Midterm Exam II - Solution

8:30-10:00pm, Tuesday, April 12, 2022

Name: _			
Section:	9:00 AM	12:00 PM	3:00 PM
NetID: _		_	
Score: _		-	

Problem	Pts.	Score
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1	10	
2	8	
3	16	
4	8	
5	8	
6	12	
7	16	
8	10	
9	12	
Total	100	

Instructions

- You may not use any books, calculators, or notes other than two <u>handwritten</u> two-sided sheets of 8.5" x 11" paper.
- Show all your work to receive full credit for your answers.
- When you are asked to "calculate", "determine", or "find", this means providing closed-form expressions (i.e., without summation or integration signs).
- Neatness counts. If we are unable to read your work, we cannot grade it.
- Turn in your entire booklet once you are finished. No extra booklet or papers will be considered.

(10 Pts.)

1. Find a closed-form expression for the frequency response $H_d(\omega)$ of a system with unit pulse response:

$$h[n] = \delta[n] + \delta[n-1] + \delta[n-2].$$

Find $|H_d(0)|$, $|H_d(\frac{\pi}{2})|$, $|H_d(\pi)|$, $\angle H_d(0)$, $\angle H_d(\frac{\pi}{2})$, and $\angle H_d(\pi)$.

Solution: By definition of the DTFT:

$$H_d(\omega) = \sum_{n = -\infty}^{\infty} h[n]e^{-j\omega n} = 1 + e^{-j\omega} + e^{-2j\omega}$$
$$= e^{-j\omega}(e^{j\omega} + 1 + e^{-j\omega}) = e^{-j\omega}(1 + 2\cos(\omega))$$

where the last equality uses the fact that $\cos(\omega) = \frac{e^{-j\omega} + e^{j\omega}}{2}$. Then, the mangitude and phase of the frequency response is given as

$$|H_d(\omega)| = |1 + 2\cos(\omega)|, \quad \angle H_d(\omega) = \begin{cases} -\omega, & 1 + 2\cos(\omega) \ge 0\\ -\omega + \pi, & \text{otherwise} \end{cases}$$

By plugging the values of ω into $|H_d(\omega)|$ and $\angle H_d(\omega)$ we get the following:

$H_d(\omega) = e^{-j\omega} (1 + 2\cos(\omega))$	(4 Pts)
$ H_d(0) = 3$	(1 Pts)
$ H_d(\frac{\pi}{2}) = 1$	(1 Pts)
$ H_d(\pi) = 1$	(1 Pts)
$\angle H_d(0) = 0$	(1 Pts)
$\angle H_d(\frac{\pi}{2}) = -\pi/2$	(1 Pts)
$\angle H_d(\pi) = 0$	(1 Pts)

(8 Pts.)

- 2. Recall that an eigenfunction of a system is an input signal which appears at the output of the system scaled by a complex constant. Which of the following discrete-time signals could be eigenfunctions of any (both real and complex) stable LSI system? (note that ω_0 is a constant; -2 pt for each incorrect choice; minimum score for the whole problem =0).
 - (a) $5^{-n}u[n]$
 - (b) $e^{j2\omega_0 n}$
 - (c) $e^{j\omega_0 n} + e^{j2\omega_0 n}$
 - (d) $\cos(\omega_0 n)$

Solution: By definition, a sequence x[n] is called an *eigenfunction* of a system if the output of that system when inputting x[n] can be written as

$$y[n] = \lambda \cdot x[n]$$

for some real or complex-valued constant $\lambda \in \mathbb{C}$, which is called the *eigenvalue* correspondingly. Then we have:

(a) is **NOT** an eigenfunction for any stable LSI system. **Example**: consider an LSI system with difference equation y[n] = x[n-1]. Then the output when inputting $x[n] = 5^{-n}u[n]$ is

$$y[n] = 5^{-n+1}u[n-1] \neq \lambda \cdot 5^{-n}u[n], \ \forall \lambda \in \mathbb{C}$$

Then (a) is not an eigenfunction in general.

