Fourier Transform

- Fourier transform converts a physical-space (or time series) representation of a function into frequency-space
 - Equivalent representation of the function, but gives a new window into its behavior
 - Inverse operation exists

$$F(k) = \int_{-\infty}^{\infty} f(x)e^{-2\pi ixk}dx$$

$$f(x) = \int_{-\infty}^{\infty} F(k)e^{2\pi ixk}dk$$

• You can think of F(k) as being the amount of the function f represented by a frequency k

Fourier Transform

- For discrete data, the discrete analog of the Fourier transform gives:
 - Amplitude and phase at discrete frequencies (wavenumbers)
 - Allows for an investigation into the periodicity of the discrete data
 - Allows for filtering in frequency-space
 - Can simplify the solution of PDEs: derivatives change into multiplications in frequency space.

Discrete Fourier Transform

- The discrete Fourier transform (DFT) operates on discrete data
 - Usually we have evenly-spaced, real data
 - E.g. a time-series from an experiment
 - Simulation data for a velocity field
- DFT transforms the N spatial/temporal points into N frequency points

- Transform:
$$F_k = \sum_{n=0}^{N-1} f_n e^{-2\pi i n k/N}$$

- Inverse:
$$f_n = \frac{1}{N} \sum_{k=0}^{N-1} F_k e^{2\pi i n k/N}$$

 This is the form used in NumPy, Garcia, and others. Pang splits the normalization

Notation

- Many different notations are used in various texts
 - Original function: f(x) or f(t)
 - Transformed function: $F(k), \mathcal{F}(k), \hat{f}(k)$
- For the discrete version:
 - Original function: f_n
 - Transformed function: $F_k, \mathcal{F}_k, \hat{f}_k$

Real Space vs. Frequency Space

- What are the significance of the Re and Im parts?
 - Recall that we are integrating with $e^{-2\pi i k x}$
 - Euler's formula: $e^{ix} = \cos(x) + i\sin(x)$
 - Real part represents the cosine terms, symmetric functions
 - Imaginary part represents the sine terms, antisymmetric functions
 - Can also think in terms of an amplitude and a phase
 - If f_n is Real, then:

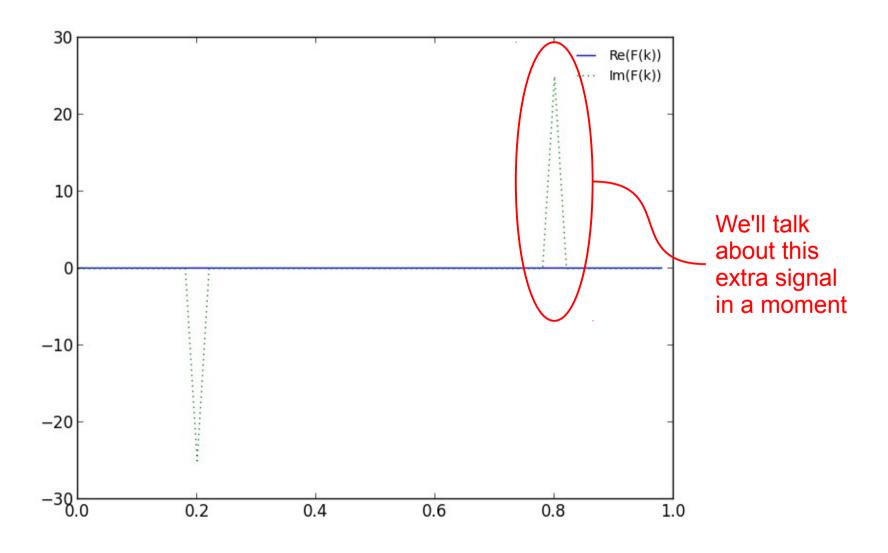
$$\operatorname{Re}(F_k) = \sum_{n=0}^{N-1} f_n \cos\left(\frac{2\pi nk}{N}\right)$$

$$\operatorname{Im}(F_k) = \sum_{n=0}^{N-1} f_n \sin\left(\frac{2\pi nk}{N}\right)$$

