Fourier transforms in digital signal processing

Real space: numbers at points in time or space



3D array







2D array

shape: (2, 3)





Real space is an intuitive representation of data



How fast is the signal decaying? How fast is the signal oscillating? How many oscillators make up the signal?



Where are the particles located? What is the shape of the particles? Is this image blurred and distorted? How does this image relate to the structure?

...but some questions are harder to pose

Main ideas of today's lecture

(1) There many ways of representing numerical data

(2) Fourier space is an alternative representation based on waves of different frequency

(3) Many find Fourier space initially unintuitive

(4) Many hard problems become easier to understand and solve in Fourier space

Linear combinations

Most functions can be represented as weighted sums of other functions.

$$S(x) = w_1 F_1(x) + w_2 F_2(x) \dots + w_N F_N(x) = \sum_{i=1}^N w_i F_i(x)$$

Some function of
the variable **x** Basis functions:
a set of functions that can be combined

a set of functions that can be combined to form other functions. A.k.a. components, dimensions

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Some function of
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Basis functions:
a set of functions that can be combined
to form other functions.
A.k.a. weights or coordinates

Reminder of dot product and inner product

Discrete

dot product:
$$\mathbf{c} = \Sigma_k \mathbf{u}_k \mathbf{v}_k = \mathbf{u} \cdot \mathbf{v}_k$$

Example: x = [141] y = [726]

dot_product(**x**,**y**) = 1*7 + 4*2 + 1*6 = 21

Continuous

inner product $c = \int f(t) g(t) dt = (f,g)$

Orthonormal linear basis functions

Orthonormality conditions

Dot product of F1 and F2 in a basis is always zero. Dot product of F1 and F1 is always one' ex. $x=[1 \ 0 \ 0]$, $y=[0 \ 1 \ 0]$, and $z=[0 \ 0 \ 1]$

dot_product(\mathbf{x} , \mathbf{y}) = 1*0 + 0*1 + 0*0 = 0 dot_product(\mathbf{x} , \mathbf{z}) = 1*0 + 0*0 + 0*1 = 0 dot_product(\mathbf{y} , \mathbf{z}) = 0*0 + 1*0 + 0*1 = 0 dot_product(\mathbf{x} , \mathbf{x}) = 1*1 + 0*0 + 0*0 = 1

If basis vectors are mutually orthonormal, we can determine the coefficients for a function S simply by taking the dot product of the function and the basis functions:

ex. $\mathbf{S} = [3 \ 2 \ 6]$ $w_x = dot_product(S, \mathbf{x}) = 3^{*1} + 2^{*0} + 6^{*0} = 3$ $w_y = dot_product(S, \mathbf{y}) = 3^{*0} + 2^{*1} + 6^{*0} = 2$ $w_z = dot_product(S, \mathbf{z}) = 3^{*0} + 2^{*0} + 6^{*1} = 6$ $\mathbf{S} = w_x^* \mathbf{x} + w_y^* \mathbf{y} + w_z^* \mathbf{z} = [3 \ 0 \ 0] + [0 \ 2 \ 0] + [0 \ 0 \ 6] = [3 \ 2 \ 6]$



<u>Geometric interpretation:</u> dot_product(\mathbf{x} , \mathbf{y}) = 0 means arccos(x,y) = pi / 2 = 90° "a right angle between x and y"

The linear orthogonal basis for Fourier space: Waves with different frequencies



Square wave function

The linear orthogonal basis for Fourier space: Waves with different frequencies



Waves are represented with sine and cosine functions over time or space.



Sine waves can have different phases



Wave phase (offset)

 $F_i(x) = \sin(2\pi k x + \phi_i)$

Adding to the argument of the sine function adds an offset to the wave called the 'phase'

Phase shifts are less than 2π

Sines and cosines are related by a 90° (π/2) phase shift

$$\sin(heta+\pi)=-\sin heta\ \sin(heta+2\pi)=+\sin heta$$

$$\sin(heta+rac{\pi}{2})=+\cos heta\ \cos(heta+rac{\pi}{2})=-\sin heta$$

Sine waves can have **different amplitudes**: these will be coefficients of our combination linear combinations





Polar coordinates 'Amplitude - phase' coordinates

Rectangular coordinates 'Sine - cosine' coordinates

A*cos(x) + B*sin(x)

$$M^* \cos(x + \theta) =$$

Convenient identities
$$M = (A^2 + B^2)^{\frac{1}{2}}$$
 $\theta = arctan(B/A)$ $A = M^*cos(\theta)$ $B = M^*sin(\theta)$



Scientists prefer to think in polar coordinates

Computer programs typically output rectangular coordinates.



$e^{ix} = \cos x + i \sin x$

Waves are also commonly represented by exponential functions using Euler's formula.

