

March 18-22, 2024: Quantum Computing, Quantum Machine Learning and Quantum Information Theories

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Week of March 18-22

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Plans for the week of March 18-22, 2024

1. Discussion of project 1 and possible paths for project 2
2. Start discussion of discrete Fourier transforms as prelude to Quantum Fourier transforms, basic mathematical expressions
3. Reading recommendation Hundt, Quantum Computing for Programmers, sections 6.1-6.4 on QFT.
4. Video of lecture
5. Whiteboard notes

Possible paths for project 2

- ▶ Implement QFTs and study the phase estimation algorithm and eventually Shor's algorithm for factorization of numbers.
- ▶ Study other algorithms
 1. Deutsch-Jozsa algorithm: Determine if a function is constant or balance using the fewest number of queries.
 2. Simon's algorithm: Determine if a function (Oracle) is one-to-one or two-to-one
 3. Grover's algorithm: Search unstructured list of data fast
 4. Shor's algorithm : Factorize integers efficiently
- ▶ Study the solution of quantum mechanical eigenvalue problems with systems from atomic/molecular physics and quantum chemistry
- ▶ Quantum machine learning projects
- ▶ Other paths

For project 2, in order to be time efficient, you can use software like Qiskit, PennyLane, qBraid and/or other

Overarching motivation

After Simon's algorithm (to be discussed after the Easter break), the next big breakthrough in quantum algorithms occurred when Peter Shor discovered his algorithm for efficiently factoring numbers. This algorithm makes use of the quantum Fourier transform which is the topic of this sets of lectures.

We will start with discrete Fourier transforms (DFT). There are many motivations for the DFT. For those of you familiar with signal processing, harmonic oscillations, and many other areas of applications, Fourier transforms are almost standard kitchen items. For those of you who have studied quantum theory, you have probably met Fourier transforms when studying Heisenberg's uncertainty relation between momentum and position.

A familiar case

For problems with so-called harmonic oscillations, given by for example the following differential equation

$$m \frac{d^2 x}{dt^2} + \eta \frac{dx}{dt} + x(t) = f(t),$$

where $f(t)$ is an applied external force acting on the system (often called a driving force), one can use the theory of Fourier transformations to find the solutions of this type of equations.

Several driving forces

If one has several driving forces, $f(t) = \sum_n f_n(t)$, one can find the particular solution $x_{pn}(t)$ to the above differential equation for each f_n . The particular solution for the entire driving force is then given by a series like

$$x_p(t) = \sum_n x_{pn}(t).$$

This is known as the principle of superposition. It only applies when the homogenous equation is linear. Superposition is especially useful when $f(t)$ can be written as a sum of sinusoidal terms, because the solutions for each sinusoidal (sine or cosine) term is analytic.

Periodicity

Driving forces are often periodic, even when they are not sinusoidal. Periodicity implies that for some time t our function repeats itself periodically after a period T , that is

$$f(t + \tau) = f(t).$$

One example of a non-sinusoidal periodic force is a square wave. Many components in electric circuits are non-linear, for example diodes. This makes many wave forms non-sinusoidal even when the circuits are being driven by purely sinusoidal sources.

Simple Code Example

The code here shows a typical example of such a square wave generated using the functionality included in the **scipy** Python package. We have used a period of $\tau = 0.2$.

```
import numpy as np
import math
from scipy import signal
import matplotlib.pyplot as plt

# number of points
n = 500
# start and final times
t0 = 0.0
tn = 1.0
# Period
t = np.linspace(t0, tn, n, endpoint=False)
SqrSignal = np.zeros(n)
SqrSignal = 1.0+signal.square(2*np.pi*5*t)
plt.plot(t, SqrSignal)
plt.ylim(-0.5, 2.5)
plt.show()
```


Continuous Fourier transforms and the principle of Superposition

It was Fourier's idea to expand a continuous and periodic function $f(t)$ in terms of sums sinus and cosinus ordered functions (we will use exponentials however) as

$$f(t) = \sum_{k=1}^n a_k \sin(2\pi kt + \phi_n),$$

with ϕ_n being a constant phase. The function f is assumed to be bounded in order to be able to define properly the error in truncating the sum over n , that is

$$\int_a^b dt |f(t)|^2 \leq M < \infty.$$

Below we discuss how to find the coefficients a_n .