- Implementing the discrete Fourier transform is simple
 - Double sum: for each wavenumber, we sum over all the spatial points

```
def dft(f n):
N = len(f n)
 f k = numpy.zeros((N), dtype=numpy.complex128)
k = 0
 while k < N:
     n = 0
     while n < N:
         f k[k] += f n[n]*numpy.exp(-2.0*math.pi*1j*n*k/N)
         n += 1
     k += 1
 return f k
```

• DFT of $\sin(2 \pi f_0 x)$ with $f_0 = 0.2$



- Note that in the DFT, nowhere does the physical coordinate value, x_n , enter—instead, we just look at the index n itself
 - This assumes that the data is regularly gridded
- In this index space, the smallest wavelength is from one cell to the next, and the smallest frequency is 1/N
 - Note that this means that if we add points to our data, then we open up higher and higher frequencies (in terms of index space)
- Clearly there is a physical scale for the frequency

$$\nu_k = \frac{k}{N} \cdot \frac{1}{\Delta x} = \frac{k}{N} \cdot \frac{N}{L}$$

- Lowest frequency: 1/L
- Highest frequency ~ $1/\Delta x$

• k=0 is special:

$$\operatorname{Re}(F_0) = \sum_{n=0}^{N-1} f_n \cos\left(\frac{2\pi n0}{N}\right) = \sum_{n=0}^{N-1} f_n$$

$$\operatorname{Im}(F_k) = \sum_{n=0}^{N-1} f_n \sin\left(\frac{2\pi n0}{N}\right) = 0$$

This is sometimes termed the DC offset

FFT = DFT

- The Fast Fourier Transform (FFT) is equivalent to the discrete Fourier transform
 - Faster because of special symmetries exploited in performing the sums
 - O(N log N) instead of O(N²)
- Both texts offer a reasonable discussion on how the FFT works
 —we'll defer it to those sources.

Normalization

 Normalization of the forward and inverse transforms follows from Parseval's theorem:

$$\int_{-\infty}^{\infty} |f(x)|^2 dx = \int_{-\infty}^{\infty} |F(k)|^2 dk$$

- Discrete form(taking $\Delta x = 1$ and $\Delta k = 1/N$)

$$\sum_{n=0}^{N-1} |f_n|^2 = \frac{1}{N} \sum_{k=0}^{N-1} |F_k|^2$$

The definition of the inverse transformation compatible with this condition is

$$f_n = \frac{1}{N} \sum_{k=0}^{N-1} F_k e^{2\pi i n k/N}$$

Normalization

- To illustrate this, consider the transform of 1
 - Analytically: $f(x) = 1 \iff F(k) = \delta(k)$
 - Discrete transform:

$$F_k = \sum_{n=0}^{N-1} 1 \cdot e^{-2\pi i k n/N} = \sum_{n=0}^{N-1} [\cos(2\pi k n/N) + i \sin(2\pi k n/N)]$$

- For k = 0, cos(0) = 1, sin(0) = 0, so $F_0 = N$
- For k > 0, we are essentially doing a discrete sum of the cosine and sine using equally-spaced points, and the sum is always over a multiple of a full wavelength, therefore, $F_k = 0$
- Discrete inverse transform:

$$f_n = \frac{1}{N} \sum_{k=0}^{N-1} F_k e^{2\pi i n k/N} = \frac{1}{N} N e^{2\pi i n 0/N} = 1$$
 only non-zero for k = 0

Normalization

- When plotting, we want to put the physical scale back in, as well as make the correspondence to the continuous representation
 - Already saw how to put the wavenumbers into a physical frequency
 - Rewrite inverse expression:

$$f_n = \frac{1}{N} \sum_{n=0}^{N-1} F_k e^{2\pi i n k/N} = \sum_{n=0}^{N-1} \left(\frac{F_k}{N}\right) e^{2\pi i n k/N}$$

This looks like the continuous form

This is what we plot

- Another interpretation: $F_k / N \Leftrightarrow F_k dk$

Let's look at code that transforms 1...

