

# Fourier Transforms

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# (Discrete) Fourier Transform

# Fourier Transform

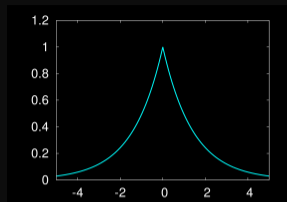
In this part of the lecture, we will discuss:

- The Fourier transform,
- The discrete Fourier transform
- The **fast** Fourier transform
- Examples using the FFTW library



# Fourier Transform recap

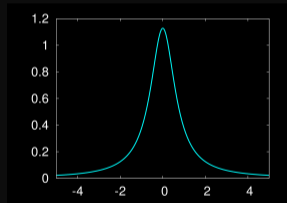
- Let  $f$  be a function of some spatial variable  $x$ .



$$f(x) = e^{-|x|}$$

- Transform to a function  $\hat{f}$  of the angular wavenumber  $k$ :

$$\hat{f}(k) \propto \int f(x) e^{\pm i k \cdot x} dx$$



$$f(x) = (1 + k^2)^{-1}$$

- Inverse transformation:

$$f(x) \propto \int \hat{f}(k) e^{\mp i k \cdot x} dk$$

# Fourier Transform

- Fourier made the claim that any function can be expressed as a harmonic series.
- The FT is a mathematical expression of that.
- Constitutes a linear (basis) transformation in function space.
- Transforms from spatial to wavenumber, or time to frequency, etc.
- Constants and signs are just convention.\*

\* some restrictions apply.

# Discrete Fourier Transform



C. F. Gauss

- Given a set of  $n$  function values on a regular grid:

$$x_j = j\Delta x; \quad f_j = f(j\Delta x)$$

- Transform to  $n$  other values

$$\hat{f}_q = \sum_{j=0}^{n-1} f_j e^{\pm 2\pi i j q/n}$$

- Easily back-transformed:

$$f_j = \frac{1}{n} \sum_{q=0}^{n-1} \hat{f}_q e^{\mp 2\pi i j q/n}$$

- Solution is periodic:  $\hat{f}_{-q} = \hat{f}_{n-q}$ . You run the risk of aliasing, as  $q$  is equivalent to  $q + \ell n$ . Cannot resolve frequencies higher than  $q = n/2$  (Nyquist).

# Slow Fourier Transform

$$\hat{f}_q = \sum_{j=0}^{n-1} f_j e^{\pm 2\pi i j q/n}$$

- Discrete Fourier transform is a linear transformation.
- In particular, it's a matrix-vector multiplication.
- Naively, costs  $\mathcal{O}(n^2)$ . Slow!

# Slow DFT

```
#include <complex>
#include <rarray>
#include <cmath>

typedef std::complex<double> complex;

void fft_slow(const rvector<complex>& f, rvector<complex>& fhat, bool inverse)
{
    int n = fhat.size();
    double v = (inverse?-1:1)*2*M_PI/n;
    for (int q=0; q<n; q++)
    {
        fhat[q] = 0.0;
        for (int m=0; m<n; m++) {
            fhat[q] += complex(cos(v*q*m), sin(v*q*m)) * f[m];
        }
    }
}
```

Even Gauss realized  $\mathcal{O}(n^2)$  was too slow and came up with ...



# Fast Fourier Transform

- Derived in partial form several times before and even after Gauss, because he'd just written it in his diary in 1805 (published later).
- Rediscovered (in general form) by Cooley and Tukey in 1965.

## Basic idea

- Write each  $n$ -point FT as a sum of two  $\frac{n}{2}$  point FTs.
- Do this recursively  $2 \log n$  times.
- Each level requires  $\sim n$  computations:  $\mathcal{O}(n \log n)$  instead of  $\mathcal{O}(n^2)$ .
- Could as easily divide into 3, 5, 7, ... parts.

# Fast Fourier Transform: How is it done?

- Define  $\omega_n = e^{2\pi i/n}$ .
- Note that  $\omega_n^2 = \omega_{n/2}$ .
- DFT takes form of matrix-vector multiplication:

$$\hat{f}_q = \sum_{j=0}^{n-1} \omega_n^{qj} f_j$$

- With a bit of rewriting (assuming  $n$  is even):

$$\hat{f}_q = \underbrace{\sum_{j=0}^{n/2-1} \omega_{n/2}^{qj} f_{2j}}_{\text{FT of even samples}} + \omega_n^q \underbrace{\sum_{j=0}^{n/2-1} \omega_{n/2}^{qj} f_{2j+1}}_{\text{FT of odd samples}}$$

- Repeat, until the lowest level (for  $n = 1$ ,  $\hat{f} = f$ ).
- Note that a fair amount of shuffling is involved.

