

Medical Imaging

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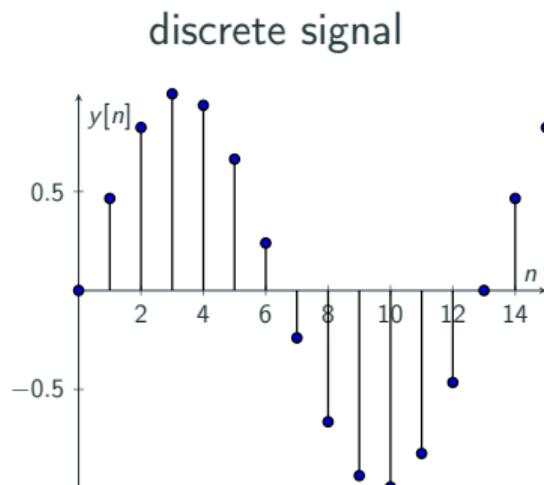
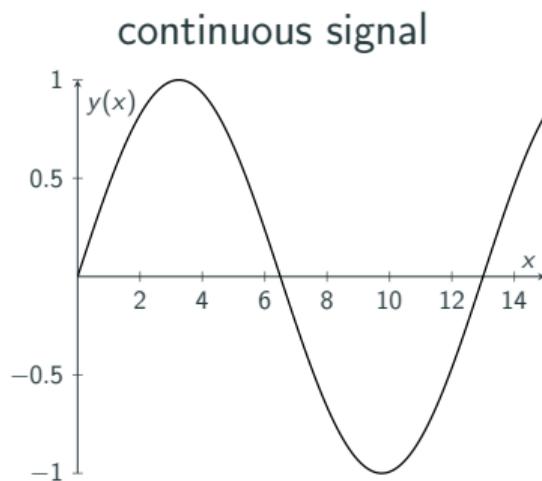
Signal Processing

- Understanding tomographic imaging techniques requires profound knowledge of signal processing
- In particular Fourier analysis plays an important role
- Basics of signal processing will be recapitulated

- A **signal** can be seen as a message that is send from a *sender* to a *recipient*.
- Usually they transport a physical quantity
- Signals can either be discrete or continuous (i.e. a function of \mathbb{N} or \mathbb{R})
- Typically a signal depends on *time* and/or *space*

Signal Examples

- $s(x)$: spatial 1D signal, e.g. detector array in CT
- $f(x, y)$: spatial 2D signal, e.g. slice through an object in CT
- $c(t)$: temporal 1D signal, e.g. ECG



Fundamental Signals

- Heaviside step function

$$\text{step}(x) = \begin{cases} 1, & \text{for } x \geq 0 \\ 0, & \text{for } x < 0 \end{cases} \quad (1)$$

- Rectangular function

$$\text{rect}(x) = \begin{cases} 1, & \text{for } |x| < \frac{1}{2} \\ 0.5, & \text{for } |x| = \frac{1}{2} \\ 0, & \text{for } |x| > \frac{1}{2} \end{cases} \quad (2)$$

- Sinc function

$$\text{si}(x) = \text{sinc}(x) = \frac{\sin(\pi x)}{\pi x} \quad (3)$$

- Dirac delta distribution

$$\begin{aligned}\delta(x - x_0) &= \begin{cases} 0, & \text{for } x \neq x_0 \\ \infty, & \text{for } x = x_0 \end{cases} \\ &= \lim_{\tilde{x} \rightarrow 0} \frac{1}{\tilde{x}} \text{rect}\left(\frac{x - x_0}{\tilde{x}}\right)\end{aligned}$$

Fundamental Signals

- The Dirac delta distribution is no function in a classical sense as ∞ can only be considered in the limes and is not a valid function value.
- The Dirac delta distribution can be defined over the integral

$$\int_{-\infty}^{\infty} \delta(x - x_0) f(x) dx = f(x_0) \quad (4)$$

In particular it holds that

$$\int_{-\infty}^{\infty} \delta(x - x_0) dx = 1 \quad (5)$$

I.e. Dirac delta distribution has unit area.

