

**EE113 Digital Signal Processing**  
Spring 2019

**Homework 2**

Due: Monday, April 15 at 12pm

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**Total: 100 points**

**Problem 1. (20 points)** Is the system linear? causal? time-invariant? stable?

(a) (10 points) Consider the moving average system with exponential weighting

$$y[n] = \frac{1}{M+1} \sum_{k=0}^M \lambda^k x[n-k]$$

where  $|\lambda| < 1$ .

(b) (10 points) Consider the system

$$y[n] = \frac{1}{|n|+1} x[-n^2]$$

**Problem 2. (10 points)** The response of a linear system to  $0.5^n u[n]$  is  $u[n]$  and to  $0.5^{n-1} u[n-1]$  is  $u[n-1]$ . Is the system time-invariant? Why or why not?

**Problem 3. (10 points)** Consider a first-order differential equation in the form

$$\frac{dy_a}{dt} + A y_a(t) = A x_a(t)$$

The derivative can be approximated using the backward difference

$$\frac{dy_a}{dt} \approx \frac{y_a(t) - y_a(t-T)}{T}$$

where  $T$  is the step size.

(a) (5 points) Using the approximation for the derivative in the differential equation, express  $y_a(t)$  in terms of  $y_a(t-T)$  and  $x_a(t)$ .

(b) (5 points) Convert the differential equation to a difference equation by defining discrete-time signals

$$\begin{aligned} x[n] &= x_a(nT) \\ y[n] &= y_a(nT) \\ y[n-1] &= y_a(nT-T) \end{aligned}$$

Show that the resulting difference equation corresponds to an exponential smoother. That is, the difference equation is of the form

$$y[n] = (1-\alpha)y[n-1] + \alpha x[n].$$

Determine the parameter  $\alpha$  in terms of  $A$  and  $T$ .

**Problem 4. (15 points)**

(a) (5 points) Compute the convolution of  $x[n]$  and  $h[n]$ ,  $y[n] = x[n] * h[n]$ :

$$x[n] = \begin{matrix} \uparrow \\ n=0 \end{matrix} \left\{ 0, 1, 2, 3, 4, \quad 3, 2, 1, 0 \right\} \text{ and } h[n] = \begin{matrix} \uparrow \\ n=0 \end{matrix} \left\{ 1, \quad 1, -1, -1 \right\}.$$

(b) **(10 points)** Let  $y[n]$  be the convolution of two discrete-time signals  $x[n]$  and  $h[n]$ , that is

$$y[n] = x[n] * h[n]$$

Show that time shifting either  $x[n]$  or  $h[n]$  by  $m$  samples causes  $y[n]$  to be time shifted by  $m$  as well. Mathematically prove that

$$x[n - m] * h[n] = y[n - m]$$

and

$$x[n] * h[n - m] = y[n - m]$$

**Problem 5. (25 points)** (Use MATLAB for part (c) of this problem) Consider the DTLTI system shown in Fig. 1:

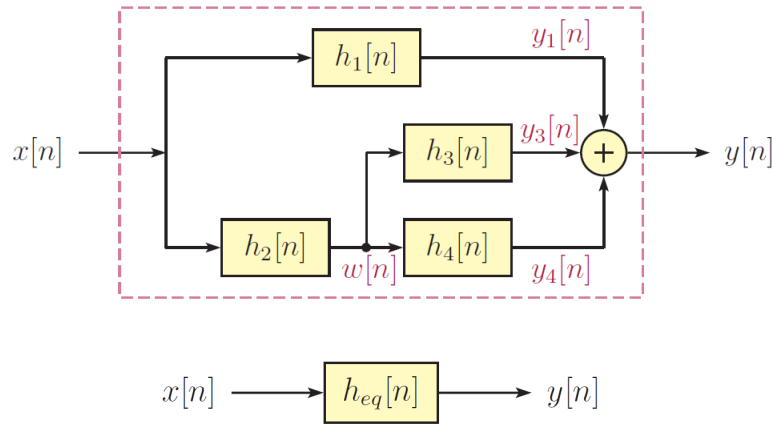


Figure 1: System Block diagram for Problem 5

(a) **(5 Points)** Express the impulse response of the system as a function of the impulse responses of the subsystems.

(b) **(10 points)** Let

$$h_1[n] = e^{-0.1n}u[n]$$

$$h_2[n] = h_3[n] = u[n] - u[n - 3]$$

and

$$h_4[n] = \delta[n - 2]$$

Determine the impulse response  $h_{eq}[n]$  of the equivalent system. Note that you need to do the convolution on paper, but are free to use MATLAB (“conv” command) to verify if your calculations are correct.

(c) **(10 points)** Let the input signal be a unit-step, that is,  $x[n] = u[n]$ . Determine and plot  $y[n]$  in MATLAB.

(Include the MATLAB plots for  $n = 0, 1, \dots, 10$  samples only, and use “stem” for the plot command)

**Problem 6. (20 points) (Use MATLAB for this problem)**

Consider the problem 4 of Exercise 1, let's say that the goal of that problem is to play the music slower and still make it sound natural. In this problem, we will implement a MATLAB program to achieve the same goal using a system of upsampler and smoother as shown in the figure below.

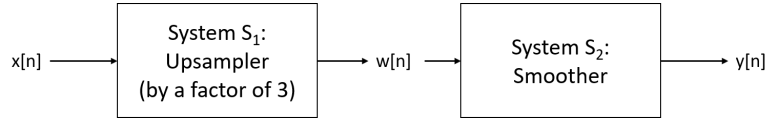


Figure 2: System block diagram for Problem 5

Write a MATLAB program and report your observations for a system that has the cascade of two systems,  $S_1$  and  $S_2$ , as shown in the Fig. 2. System  $S_1$  is an upsampler, such that the response of the system to an input of  $x[n]$  is  $w[n]$ :

$$w[n] = \begin{cases} x[n/3], & \text{if } n/3 \text{ is an integer} \\ 0, & \text{otherwise} \end{cases}$$

The system  $S_2$  is a smoother that can be implemented using:

- (a) **(10 points)** Moving average filter. Choose window lengths to be 1, 5, 10, 50, and 100, and play the output audio signal  $y[n]$  from the system corresponding to each window length. Report the window length that makes the output signal  $y[n]$  sound better, and explain the differences, based on what you hear, as you change the window lengths.
- (b) **(10 points)** Exponential smoother. Report the best  $\alpha = 0, 0.3, 0.5, 0.8, 1$  that makes the output signal  $y[n]$  sounds better, and explain the differences with the other values of  $\alpha$ .

Consider the output of the overall system is  $y[n]$ .

The snippet of the code and the audio file are available in a compressed file “HW2\_Prob6.rar” in week 2 folder.