# Fast Fourier Transform: Already done!

We've said it before and we'll say it again: Do not write your own: use existing libraries!

Why?

- Because getting all the pieces right is tricky;
- Getting it to compute fast requires intimate knowledge of how processors work and access memory;
- Because there are libraries available.

Examples:

- ▶ FFTW3 (Faster Fourier Transform in the West, version 3)
  - ▶ Intel MKL
  - ▶ IBM ESSL
- Because you have better things to do.

# Example of using a library: FFTW

Rewrite of previous (slow) ft to a fast one using fftw

```
#include <complex>
#include <rarray>
#include <fftw3.h>

typedef std::complex<double> complex;

void fft_fast(const rvector<complex>& f, rvector<complex>& fhat, bool inverse)
{
    int n = f.size();
    fftw_plan p = fftw_plan_dft_1d(n,
        (fftw_complex*)f.data(), (fftw_complex*)fhat.data(),
        inverse?FFTW_BACKWARD:FFTW_FORWARD,
        FFTW_ESTIMATE);
    fftw_execute(p);
    fftw_destroy_plan(p);
}
```

# Notes

- Always create a plan first.
- An `fftw_plan` contains all information necessary to compute the transform, including the pointers to the input and output arrays.
- Plans can be reused in the program, and even saved on disk!
- When creating a plan, you can have FFTW measure the fastest way of computing dft's of that size (FFTW\_MEASURE), instead of guessing (FFTW\_ESTIMATE).
- FFTW works with doubles by default, but you can install single precision too.

# Inverse DFT

- Inverse DFT is similar to forward DFT, up to a normalization: almost just as fast.

$$f_j = \frac{1}{n} \sum_{q=0}^{n-1} \hat{f}_q e^{\mp 2\pi i j q/n}$$

Many implementations (almost all in fact) leave out the  $1/n$  normalization.

- FFT allows quick back-and-forth between space and wavenumber domain, or time and frequency domain.
- Allows parts of the computation and/or analysis to be done in the most convenient or efficient domain.

# Consider an example

- Create a 1d input signal: a discretized  $\text{sinc}(x) = \sin(x)/x$  with 16384 points on the interval  $[-30:30]$ .
- Perform forward transform
- Write to standard out
- Compile, and linking to fftw3 library.
- Continuous FT of  $\text{sinc}(x)$  is the rectangle function:

$$\text{rect}(f) = \begin{cases} 0.5 & \text{if } \|k\| \leq 1 \\ 0 & \text{if } \|k\| > 1 \end{cases}$$

up to a normalization.

- Does it match?

# Code for the working example

```
//sincfftw.cpp
#include <iostream>
#include <complex>
#include <rarray>
#include <fftw3.h>
typedef std::complex<double> complex;
int main() {
    const int n = 16384;
    rvector<complex> f(n), fhat(n);
    for (int i=0; i<n; i++) {
        double x = 60*(i/double(n)-0.5); // x-range from -30 to 30
        if (x!=0.0) f[i] = sin(x)/x; else f[i] = 1.0;
    }
    fftw_plan p = fftw_plan_dft_1d(n,
                                   (fftw_complex*)f.data(), (fftw_complex*)fhat.data(),
                                   FFTW_FORWARD, FFTW_ESTIMATE);

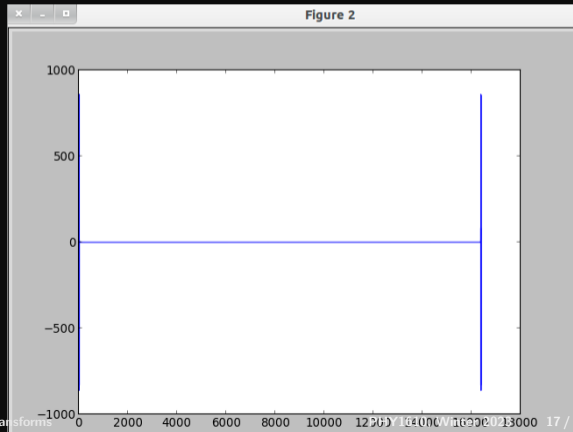
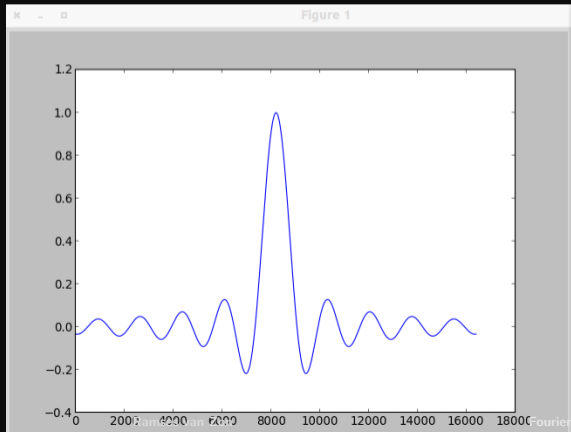
    fftw_execute(p);
    fftw_destroy_plan(p);
    for (int i=0; i<n; i++)
        std::cout << f[i] << ", " << fhat[i] << std::endl;
    return 0;
}
```