Rewriting in terms of sines and cosines

It is common to rewrite the above sum in terms of sines and cosines and to add a complex constant a_0 . This gives us

$$f(t) = \frac{a_0}{2} + \sum_{k=1}^n (a_k \cos(2\pi kt) + b_n \sin(2\pi kt)).$$

Using the standard trigonometric relations

$$\cos t = \frac{\exp it + \exp -it}{2} \quad \sin t = \frac{\exp it - \exp -it}{2i}.$$

Exponential expression

We can rewrite the Fourier expansion as

$$f(t) = \sum_{k=-n}^n c_k \exp(2\pi i k t),$$

with $c_0 = a_0/2$. The coefficients c_k are complex and satisfy $c_{-k} = c_k^*$. The constant $c_0 = c_0^*$ is a real number.

The above sum can also be rewritten in terms of the real part of the exponentials as

$$f(t) = 2\operatorname{Re} \left(\sum_{k=0}^n c_k \exp(2\pi i k t) \right),$$

where we used that $c_{-k} = c_k^*$. We leave it as an exercise to the reader to show the latter expression.

How do we determine the coefficients c_k ?

Determining c_k

Let us assume that we have a periodic function $f(t)$

$$f(t) = \sum_{k=-n}^n c_k \exp(2\pi i k t),$$

and we select the coefficient c_l , and multiply the l.h.s and r.h.s. with $\exp(-2\pi i l t)$, and isolate the c_l terms

$$c_l = f(t) \exp(-2\pi i l t) - \sum_{k=-n, k \neq l}^n c_k \exp(2\pi i (k - l) t).$$

Integrating both sides

Then we integrate both sides (note we have assumed a period of one) from 0 to 1. This gives

$$c_l = \int_0^1 dt f(t) \exp(-2\pi i l t),$$

since the terms with the sum is zero. We leave again this derivation as an exercise to the dedicated reader. If the function f is real (t and dt are real), then $c_l^* = c_{-l}$. The coefficients c_n are normally called the Fourier coefficients and it is common to relabel them in terms of the function f as

$$\hat{f}(k) = \int_0^1 dt f(t) \exp(-2\pi i k t).$$

Independence of interval length

In the discussions and equations above, we have assumed that the integration interval has a length of one, that is a period of length one. It is easy to change this length and as we show below, if we integrate from say a to $a + 1$, this is the same as integrating over an interval of length 1.

To show this, we need also to recall that our function $f(t)$ is assumed to be periodic, that is we need to satisfy

$$f(a + 1) = f(a).$$

Later we will generalize this to a period of arbitrary length. Assume that a is any number and taking the derivative of the integral with respect to a , we have

$$\frac{d}{da} \left[\int_a^{a+1} dt f(t) \exp(-2\pi i kt) \right].$$

Writing out the various terms

From the last equation we have then

$$\exp(-2\pi i k a) \exp(-2\pi i k) f(a+1) - \exp(-2\pi i k a) f(a) = 0,$$

where we used the periodicity $f(a+1) = f(a)$ and that $\exp(-2\pi i k) = 1$ since k is an integer. This shows that the expression for \hat{f} is independent of a . A common instance is

$$\hat{f}(k) = \int_{-\frac{1}{2}}^{\frac{1}{2}} dt f(t) \exp(-2\pi i k t).$$

Changing period

What if the period is not equal to one? Assume we are working with a function $f(t)$ whose period is T . We can then define a function $g(t)$ with period one as

$$g(t) = f(Tt),$$

that is

$$g(t) = \sum_{k=-n}^n c_k \exp(2\pi i k t),$$

and introducing the variable $s = Tt$ we have $g(t) = f(s)$ we have

$$f(s) = g(t) = \sum_{k=-n}^n c_k \exp(2\pi i k t) = \sum_{k=-n}^n c_k \exp(2\pi i k s / T).$$

Harmonics

The so-called harmonics are now defined as $\exp(2\pi i k s / T)$. We have

$$\hat{g}(k) = \int_0^1 dt g(t) \exp(-2\pi i k t),$$

and using $s = Tt$ we have

$$\hat{g}(k) = \int_0^1 dt g(t) \exp(-2\pi i k t) = \frac{1}{T} \int_0^T ds f(s) \exp(-2\pi i k s / T).$$

For a period T we have thus the general expression for the Fourier transform

$$\hat{f}(k) = c_k = \frac{1}{T} \int_0^T dt f(t) \exp(-2\pi i k t / T).$$

Typical interval

An often used choice of interval is

$$\hat{f}(k) = c_k = \frac{1}{T} \int_{-T/2}^{T/2} dt f(t) \exp(-2\pi i k t / T),$$

and a common choice for the harmonics is $(1/\sqrt{T}) \exp(2\pi i k t / T)$. As a small addendum, if the signal (function) $f(t)$ is real and even one has $f(-t) = f(t)$. If f is an even function, then \hat{f} is also an even function which leads to $\hat{f}(-k) = \hat{f}(k)$.

