332.501 Final Project

Fall 2021

Due date is December 19. No late work will be accepted.

1. (Verify the convolution relationship)

Let us first generate a row vector $\mathbf{a} = \begin{bmatrix} 1 & 1 & 1 & 1 & 0 & 0 & 0 \end{bmatrix}$ and make another vector $\mathbf{b} = \mathbf{a}$. Then

- 1.) Please plot the vector **a** in Matlab.
- 2.) Use the definition of the convolution in the textbook to compute the output $c = a \star b$ and plot the vector c in Matlab. That is, please derive the expression of the output $c = a \star b$ in your project document and plot your derived results.
- 3.) Use the Matlab command d = conv(a, b); to get the convolution output d and also plot d in Matlab.
- 4.) Compare with c and d in Matlab plots and make your comments.
- 2. (Use convolution concept to estimate the relative time delay)

• a)

- 1.) Let vectors $a = [1 \ 1 \ 1 \ 1 \ 0 \ 0 \ 0]$ and $b = [0 \ 0 \ 0 \ 1 \ 1 \ 1 \ 1 \ 0]$. Use subplot to plot them in one figure. What is the relationship between a and b with your observation?
- 2.) Let vector c = conv(a, b) and plot it in Matlab.
- 3.) Use Matlab command [max_val, ind] = max(c) to find the maximum output. What does ind mean? (i.e., is there any relationship between ind and the relative delay of a and b?)
- 4.) Change b to $b = [0 \ 0 \ 1 \ 1 \ 1 \ 1 \ 0 \ 0]$ and repeat the steps 2.) and 3.). What is the new ind value?
- 5.) Change b to $b = [1 \ 1 \ 0 \ 0 \ 0 \ 1 \ 1]$ and repeat the steps 2.) and 3.). What is the new ind value?
- 6.) From the observation of 4.) and 5.), what is your conclusion of the meaning for ind?

- b)

 - 2.) Repeat steps 2.) and 3.) in the previous section a) multiple times (say, 10 times). What is your observation of the output value ind? Does ind change during your 10 trials?
 - 3.) Keep a the same and change b = circshift(a, 3) + 10 * randn(8, 1).
 - 4.) Repeat steps 2.) and 3.) in the previous section a) multiple times (say, 20 times). What is your observation of the output value ind? Does ind change during your 20 trials? What are the values you get in your test?
 - 5.) Keep a the same and change b = circshift(a, -2) + 10 * randn(8, 1).
 - 6.) Repeat steps 2.) and 3.) in the previous section a) multiple times (say, 20 times). What is your observation of the output value ind? Does ind change during your 20 trials? What are the values you get in your test?

Remarks: In 2.), 4.) and 6.), even with multiple trials, you only need to print out one plot in each step in your final project document but have to keep all the records of the outputs value ind from the command $\max(c)$.

3. (Time-delay impacts on frequency domain)

- a)
 - 1.) Let vectors $a = [1 \ 1 \ 1 \ 1 \ 0 \ 0 \ 0]$ and $b = [0 \ 0 \ 1 \ 2 \ 3 \ 0 \ 0]$. Plot them in Matlab.
 - 2.) Let vector c = conv(a, b) and plot it in Matlab.
 - 3.) Let discrete Fourier transforms af = fft(a) and bf = fft(b). Plot the absolute values of af and bf, i.e., abs(af) and abs(bf) in Matlab.
 - 4.) Let cc = ifft(af. * bf) and plot cc.
 - 5.) Compare with cc and c and make comments of your results.
- b)
 - 1.) Let vectors $\mathbf{a} = [1 \ 1 \ 1 \ 1 \ 0 \ 0 \ 0]$ and $\mathbf{af1} = \mathbf{fft}(\mathbf{a}, 8)$. Plot $\mathbf{abs}(\mathbf{af1})$ in Matlab.

- 2.) Let af2 = fft(a, 16). Plot abs(af2) in Matlab.
- 3.) Compare with af1 and af2, what is your observation?
- c)
 - 1.) Let vector a = randn(8, 1) and af = fft(a).
 - 2.) Let vector b = circshift(a, 3) and bf = fft(b). Are af and bf the same?
 - 3.) Let index sequence nn = [0:7]' (column vector).
 - 4.) Let delay = $\exp(j * 2 * pi * nn * 3/8)$.
 - 5.) Let afa = delay.*bf and compare afa with af. What is difference between them?
 - 6.) Change vector b = circshift(a, -2) and bf = fft(b). Are af and bf the same?
 - 7.) Let delay = $\exp(j * 2 * pi * nn * (-2)/8)$.
 - 8.) Let afa = delay.*bf and compare afa with af. What is difference between them? Why? What is your conclusion now?

Remarks: In 8.), if the differences between two vectors are in the order 10^{-13} or even smaller, i.e., smaller than 1.0e - 14, we believe they are essentially the same.

4. From time domain to the frequency domain is just to change the view of the signal.

For example, it is very hard for one to tell what signals containing in a time series from the time domain observations. But it is much easier to distinguish the different frequency components in the frequency domain.

Filtering a signal using FFT.

Use the below reference URL to implement how you can manually filter a signal components.

For example, let us assume the frequency 7 is the signal we like to keep and the rest are interference or noise.

Use FFT and IFFT to filter out the single sinusoid with frequency 7 and eliminate the rest.

https://pythonnumericalmethods.berkeley.edu/notebooks/chapter24.04-FFT-in-Python.html