- A system \mathcal{L} takes as input a function $f(x)$ and outputs a function $g(x)$

$$g(x) = \mathcal{L}\{f(x)\}$$

- A system is linear if

$$\mathcal{L}\left\{\sum_i a_i f_i(x)\right\} = \sum_i a_i \mathcal{L}\{f_i(x)\}$$

- A system is said to be *time invariant / shift invariant* if

$$\mathcal{L}\{f(x - x_0)\} = \mathcal{L}\{f\}(x - x_0).$$

This means a shift of the input signal by x_0 leads to a shift by x_0 of the output signal.

- A system is said to be LTI or LSI if it is linear and time / shift invariant.

Impulse Response and Convolution

Any LSI system can be described by a convolution integral

$$\begin{aligned}g(x) &= \mathcal{L}\{f(x)\} \\ &= \int_{-\infty}^{\infty} f(\tilde{x})h(x - \tilde{x}) d\tilde{x} \\ &= (f * h)(x)\end{aligned}$$

h is the so-called *impulse response* or *point-spread function (PSF)* that is obtained by applying a Dirac delta to the LSI system

$$\begin{aligned}\mathcal{L}\{\delta(x)\} &= \int_{-\infty}^{\infty} \delta(\tilde{x})h(x - \tilde{x}) d\tilde{x} \\ &\stackrel{\text{subst. } y=x-\tilde{x}}{=} \int_{-\infty}^{\infty} \delta(x - y)h(y) dy \\ &\stackrel{\delta \text{ is even}}{=} \int_{-\infty}^{\infty} \delta(y - x)h(y) dy = h(x)\end{aligned}$$

Derivation of the Convolution Integral

Idea: express $f(x)$ as a sum of shifted rectangular functions

$$f(x) = \lim_{X_0 \rightarrow 0} \sum_{n=-\infty}^{\infty} f(nX_0) \text{rect} \left(\frac{x - nX_0}{X_0} \right)$$

Applying LSI properties yields

$$\begin{aligned} g(x) = \mathcal{L}\{f(x)\} &= \lim_{X_0 \rightarrow 0} \sum_{n=-\infty}^{\infty} f(nX_0) \mathcal{L} \left\{ \text{rect} \left(\frac{x - nX_0}{X_0} \right) \right\} \\ &= \int_{-\infty}^{\infty} f(\tilde{x}) \mathcal{L}\{\delta(x - \tilde{x})\} d\tilde{x} \\ &= \int_{-\infty}^{\infty} f(\tilde{x}) h(x - \tilde{x}) d\tilde{x} \end{aligned}$$

Fourier Transformation

It is known from analysis lectures that (almost) any p -periodic function $s(x)$ can be expanded into a *Fourier series*

$$s(x) = \sum_{n=-\infty}^{\infty} c_n e^{2\pi i \frac{nx}{p}}$$

consisting of complex sinus functions $e^{ix} = \cos x + i \sin x$. The Fourier coefficients c_n can be calculated by

$$c_n = \frac{1}{p} \int_{-p/2}^{p/2} s(x) e^{-2\pi i \frac{nx}{p}} dx$$

From Fourier Series to Fourier Transformation

In order to also express non-periodic functions by Fourier series (i.e. any general signal/function) one can consider the limes $p \rightarrow \infty$

$$\begin{aligned} s(x) &= \lim_{p \rightarrow \infty} \sum_{n=-\infty}^{\infty} c_n e^{2\pi i \frac{nx}{p}} \\ &= \int_{f:=-\frac{1}{p}}^{\infty} S(f) e^{2\pi i f x} df =: \mathcal{F}^{-1}\{S(f)\} \end{aligned}$$

Here, $f := \frac{n}{p}$ is the frequency and $S(f) = S(\frac{n}{p}) = c_n$ is a continuous function of frequency. This is in contrast to the discrete spectrum of the periodic Fourier series.