Sinusoidal example

For the sinusoidal example the period is $T = 2\pi/\omega$. However, higher harmonics can also satisfy the periodicity requirement. In general, any force that satisfies the periodicity requirement can be expressed as a sum over harmonics,

$$f(t) = \frac{a_0}{2} + \sum_{n>0} a_n \cos(2n\pi t/T) + b_n \sin(2n\pi t/T).$$

Square well example

In the code discussed earlier, we used a square well as a our example. If we assume our period has length 1, we can define the square well function $f(t)$ as

$$f(t) = \begin{cases} 1 & 0 \leq t \leq \frac{1}{2} \\ -1 & \frac{1}{2} < t \leq 1 \end{cases}$$

The zeroth coefficient a_0 is zero since it is the average of the function $f(t)$ over the intergration domain $t \in [0, 1]$.

The coefficients

We find that the other coefficients are

$$\hat{f}(n) = c_n = \int_0^1 dt f(t) \exp(-2\pi i n t) = \frac{1}{i\pi n} (1 - \exp -i\pi n).$$

We note that $(1 - \exp -i\pi n)$ is zero when n is an even number and 2 when n is an odd number. We can then combine the positive and negative values of n using

$$\exp(2\pi i n t) - \exp(-2\pi i n t) = 2i \sin(2\pi n t),$$

and we have

$$f(t) = \frac{4}{\pi} \sum_{k=0}^{\infty} \frac{1}{2k+1} \sin(2\pi(2k+1)t).$$

Code for the Fourier Transforms

The code here uses the Fourier series applied to a square wave signal. The code here visualizes the various approximations given by Fourier series compared with a square wave with period $T = 0.2$ (dimensionless time), width 0.1 and max value of the force $F = 2$. We see that when we increase the number of components in the Fourier series, the Fourier series approximation gets closer and closer to the square wave signal.

```
import numpy as np
import math
from scipy import signal
import matplotlib.pyplot as plt

# number of points
n = 500
# start and final times
t0 = 0.0
tn = 1.0
# Period
T = 0.2
# Max value of square signal
Fmax = 2.0
# Width of signal
Width = 0.1
t = np.linspace(t0, tn, n, endpoint=False)
SquareSignal = np.zeros(n)
```

Inverse Fourier transform

The inverse Fourier transform is given by

$$\mathbf{F}^{-1}[g(y)] = \frac{1}{2\pi} \int_{-\infty}^{\infty} d\omega \exp i\omega y g(\omega).$$

The inverse Fourier transform of the product of the two functions $\hat{f}\hat{g}$ can be written as

$$\mathbf{F}^{-1}[(\hat{f}\hat{g})(x)] = \frac{1}{2\pi} \int_{-\infty}^{\infty} d\omega \exp i\omega x \hat{f}(\omega) \hat{g}(\omega).$$

Rewriting

We can rewrite the latter as

$$\mathbf{F}^{-1}[(\hat{f}\hat{g})(x)] = \int_{-\infty}^{\infty} d\omega \exp i\omega x \hat{f}(\omega) \left[\frac{1}{2\pi} \int_{-\infty}^{\infty} g(y) dy \exp -i\omega y \right] = \frac{1}{2\pi} \int_{-\infty}^{\infty} g(y) dy \int_{-\infty}^{\infty} d\omega \exp i\omega (x-y)$$

which is simply

$$\mathbf{F}^{-1}[(\hat{f}\hat{g})(x)] = \int_{-\infty}^{\infty} dy g(y) f(x-y) = (f * g)(x),$$

the convolution of the functions f and g .

Transforming to discrete variables

In the fourier transform c_n is transformed from a discrete variable to a continuous one as $L \rightarrow \infty$. We then replace A_n with $f(k)dk$ and let $n/L \rightarrow k$, and the sum is changed to an integral. This gives

$$f(x) = \int_{-\infty}^{\infty} dk F(k) \exp i(2\pi kx)$$

and

$$F(k) = \int_{-\infty}^{\infty} dx f(x) \exp -i(2\pi kx)$$

One way to interpret the Fourier transform is then as a transformation from one basis to another.

