Discrete Fourier Transform (DFT)

Chapter 8 in the textbook

Course at a glance



The discrete-time Fourier transform (DTFT)

- The DTFT is useful for the theoretical analysis of signals and systems.
- But, according to its definition

$$X(e^{j\omega}) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n}$$

computation of DTFT by computer has several problems:

- The summation over n is infinite
- The independent variable w is continuous

The discrete Fourier transform (DFT)

- In many cases, only finite duration is of concern
 - The signal itself is finite duration
 - Only a segment is of interest at a time
 - Signal is periodic and thus only finite unique values
- For finite duration sequences, an alternative Fourier representation is DFT
 - The summation over n is finite
 - DFT itself is a sequence, rather than a function of a continuous variable
 - Therefore, DFT is computable and important for the implementation of DSP systems
 - DFT corresponds to samples of the Fourier transform

Part I: The discrete Fourier series

- The discrete Fourier series
- The Fourier transform of periodic signals
- Sampling the Fourier transform
- The discrete Fourier transform
- Properties of the DFT
- Linear convolution using the DFT

- A periodic sequence with period N $\widetilde{x}[n] = \widetilde{x}[n+rN]$
- Periodic sequence can be represented by a Fourier series, i.e. a sum of complex exponential sequences with frequencies being integer multiples of the fundamental frequency $(2\pi/N)$ associated with the $\tilde{x}[n]$ $\tilde{x}[n] = \frac{1}{N} \sum_{k} \tilde{X}[k] e^{j(2\pi/N)kn}$ The frequency of the periodic sequence.
 - Only N unique harmonically related complex exponentials since

$$e^{j(2\pi/N)(k+mN)n} = e^{j(2\pi/N)kn}e^{j2\pi mn} = e^{j(2\pi/N)kn}e^{j2\pi mn}$$

• so
$$\widetilde{x}[n] = \frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] e^{j(2\pi/N)kn}$$

The Fourier series coefficients

The coefficients

$$\widetilde{x}[n] = \frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] e^{j(2\pi/N)kn}$$
$$\widetilde{X}[k] = \sum_{n=0}^{N-1} \widetilde{x}[n] e^{-j(2\pi/N)kn}$$

• The sequence is periodic with period N $\widetilde{X}[k+N] = \sum_{n=0}^{N-1} \widetilde{x}[n]e^{-j(2\pi/N)(k+N)n} = \widetilde{X}[k]$

For convenience, define $W_N = e^{-j(2\pi/N)}$

Synthesis equation
$$\widetilde{x}[n] = \frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] W_N^{-kn}$$

Analysis equation
$$\widetilde{X}[k] = \sum_{n=0}^{N-1} \widetilde{x}[n] W_N^{kn}$$

Very similar equations → duality

Periodic Convolution



Figure 8.3 Procedure for forming the periodic convolution of two periodic sequences.

DFS of a periodic impulse train

Periodic impulse train $\widetilde{x}[n] = \sum_{r=-\infty}^{\infty} \delta[n-rN]$

- The discrete Fourier series coefficients $\widetilde{X}[k] = \sum_{n=0}^{N-1} \delta[n] W_N^{kn} = 1$
- By using synthesis equation, an alternative representation of x̃[n] is

$$\widetilde{x}[n] = \frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] W_N^{-kn} = \frac{1}{N} \sum_{k=0}^{N-1} W_N^{-kn} = \frac{1}{N} \sum_{k=0}^{N-1} e^{j(2\pi/N)kn}$$

Part II: The Fourier transform of periodic signals

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The Fourier transform of periodic signals

Fourier transform of complex exponentials

$$x[n] = \sum_{k} a_{k} e^{j\omega_{k}n}, \quad -\infty < n < \infty$$
$$X(e^{j\omega}) = \sum_{r=-\infty}^{\infty} \sum_{k} 2\pi a_{k} \delta(\omega - \omega_{k} + 2\pi r)$$

Fourier transform of x̃[n]

$$\widetilde{X}[n] = \frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] e^{j(2\pi/N)kn}$$
$$\widetilde{X}(e^{j\omega}) = \sum_{k=-\infty}^{\infty} \frac{2\pi}{N} \widetilde{X}[k] \delta(\omega - \frac{2\pi k}{N})$$