Real Space vs. Frequency Space

- Imagine transforming a real-valued function, f , with N data points
 - The FFT of f will have N/2 complex data points
 - Same amount of information in each case, but each complex point corresponds to 2 real numbers.
 - This is a consequence of the analytic Fourier transform satisfying $F(-k) = F^*(k)$ if f(x) is real
 - Most FFT routines will return N complex points—half of them are duplicate, i.e. they don't add to the information content
 - Often, there are specific implementations optimized for the case where the function is real (e.g. rfft)
 - This affects normalization (note k=0 different)
- This is also referred to as aliasing. The maximum frequency, $1/(2 \Delta x)$ is called the Nyquist frequency

Power Spectrum

The power spectrum is simply defined as:

$$P(k) = |F(k)|^2 = F(k)F^*(k)$$

- Single number showing the importance/weight of a given wavenumber
- Also useful to look at the phase at each wavenumber

Python/NumPy's FFT

- numpy.fft: http://docs.scipy.org/doc/numpy/reference/routines.fft.html
- fft/ifft: 1-d data
 - By design, the k=0, ... N/2 data is first, followed by the negative frequencies. These later are not relevant for a real-valued f(x)
 - k's can be obtained from fftfreq(n)
 - fftshift(x) shifts the k=0 to the center of the spectrum
- rfft/irfft: for 1-d real-valued functions. Basically the same as fft/ifft, but doesn't return the negative frequencies
- 2-d and n-d routines analogously defined

Python's FFT

- It's always a good idea to run some simple tests to make sure the FFT is behaving the way you expect
 - $-\sin(2\pi f_0 x)$ —should be purely imaginary at a single wavenumber
 - $-\cos(2\pi f_0 x)$ —should be purely real at a single wavenumber
 - $-\sin(2\pi f_0 x + \pi/4)$ —should have equal magnitude real and imaginary parts at a single wavenumber

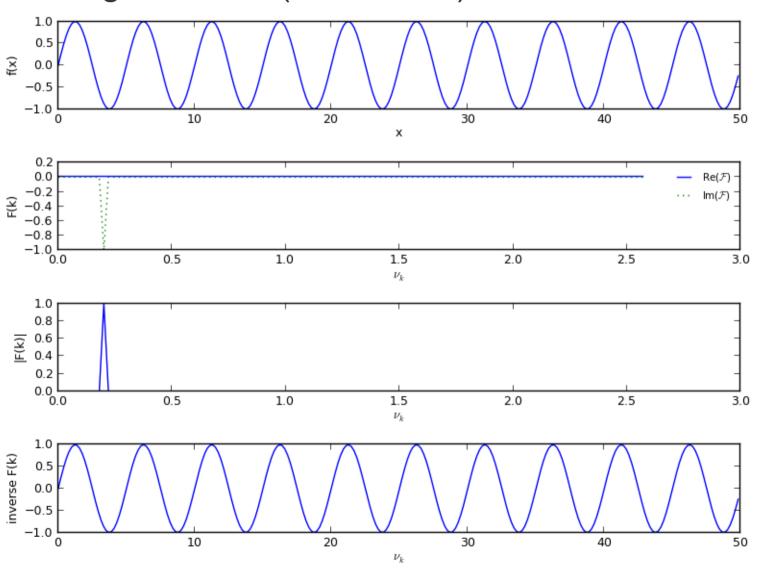
Ex: Single Frequency Sine

Transform a single mode sine ($f_0 = 0.2$) and inverse transform back—do you get the original result (to roundoff)?

Notice that the since sine is odd, the FFT is only non-zero in the imaginary component.

And... we get back our original component.

What happens when we increase the # of points?



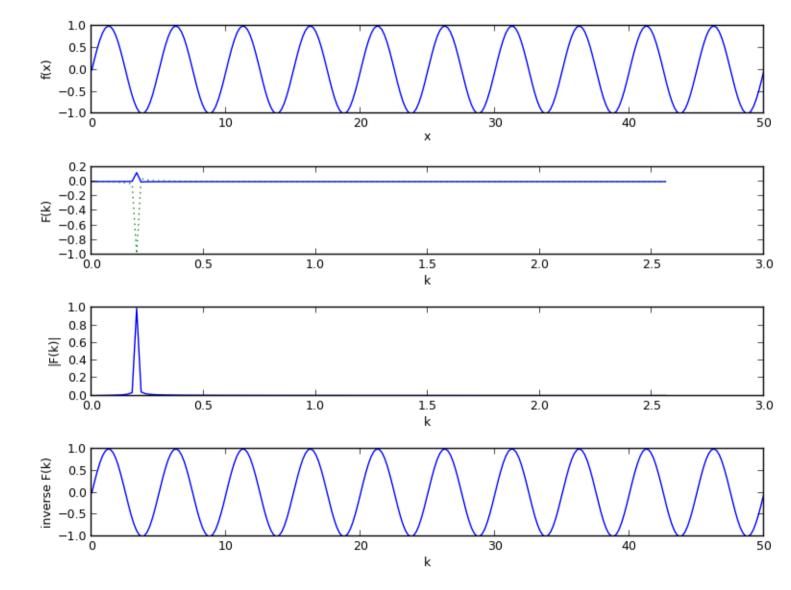
Gotyas...

