EECS 120 Signals & Systems Ramchandran

## Homework 7 Due: Thursday, October 26, 2006, at 5pm Homework 7 GSI: Mark Johnson

Reading OWN Chapter 7.

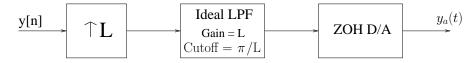
Problem 1 (Downsampling.) OWN Problem 7.35

**Problem 2** (Upsampling.) OWN Problem 7.49

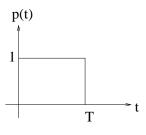
**Problem 3** (Zero-Order and First-Order Holds.) OWN Problem 7.50

**Problem 4** (Oversampled D/A.)

An oversampled D/A is implemented using a digital interpolator and a zero-order hold, as shown below.



The interpolator inserts L-1 zeros between each sample of y[n]. The zero-order hold has period T = 1/L, i.e., the duration of the ZOH is dependent on the upsampling factor. As usual, the impulse response of the ZOH is given by



The discrete time signal y[n] has DTFT  $Y(e^{j\Omega}) = \begin{cases} 1 & |\Omega| < 3\pi/4 \\ 0 & 3\pi/4 < |\Omega| < \pi \end{cases}$ 

(a) Plot the magnitude spectrum  $|Y_a(j\omega)|$  for three cases: L = 1, L = 2, and L = 4.

Plot each spectrum over the interval  $-5\pi/T \le \omega \le 5\pi/T$  (remember that T is dependent on L). You may either use Matlab or sketch the plots by hand. Be sure to accurately label all important points on the  $\omega$ -axis.

(b) Plot the magnitude spectrum  $|Y_a(j\omega)|$  when the ZOH is replaced with an ideal D/A converter.

(c) For the four plots in parts (a) and (b), calculate the magnitude of the largest component of  $Y_a(j\omega)$  outside the band  $|\omega| \leq \pi/T$ 

## Problem 5 (Aliasing.)

The signal x(t) has a real-valued spectrum

$$X(j\omega) = e^{-|\omega|}, \ -\infty \le \omega \le \infty.$$
(1)

(a) To avoid aliasing, we could pass the signal x(t) through an ideal low-pass filter of width  $W = \frac{\pi}{T}$  before sampling with a sampling interval T. Denote the resulting samples by  $x_T[n]$ . As we have seen, the spectrum  $X_T(e^{j\Omega})$  of the discrete-time signal  $x_T[n]$  can be expressed as

$$X_T(e^{j\Omega}) = \frac{1}{T} e^{-\left|\frac{\Omega}{T}\right|}, \quad \text{for } -\pi \le \Omega \le \pi.$$
(2)

Sketch the spectrum  $X_T(e^{j\Omega})$  for  $-2\pi \leq \Omega \leq 2\pi$ . Carefully label the frequency axis.

(b) Suppose we directly sample x(t) (without prior low-pass filtering) with a sampling interval T. Denote the samples by  $\tilde{x}_T[n] = x(nT)$ . Since x(t) is not bandlimited, aliasing occurs. Draw the shifted copies of  $X(j\omega)$  that make up the spectrum  $\tilde{X}_T(e^{j\Omega})$  of the discrete-time signal  $\tilde{x}_T[n]$ , without summing them up. Identify the parts which cause aliasing.

(c) Based on your sketch from Part (b), the spectrum  $\tilde{X}_T(e^{j\Omega})$  of the discrete-time signal  $\tilde{x}_T[n]$  can be written as

$$\tilde{X}_T(e^{j\Omega}) = \frac{1}{T} \sum_{n=-\infty}^{\infty} e^{-\left|\frac{\Omega - n2\pi}{T}\right|}.$$
(3)

Decompose the sum into three parts: one term for n = 0, an infinite sum for negative n, and an infinite sum for positive n. Assuming  $-\pi \leq \Omega \leq \pi$ , evaluate the two infinite sums. *Hint:* In each of the two infinite sums *individually*, carefully get rid of the magnitude  $|\cdot|$  in the exponent.

## **Problem 6** (Aliasing - continued.)

This problem refers to the signals  $x_T[n]$  and  $\tilde{x}_T[n]$  from Problem 5. A good measure of the error due to aliasing is the mean-squared error between  $x_T[n]$  and  $\tilde{x}_T[n]$ , defined as

$$E = \sum_{n=-\infty}^{\infty} |\tilde{x}_T[n] - x_T[n]|^2.$$
(4)

(a) Calculate E.

(b) Determine the limiting value of E as T tends to zero. Explain why your result makes sense.

**Problem 7** (AM Communication Systems.) OWN Problem 8.22