Discrete Fourier transform

Next we make another generalization by having a discrete function, that is $f(x) \rightarrow f(x_k)$ with $x_k = k\Delta x$ for $k = 0, \dots, N-1$. This leads to the sums

$$f_x = \frac{1}{N} \sum_{k=0}^{N-1} F_k \exp i(2\pi kx)/N,$$

and

$$F_k = \sum_{x=0}^{N-1} f_x \exp -i(2\pi kx)/N.$$

Although we have used functions here, this could also be a set of numbers.

Simple example

As an example we can have a set of complex numbers $\{x_0, \dots, x_{N-1}\}$ with fixed length N , we can Fourier transform this as

$$y_k = \frac{1}{\sqrt{N}} \sum_{j=0}^{N-1} x_j \exp i(2\pi jk)/N,$$

leading to a new set of complex numbers $\{y_0, \dots, y_{N-1}\}$.

Discrete Fourier Transformations

Consider two sets of complex numbers x_k and y_k with $k = 0, 1, \dots, n-1$ entries. The discrete Fourier transform is defined as

$$y_k = \frac{1}{\sqrt{n-1}} \sum_{j=0}^{n-1} \exp\left(\frac{2\pi i j k}{n}\right) x_j.$$

As an example, assume $x_0 = 1$ and $x_1 = 1$. We can then use the above expression to find y_0 and y_1 .

With the above formula we get then

$$y_0 = \frac{1}{\sqrt{2}} \left(\exp\left(\frac{2\pi i 0 \times 1}{2}\right) \times 1 + \exp\left(\frac{2\pi i 0 \times 1}{2}\right) \times 2 \right) = \frac{1}{\sqrt{2}}(1+2) = \frac{3}{\sqrt{2}}$$

and

$$y_1 = \frac{1}{\sqrt{2}} \left(\exp\left(\frac{2\pi i 0 \times 1}{2}\right) \times 1 + \exp\left(\frac{2\pi i 1 \times 1}{2}\right) \times 2 \right) = \frac{1}{\sqrt{2}}(1+2 \exp(\pi i))$$

More details on Discrete Fourier transforms

Suppose that we have a vector f of N complex numbers, $f_k, k \in \{0, 1, \dots, N-1\}$. Then the discrete Fourier transform (DFT) is a map from these N complex numbers to N complex numbers, the Fourier transformed coefficients \tilde{f}_j , given by

$$\tilde{f}_j = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \omega^{-jk} f_k$$

where $\omega = \exp\left(\frac{2\pi i}{N}\right)$.

Invert DFT

The inverse DFT is given by

$$f_j = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \omega^{jk} \tilde{f}_k$$

To see this consider how the basis vectors transform. If $f_k^l = \delta_{k,l}$, then

$$\tilde{f}_j^l = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \omega^{-jk} \delta_{k,l} = \frac{1}{\sqrt{N}} \omega^{-jl}$$

Orthonormality

These DFT vectors are orthonormal:

$$\sum_{j=0}^{N-1} \tilde{f}_j^{l*} \tilde{f}_j^m = \frac{1}{N} \sum_{j=0}^{N-1} \omega^{jl} \omega^{-jm} = \frac{1}{N} \sum_{j=0}^{N-1} \omega^{j(l-m)}$$

This last sum can be evaluated as a geometric series, but beware of the $(l - m) = 0$ term, and yields

$$\sum_{j=0}^{N-1} \tilde{f}_j^{l*} \tilde{f}_j^m = \delta_{l,m}$$

From this we can check that the inverse DFT does indeed perform the inverse transform:

$$f_j = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \omega^{jk} \tilde{f}_k = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \omega^{jk} \frac{1}{\sqrt{N}} \sum_{l=0}^{N-1} \omega^{-lk} f_l = \frac{1}{N} \sum_{k,l=0}^{N-1} \omega^{(j-l)k} f_l = \sum_{l=0}^{N-1} \delta_{j,l} f_l = f_j$$

Plans for the week of April 1-5

1. Finalize our discussion of QFTs
2. Set up calculations of QFTs
3. Implementing the phase estimation algorithm for finding eigenvalues