From Fourier Series to Fourier Transformation

The Fourier transform $S(f)$ of $s(x)$ can be calculated by

$$S(f) = \int_{-\infty}^{\infty} s(x)e^{-2\pi ifx} dx =: \mathcal{F}\{s(x)\}$$

$S(f)$ is often called the *spectrum* of $s(x)$. The Fourier relation is often indicated by

$$s(x) \text{ ---} \circ S(f)$$

Example

The Fourier transformation of the rectangular function can be calculated to be

$$\begin{aligned}\mathcal{F}\{\text{rect}(x)\} &= \int_{-\infty}^{\infty} \text{rect}(x)e^{-2\pi ifx} dx \\ &= \int_{-\frac{1}{2}}^{\frac{1}{2}} e^{-2\pi ifx} dx \\ &= \frac{i}{2\pi f} \left(e^{-i\pi f} - e^{i\pi f} \right) \\ &= \frac{\sin(\pi f)}{\pi f} = \text{sinc}(f)\end{aligned}$$

In the last step the euler formula $\sin(f) = \frac{e^{if} - e^{-if}}{2i}$ has been used.

Transfer Function

Instead of describing an LSI system as a convolution with the PSF $h(x)$ one can equivalently describe it in Fourier space by considering the *transfer function* $H(f) = \mathcal{F}\{h(x)\}$.

Theorem

A convolution $g(x) = (s * h)(x)$ in spatial space corresponds to a multiplication in Fourier space:

$$G(f) = S(f)H(f)$$

where $G(f) = \mathcal{F}\{g(x)\}$, $S(f) = \mathcal{F}\{s(x)\}$, and $H(f) = \mathcal{F}\{h(x)\}$.

$$\begin{aligned}G(f) &= \mathcal{F}\{g(x)\} = \mathcal{F}\{(s * h)(x)\} \\&= \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} s(\tilde{x})h(x - \tilde{x}) d\tilde{x} e^{-2\pi ifx} dx \\&= \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} s(\tilde{x})h(x - \tilde{x}) e^{-2\pi if(x-\tilde{x})} e^{-2\pi if\tilde{x}} d\tilde{x} dx \\&= \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} s(\tilde{x})h(z) e^{-2\pi if\tilde{x}} e^{-2\pi ifz} d\tilde{x} dz \\&= \int_{-\infty}^{\infty} s(\tilde{x}) e^{-2\pi if\tilde{x}} d\tilde{x} \int_{-\infty}^{\infty} h(z) e^{-2\pi ifz} dz \\&= S(f)H(f)\end{aligned}$$

Application of the Convolution theorem

- Efficient application of filter (low pass, high pass)
- Deconvolution / image sharpening

Discrete Fourier transformation

In order to numerically calculate Fourier coefficients one has to sample the Fourier integral a discrete and equidistant sampling points.

This makes the Fourier coefficients periodic so that the data in both domains (spatial and frequency) is discrete (line spectrum) and periodic.

Given a sequence s_0, \dots, s_{N-1} the discrete Fourier transformation (DFT) is defines as

$$S_m = \sum_{n=0}^{N-1} s_n e^{-2\pi i \frac{nm}{N}}, \quad m = 0, \dots, N - 1$$

Discrete Fourier transformation

The DFT is a unitary transformation and its inverse can be calculated by

$$s_n = \frac{1}{N} \sum_{m=0}^{N-1} S_m e^{2\pi i \frac{nm}{N}}, \quad n = 0, \dots, N-1$$

When defining the vectors $\mathbf{S} := (S_m)_{m=0}^{N-1}$ and $\mathbf{s} := (s_n)_{n=0}^{N-1}$, and the discrete Fourier matrix $\mathbf{F} := \left(e^{-2\pi i \frac{nm}{N}} \right)_{m=0, \dots, N-1; n=0, \dots, N-1}$ the DFT can be written in matrix-vector form

$$\mathbf{S} = \mathbf{F} \mathbf{s}$$

Fast Fourier Transformation

A naive implementation of the DFT would require $\mathcal{O}(N^2)$ arithmetic operations.