 $\widetilde{X}(e^{j\omega})$ has the required periodicity with period 2π

Fourier transform of a periodic impulse train

- Periodic impulse train
 $\widetilde{p}[n] = \sum_{r=-\infty}^{\infty} \delta[n-rN]$ The discrete Fourier series coefficients
 $\widetilde{P}[k] = \sum_{n=0}^{N-1} \delta[n] W_N^{kn} = 1$ Fourier transform
 $\widetilde{P}(e^{j\omega}) = \sum_{k=-\infty}^{\infty} \frac{2\pi}{N} \delta(\omega \frac{2\pi k}{N})$
- Finite duration signal x[n] (x[n] = 0 outside of [0, N-1]) Construct $\tilde{x}[n]$ $\tilde{x}[n] = x[n] * \tilde{p}[n] = x[n] * \sum_{r=-\infty}^{\infty} \delta(n-rN) = \sum_{r=-\infty}^{\infty} x(n-rN)$ Its Fourier transform

$$\widetilde{X}(e^{j\omega}) = X(e^{j\omega})\widetilde{P}(e^{j\omega}) = \sum_{k=-\infty}^{\infty} \frac{2\pi}{N} X(e^{j(2\pi/N)k}) \delta(\omega - \frac{2\pi k}{N})$$

The Fourier transform of periodic signals

Compare

$$\widetilde{X}(e^{j\omega}) = X(e^{j\omega})\widetilde{P}(e^{j\omega}) = \sum_{k=-\infty}^{\infty} \frac{2\pi}{N} X(e^{j(2\pi/N)k}) \delta(\omega - \frac{2\pi k}{N})$$
$$\widetilde{Y}(e^{j\omega}) = \sum_{k=-\infty}^{\infty} \frac{2\pi}{N} X(e^{j(2\pi/N)k}) \delta(\omega - \frac{2\pi k}{N})$$
First concernt it as For

$$\widetilde{X}(e^{j\omega}) = \sum_{k = -\infty} \frac{2\pi}{N} \widetilde{X}[k] \delta(\omega - \frac{2\pi}{N})$$

→ First represent it as Fourier series and then calculate Fourier transform

Conclude that

$$\widetilde{X}[k] = X(e^{j(2\pi/N)k}) = X(e^{j\omega})|_{\omega - (2\pi/N)k}$$

i.e. the DFS coefficients of $\tilde{x}[n]$ are samples of the Fourier transform of the one period of $\tilde{x}[n]$

$$x[n] = \begin{cases} \widetilde{x}[n], & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$$

<u>**Part III</u>: Sampling the Fourier transform**</u>

- The discrete Fourier series
- The Fourier transform of periodic signals
- Sampling the Fourier transform
- The discrete Fourier transform
- Properties of the DFT
- Linear convolution using the DFT

Sampling the Fourier transform

An aperiodic sequence and its Fourier transform

$$X(e^{j\omega}) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n} \leftrightarrow x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega})e^{j\omega n} d\omega$$

- Sampling the Fourier transform $\widetilde{X}[k] = X(e^{j\omega})|_{\omega = (2\pi/N)k} = X(e^{j(2\pi/N)k})$
 - generates a periodic sequence in k with period N since the Fourier transform is periodic in ω with period 2π



Figure 8.7 Points on the unit circle at which X(z) is sampled to obtain the periodic sequence $\tilde{X}[k] (N = 8)$.

Sampling the Fourier transform

Now we want to see if the sampling sequence X̃[k] is the sequence of DFS coefficients of a sequence x̃[n] this can be done by using the synthesis equation

$$\frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] W_N^{-kn}$$

$$= \frac{1}{N} \sum_{k=0}^{N-1} \left[\sum_{m=-\infty}^{\infty} x[m] e^{-j(2\pi/N)km} \right] W_N^{-kn}$$

$$= \sum_{m=-\infty}^{\infty} x[m] \left[\frac{1}{N} \sum_{k=0}^{N-1} W_N^{-k(n-m)} \right] = \sum_{m=-\infty}^{\infty} x[m] \widetilde{p}[n-m]$$

$$= \sum_{r=-\infty}^{\infty} x[n-rN]$$

$$= \widetilde{x}[n]$$
 A periodic sequence resulting from aperiodic convolution



Figure 8.8 (a) Finite-length sequence x[n]. (b) Periodic sequence $\bar{x}[n]$ corresponding to sampling the Fourier transform of x[n] with N = 12.

 In this case, the Fourier series coefficients for a periodic sequence are samples of the Fourier transform of one period





Figure 8.9 Periodic sequence $\bar{x}[n]$ corresponding to sampling the Fourier transform of x[n] in Figure 8.8(a) with N = 7.

- In this case, still the Fourier series coefficients for x[n] are samples of the Fourier transform of x[n]. But, one period of x[n] is no longer identical to x[n]
- This is just sampling in the frequency domain as compared in the time domain discussed before.

Sampling in the frequency domain

- The relationship between x[n] and one period of x̃[n] in the undersampled case is considered a form of time domain aliasing.
- Time domain aliasing can be avoided only if x[n] has finite length, just as frequency domain aliasing can be avoided only for signals being bandlimited.
- If x[n] has finite length and we take a sufficient number of equally spaced samples of its Fourier transform (specifically, a number greater than or equal to the length of x[n]), then the Fourier transform is recoverable from these samples, equivalently x[n] is recoverable from x̃[n].