(b) is an eigenfunction for any stable LSI system, since by eigenfunction property

$$y[n] = H_d(2\omega_0) \cdot x[n]$$

where $\lambda = H_d(2\omega_0)$ is the corresponding eigenvalue and $H_d(\cdot)$ is the frequency response of this system. To prove this, let h[n] be the impulse response of the system, then the output y[n] when inputting $x[n] = e^{j2\omega_0 n}$ is

$$y[n] = x[n] * h[n] = \sum_{k=-\infty}^{\infty} h[k]x[n-k] = \sum_{k=-\infty}^{\infty} h[k] \cdot e^{j2\omega_0(n-k)}$$
$$= \underbrace{e^{j2\omega_0 n}}_{=x[n]} \cdot \underbrace{\sum_{k=-\infty}^{\infty} h[k]e^{-j2\omega_0 k}}_{=H_d(2\omega_0)}$$
$$= H_d(2\omega_0) \cdot x[n]$$

which holds for any sequence h[n] as long as the DTFT $H_d(\cdot)$ exists (which is guaranteed by the fact that the system is stable).

(c) is **NOT** an eigenfunction for any stable LSI system. Since from part (b), in this case the output is

$$y[n] = H_d(\omega_0)e^{j\omega_0 n} + H_d(2\omega_0)e^{j2\omega_0 n} \neq \lambda \cdot (e^{j\omega_0 n} + e^{j2\omega_0 n})$$

for some $\lambda \in \mathbb{C}$, where the last inequality holds as long as $H_d(\omega_0) \neq H_d(2\omega_0)$. Therefore, $e^{j\omega_0 n} + e^{j2\omega_0 n}$ is not an eigenfunction in general.

(d) is **NOT** an eigenfunction for any stable LSI system. This is because, similar to part (c), in this case the input is

$$x[n] = \cos(\omega_0 n) = \frac{e^{j\omega n} + e^{-j\omega n}}{2}.$$

Then the output is

$$y[n] = \frac{1}{2}H_d(\omega_0)e^{j\omega n} + \frac{1}{2}H_d(-\omega_0)e^{-j\omega n} \neq \lambda \cdot \frac{e^{j\omega n} + e^{-j\omega n}}{2}$$

for some $\lambda \in \mathbb{C}$, where the last inequality holds as long as $H_d(-\omega_0) \neq H_d(\omega_0)$. Therefore, $\cos(\omega_0 n)$ is not an eigenfunction in general.

(16 Pts.)

3. An ideal low-pass filter has unit pulse response $h_{lp}[n]$ and frequency response:

$$H_{\rm lp}(\omega) = \left\{ \begin{array}{ll} 1, & |\omega| < \frac{\pi}{5} \\ 0, & \frac{\pi}{5} \le |\omega| \le \pi \end{array} \right.$$

- (a) Determine the frequency response, $H_1(\omega)$, of a filter whose unit pulse response is $h_1[n] = (-1)^n h_{lp}[n]$, and plot it for $|\omega| \leq \pi$. What type of filter is this?
- (b) Determine the frequency response, $H_2(\omega)$, of a filter whose unit pulse response is $h_2[n] = 2h_{\rm lp}[n]\cos(\frac{\pi}{2}n)$, and plot it for $|\omega| \le \pi$. What type of filter is this?

Solutions

(a) We do not need to find $h_{\rm lp}[n]$ to complete this problem if we artfully use pairs and properties of the DTFT. In particular, we note that, given $e^{j\pi n} \leftrightarrow 2\pi \sum_{l=-\infty}^{\infty} \delta(\omega - \pi - 2\pi l)$,

$$h_{1}[n] = (-1)^{n} h_{lp}[n] = e^{j\pi n} h_{lp}[n] \Longrightarrow$$

$$H_{1}(\omega) = \frac{1}{2\pi} \int_{0}^{2\pi} H_{lp}(\theta) 2\pi \sum_{l=-\infty}^{\infty} \delta(\omega - \theta - \pi - 2\pi l) d\theta$$

