interest



Measurement (Fourier space)

www.fmrib.ox.ac.uk/~saad/GP/fourier.html

Prac Maths - Signal and Image Processing Practical

Overview

This practical requires Matlab. Go through the instructions and execute the listed commands in a Matlab command window (you can copy-paste). Don't click on the "answers" links until you have thought hard about the questions. Raise your hand if you need help, but perhaps try first the "help" command in Matlab if you are unsure about Matlab syntax issues.

Contents:

- Fourier analysis
 Learn basics of FFT in 1D (signals) and 2D (images)

 Filtering
- Learn to implement linear filters using convolution
- Image reconstruction Fourier/Radon transforms and image reconstruction

Fourier analysis

To start with, let us create some simple 1D signals and examine their Fourier transforms. In this first example, we will check the frequency content of a simple periodic signal.

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Generate a cosine signal with given magnitude, frequency, sampling rate, and duration:

```
mag = 2; % magnitude (arbitrary units)
freq = 5; % frequency in Hz
samp = 100; % sampling rate in Hz
t = 0:1/samp:1; % time (ls of data)
t = t(1:end-1); % remove last time point
N = length(t); % store the number of time points
x = mag*cos(2*pi*freq*t); % the signal equation
figure
plot(t,x,'.-');
```

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