

Faculty of Information Technology

Computer Systems Engineering

Second Exam [Time Allowed: 1.5 hours]

ENCS431 (DSP)		Fall Semester 9/12/2012
Name:	ID:	section:

<u>Note</u>: this exam consists of four questions, every candidate has to answer question 1, and any other two questions only.

Question 1: [40 marks] (sampling and reconstruction)

a) In the context of discretize continuous-time signals, what do we mean be aliasing? Support your answer with an example. [6] (or distortion on Maximul)

Aliasing is an Overlap between Shifted copies of spectra of the discretized signal because sampling freques less than the max. freq. in the sampled signal.

b) Explain how can we avoid aliasing? [6]

Sampling freq. must be grober or equal to the double of of largest freq. in the sampled the signal S25 7 2 - 2m., when Sam is make frog. in sampled sig and Is is the sampling freq.

Consider the system below

 $x_c(t) \rightarrow C/D$ y[n] D/C $H(e^{jw})$

c) Let the continuous-time input signal $x_c(t) = \cos(150\pi t) + \cos(400\pi t) + \cos(600\pi t)$, what is the range of sampling period T1 that will avoid aliasing in the C/D converter? [8] 11

$$SLs = \frac{2\pi}{T} > 2 SL \cdot SLN \text{ is max. freq. in Xelt) which is}$$

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$$SLN = 600\pi$$

$$SLs = 1200\pi$$

d) Now, assume the sampling period T1=0.001 sec and the discrete system is an ideal low-pass filter with frequency response as follow:

$$H(e^{j\omega}) = \begin{cases} 1 & |\omega| \le 0.5\pi \\ 0 & \text{otherwise} \end{cases} \quad \mathcal{D}_{S} = \frac{2\pi}{T} = \frac{2\pi}{0.001} = 2000\pi.$$

$$\int in G \cdot \mathcal{D}_{S} = 2000\pi > 2\mathcal{D}_{N} (120\pi)$$
For $-\pi \le \omega \le \pi$, Find the discrete output $y(n)$ if the input signal $x_{o}(t)$ is the same as specified in part (c) above?

F XsCi [10]



50, output of L.P.F is (y(n) = cos(0.15710) + cos(0.14710).



e) If the sampling period of D/C converter T2 = (1/300) sec, Find the reconstructed continuous-time signal $y_r(t)$ if the reconstruction filter in the D/C converter, $H_r(j\Omega)$, has a gain of (1/T2) and cut-off frequency, Ω_c half of the sampling frequency. [10]



 $y_r(t) = \frac{T_2}{T_1} C_0 s(150 pt) + \frac{T_2}{T_1} C_0 s(200 pt)$

Question 2: [30 marks, 6 pts each] (Discrete Fourier Transform-DFT)

Suppose we have two four-point sequences x(n) and h(n) as follows:

 $x(n) = \{1, 0, 0, 1\}$, and $h(n) = \{1, 1, 0, 0\}$, for n=0,1,2,3 respectively.

a) Calculate the 4-point DFT X(k).
$$X(k) = \sum_{n=0}^{N-1} 2e(n) W_{N}^{2n}$$
 $J \leq k \leq N-1$
 $X(0) = \sum_{n=0}^{3} k(n) W_{Y}^{0} = \frac{3}{2} 2e(n) = 2$
 $X(1) = \sum_{n=0}^{3} 2e(n) W_{Y}^{0} = (1) W_{Y}^{0} + (1) W_{Y}^{0} = 1 + e = 1 + j$
 $X(2) = \sum_{n=0}^{3} 2e(n) W_{Y}^{2n} = 0 W_{Y}^{0} + 0 W_{Y}^{0} = 1 + e^{1} = 1 - j = 0$
 $X(3) = \sum_{n=0}^{3} 2e(n) W_{Y}^{3n} = 0 W_{Y}^{0} + 0 W_{Y}^{0} = 1 + e^{1} = 1 - j = 0$
 $S_{1}(k) = \sum_{n=0}^{3} 2e(n) W_{Y}^{3n} = 0 W_{Y}^{0} + 0 W_{Y}^{0} = 1 + e^{1} = 1 - j = 0$
 $S_{1}(k) = \sum_{n=0}^{3} 2e(n) W_{Y}^{3n} = 0 = 0$

b) Calculate the 4-point DFT H(k).