Waves of different frequency form an orthonormal basis

$$\frac{1}{L} \int_{-L}^{L} \cos(n\frac{\pi}{L}t) \cos(m\frac{\pi}{L}t) dt = \begin{cases} 1 & n = m \neq 0 \\ 0 & n \neq m \\ 2 & n = m = 0 \end{cases}$$
$$\frac{1}{L} \int_{-L}^{L} \cos(n\frac{\pi}{L}t) \sin(m\frac{\pi}{L}t) dt = 0$$
$$\frac{1}{L} \int_{-L}^{L} \sin(n\frac{\pi}{L}t) \sin(m\frac{\pi}{L}t) dt = \begin{cases} 1 & n = m \neq 0 \\ 0 & n \neq m \end{cases}$$

In words: "The inner product (dot product) of two sine or cosine functions is zero if they have different frequencies."

Proof of the orthogonality relations: This is just a straightforward calculation using the periodicity of sine and cosine and either (or both) of these two methods:

Method 1: use $\cos at = \frac{e^{iat} + e^{-iat}}{2}$, and $\sin at = \frac{e^{iat} - e^{-iat}}{2i}$. Method 2: use the trig identity $\cos(\alpha)\cos(\beta) = \frac{1}{2}(\cos(\alpha + \beta) + \cos(\alpha - \beta))$, and the similar trig identies for $\cos(\alpha)\sin(\beta)$ and $\sin(\alpha)\sin(\beta)$.



Any periodic function can be represented by a linear combination of sine and cosine wave functions.



Continuous functions may require infinite waves. Discrete functions (real-world data) can be exactly represented with a finite sum of waves. (N/2+1 sines and N/2+1 cosines)

The Fourier synthesis equation



The function S(x) is equal to a weighted sum of sines and cosines of increasing frequency, k. The weights are the coefficients a_{ν} and b_{ν}

$$S(x) = rac{1}{N}\sum_{k=0}^{N-1} a_k \sin(2\pi k x/N) + i b_k \cos(2\pi k x/N) = rac{1}{N}\sum_{k=0}^{N-1} A_k e^{i2\pi k x + \phi_k}$$

Rectangular coordinates

Exponential coordinates

The Fourier analysis equation

We can solve for the coefficients a_k and b_k by calculating the dot product of S(x) with a wavefunction at the frequency k

$$X(k) = a_k + i b_k = \sum_{x=0}^{N-1} S(x) e^{-i 2 \pi k x / N}$$

X(k) is a **frequency-domain** representation of the **real-space** function S(x) You might also hear **Fourier-space** or **reciprocal space** representation

Shannon's sampling theorem

For a discrete FT, what are the frequencies k?

First we need the real-space **sampling rate**, d examples d = 1 second / sample (temporal signal, like a sound)

d = 1 angstrom / pixel (spatial signal, like an image)

We also need the number of samples, N

The FT will have N/2+1 **frequencies**, k. The units will be 1/d and they will run linearly from 0 to 1/2d.

The frequency k=1/(2d) is the **Nyquist frequency**. It is the highest possible frequency sinusoid that can be correctly represented at sampling rate d.

Examples of discrete wavefunctions



cos(k=0) is a constant value. It's coefficient is the mean of the real-space data. Sometimes it's called the DC component.

sin(k=0) is always zero.

The component at the nyquist frequency, $\cos(k=1/(2d))$ is a function that alternates each pixel between -1 and 1.

sin(k=1/(2d)) is always zero.

The Fourier analysis equation, aka. The Fourier Transform

<u>Real-space representation of data</u> Values at points in time/space **Fourier-space representation of data:** Coefficients of waves of different frequencies.

The Fourier synthesis equation, aka. The Inverse Fourier Transform

Fourier transforms can also be calculated for 2D functions like **images**: S(x,y)

$$X(k,l) = \sum_{x=0}^{N_x-1} \sum_{y=0}^{N_y-1} S(x,y) e^{-i2\pi (kx/N_x+ly/N_y)}$$



Usually when we represent a 2D Fourier transform, we put **low spatial frequencies** near the center, **high spatial frequencies** farther away



FT

Real-space image



Amplitudes of Fourier transform



Amplitudes of Fourier transform



Amplitudes of Fourier transform



The Phase Problem: Animal Magic



The Phase Problem: Animal Magic

Combine the magnitudes from the Duck transform with the phases from the Cat transform



The Phase Problem: Animal Magic



The phase contains the bulk of the information!

Kevin Cowtan http://www.ysbl.york.ac.uk/~cowtan/

John Tukey's Fast Fourier Transform (FFT) algorithm One of the greatest algorithm of all time

For the discrete Fourier transform, we convert an array with N elements to N/2+1 sines and cosines. Each sine/cosine pair requires the dot product over all N elements, we require N*(N/2) operations. The FFT solves the same problem in N*log(N) operations, making it 'cheap' even for very large N.



Scientific computing libraries have highly optimized implementations of the FFT



real FFT: Produces N/2-1 coefficients

complex FFT:

Produces N coefficients, but one side of the FFT is exactly the same as the other side.