# Compile, link, run, plot

```
$ module load gcc/12 fftw/3 python/3
$ g++ -std=c++17 -c -O2 sincfftw.cpp -o sincfftw.o
$ g++ sincfftw.o -o sincfftw -lfftw3
$ ./sincfftw > output.dat
$ ipython --pylab
```

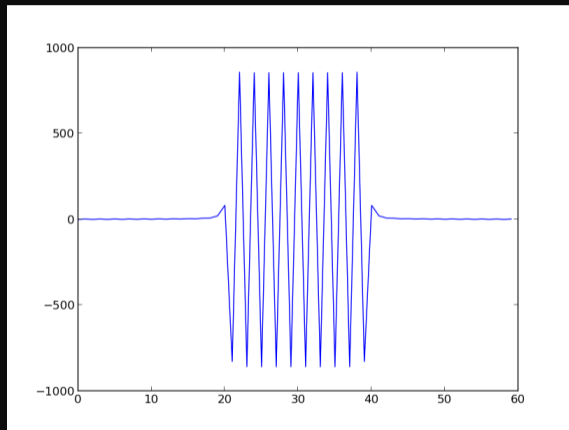
```
>>> data = genfromtxt('output.dat')
>>> plot(data[:,0])
>>> figure()
>>> plot(data[:,2])
```



# Plots of the output, rewrapped

Pick the first and the last 30 points.

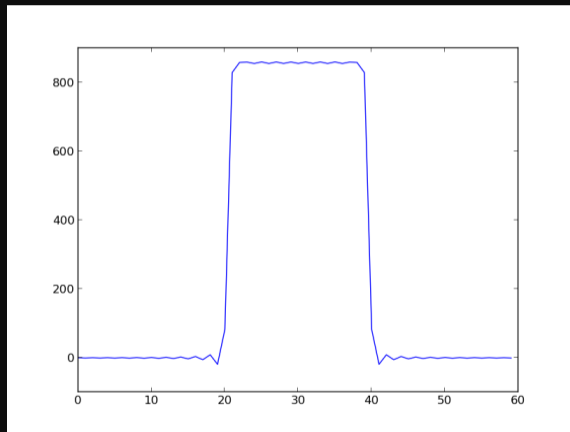
```
>>> x1=range(30)
>>> x2=range(len(data)-30,len(data))
>>> y1=data[x1,2]
>>> y2=data[x2,2]
>>> figure()
>>> plot(hstack((y2,y1)))
```



# Undo phase factor due to shifting

```
>>> plot(hstack((y2,y1))*array([1,-1]*30))
```

We retrieved our rectangle function!



# Precise Relation FT and DFT

- Consider a function on  $f(x)$  an interval  $[x_1, x_2]$ .
- The fourier analysis will express this in terms of periodic functions, so think of  $f$  as periodic.
- We will approximate this function with  $n$  discrete points on  $x_1 + j\Delta x$ , where  $\Delta x = (x_2 - x_1)/n$ , and  $j = 0..n - 1$ , i.e.

$$f(x) = \sum_{j=0}^{n-1} f_j \delta(x - (x_1 + j\Delta x)) \Delta x$$

- Consider its continuous FT:

$$\hat{f}(k) = \int_{x_1}^{x_2} e^{ikx} f(x) dx$$

- $e^{ikx}$  must have period  $(x_2 - x_1)$ :  $k = q \times 2\pi / (x_2 - x_1)$  with  $q$  integer.