$$H(0) = \sum_{n=0}^{2} h(n) W_{n}^{0} = \sum_{n=0}^{3} h(n) = 2$$

$$H(1) = \sum_{n=0}^{2} h(n) W_{n}^{0} = WW_{n}^{0} = (1) W_{n}^{0} + W_{n}^{1} = HC^{-1} = 1 - j$$

$$H(2) = \sum_{n=0}^{2} h(n) W_{n}^{2n} = (1) W^{0} + (1) W_{n}^{2} = 1 + C^{-1} = 0$$

$$H(3) = \sum_{n=0}^{3} h(n) W_{n}^{3n} = U(W_{n}^{0} + U(W_{n}^{0} + U(W_{n}^{3} = 1 + C^{-1}))$$

c) Prove the circular convolution property of the DFT, i.e. circular convolution in time-domain corresponds to multiplication in frequency-domain, and vice versa.

$$Y(k) = \sum_{m=0}^{N-1} \chi_{1}(m) \left[\sum_{m=0}^{N-1} \chi_{2}(\ell(m-m)) \wedge W_{N}^{km} \right]$$

$$= \sum_{m=0}^{N-1} \chi_{2}(m) \left[\sum_{l=0}^{N-1} \chi_{2}(l) W_{N}^{k(l+m+Nr)} \right] = \left(\sum_{m=0}^{N-1} \chi_{2}(m) W_{N}^{km} \right) \chi_{2}(k)$$

$$= \chi_{1}(k) \chi_{2}(k)$$

d) Calculate y(n) = x(n) 4 h(n) by doing the circular convolution directly.

Y

Y(n) = 22,1,0,13, h=0,1,2,3.

e) Calculate y(n) of Part (c) by multiplying the DFTs of x(n) and h(n) and performing an inverse DFT.

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$$\begin{split} \gamma(k) &= \chi(k) H(k) \\ &= \int_{1+j}^{1+j} \int_{1-j}^{2} \cdot \star \begin{bmatrix} 2 \\ i+j \\ i-j \end{bmatrix} = \int_{2-j}^{1+j} \begin{bmatrix} 4 \\ 2 \\ 0 \\ 2 \end{bmatrix} \\ \psi_{4}^{2} &= e^{-\frac{1}{2}} = e^{-\frac{1}{2}} \\ \psi_{4}^{2} &= e^{-\frac{1}{2}} = e^{-\frac{1}{2}} \\ \psi_{4}^{2} &= e^{-\frac{1}{2}} \\ \psi$$

Question 3: [30 marks] (Fast Fourier Transform - FFT)

The N-point DFT, X(k), of discrete sequence x(n) is given by

$$X(k) = \sum_{n=0}^{N-1} x(n) W_N^{nk}, k = 0, 1, \dots N - 1, W_N^{nk} = e^{-j2\pi k n/N}$$

a) Derive an expression for N/2-point DFT X(k). [8]

$$X(k) = \sum_{N=0}^{\infty} 2c(n) W_{N}^{nk} = \sum_{N=0}^{\infty} 2c(n) W_{N}^{nk} + \sum_{N=0}^{\infty} 2c(n) W_{N}^{nk}$$

$$X(k) = \sum_{Y=0}^{N-1} 2c(2r) W_{N}^{2rk} + \sum_{Y=0}^{N-1} Gr(k) W_{N}^{nk}$$

$$W_{N}^{2} = W_{N}^{k} W_{N}^{rk} W_{N}^{2}$$

$$W_{N}^{2} = \frac{1}{2} \sum_{X=0}^{2r/N} 2 -\frac{1}{2} \sum_{X=0}^{2r/N} W_{N}^{rk}$$