The fast Fourier transformation is a fast algorithm capable of carrying out FFT in only $\mathcal{O}(N \log N)$. It uses a recursive divide and conquer principle.

Important is that the FFT can only be applied to equidistant sampling positions.

Fast Fourier Transformation

Assumption: We will derive the DFT for $N = 2^r$.

Basic idea: split the sum into two sums, which are itself regular DFTs.

$$\begin{aligned} S_m &= \sum_{n=0}^{N-1} s_n e^{-2\pi i \frac{nm}{N}} \\ &= \sum_{n=0}^{\frac{N}{2}-1} s_n e^{-2\pi i \frac{nm}{N}} + \sum_{n=0}^{\frac{N}{2}-1} s_{n+\frac{N}{2}} e^{-2\pi i \frac{(n+\frac{N}{2})m}{N}} \end{aligned}$$

Now we will discuss two cases. $m = 2l$ (even) and $m = 2l + 1$ (odd) for $l = 0, \dots, \frac{N}{2} - 1$.

Fast Fourier Transformation

Case $m = 2l$:

$$\begin{aligned} S_{2l} &= \sum_{n=0}^{\frac{N}{2}-1} s_n e^{-2\pi i \frac{n2l}{N}} + \sum_{n=0}^{\frac{N}{2}-1} s_{n+\frac{N}{2}} e^{-2\pi i \frac{(n+\frac{N}{2})2l}{N}} \\ &= \sum_{n=0}^{\frac{N}{2}-1} s_n e^{-2\pi i \frac{nl}{\frac{N}{2}}} + \sum_{n=0}^{\frac{N}{2}-1} s_{n+\frac{N}{2}} e^{-2\pi i \frac{nl}{\frac{N}{2}}} \underbrace{e^{-2\pi i l}}_1 \\ &= \underbrace{\sum_{n=0}^{\frac{N}{2}-1} (s_n + s_{n+\frac{N}{2}}) e^{-2\pi i \frac{nl}{\frac{N}{2}}}}_{\text{DFT length } \frac{N}{2}} \end{aligned}$$

Fast Fourier Transformation

Case $m = 2l + 1$:

$$\begin{aligned} S_{2l+1} &= \sum_{n=0}^{\frac{N}{2}-1} s_n e^{-2\pi i \frac{n(2l+1)}{N}} + \sum_{n=0}^{\frac{N}{2}-1} s_{n+\frac{N}{2}} e^{-2\pi i \frac{(n+\frac{N}{2})(2l+1)}{N}} \\ &= \sum_{n=0}^{\frac{N}{2}-1} s_n e^{-2\pi i \frac{nl}{N}} e^{-2\pi i \frac{n}{N}} + \sum_{n=0}^{\frac{N}{2}-1} s_{n+\frac{N}{2}} e^{-2\pi i \frac{nl}{N}} \underbrace{e^{-2\pi i l}}_1 e^{-2\pi i \frac{n}{N}} \underbrace{e^{-2\pi i \frac{1}{2}}}_{-1} \\ &= \sum_{n=0}^{\frac{N}{2}-1} s_n e^{-2\pi i \frac{nl}{N}} e^{-2\pi i \frac{n}{N}} - \sum_{n=0}^{\frac{N}{2}-1} s_{n+\frac{N}{2}} e^{-2\pi i \frac{nl}{N}} e^{-2\pi i \frac{n}{N}} \\ &= \underbrace{\sum_{n=0}^{\frac{N}{2}-1} e^{-2\pi i \frac{n}{N}} (s_n - s_{n+\frac{N}{2}}) e^{-2\pi i \frac{nl}{N}}}_{\text{DFT length } \frac{N}{2}} \end{aligned}$$

Fast Fourier Transformation

Instead of one length N DFT we can apply two length $\frac{N}{2}$ DFTs.