$$= \frac{2\pi}{2\pi} \int_{0}^{2\pi} H_{lp}(\theta) \delta((\omega - \pi) - \theta) d\theta$$

$$= H_{lp}(\omega - \pi).$$

This can be seen more readily by the frequency-shifting property. Effectively, the multiplication of $(-1)^n$ ends up shifting $H_{lp}(\omega)$ to the right by π , with no additional scaling. This ends up forming a **high-pass filter**, as shown in the figure below. It is high pass because, on the interval of $[-\pi, \pi]$, it only passes "high frequencies", ones above a particular cutoff $|\omega| > \omega_c = 4\pi/5$.

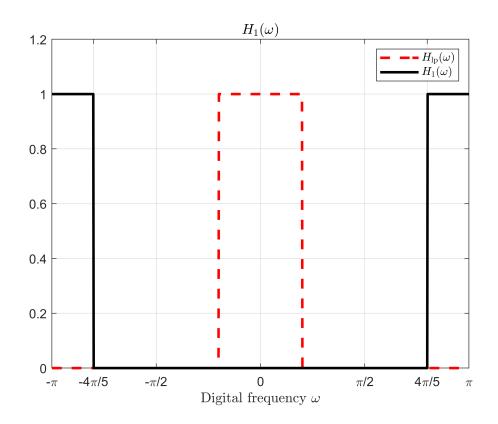
(b) The same process can be followed, noting that $2\cos\left(\frac{\pi}{2}n\right) = e^{j\frac{\pi}{2}n} + e^{-j\frac{\pi}{2}n}$. As such,

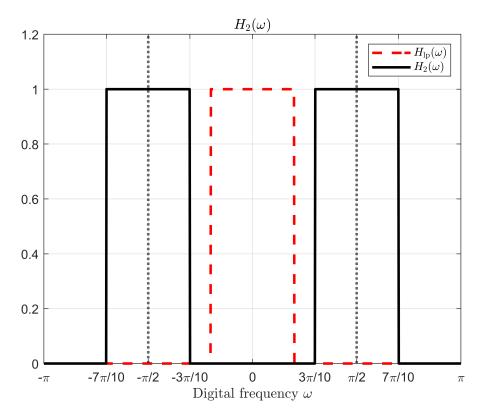
$$h_2[n] = 2\cos\left(\frac{\pi}{2}n\right)h_{\rm lp}[n] = e^{j\frac{\pi}{2}n}h_{\rm lp}[n] + e^{-j\frac{\pi}{2}n}h_{\rm lp}[n] \implies$$

$$H_2(\omega) = H_{\rm lp}\left(\omega - \frac{\pi}{2}\right) + H_{\rm lp}\left(\omega + \frac{\pi}{2}\right).$$

This results in a **band-pass filter**, where only frequencies in a particular pair of non-touching bands on the interval $\omega \in [-\pi, \pi]$ are allowed to pass, the rest being rejected.

Points are assigned as follows: +2 for each correct filter interpretation, +2 for each correct plot and +2 for each correct closed-form $H(\omega)$. If a student had a correct plot with preliminary work but no closed-form $H(\omega)$, points were awarded for $H(\omega)$. One point was deducted for each minor error, such as mislabelling an axis or having an incorrect scale factor.





(8 Pts.)

- 4. Let $(X[m])_{m=0}^{99}$ be the 100-point **DFT** of a **real-valued** sequence $(x[n])_{n=0}^{99}$ and $X_d(\omega)$ be the **DTFT** of x[n] zero-padded to infinite length. Choose all the correct answers (-2 pt for each incorrect choice; minimum score for the whole problem =0).
 - (a) $X[70] = X_d\left(-\frac{6\pi}{10}\right)$
 - (b) $X[70] = X_d\left(\frac{70\pi}{50}\right)$
 - (c) $|X[70]| = |X_d\left(\frac{70\pi}{100}\right)|$
 - (d) $\angle X[70] = -\angle X_d\left(\frac{3\pi}{5}\right)$