# Precise Relation FT and DFT

## Input

$$f(x) = \sum_{j=0}^{n-1} f_j \delta(x - (x_1 + j\Delta x)) \Delta x$$

$$\Delta x = \frac{x_2 - x_1}{n}$$

$$\hat{f}(k) = \int_{x_1}^{x_2} e^{ikx} f(x) dx$$

$$k = \frac{2\pi}{x_2 - x_1} q = \frac{2\pi}{n\Delta x} q$$

## Result

$$\hat{f}(k) = e^{ikx_1} \Delta x \hat{f}_q$$

$$\hat{f}(k) = \int_{x_1}^{x_2} \sum_{j=0}^{n-1} e^{ikx} f_j \delta(x - (x_1 + j\Delta x)) \Delta x dx$$

$$= \sum_{j=0}^{n-1} f_j e^{ik(x_1 + j\Delta x)} \Delta x$$

$$= e^{ikx_1} \Delta x \sum_{j=0}^{n-1} f_j e^{ikj\Delta x}$$

$$= e^{ikx_1} \Delta x \sum_{j=0}^{n-1} f_j e^{2\pi i q j / n}$$

# Multidimensional transforms

In principle a straightforward generalization:

- Given a set of  $n \times m$  function values on a regular grid:

$$f_{ab} = f(a\Delta x, b\Delta y)$$

- Transform these to  $n$  other values  $\hat{f}_{kl}$

$$\hat{f}_{kl} = \sum_{a=0}^{n-1} \sum_{b=0}^{m-1} f_{ab} e^{\pm 2\pi i (a k + b l)/n}$$

- Easily back-transformed:

$$f_{ab} = \frac{1}{nm} \sum_{k=0}^{n-1} \sum_{l=0}^{m-1} \hat{f}_{kl} e^{\mp 2\pi i (a k + b l)/n}$$

- Negative frequencies:  $f_{-k, -l} = f_{n-k, m-l}$ .

# Multidimensional FFT

- We could successive apply the FFT to each dimension
- This may require transposes, can be expensive.
- Alternatively, could apply FFT on rectangular patches.
- Mostly should let the libraries deal with this.
- FFT scaling still  $n \log n$ .

# Symmetries for real data

- All arrays were complex so far.
- If input  $f$  is real, this can be exploited.

$$f_j^* = f_j \leftrightarrow \hat{f}_k = \hat{f}_{n-k}^*$$

- Each complex number holds two real numbers, but for the input  $f$  we only need  $n$  real numbers.
- If  $n$  is even, the transform  $\hat{f}$  has real  $\hat{f}_0$  and  $\hat{f}_{n/2}$ , and the values of  $\hat{f}_k > n/2$  can be derived from the complex valued  $\hat{f}_{0 < k < n/2}$ : again  $n$  real numbers need to be stored.



# Symmetries for real data

- A different way of storing the result is in “half-complex storage’’. First, the  $n/2$  real parts of  $\hat{f}_{0 < k < n/2}$  are stored, then their imaginary parts in reversed order.
- Seems odd, but means that the magnitude of the wave-numbers is like that for a complex-to-complex transform.
- These kind of implementation dependent storage patterns can be tricky, especially in higher dimensions.

# Applications?

# Application of the Fourier transform

- Signal processing, certainly.
- Many equations become simpler in the fourier basis.
  - ▶ Reason:  $\exp(ik \cdot x)$  are eigenfunctions of the  $\partial/\partial x$  operator.
  - ▶ Partial diferential equation become algebraic ones, or ODEs.
  - ▶ Thus avoids matrix operations.

# Example: Solving a 1D diffusion with FFT

$$\frac{\partial \rho}{\partial t} = \kappa \frac{\partial^2 \rho}{\partial x^2}$$

for  $\rho(x, t)$  on  $x \in [0, L]$ , with boundary conditions  $\rho(0, t) = \rho(L, t) = 0$ , and  $\rho(x, 0) = f(x)$ .

Write

$$\rho(x, t) = \sum_{k=-\infty}^{\infty} \hat{\rho}_k(t) e^{2\pi i k x / L}$$

then the PDE becomes an ODE:

$$\frac{d\hat{\rho}_k}{dt} = -\kappa \frac{4\pi^2 k^2}{L^2} \hat{\rho}_k; \quad \text{with } \hat{\rho}_k(0) = \hat{f}_k.$$

Alternatively, one can first discretize the PDE, then take an FFT. This is numerically different.