

## Signals and Systems MATLAB HW2

Deadline: 2019/04/26 before 23:59

### Discrete Fourier Transform

The objective of this section is to learn how to use MATLAB **fft** function.

#### 1. Background

In order to analyze the frequency domain of a finite duration and discrete-time signal  $x[n]$ ,  $n=1,2,\dots,N$ , its discrete Fourier transform (DFT) is defined as

$$X_k = \sum_{n=1}^N x[n] e^{-j \frac{2\pi}{N} (n-1)(k-1)}, k = 1, 2, \dots, N$$

It is observed that DFT is the sampled Fourier transform of a finite duration signal with frequency  $\omega = \frac{2\pi k}{N}$ . On the other hand, the inverse DFT (IDFT)

of  $X_k$  is defined as

$$x[n] = \frac{1}{N} \sum_{k=1}^N X_k e^{j \frac{2\pi}{N} (n-1)(k-1)}, n = 1, 2, \dots, N$$

The fast Fourier transform (FFT) is equivalent to DFT with reduced computational complexity as well as inverse FFT (IFFT) to IDFT. To calculate the DFT of the signal  $x[n]$  in MATLAB function, you may type:

$$\mathbf{X} = \text{fft}(\mathbf{x});$$

If you want to explicitly specify the length  $M$ , then you can type:

$$\mathbf{X} = \text{fft}(\mathbf{x}, M);$$

and the **fftshift** command swaps the first and the second half of the vector

$\mathbf{X}$  so that the frequency range is in  $[-\frac{N}{2}, \frac{N}{2}]$  (assuming  $N$  is even.)

However, for signals with infinite length, we have to truncate it so that it can be computed with MATLAB. Such truncation causes *Gibbs phenomenon* (pp. 200-201 of the textbook).

## 2. Questions

Program a MATLAB script (save as **fftsinc.m** file) to achieve the question 1.(a)(b) and 2.(c)(d)(e).

1. Let  $x(t)$  be a sinc function written as

$$x(t) = \frac{\sin(\pi t)}{\pi t}$$

Now,  $x(t)$  is sampled at the rate  $T_s = T / N_1$  so that  $x[n] = x(nT_s)$ ,

$n \in \{-N_1, -N_1 + 1, \dots, 0, \dots, N_1 - 1, N_1\}$  and  $N = 2N_1 + 1$ . Let  $N = 1001$  and  $T = 100$ .

(a) (20%) Use the MATLAB function **stem** to plot  $x[n]$  vs  $n$ .

(b) (20%) Use the MATLAB function **fft** to compute  $x[n]$ , and use the MATLAB function **plot** to plot the magnitude of the **fft** output vs frequency  $\omega$ . The zero frequency should be centered in your plot. Observe the *Gibbs phenomenon* in (b) and give some explanation for it in your report.

2. A way of mitigating *Gibbs phenomenon* is to multiply  $x(t)$  by a finite-duration signal  $w(t)$ , i.e.,  $y(t) = x(t)w(t)$ . The signal  $w(t)$  is called as the window function. A famous one is *Hanning* window, which is specifically written as

$$w(t) = \begin{cases} \frac{1}{2} [1 + \cos(\frac{2\pi |t|}{T_w})], & |t| \leq T_w / 2 \\ 0 & , \text{else} \end{cases}$$

where  $T_w$  denotes the duration of the window function.

Suppose  $w(t)$  is also sampled at a rate  $T_s = T / N_1$  so that  $w[n] = w(nT_s)$ ,

$n \in \{-N_1, -N_1 + 1, \dots, 0, \dots, N_1 - 1, N_1\}$  and  $N = 2N_1 + 1$ . Let  $T_w = T/2$ ,  $N = 1001$ , and  $T = 100$ .

(c) (20%) Use the MATLAB function **stem** to plot  $w[n]$  vs  $n$ .

(d) (20%) Use the MATLAB function **stem** to plot  $y[n]$  vs  $n$ , where  $y[n] = x[n]w[n]$ , and  $x[n]$  is the signal plotted in 1(a).

- (e) (20%) Use the MATLAB function **fft** to compute  $y[n]$  in (d), and use the MATLAB function **plot** to plot the magnitude of the **fft** output vs frequency  $\omega$ . The zero frequency should be centered in your plot. Observe the *Gibbs phenomenon* here and give some explanation for comparison with 1.(b) in your report.

Note: We expect that if executing your **fftsinc.m** file, there will be total five figures come out in order. (Question (a)~(e) has one figure respectively).

### 3.CEIBA Submission

- Please upload a compressed file (.zip, .rar or .tar), which includes your **m-files** (save as **fftsinc.m** file) and a **word file** (save as **report.doc** file). Please show the relevant plots mentioned above in the word file (report.doc) and some explanation.
- The compressed file name should be **ID\_MATLAB2**.  
(ex: B07901xxx\_MATLAB2)