- Be careful about how you define the points
 - The FFT considers the data without any spatial coordinates—it just considers distance in terms of the number of points
 - Using the NumPy linspace() routine puts a point at both the start and end of the interval
 - e.g., np.linspace(0, 1, 5) = [0.0.250.50.751.]
 - The FFT routine treats the first and last point as distinct
 - If you define $\sin(2\pi x)$ on this data, the first and last point will be equivalent, and the FFT picks up an extra (non-periodic) signal
 - You should do np.linspace(0, 1, 5, endpoint=False) instead

Gotyas...

FFT with linspace putting a point at both endpoints.

Note the blip in the real portion of the FFT.

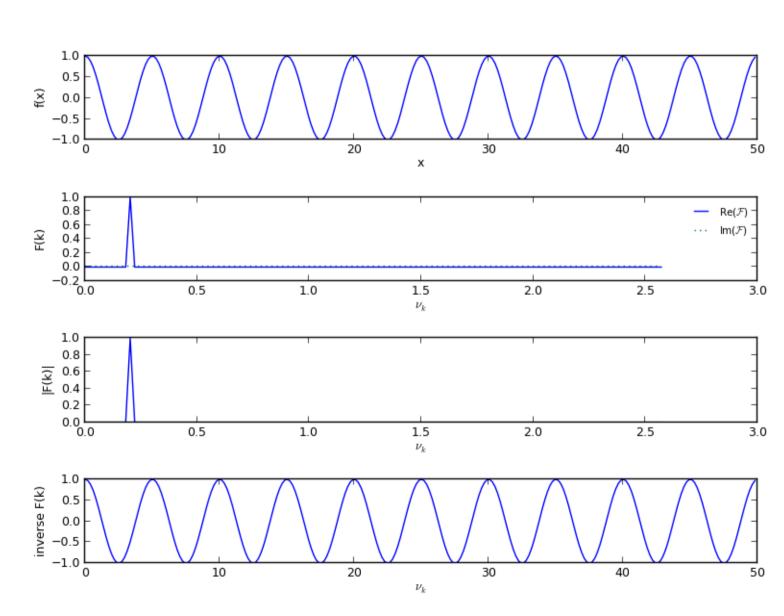


Ex: Single Frequency Cosine

And a cosine is "opposite" of the sine

Notice that the since cosine is even, the FFT is only non-zero in the real component.

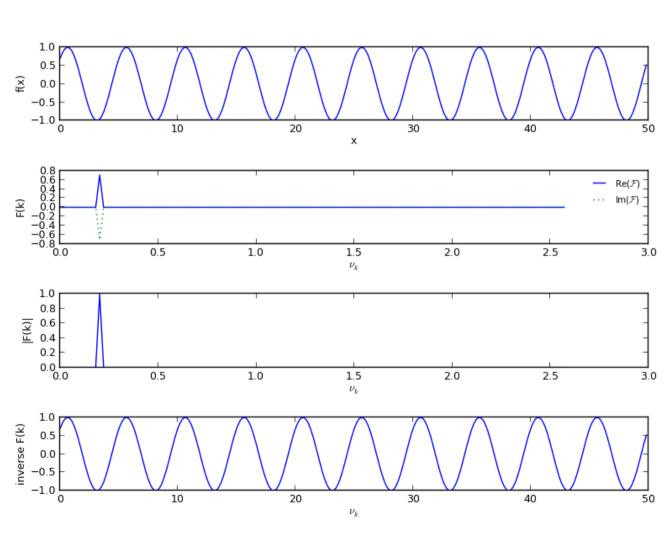
Also note the normalizations



Ex: Phase Shift

- Amplitude and phase. Consider: $\sin(2\pi \ f_0 \ x + \pi/4)$
- Phase:

$$\phi = \tan^{-1} \left(\frac{\operatorname{Im}(F_k)}{\operatorname{Re}(F_k)} \right)$$