Spectral analysis: which waves are in a signal?



The fastest wave that can be represented with 500 samples has 250 oscillations.

Nyquist frequency = 0.5 = 250

The frequency of each plotted wave is:

wave1 = 100/500 = 0.2 wave2 = 50/500 = 0.1 wave3 = 80/500 = 0.16

Can we determine these values from the data itself using the FFT?

Spectral analysis: which waves are in a signal?

1 noise = np.random.rand(500) - 0.5
2 noisy_wave = (wave_sum) + 2*noise
3 plt.plot(noisy_wave, c='r')
4 plt.show()



In a realistic case, we'll also have noise or other processes occuring.

We can thus add random noise to our wave sum to simulate these processes.

Now there's no way you could guess the frequencies from just looking!

Spectral analysis: which waves are in a signal?



Power of a signal at frequency k = squared amplitude at frequency k

The **power spectrum** is the power as a function of k

Calculate the power of a Fourier coefficient by multiplying a+ib by its complex conjugate a-ib:

$$(a+bi)(a-bi)=a^2+b^2$$

Welch's algorithm for estimating the power spectrum:

- 1. Divide signal up into overlapping patches.
- 2. Calculate the FT of each patch
- 3. Calculate PS = FT*FT.conj() (python syntax)
- 4. Average all the PS together

Using python's implementation of Welch's algorithm, we see peaks at 0.1, 0.16, and 0.2 as expected!

Why would we want to do all this work?

FT applications in cryo-EM

Linear systems and convolution





Time/space

Point-spread function p(x)



Describes how a point is transformed by a linear system. Also called the impulse-response function or the convolution kernel

Convolution of f(x) with p(x):

Each point is multiplied by the point-spread function:

 $f(x)^*p(x) = y(x)$

* stands for convolution, not multiplication



The Fourier convolution theorem

 $f(x)^*p(x) = y(x)$

F(k) = FT(f(x))P(k) = FT(f(x))

 $\mathsf{F}(\mathsf{k})\mathsf{P}(\mathsf{k})=\mathsf{Y}(\mathsf{k})$

 $\mathsf{IFT}(\mathsf{Y}(\mathsf{k})) = \mathsf{y}(\mathsf{x})$

Convolution in real-space is computationally challenging.

However, in Fourier space, convolution is an elementwise multiplication of the FT of the signal and FT of the PSF.

In cryo-EM, the FT of the PSF is called the **Contrast Transfer Function** (CTF).

The CTF corresponds to the way the electron optical system distorts

Fourier transforms let us do efficient convolutions







Fourier transforms let us do efficient convolutions



Scattering and lensing are like Fourier transforms

The linear Fourier imaging model

f(x) = projection image of electrons through the specimen

F(k) = FT(f(x))

Y(k) = F(k)CTF(k) + noise(k)

y(x) = IFT(Y(k))

y(x) = image we record on our electron cameras



original



f(x,y)

|F(u,v)|











FFT (log transformed)



Lowpass filter









FFT (log transformed) Filtered image (histogram equalised)











FFT



Bandpass filter





Filtered image (histogram equalised)



Fourier transforms let us align noisy images

The **matched-filter** or **cross-correlation** algorithm: $FT^{-1}[FT(M) \times FT(T)^*] = cross-correlation map$



The better the match, the larger the peak

We can divide an image into resolution shells and compute cross-correlation for each shell

If we have two images, we can use the **Fourier Shell Correlation (FSC)** curve to find the resolution where they become inconsistent with each other



$$FSC(r) = rac{\displaystyle \sum_{r_i \in r} F_1(r_i) \cdot F_2(r_i)^*}{\displaystyle \sqrt[2]{\displaystyle \sum_{r_i \in r} |F_1(r_i)|^2 \cdot \sum_{r_i \in r} |F_2(r_i)|^2}}$$

FSC curves:

- resolution-dependent cross-correlation
- resolution-dependent consistency metric for structures
- commonly used as measure of resolution

In electron microscopy, we want to average 2D projection images together to form a 3D volumetric ('density') image... **but how?**



Forward process: e- beam forms projection image from 3D structure

Inverse problem: Can we recover 3D structure from projection images?

Images can be averaged in real-space or averaged in reciprocal space and then inverse Fourier transformed

$S_1(x) + S_2(x) = FT^{-1}(X_1(k) + X_2(k))$

The projection-slice theorem:

If I take a **2D projection** through a **3D object** and fourier transform it, I get a **2D slice** of the object's **3D fourier transform**.



If we project from a different viewpoint, the slice we get is from that viewpoint

This gives leads to a **reconstruction algorithm** for solving EM structures:

1. Project an initial 3D density from many views use the projection-slice theorem

2. Match experimental images to projections use the cross-correlation algorithm

3. Calculate a new 3D density from aligned images *use the projection-slice theorem*

4. Iterate this process until 3D density stops improving





Space of all 3D structures