# ECE 538 Digital Signal Processing I Final Exam 2015 Test Date: Tuesday, 16 December 2015 

## Cover Sheet

Test Duration: 120 minutes.
Open Book but Closed Notes.
Four $8.5 \times 11$ handwritten crib sheets allowed.
Calculators NOT allowed.
This test contains FOUR problems.
All work should be done in the proper space provided.

Good luck in this final exam!
Great having you in class this semester!
Have a joyous holiday season!

Problem 1. Consider the upsampler system below in Figure 1 and consider that the signal input to this system is

$$
\begin{equation*}
\text { Input Signal: } \quad x[n]=\frac{\sin \left(\frac{3 \pi}{4} n\right)}{\pi n} \frac{\sin \left(\frac{\pi}{4} n\right)}{\pi n} \tag{1}
\end{equation*}
$$



## Figure 1.

(a) Draw block diagram of efficient implementation of the upsampler system in Figure 1.
(b) For the ideal case where $h[n]=3 \frac{\sin \left(\frac{\pi}{3} n\right)}{\pi n}$, consider the filter $h_{0}[n]=h[3 n]$ and determine an expression for the output of this filter, denoted $y_{0}[n]=x[n] * h_{0}[n]$, when the input is the signal defined in Eqn (1) above.
(c) For the ideal case where $h[n]=3 \frac{\sin \left(\frac{\pi}{3} n\right)}{\pi n}$, consider the filter $h_{1}[n]=h[3 n+1]$ and determine an expression for the output of this filter, denoted $y_{1}[n]=x[n] * h_{1}[n]$, when the input is the signal defined in Eqn (1) above.
(d) For the ideal case where $h[n]=3 \frac{\sin \left(\frac{\pi}{3} n\right)}{\pi n}$, consider the filter $h_{2}[n]=h[3 n+2]$ and determine an expression for the output of this filter, denoted $y_{2}[n]=x[n] * h_{2}[n]$, when the input is the signal defined in Eqn (1) above.
(e) For the ideal case where $h[n]=3 \frac{\sin \left(\frac{\pi}{3} n\right)}{\pi n}$, determine an expression for the overall output of this system, denoted $y[n]$, when the input is the signal defined in Eqn (1) above.

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## Digital Signal Processing I

Problem 2. A second-order digital filter is to be designed from an analog filter having two poles in the s-plane at $p_{1}=-1+2 j$ and $p_{2}=-1-2 j$ and two zeros at $z_{1}=j$ and $z_{2}=-j$,

$$
H_{a}(s)=G \frac{\left(s-z_{1}\right)\left(s-z_{2}\right)}{\left(s-p_{1}\right)\left(s-p_{2}\right)}
$$

via the bilinear transformation method characterized by the mapping

$$
s=\frac{z-1}{z+1}
$$

(a) Is the resulting digital filter (BIBO) stable? Briefly explain why or why not.
(b) Denote the frequency response of the resulting digital filter as $H(\omega)$ (the DTFT of its impulse response). You are given that in the range $0<\omega<\pi$, there is only one value of $\omega$ for which $H(\omega)=0$. Determine that value of $\omega$.
(c) Draw a pole-zero diagram for the resulting digital filter. Give the exact locations of the poles and zeros of the digital filter in the z-plane.
(d) Plot the magnitude of the DTFT of the resulting digital filter, $|H(\omega)|$, over $-\pi<\omega<\pi$. You are given that $H(0)=0.8$. Be sure to indicate any frequency for which $|H(\omega)|=0$. Also, specifically note the numerical value of $|H(\omega)|$ for $\omega=\frac{\pi}{2}$ and $\omega=\pi$.
(e) Determine the difference equation for the resulting digital filter.

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## Digital Signal Processing I

Problem 3. Consider the length-7 DT signal below where the first value corresponds to $n=0$.

$$
x[n]=\{+1,+1,+1,-1,-1,+1,-1\}
$$

One way to compute autocorrelation is $r_{x x}[n]=x[n] * x^{*}[-n]$. Suppose we want to use DFT-based processing to compute autocorrelation. One issue is that the DFT assumes that signals start at $n=0$. So, let's time-shift the time-reversed signal to the right by $7-1=6$ to form $x[-(n-6)]$ so that it starts at $n=0 ; h[n]=x[-(n-6)]$ is the causal matched filter. Since $x[n]$ is real-valued,

$$
h[n]=x[-(n-6)] \stackrel{D T F T}{\longleftrightarrow} H(\omega)=X^{*}(\omega) e^{-j 6 \omega}
$$