The transformation can be applied again and in the second step we need to apply four length $\frac{N}{4}$ DFTs.

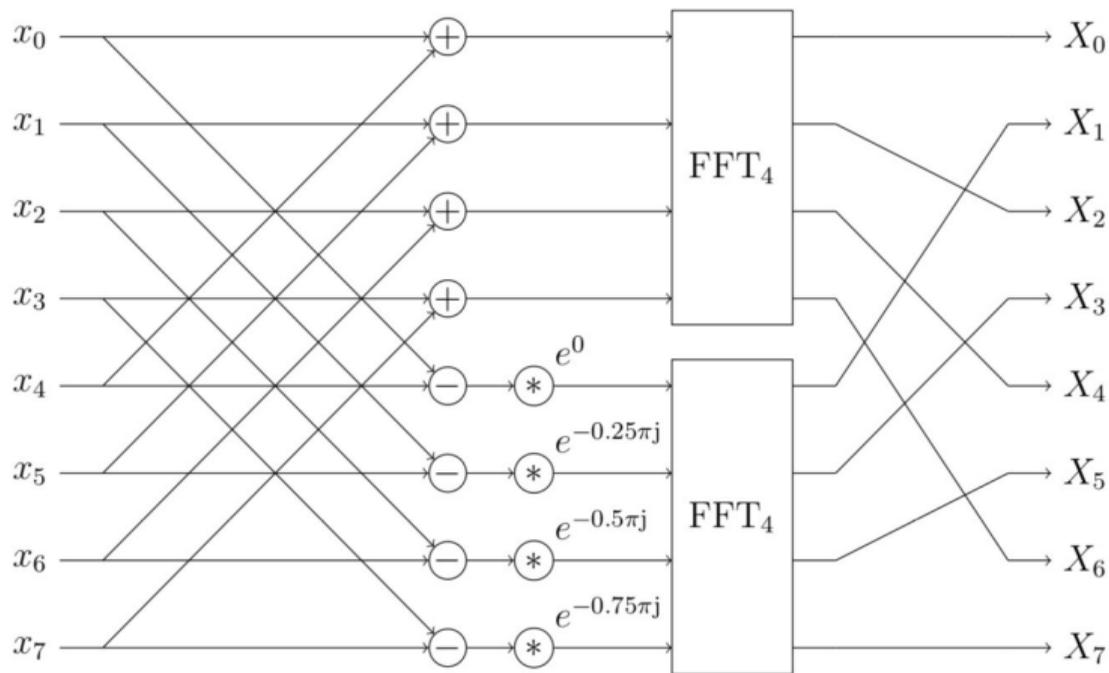
After $r = \log(N)$ steps we need to apply N DFTs of length 1.

In each step the algorithm needs $\frac{N}{2}$ additions, $\frac{N}{2}$ subtractions, and $\frac{N}{2}$ multiplications. The algorithmic complexity is thus $\mathcal{O}(N)$ arithmetic operations.

Since the DFT requires $\log(N)$ steps, the overall time complexity is $\mathcal{O}(N \log(N))$ compared to $\mathcal{O}(N^2)$ of an ordinary DFT.

Fast Fourier Transformation

The FFT is usually implemented inplace and can be visualized as follows



Fast Fourier Transformation

- We have derived the FFT for $N = 2^r$
- The same can be done for other basis 3,5,7,11,13, ...
- By prime factorization one can apply the FFT to general N
- Alternatively, one can pad the vector with zeros to the next power of 2
- The most popular FFT library is the FFTW (Fastest Fourier Transform in the West). It is used in the Julia package `FFTW.jl`