Sampling in the frequency domain

Recovering x[n]

$$x[n] = \begin{cases} \widetilde{x}[n], & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$$

i.e. recovering x[n] does not require to know its Fourier transform at all frequencies

 Application: represent finite length sequence by using Fourier series (coefficients) → DFT

 $x[n] \rightarrow \widetilde{x}[n] \rightarrow DFS, \widetilde{X}[k] \rightarrow \widetilde{x}[n] \rightarrow x[n]$

Sampling the Fourier transform

• Fourier transform $X(j\Omega) = \int_{-\infty}^{\infty} x(t)e^{-j\Omega t} dt$ $x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\Omega)e^{j\Omega t} d\Omega$

Discrete-time Fourier transform

$$X(e^{j\omega}) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n}$$

$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) e^{j\omega n} d\omega$$

Discrete Fourier transform

$$X[k] = \sum_{n=0}^{N-1} x[n] e^{-j(2\pi/N)kn}$$
$$x[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{j(2\pi/N)kn}$$

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The discrete Fourier transform

Consider a finite length sequence x[n] of length N samples (if smaller than N, appending zeros) Construct a periodic sequence $\widetilde{x}[n] = \sum x[n-rN]$ Assuming no overlap btw x[n-rN] $\widetilde{x}[n] = x[(n \mod 0 N)] = x[((n))_N]$ Recover the finite length sequence $x[n] = \begin{cases} \widetilde{x}[n], & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$

To maintain a duality btw the time and frequency domains, choose one period of $\widetilde{X}[k]$ as the DFT $X[k] = \begin{cases} \widetilde{X}[k], & 0 \le k \le N-1 \\ 0, & \text{otherwise} \end{cases}$

The DFT

Periodic sequence and DFS coefficients $\widetilde{X}[k] = \sum_{n=0}^{N-1} \widetilde{X}[n] W_N^{kn}$ $\widetilde{x}[n] = \frac{1}{N} \sum_{k=0}^{N-1} \widetilde{X}[k] W_N^{-kn}$ Since summations are calculated btw 0 and (N-1) $X[k] = \begin{cases} \sum_{n=0}^{N-1} x[n] W_N^{kn}, & 0 \le k \le N-1 \\ 0, & \text{otherwise} \end{cases}$ Generally $X[k] = \sum_{n=0}^{N-1} x[n] W_N^{kn}$ $x[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k] W_N^{-kn}$ $x[n] = \begin{cases} \frac{1}{N} \sum_{k=0}^{N-1} X[k] W_N^{-kn}, & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$

The DFT

- A finite or periodic sequence has only N unique values, x[n] for 0<=n<N</p>
- Spectrum is completely defined by N distinct frequency samples
- DFT: uniform sampling of DTFT spectrum

 The DFT corresponds to sampling the z-transform of X(z) at N-points equally spaced in angle around the unit circle.





N points equally spaced on the unit circle



Example:

Let $f_s = 8000 \text{ Hz}$, Number of sample(N) = 1000 Frequency resolution = $\frac{f_s}{N} = \frac{8000}{1000} = 8Hz$ $f_0 = 0$, $f_1 = 8\text{Hz}$, $f_2 = 16 \text{ Hz}$, ..., $f_{999} = 8000\text{Hz}$

$$X[k] = \sum_{0}^{999} x[n] e^{-jn\left(\frac{2\pi k}{1000}\right)}$$

$$k = 0, 1, 2, 3, \ldots$$

Example :

 A speech signal is sampled at a rate of 20000 samples/sec. A sequence of length (N) 1024 samples is selected and the 1024-point DFT is computed.

(1) What is the time duration of segment of speech?
Duration = no of samples × sampling period.
= 1024 (1/20000) = 51.2 ms
(2) What is the frequency resolution (spacing in Hz) between the DFT values.

Resolution =
$$\frac{f_s}{N} = \frac{20000}{1024} = 19.531 Hz$$



The frequency resolution (Δf) can be made as small as desired by increasing the value of N (window size being analysed)

Padding with Zeros and frequency Resolution

DFT:
$$X[k] = \sum_{n=0}^{N-1} x[n] e^{-jn \frac{2\pi k}{N}}$$
 $k = 0, 1, 2, 3, \dots N$

- To obtain more points in the DFT sequence, we can always increase the duration of x[n] by adding additional zero-valued elements. This procedure is called padding with zeros.
- These zero-valued elements contribute nothing to the sum in the above equation, but act to decrease the frequency spacing (2π/N).

- The zero padding gives us a highdensity spectrum and provided a better displayed version for plotting.
- But it does not give us a high resolution spectrum because no new information is added to the signal.
- Only additional zeros are added in the data.