Solution

We know that by definition of the DFT,

$$X[k] = \sum_{n=0}^{99} x[n]e^{-j\frac{2\pi}{100}nk},$$

and by definition of the DTFT of the zero-padded version,

$$X_d(\omega) = \sum_{n=-\infty}^{\infty} x_{zp}[n]e^{-j\omega n} = \sum_{n=0}^{99} x[n]e^{-j\omega n},$$

as each term in x_{zp} outside the range $n \in \{0, 1, 2, ..., 99\}$ is zero. As such, we can write

$$X[k] = X_d \left(\omega_k = \frac{2\pi k}{100} \right).$$

This relationship will form the basis of how to analyze these four options.

(a) Notice that, from the above analysis, $X[70] = X_d\left(\frac{2\pi\times70}{100}\right) = X_d\left(\frac{140\pi}{100}\right) = X_d\left(\frac{14\pi}{10}\right)$. Using the fact that any DTFT is 2π -periodic, we know that for any integer l,

$$X_d \left(\frac{14\pi}{10} \right) = X_d \left(\frac{14\pi}{10} + 2\pi l \right) = X_d \left(\frac{14\pi - 20\pi l}{10} \right).$$

If we let l = 1, this shows

$$X_d\left(\frac{14\pi}{10}\right) = X_d\left(\frac{14\pi - 20\pi}{10}\right) = X_d\left(\frac{-6\pi}{10}\right).$$

Therefore, $X[70] = X_d\left(\frac{6\pi}{10}\right)$, so (a) is true.

- (b) Similar analysis shows that $X_d\left(\frac{70\pi}{50}\right) = X_d\left(\frac{140\pi}{100}\right)$, which is equal to X[70] as proved in (a). Thus, (b) is true.
- (c) Similar analysis shows that $|X_d\left(\frac{70\pi}{100}\right)| = |X_d\left(\frac{70\pi}{100} + 2\pi l\right)| = |X_d\left(\frac{70\pi + 200\pi l}{100}\right)|$, and given that x[n] is real, $|X_d(\omega)| = |X_d(-\omega)|$. However, no manipulation of the argument can show equality to $|X[70]| = |X_d\left(\frac{70\pi}{50}\right)|$, so (c) is false.
- $|X[70]| = |X_d(\frac{70\pi}{50})|$, so (c) is false. (d) Converting $X[70] = X_d(\frac{140\pi}{100})$, we can use conjugate symmetry to know that given x[n] is real, $\angle X_d(\omega) = -\angle X_d(-\omega)$. Upon similar manipulations as above,

$$\angle X_d \left(\frac{140\pi}{100} \right) = -\angle X_d \left(-\frac{140\pi}{100} \right) = -\angle X_d \left(-\frac{7\pi}{5} \right) = -\angle X_d \left(-\frac{7\pi}{5} + 2\pi \right) = -\angle X_d \left(\frac{3\pi}{5} \right),$$

so (d) is true.

(8 Pts.)

5. The following linear convolution

$${x_n}_{n=0}^8 * {h_n}_{n=0}^{16}$$

is to be evaluated using the DFT method. Namely,

$$\{x_n\}_{n=0}^8 * \{h_n\}_{n=0}^{16} = DFT^{-1}\{DFT\{x_n\} \cdot DFT\{h_n\}\}$$
 (1)

- (a) Determine the minimum number of zeros that should be padded to $\{x_n\}$ and $\{h_n\}$, respectively, before the DFTs are applied.
- (b) If the DFTs are to be calculated with a radix-2 FFT algorithm, how many zeros should now be padded to $\{x_n\}$ and $\{h_n\}$, respectively.

