EEE401F EXAM DIGITAL SIGNAL PROCESSING

University of Cape Town Department of Electrical Engineering

June 2004

3 hours

PART A

Answer all of the following questions.

1. You are given the two signals $x_1[n]$ and $x_2[n]$ shown below. The signal values are zero outside of the range of n shown:



(a) Use graphical methods to find the linear convolution $y[n] = x_1[n] * x_2[n]$.

(b) Use the expression for the convolution sum directly to find the values of y[n] for n = -1 and n = 1.

(10 marks)

Information

- The exam is closed-book.
- There are two parts to this exam.
- Part A has six questions totalling 70 marks. You must answer all of them.
- Part B has two questions totalling 30 marks. You must answer all of them.
- You have 3 hours.





- (a) Find the Fourier transform of the x[n] and sketch its magnitude over the range -2π to 2π. Does the signal have predominantly low or high-frequency content?
- (b) Calculate the 4-point DFT X[k] of the signal.

(c) If $\tilde{x}[n]$ is the inverse DFT of X[k], what is the value of $\tilde{x}[-2]$ and $\tilde{x}[4]$?

(10 marks)

3. (a) Find the z-transform of the exponentially-decaying signal

$$\begin{array}{c|c} & 1 & 0.8 \\ 0.64 & & x[n] \\ \hline & 0 & & & \\ \end{array}$$

(b) What is the z-transform of the signal y[n] = x[n] * x[n]? Use this result, along with the transform pair

$$na^n u[n] \xleftarrow{\mathcal{Z}} \frac{az^{-1}}{(1-az^{-1})^2},$$

to find y[n].

(c) Use either a power-series expansion or the method of long division to find the first 4 values of the causal signal corresponding to the z-transform

$$X(z) = \frac{1}{1.2 + z}.$$

Where is the region of convergence in the z-plane?

(15 marks)

4. The figure below shows a second-order system:



- (a) Write down an expression for the z-transform of the system.
- (b) Under the assumption that the system is causal, draw a pole-zero plot of the system in the z-plane. Is the system stable?
- (c) Find the impulse response of the system under the assumptions given. You may need the transform pair

$$a^n u[n] \xleftarrow{\mathcal{Z}} \frac{1}{1 - az^{-1}}, \qquad |z| > |a|.$$

(15 marks)

5. The pole-zero plot for a filter is as follows:



(a) Use graphical methods to roughly sketch the magnitude and phase response of the filter.

- (b) Describe the effect the filter will have on an input signal.
- (c) What are the possible regions of convergence for the filter? Which will correspond to a causal filter, and which to a stable filter?

(10 marks)

6. In the system



 $X_c(j\Omega)$ and $H(e^{j\omega})$ are as shown below:



Sketch and label the Fourier transform of $y_c(t)$ for each of the following cases:

(a) $1/T_1 = 1/T_2 = 10^4$. (b) $1/T_1 = 1/T_2 = 2 \times 10^4$. (c) $1/T_1 = 2 \times 10^4, 1/T_2 = 10^4$. (d) $1/T_1 = 10^4, 1/T_2 = 2 \times 10^4$.



PART B

Answer both of the following two questions. Each question counts 15 marks.

1. Multidimensional signal and image processing

(a) Explain why a periodic sequence in 2-D is more complex than for 1-D.

(5 marks)

(b) Use the method of your choice to find the 2-D convolution between the following two signals:



(5 marks)

(c) Describe in detail how a charge-coupled device (CCD) creates a signal.

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(5 marks)

2. Speech processing

(a) Radix-3 Decimation in time FFT algorithm:

- i. Write an expression for X[k] involving three N/3-DFTs.
- ii. What is the advantage of a radix-3 FFT for a $N = 3^{\nu}$ length signal as compared to a direct computation of the DFT?

(5 marks)

(b) True or False:

- i. A radix-2 Decimation-in-time FFT algorithm will be faster than a radix-2 Decimation-in-frequency FFT algorithm.
- ii. A spectrogram contains 3-D information about a speech signal.
- iii. Estimating nasals parameters is one of the weakness of LPC model.
- iv. Zero padding a signal before computing its DFT will increase its frequency resolutions.
- v. Vowels have distinct formants.

(5 marks)

(c) The following are three different signals $x_i[n]$ that are the sum of two sinusiods:

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 $\begin{aligned} x_1[n] &= \cos(\pi n/4) + \cos(17\pi n/64) \\ x_2[n] &= \cos(\pi n/4) + 0.8\cos(21\pi n/64) \\ x_3[n] &= \cos(\pi n/4) + 0.0001\cos(21\pi n/64) \end{aligned}$

We wish to estimate the spectrum of each of these signals using the 64-point DFT with a 64-point rectangular window w[n]. Indicate for which of the signals you would expect to see two distinct spectral peaks.

(5 marks)