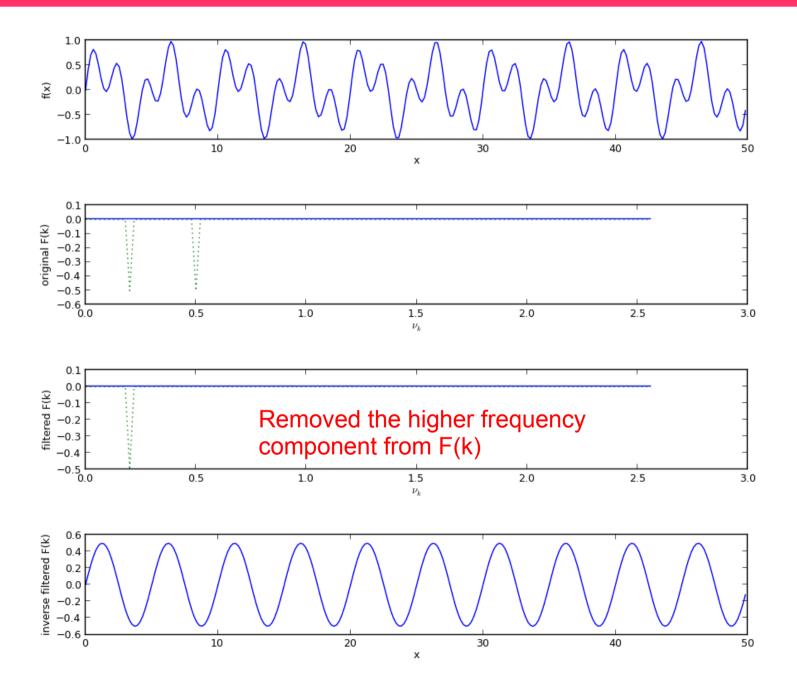
Example: Filtering

- In frequency-space, we can filter out high or low frequency components.
- Consider:

$$f(x) = \frac{1}{2} \left[\sin(2\pi f_0 x) + \sin(2\pi f_1 x) \right]$$

- With
$$f_0 = 0.2, f_1 = 0.5$$

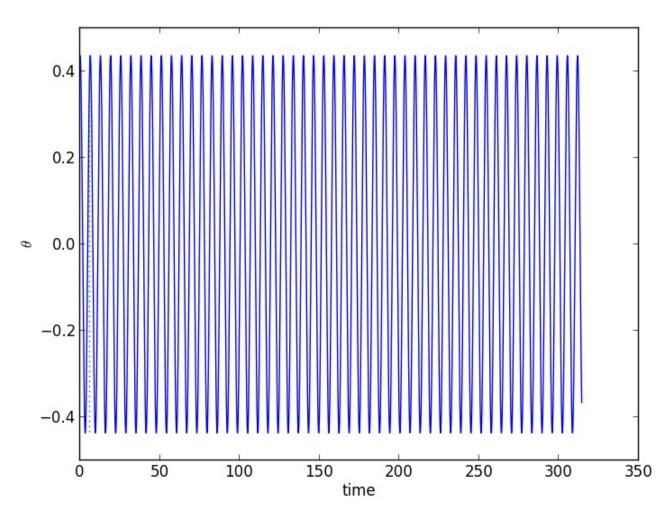
Example: Filtering



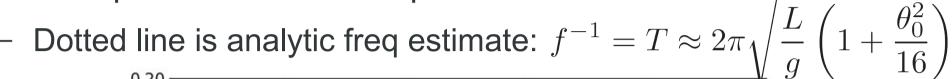
- Remember our pendulum problem from homework #2
 - Finite-amplitude system
 - Here's the solution with $\theta_0 = 25^\circ$