(a) A 16-point DFT of $x[n]$ is computed and denoted as $X_{16}(k), k=0,1, \ldots 15$. We then form $Y_{16}(k)$ via the point-wise product defined below

$$
Y_{16}(k)=X_{16}(k) X_{16}^{*}(k) e^{-j 6 \frac{k 2 \pi}{16}} \quad k=0,1, \ldots, 15
$$

Finally, compute $y_{16}[n]$ as the 16-point Inverse DFT of $Y_{16}(k)$. Determine and list all 16 values of $y_{16}[n]$, for $n=0,1, \ldots, 15$. Show all work.
(b) A 13-point DFT of $x[n]$ is computed and denoted as $X_{13}(k), k=0,1, \ldots 12$. We then form $Y_{13}(k)$ via the point-wise product defined below

$$
Y_{13}(k)=X_{13}(k) X_{13}^{*}(k) e^{-j 6 \frac{k 2 \pi}{13}} \quad k=0,1, \ldots, 12
$$

Finally, compute $y_{13}[n]$ as the 13-point Inverse DFT of $Y_{13}(k)$. Determine and list all 13 values of $y_{13}[n]$, for $n=0,1, \ldots, 12$.
(c) An 10-point DFT of $x[n]$ is computed and denoted as $X_{10}(k), k=0,1, \ldots, 9$. We then form $Y_{10}(k)$ via the point-wise product defined below

$$
Y_{10}(k)=X_{10}(k) X_{10}^{*}(k) e^{-j 6 \frac{k 2 \pi}{10}} \quad k=0,1, \ldots, 9
$$

Finally, compute $y_{10}[n]$ as the 10-point Inverse DFT of $Y_{10}(k)$. Determine and list all 10 values of $y_{10}[n]$, for $n=0,1, \ldots, 9$.
(d) An 16 -point DFT of $x[n]$ is computed and denoted as $X_{16}(k), k=0,1, . ., 15$. We then form $Y_{16}(k)$ via the point-wise product defined below

$$
Y_{16}(k)=X_{16}(k) X_{16}^{*}(k) e^{-j 6 \frac{k 2 \pi}{16}} G_{16}(k) \quad k=0,1, \ldots, 15
$$

where $G_{16}(k)$ is the 16 -point DFT of the 2-tap filter $g[n]=\{1,1\}$. Finally, compute $x_{16}[n]$ as the 16 -point Inverse DFT of $Y_{16}(k)$. Determine and list all 16 values of $x_{16}[n]$, for $n=0,1, \ldots, 15$.

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## Digital Signal Processing I

## Problem 4.

(a) Consider sampling the DTFT below at $N=8$ equi-spaced frequencies over $0 \leq \omega<2 \pi$.

$$
X(\omega)=\frac{1}{1-0.8 e^{-j \omega}}+\frac{1}{1-0.9 e^{-j \omega}}
$$

That is, let $X_{8}(k)=X(2 \pi k / 8), k=0,1, \ldots 7$, where $X(\omega)$ is defined above. Now, compute $x_{8}[n]$ as the 8 -point Inverse DFT of $X_{8}(k)$. Determine a closed-form ("plug-and-chug") expression for $x_{8}[n]$ that prescribes the value of $x_{8}[n]$ for $n=0,1, \ldots, 7$. Show all work. Note: A closed-form expression contains NO summations and it is NOT a listing of numbers. Hint:

$$
\left.\frac{1}{1-(.9)^{8}}=1.7558\right] \quad \frac{1}{1-(.8)^{8}}=1.2016
$$

(b) Show how your answer changes when everything is the same as defined in part (a) except that the DTFT is now defined as below.

$$
X(\omega)=\frac{8}{1-0.8 e^{-j \omega}}-\frac{4}{1-0.9 e^{-j \omega}}
$$

(c) Show how your answer changes when everything is the same as defined in part (a) except that the DTFT is now defined as below.

$$
X(\omega)=\frac{e^{j 3 \omega}}{1-0.8 e^{-j \omega}}+\frac{e^{j 5 \omega}}{1-0.9 e^{-j \omega}}
$$

(d) Show how your answer changes when everything is the same as defined in part (a) except that the DTFT is now defined below. (That is, the two terms are now multiplied rather than being summed as in part (a).) Show all work.

$$
X(\omega)=\frac{1}{1-0.8 e^{-j \omega}} \frac{1}{1-0.9 e^{-j \omega}}
$$

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