Solution

- (a) We must pad zeros onto x and h until their circular convolution equals their linear convolution. Denote the length of x as L=9 (x_n defined as $\{x_n\}_{n=0}^{L-1}$), and similarly the length of h as M=17. We must pad zeros onto both until they are of length N=L+M-1=25 in order for the circular convolution of these zero-padded signals to equal the linear convolution of the originals. As such, we pad N-L=M-1=16 zeros to x, and N-M=L-1=8 zeros to x.
- (b) The same process as in (a) can be followed, where we now must pad the sequences to a length both greater than L+M-1=25 and the length must be a power of two, i.e. the smallest integer ν such that $N=2^{\nu}>25$. The smallest ν such that this is the case is $\nu=5$ yielding $N=2^5=32$. As such, we pad N-L=32-9=23 zeros to x, and N-M=32-17=15 zeros to h.

Points were awarded as follows: 2 points were deducted for misidentifying the condition ($N_a = L + M - 1 = 25$, $N_b = 32$) for each sub-problem. One point was deducted for the mis-identification of L and M as 8 and 16, and each small error.

(12 Pts.)

- 6. Let $\{y_0, y_1, y_2, y_3, y_4\} = DFT^{-1}\{DFT\{x_0, x_1, 0, 0, 0\} \cdot DFT\{h_0, h_1, h_2, 0, 0\}\}.$
 - (a) Determine linear convolution $\{x_0, x_1\} * \{h_0, h_1, h_2\}$ (express your results in terms of y_0, y_1, \dots, y_4 .) Solution:

Since DFT length $5 \ge 2 + 3 - 1$, the linear convolution length, linear convolution will result in the same sequence as inverse DFT: $\{y_0, y_1, y_2, y_3, y_4\}$ (6 Pt)

(b) Determine DFT⁻¹{DFT $\{0,0,0,x_0,x_1\}$ · DFT $\{h_0,h_1,h_2,0,0\}$ } (express your results in terms of y_0,y_1,\cdots,y_4 .)

Solution:

$$DFT\{0,0,0,x_0,x_1\} = DFT\{x_0,x_1,0,0,0\} \cdot e^{-j\frac{6\pi k}{5}}$$
(3 Pt)

So our result

$$\tilde{Y} = \text{DFT}^{-1}\{\text{DFT}\{x_0, x_1, 0, 0, 0\} \cdot \text{DFT}\{h_0, h_1, h_2, 0, 0\} \cdot e^{-j\frac{6\pi k}{5}}\} = Y[\langle n-3 \rangle_5]$$

$$\tilde{Y} = \{y_2, y_3, y_4, y_0, y_1\}$$
(3 Pt)

(16 Pts.)

7. The input to a D/A converter is $\{x[n]\} = \{2, \frac{1}{1}, 0, 0, -2, 0, 4\}$ with sampling interval T. Determine the output of the D/A converter if the D/A converter is (a) a ZOH, and (b) an ideal D/A.

Solution:

D/A process of a signal is given by:

$$x_c(t) = \sum_{n = -\infty}^{\infty} x[n]g_a(t - nT)$$
(4 Pt)

For a ZOH, $g_a(t - nT) = rect(\frac{t - nT - \frac{T}{2}}{T})$ so that (4 Pt)

$$x_c(t) = x[-1]rect(\frac{t - \frac{T}{2} + T}{T}) + x[0]rect(\frac{t - \frac{T}{2}}{T}) + x[3]rect(\frac{t - \frac{T}{2} - 3T}{T}) + x[5]rect(\frac{t - \frac{T}{2} - 5T}{T})$$

$$=2rect(\frac{t-\frac{T}{2}+T}{T})+rect(\frac{t-\frac{T}{2}}{T})-2rect(\frac{t-\frac{T}{2}-3T}{T})+4rect(\frac{t-\frac{T}{2}-5T}{T})$$
 (2 Pt)

For an ideal D/A, $g_a(t - nT) = sinc(\frac{\pi(t - nT)}{T})$ (Here it's the **unnormalized** sinc function) (4 Pt)

$$x_c(t) = x[-1]sinc(\frac{\pi(t+T)}{T}) + x[0]sinc(\frac{\pi t}{T}) + x[3]sinc(\frac{\pi(t-3T)}{T}) + x[5]sinc(\frac{\pi(t-5T)}{T})$$

$$= 2sinc(\frac{\pi(t+T)}{T}) + sinc(\frac{\pi t}{T}) - 2sinc(\frac{\pi(t-3T)}{T}) + 4sinc(\frac{\pi(t-5T)}{T})$$

$$(2 \text{ Pt})$$

(10 Pts.)