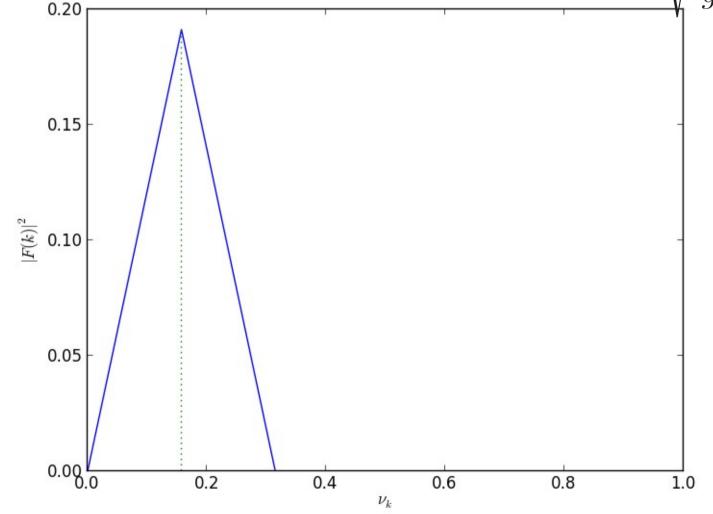
50 periods integrated (using small-angle estimate of period)

