

#### Lab 7 Data Acquisition, Aliasing, FFT, and the Optical Fourier Transform



Wente Yin

February 26<sup>th</sup>, 2015





#### Agenda

- DAQ
- DFTs
- DFTs in MATLAB
- Frequency Shifting
- Gibbs Phenomenon
- Aliasing
- Advice





Input

Output





## Data Acquisition (DAQ)

- Two Sets of Inputs
  - Analogue (for...analogue)
  - Digital (for data transfer, digital sensors, etc.)
- Requires Proper Drivers for Function







#### Data Acquisition (DAQ, Cont'd)







#### **Discrete Fourier Transforms**

- Computers are Discrete in Nature
  - Discrete → Not continuous, requires individual input (samples)
  - Continuous functions would require infinite amount of RAM, which cannot be downloaded (non-countable infinity).







• Continuous Fourier Transform

$$F(\xi) = \int_{-\infty}^{\infty} f(x) \exp(-j2\pi x\xi) dx$$
$$f(x) = \int_{-\infty}^{\infty} F(\xi) \exp(j2\pi x\xi) dx$$

azengineering





• Discrete Fourier Transform









- Functions Must be Discretely Sampled
  - Allows a simulated Fourier transform
  - DFT is cyclic (periodic), and multiple cycles are required for the function to appear in its appropriate dimensions
  - Several issues arise with MATLAB's implementation of the DFT







#### • DFT in MATLAB

- *fft* (Single-Dimension), *fftn* (N-Dimension)
- The *fft* function will carry out the DFT in a manner that places zero frequencies to the left (DC Bias, Single Dimension) or corners (Two-Dimension)
- *fft* also limits DFT to a single cycle (appears more like a continuous FT), which is the source of problems like aliasing







- *fft* (Fast Fourier Transform)
  - Function makes use of symmetry of input to reduce computational complexity from N<sup>2</sup> to Nlog<sub>2</sub>(N)
  - As such, it is best to have inputs sized to powers of 2.







### Rules of DFTs

- Inherent Rules of Discrete Fourier Transforms
  - 1. The entirety of the span of a FT is equivalent to the inverse of the sampling rate, centred if necessary.
  - 2. Said sampling rate also determines normalisation of the dependent axis, due to nature of the summation in DFTs
  - 3. Always sample by a power of 2
  - 4. It is recommended the final value of an even function be dropped when performing DFTs (cyclic nature)







1. The entirety of the span of a FT is equivalent to the inverse of the sampling rate, centred if necessary. Inverse is true.



 $= \frac{1}{ALL \ OF \ THE \ NORMAL \ SPACE}$ 





2. Said sampling rate also determines normalisation of the dependent axis, due to the nature of the summation in DFTs

Normalisation Factor =  $\Delta x$ 







- 3. Always sample by a power of 2
  - MATLAB adores symmetry and improves calculation speed for large arrays
  - MATLAB also has crazy, mysterious, trademarked, copyrighted, patented, and highly lucrative optimisation techniques
  - Guideline for efficiency and good form







• Effects of Optimisation Apparent with *tic/toc* 

Average Time for Powers of 2 1.0e-03 \*

0.0304 0.0295 0.0384 0.1844

Average Time for Powers of 2 Minus 1 1.0e-03 \*

0.0257 0.0327 0.0451 0.5386





- It is recommended the final value of an even function be dropped when performing DFTs (cyclic nature)
  - Final point is unnecessary and potentially problematic in repeating signals







#### DFTs in MATLAB

• Plot a Gaussian Function

$$x = linspace (-15, 15, 257);$$
  

$$x = x(1: 256);$$
  

$$y = \exp\left(-\frac{x^2}{2}\right);$$

azengineering









• Use the *fft* Function, Plot Reals

Y = fft(y);<br/>plot(real(Y));











- Previous Figure Obviously Incorrect
  - Fourier transform of a Gaussian function is always another Gaussian...which it was obviously not.
  - Strange figure is because of the DC bias of the *fft* function
  - We must place zero frequency in middle







## **Frequency Shifting**

- Shifting Frequencies
  - fftshift (A)
  - Places zero frequency between negative and positive frequencies, essentially "flipping" the result







• Use Function on Previous Result, Plot Reals

$$Y1 = fftshift(Y);$$
  
$$plot(real(Y1));$$











- Still Horribly Wrong
  - Oscillating amplitudes indicative of phase issues risen from input Gaussian before *fft*
  - Can be fixed by applying *fftshift* to the input as well
  - Allows for function to be represented properly in time, which is the preferred input of the *fft* function, fixing phase issues







• Apply *fftshift* to Original Gaussian

• Now apply *fft*, Shift to Centre

$$Y2 = fftshift (fft(y1);$$
  
 $plot (real(Y2));$ 















# Scaling

- Axes Currently Convey no Useful Information
  - x-axis indicates relative position of value plotted
  - y-axis requires normalisation
  - We must understand the nature of normal and frequency space







# Scaling (Cont'd)

- Applying Scaling to Result Results in Proper Results
  - The correct way to do an *fft* x = linspace(-15, 15, 257);x = x(1:256);delx = x(2) - x(1);freq = linspace (-1/(2 \* delx), 1/(2 \* delx), 257);freq = freq (1:256);y1 = fftshift(y);Y2 = fftshift(fft(y1) \* delx;plot (freq,real(Y2));









## Gibbs Phenomenon

- Occurs Near Sharp Edge in Fourier Space
  - "Jump" apparent in Fourier transform
  - As number of samples increase, "jump" does not go away, merely approaches infinitely closer to the discontinuity
  - Always hovers at 1.09 magnitude of max
  - Occurs as the DFT is carried out with a limited number of frequencies of sines and cosines, which are incapable of representing the discontinuity











# Aliasing

- Sampling Frequency of DFT > Nyquist Frequency
  - Nyquist frequency = 2x Sampling Frequency of Function
  - Nyquist frequency is regarded as minimum sampling frequency required to reconstruct a function
  - If sampling frequency of DFT is less than ideal, an identical function of lower frequency may be observed







- Demonstration
  - Cosine was sampled and fitted with a curve fit to give better appearance (do this in radiometry or any class with Barrett or Dubin and burn like the unwashed heathen you are)
  - Frequency of the cosine function was then increased, but sampling for the DFT was not (this is dangerous in optics)
  - Perceived frequency in frequency space was then plotted















DFT at Various Frequencies





#### Advice

- Low-Pass Filter  $\rightarrow$  Zero Higher Frequencies
- High-Pass Filter  $\rightarrow$  Zero Lower Frequencies
- Reconstruct to See Results: Inverse DFT
- Use "help" Command in MATLAB to Understand "interp" and "interp2" or learn to zero-pad for final section of lab (padarray)







# Advice (Cont'd)

- Method to Import Image into MATLAB
- Think About Effects of Filtering
- Use a Black-White Image, or Greyscale a Coloured One
- Consider the Effects of Multiple Dimensions on Scaling







#### Quiz

 Using what you've learned of DFTs, explain the effects of adding a series of zeros in front of and after your function before attempting the DFT in MATLAB