8. Consider the following system with uniform sampling

$$x_a(t)$$
 T x_n $H_d(\omega)$ y_n D/A $y_a(t)$ (ideal)

The discrete-time system $H_d(\omega)$ is an ideal low-pass filter with cutoff frequency $\frac{\pi}{4}$.

- (a) If $x_a(t)$ is bandlimited to 4 kHz, what is the maximum value of T that will avoid aliasing in the A/D converter?
- (b) If $\frac{1}{T} = 8$ kHz, determine the maximum bandwidth of $x_a(t)$ allowed such that the overall system from $x_a(t)$ to $y_a(t)$ behaves as an LTI system.

Solution.

(a)

By Nyquist Criteria:

If the signal is bandlimited by B, or $F = \frac{B}{2\pi}$ Hz,

then we refer to twice the highest frequency in
$$x(t)$$
 as the Nyquist frequency, i.e. $2F$. (2 Pts)

When the A/D converter satises $\frac{1}{T} \geq 2F$, i.e. $T \leq \frac{1}{2F}$,

then it is possible to exactly recover the signal x(t) from its samples. (2 Pts)

therefore,
$$T_{\text{max}} = \frac{1}{8000} \text{ sec.}$$
 (1 Pt)

(b)

$$\Omega_0 T = 2\pi B_0 T = \omega_0 \tag{1 Pt}$$

$$-\Omega_0 T + 2\pi \ge \text{cutoff frequency} = \frac{\pi}{4}$$
 (2 Pts)

 $\Omega_0 \leq 14000\pi \text{ rad/s}$

or,
$$B_0 \le 7000 \text{ Hz}$$
 (2 Pts)

(12 Pts.)

- 9. A continuous-time signal $x_c(t) = \cos(8\pi t)$ is sampled at a rate of 80 Hz for five seconds to produce a discrete-time signal x[n] with length L = 400.
 - (a) Let X[k] be the length-L DFT of x[n]. At what value(s) of k will X[k] have the greatest magnitude?
 - (b) Suppose that x[n] is zero-padded to a total length of N = 1024. At what value(s) of k does the length-N DFT have the greatest magnitude?

Solution.

(a)
$$x[n] = x_c(nT) = cos(\frac{\pi}{10}n)$$
 (2 Pts)

$$X_d(\omega) = \pi \sum_{-\infty}^{\infty} \delta(\omega - \frac{\pi}{10} + 2\pi l) + \delta(\omega + \frac{\pi}{10} + 2\pi l)$$
(2 Pts)

Then, there will be two peaks in the DTFT $X_d(\omega)$ between $[0, 2\pi)$ which locate at $\frac{\pi}{10}$ and $\frac{19\pi}{10}$ respectively. Regarding the DFT, since it only gives samples of DTFT at frequencies $\frac{2\pi k}{L}$ for $k = 0, 1, \dots, L-1$,

we see that k = 20 and k = 380 correspond to the two peaks respectively. Therefore, both k = 20 and k = 380 will have the greatest magnitude of X[k].

(b) After zero-padding, now the length-L DFT have samples at frequencies $\frac{2\pi k}{L}$ for $k=0,1,\cdots,L-1$. As a result, we don't have the samples right at the peaks $\frac{\pi}{10}$ and $\frac{19\pi}{10}$ anymore. Instead, the largest magnitude of X[k] is located at the frequency which is closest to the peaks. This leads to $2\pi \cdot \frac{51}{1024}$ (that is closest to the peak at $\frac{19\pi}{10}$),

and the corresponding k choices are k = 51 and k = 973. (2 Pts)

Remark: One might be confused that how to show both k = 51 and k = 973 share the same amount of magnitude. By symmetry, since x[n] is real, we have |X[k]| = |X[N-k]| and therefore |X[51]| = |X[973]|.