Three Sample Averager







Part V: Properties of the DFT

- The discrete Fourier series
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Properties of the DFT – linearity

Linearity

 $ax_1[n] + bx_2[n] \stackrel{DFT}{\leftrightarrow} aX_1[k] + bX_2[k]$

The lengths of sequences and their DFTs are all equal to the maximum of the lengths of $x_1[n]$ and $x_2[n]$

Circular shift of a sequence

Given

$$x[n] \stackrel{DFT}{\leftrightarrow} X[k]$$

$$x_{1}[n] \stackrel{DFT}{\leftrightarrow} X_{1}[k] = e^{-j(2\pi k/N)m} X[k]$$
Then

$$x_{1}[n] = \begin{cases} \widetilde{x}_{1}[n] = \widetilde{x}[n-m] = x[((n-m))_{N}], & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$$



Duality

DFT $x[n] \leftrightarrow X[k]$ DFT $X[n] \leftrightarrow Nx[((-k))_N], \quad 0 \le k \le N-1$

$$\begin{aligned} x_3[n] &= \sum_{m=0}^{N-1} \widetilde{x}_1[m] \widetilde{x}_2[n-m], \quad 0 \le n \le N-1 \\ &= \sum_{m=0}^{N-1} x_1[((m))_N] x_2[((n-m))_N], \quad 0 \le n \le N-3 \\ &= \sum_{m=0}^{N-1} x_1[m] x_2[((n-m))_N], \quad 0 \le n \le N-1 \end{aligned}$$

 In linear convolution, one sequence is multiplied by a time –reversed and linearly shifted version of the other. For convolution here, the second sequence is circularly time reversed and circularly shifted. So it is called an N-point circular convolution

 $x_3[n] = x_1[n] N x_2[n]$





•Circular convolution of x(n) and h(n) is defined as the convolution of h(n) with a periodic signal $x_p(n)$:

$$y_p(n) = x_p(n) * h(n)$$

where

$$x_p(n) = x (n \mod N), \qquad -\infty < n < \infty$$















N-point circular convolution can be computed using Matrix form:

$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ \vdots \\ y(N-1) \end{bmatrix} = \begin{bmatrix} h(0) & h(N-1) & h(N-2) & \dots & h(1) \\ h(1) & h(0) & h(N-1) & \dots & h(2) \\ h(2) & h(1) & h(0) & \dots & h(3) \\ \vdots \\ h(N-1) & h(N-2) & h(N-3) & \dots & h(0) \end{bmatrix} \begin{bmatrix} x(0) \\ x(1) \\ x(2) \\ \vdots \\ x(N-1) \end{bmatrix}$$

- Elements in each row are obtained by circularly rotating the elements of the previous row to the right by 1.

Example: Determine 4-point circular convolution of the two length-4 sequences $x(n) = \{1 \ 2 \ 0 \ 1\}$ and $h(n)=\{2 \ 2 \ 1 \ 1\}, 0 \le n \le 3$.

Method1: use DFT equation

Method2: Graphical (cylinders)

Method3: use Matrix computation method

Example: circular convolution of two rectangular pulses



N-point circular convolution of two sequences of length N.

Example: circular convolution of two rectangular pulses (continue)

Given two sequences of length L, assume that we add L zeros on its end, making an N=2L point sequence - referred to as zero padding



N-point circular convolution of two sequences of length L, where N=2L.



N-point circular convolution of two sequences of length L, where N=2L (continue). Note that by zero padding, we can use circular convolution to compute convolution of two finite length sequences.

Part VI: Linear convolution of the DFT

- The discrete Fourier series
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Linear convolution using the DFT

- Procedure
 - Compute the N-point DFTs X₁[k] and X₂[k] of two sequences x₁[n] and x₂[n], respectively
 - Compute the product of $X_3[k] = X_1[k]X_2[k]$ for $0 \le k \le N-1$
 - Compute the sequence x₃[n] = x₁[n] Nx₂[n] as the inverse DFT of X₃[k]
- As we know, the multiplication of DFTs corresponds to a circular convolution of the sequences. To obtain a linear convolution, we must ensure that circular convolution has the effect of linear convolution.

Example of circular convolution of two sequences



An Interpolation of circular convolution



Re-Arrangement of the operations of forming circular convolution



"Circular Convolution = Linear Convolution+Aliasing" $\hat{x}_{3}(n) = x_{1}(n) + x_{2}(n)$ $x_3(n) = x_1(n) \otimes x_2(n)$ $x_3(n) = \begin{bmatrix} +\infty \\ \sum_{r=-\infty}^{\infty} \hat{x}_3(n+rN) \\ R_N(n) \end{bmatrix}$

<u>Example</u> of forming circular convolution by linear convolution followed by aliasing:





Obtaining linear convolution through the use of circular convolution





Overlap – Add Method