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$$So_{1} X(k) = \sum_{Y=0}^{N-1} 2c(2r) W_{N}^{rk} + W_{N}^{k} \sum_{Y=0}^{N-1} 2c(2r+1) W_{N}^{rk}$$

$$\sum_{Y=0}^{r=0} \frac{1}{2} \sum_{X=0}^{N-1} 2c(2r) W_{N}^{rk} + W_{N}^{k} \sum_{Y=0}^{N-1} 2c(2r+1) W_{N}^{rk}$$

$$\sum_{Y=0}^{N-1} 2c(2r) W_{N}^{rk} + W_{N}^{rk} \sum_{Y=0}^{N-1} 2c(2r+1) W_{N}^{rk}$$

b) Explain the advantage of computing X(k) by this method compared to the N-point DFT method. [6]

- N-point DFT computation requires
$$N^2$$
 MADS.
- $\frac{N}{2}$ point DFT requires $2\left(\frac{N}{2}\right)^2 + N = \frac{N^2}{2} + N$ MADS
Which is less than N^2 .





d) If the input signal $x(n) = \{1, 2, 4, 8\}$, n=0,1,2,and 3. Compute the outputs of the FFT flow-graph shown in part (c) for k=0 and k=3. [8]

from the flow-graph above.

$$X(0) = x(0) + W_{4}^{0} \times (2) + (x(1) + W_{4}^{0} \times (3)) W_{4}^{0} = 1 + 2 + 4 + 8 = 15$$

$$X(3) = x(0) + W_{4}^{2} \times (2) + [x(1) + W_{4}^{2} \times (3)] W_{4}^{3} =$$

Question 4: [30 marks] (Discrete Fourier Transform-DFT)

Consider a discrete time system shown below, where x(n), h(n) and y(n) are input signal, impulse response and output signal of the system. The magnitude and phase of the complex DFT H(k) of h(n) are shown in fig (a) and fig (b) below, respectively.



b) If X(k) is the DFT of x(n), write down expressions of DFTs of the following signals: [10, 5 each]

(i) x[((n-2))N]

Using Circular shift properity of DFT, circular shift in the time domain is a multiplication of complex exponential in the freq. domain. i.e. $\Sigma[(n-m)_N] \in DFT \rightarrow W_N X(k)$

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$$DFT. \int = \{(n-2)\}_{N} = W_{N}^{-2} X(k).$$

= $\frac{-j^{2}T}{N}^{(2)} X(k).$

(ii) 2x(n) + x[((n+1))N]

Using linearity of DFT and Circular shift properity:
We get:

$$DFT \left[\sum_{k=1}^{\infty} \sum_{k=1}^{\infty} \frac{1}{2} \sum_{k=1}^{\infty} \sum_{k=1}^{\sum$$

c) Explain with an example how linear convolution can be used to compute the circular convolution, and how circular convolution can be used to compute the linear convolution of two discrete signals x1(n) and x2(n). [10]

Set x3(MJ = 2Ri(MJ (N) X2(M) = $\sum_{n=1}^{N-1} x_i(m) x_2 [\alpha_n - m] N [N] RN(n)$ $>_{3(n)} = \left\{ (n) \neq >_{2(n)} \right\} R_{N(n)}.$ So Circular convolution is linear conv. of xich and periodic seq. X2(n) with period N. (i.e. Length of Circ. Conv.). or it can be thought as linear convolution convolved with impulse train, and extract one ponod, so it is line conv. + orliasing. (xicu) * xzcu) * PN(n). sei(n) (N) sez(n)

To get linear conv. from circular conv., we need to do Circular conv. on 2N point. i.e. $\chi_1(n) \neq \chi_2(n) = \chi_1(n) (2N) \chi_2(n)$ $1 = \chi_1(n) (2N) \chi_2(n)$ $\chi_1(n) (2N) \chi_2(n) = \chi_1(n) (2N) \chi_2(n)$

-Good Luck-