

Faculty of Engineering Electrical and Computer Engineering Department Digital Signal Processing, ENCS 4310 Suggested Problems on Chapter Two Dr. Ashraf Al-Rimawi

## Problem #1:

- a) For what values of  $\omega$  is the signal  $x[n] = \cos(n\omega)$  periodic with period 8.
- b) Consider the following discrete-time signal x[n]

$$x[n] = 2\cos\left(\frac{\pi n}{2}\right) + 2\sin\left(\frac{\pi n}{5}\right) + 3\cos\left(\frac{\pi n}{4} - \frac{\pi}{6}\right)$$

- 1. Determine whether the following discrete-time signal x[n] is periodic or a periodic? If period, find the fundamental period and the fundamental frequency
- 2. Find the output y[n] for LTI system with the frequency response  $H(e^{j\omega}) = (1 + 2\cos(2\omega))e^{-j2\omega}$

## Problem #2:

Determine the following properties (linearity, time invariant, causality, stability, and invertibility) for the following systems

- a) y[n] = 30 x[n] 15 x[n-1]
- b) y[n] = n x[n]

c) 
$$y[n] = \sin(x[n])$$

- d) y[n] = 2x[n] x[n-1] + x[n+4] + 2y[n-1]
- e)  $y[n] = \sum_{k=-\infty}^{n} x[k] + u[n-1]$

## Problem #3:

(a) Apply one of the methods you have learnt to compute discrete-time convolution of the following two finiteduration signals:

 $h(n) = \{ 0.25, 0.25, 0.25, 0.25 \}$ , and  $x(n) = \{ 5, 0, 2, 0, -1, 0, 1, 0, 2, 0, 4, 0, 6, 0 \}$ , noting that every other sample of x(n) is zero. Suggest a potential DSP application of this particular convolution?

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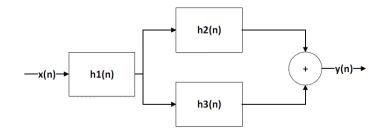
**Problem#4**: A digital filter has the impulse response  $h[n] = \begin{bmatrix} -1 & 0.5 & 0 & 2 \end{bmatrix}$  and is implemented at a sampling frequency of 8 KHz, an input signal  $x[n] = \begin{bmatrix} 0.5 & 1 & 0 \end{bmatrix}$  is sent through the filter, given the output y[n]. Note, the underline sample is the sample at n = 0.

- a. Find the output y[n].
- b. Find the frequency response of this filter, i.e,  $H(e^{j\omega})$ .
- c. Find the output response of this filter when input is a cosine signal with unity amplitude and frequency 1KHz.

#### Problem #5:

A signal x[n] has the Fourier transform  $X(e^{j\omega}) = \frac{1}{(1-0.5e^{-j\omega})}$ , determine the Fourier Transform of  $x_1[n] = e^{-j3\pi n}x(n-4)$ .

**Problem #6:** Consider the system in the following system



 $h_1[n] = \delta[n-2], h_2[n] = \left(\frac{2}{5}\right)^n u[n], \text{ and } h_3[n] = 2\delta[n].$ 

- a) Find the expression of the total impulse response, h[n]
- b) Find the frequency response of the whole system,  $H(e^{j\omega})$
- c) Check if this system stable and/ or causal? Justify your answer.
- d) Use the frequency response expression in part (b) to write the difference equation that characterize this system.

**Problem #7:** Consider the following sequences

x[n] = [1 3 - 2 4] and y[n] = [2 - 1 4 - 2]

- a) Determine the autocorrelation  $R_{xx}(k)$ , and  $R_{yy}(k)$
- b) Determine the cross-correlation  $R_{xy}(k)$
- c) Determine the Energy for each signal
- d) Determine the autocorrelation  $C_{xx}$ , and  $C_{xy}$

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#### Problem #8 (Matlab Assignment):

A. Consider the impulse response of digital filter is given by

 $h(n) = [0.25\ 0.25\ 0.25\ 0.25]$ 

Write code matlab to recover samples of an audio signal. Use the following steps:

- Record your voice, say Good Morning.
- Read your recorded file using WAVREAD(.) command.
- Set everyother sample to zero to produce an array. Call this x(n).
- Assume x(n) is the input of the proposed digital filter. Evaluate the output of this filter by using Convolution theorem between x(n) and h(n). Use CONV(.) command.
- Plot (or stem) a 20 samplessection of y(n) together with the same 20 samplessection of x( n) and compare them.
- Submit the plot and Matlab code and any other observation.
- B. Consider the following sequences  $x[n] = [1 \ 3 2 \ 4]$  and  $y[n] = [2 1 \ 4 2]$

Write a Matlab code to calculate the correlation coefficient between these two signals.

GOOD Luck