12,568 points from 4th-order Runge-Kutta



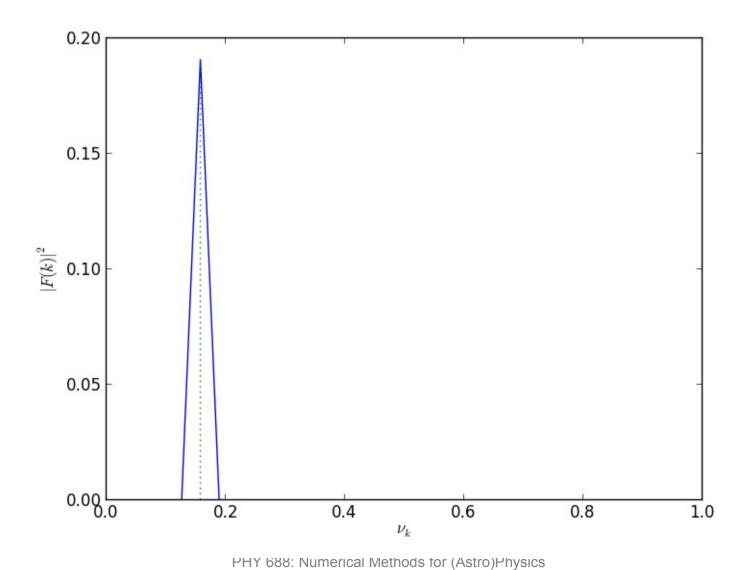
Power spectrum of the first period



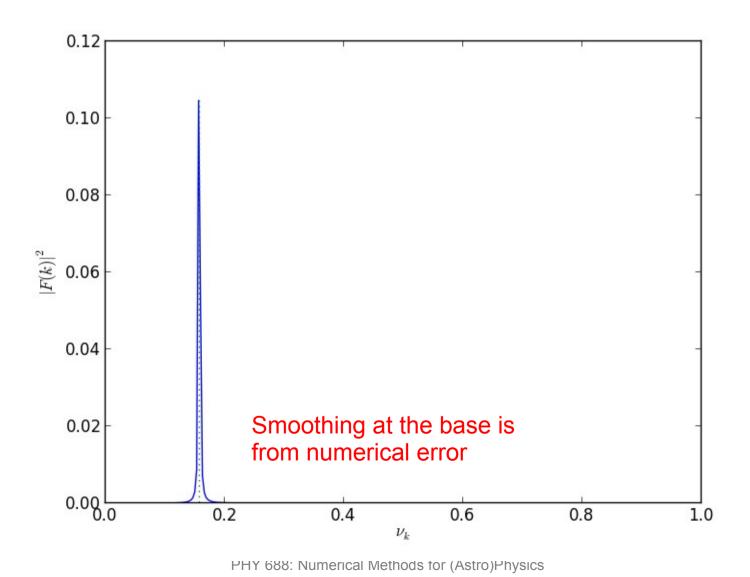


PHY 688: Numerical Methods for (Astro)Physics

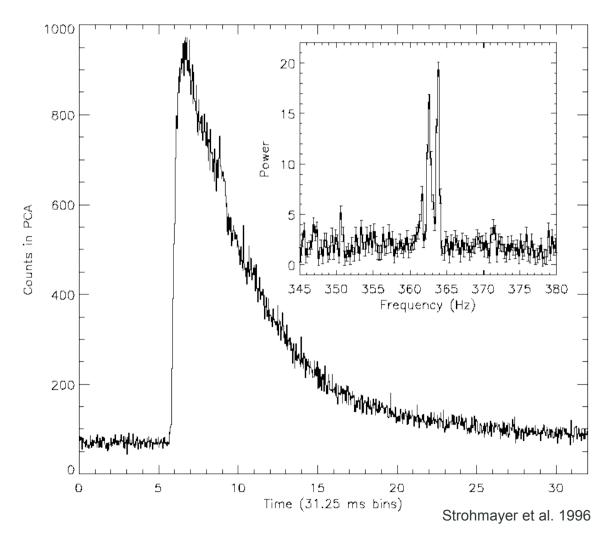
Power spectrum of the first 5 periods



Power spectrum of the all 50 periods



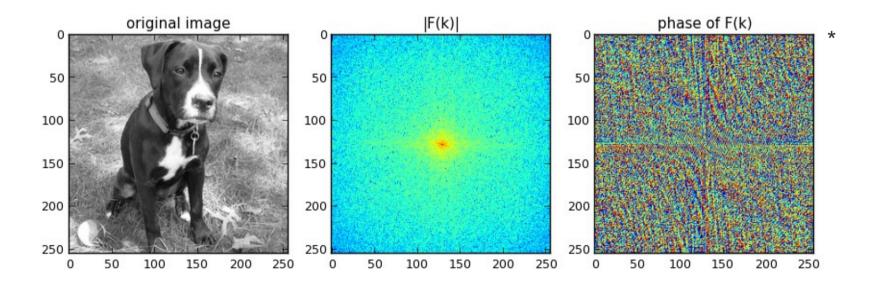
 Here's an astronomical example from an X-ray burst timeseries



 Multidimensional FFTs decompose into a sequence of onedimensional FFTs

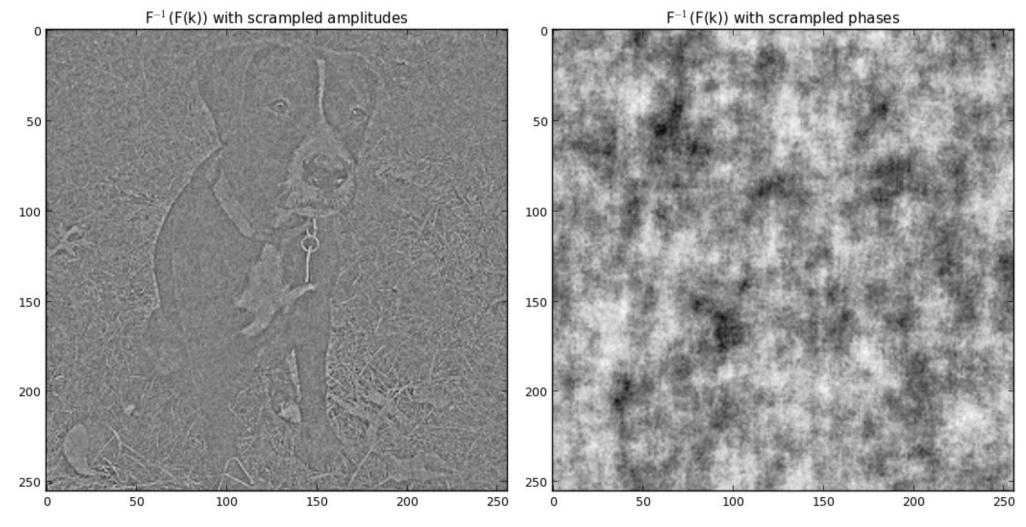
$$\begin{split} F_{k_x,k_y} &= \sum_{m=0}^{N_x-1} \sum_{n=0}^{N_y-1} f_{mn} e^{-2\pi i (k_x m/N_x + k_y n/N_y)} \\ &= \sum_{m=0}^{N_x-1} e^{-2\pi i k_x m/N_x} \sum_{n=0}^{N_y-1} f_{mn} e^{-2\pi i k_y n/N_y} \end{split}$$
 This is the transform in the y-direction

- To visualize what is going on, we need to look at both the amplitude and the phase
 - Note that only 1 quadrant is significant, because our input was real-valued.



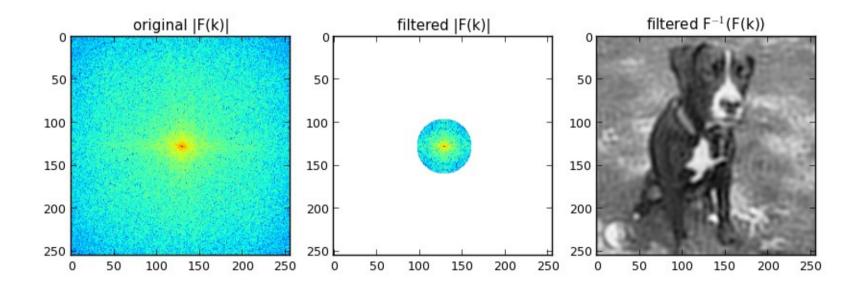
*No animals were harmed in making these slides

 To understand what the magnitude and phase influence, here we scramble each of these in turn



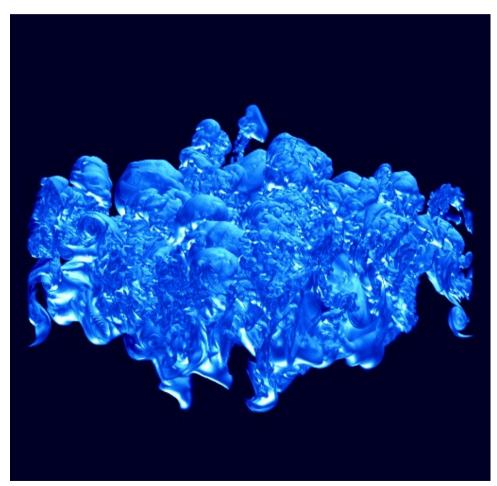
Example based on :http://matlabgeeks.com/tips-tutorials/how-to-do-a-2-d-fourier-transform-in-matlab/ PHY 688: Numerical Methods for (Astro)Physics

 In phase space we can filter out frequency components to do image processing

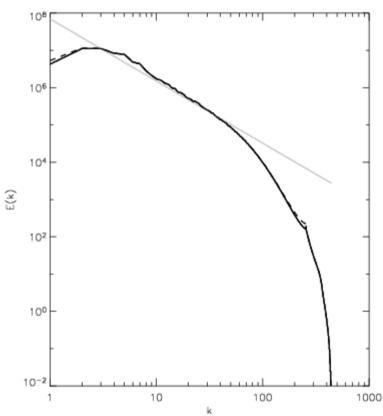


Example: Turbulence

 Power spectrum of the velocity field is used to understand the turbulent energy cascade



(Zingale et al. 2004)



$$E(K) = \int_{K=|k|} dk [\hat{u}(k)^2 + \hat{v}(k)^2 + \hat{w}(k)^2]$$

$$k^2 = k_x^2 + k_y^2 + k_z^2$$

Beyond Data Analysis

- So far we've focused on using the FFT for data analysis
- We can also use it directly for solving (some) PDEs.
- Consider the Poisson equation:

$$\frac{d^2\phi_n}{dx^2} = f_n$$

Express things in terms of the transforms

$$\phi(x) = \int \Phi(k)e^{-2\pi ikx}dk$$

$$f(x) = \int F(k)e^{-2\pi ikx}dk$$

Beyond Data Analysis

Easy to differentiate:

$$\frac{d^2\phi(x)}{dx^2} = -4\pi^2 k^2 \int \Phi(k)e^{2\pi ikx}dk$$

Then:

$$-4\pi^2 k^2 \Phi(k) = F(k)$$

- Easy to solve:

$$\Phi(k) = -\frac{F(k)}{4\pi^2 k^2}$$

- Solve algebraically in Fourier space and then transform back
 - Only works for certain boundary conditions