

College of Engineering, Pune-5

(An autonomous institute of government of Maharashtra)

Comp

End Semester Examination

(Code: CT355) Computer Algorithm in Signal Processing

Programme: T.Y. B.Tech. (Computer Engineering)

Year: 2012-13

Duration : 3 Hrs

Date: 30/11/2012

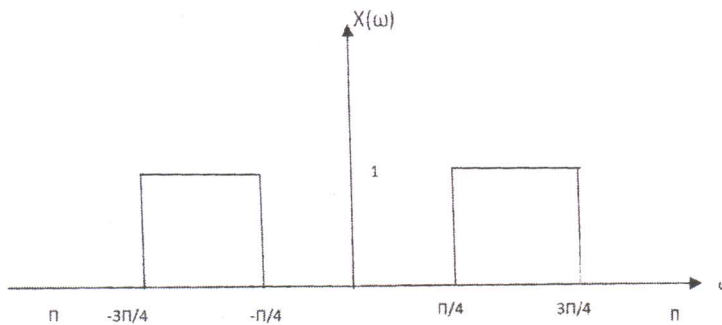
[Max. Marks: 50]

Instructions to candidates:

- 1) Question No. 5 is Compulsory. Out of remaining five, solve only four
- 2) Only first five questions will be evaluated.
- 3) When one question is attempted solve all the sub-questions serially.
- 4) Assume suitable data, if necessary.
- 5) Figures to the right indicate full marks.
- 6) Use of logarithmic tables, mollier charts, non-programmable electronic calculators and steam tables are allowed.

Q.1 A Find the inverse DTFT of $X(\omega)$ shown in figure below

4 mark



B Write a difference equation to implement the system with a frequency response 3 marks

$$H(\omega) = \frac{1 - 0.5e^{-j\omega} + e^{-3j\omega}}{1 + 0.5e^{-j\omega} + 0.75e^{-2j\omega}}$$

C Consider the eight point periodic sequence 3 marks

$$X(n) = (1, 3, 4, 5, 6, 5, 4, 3)$$

- 1) Is the sequence even symmetric
- 2) Find out $x(n-3) \bmod 8$
- 3) Find out $x(n-5) \bmod 8$
- 4) Find out time reversal of sequence $x(n)$

Q.2 A Given $x(n) = \{0, 1, 2, 3\}$. Find $X(k)$ using DIT FFT algorithm. 4 marks

Draw the flowgraph clearly and write down the values at intermediate stages

B Let $G[k]$ and $H[k]$ denote the 7 point DFTs of two sequences $g(n)$ and $h(n)$, 4 mark

$0 \leq n \leq 6$, respectively. If $G[K] = \{1+j2, -2+j3, -1-j2, 0, 8+j4, -3+j, 2+j5\}$.

$h(n) = g((n-3)_7)$.

Determine $H[K]$ without calculating IDFT of $G[k]$, i.e $g(n)$.

C The Z transform of a sequence $x(n)$ is given as 2 mark

$$X(z) = \frac{Z + 2Z^{-2} + Z^{-3}}{1 - 3Z^{-4} + Z^{-5}}$$

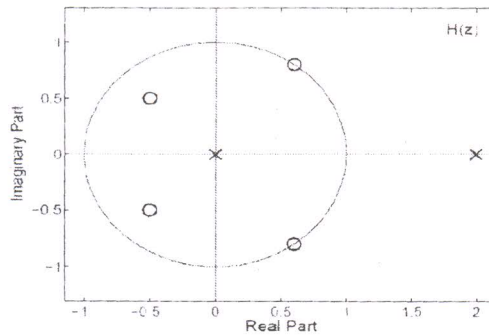
If the region of convergence include the unit circle, find the DTFT of $x(n)$ at $\omega = \pi$.

Q.3 A Consider the sequence 4 mark

$$X(n) = \delta(n) + 2\delta(n-2) + \delta(n-3)$$

- 1) Find the four point DFT of $x(n)$
- 2) Let $y(n)$ is the four point circular convolution of $x(n)$ with itself, find $Y(K)$ first and hence find out $y(n)$

B 03 marks



3-1: Pole-Zero Diagram

For the pole zero plot above answer the following question. Justify your answer with at most two statements.

- A) If ROC is $|Z| > 2$
- 1) Is the system stable?
 - 2) Is the system causal ?

- B) If ROC $|z| < 2$
- 1) Is the system stable?
 - 2) Is the system causal?

C A signal $X_a(t)$ is band limited to 10 KHz is sampled with a sampling frequency 3mark

of 20 KHz. The DFT of $N=1000$ samples of $x(n)$ is then Computed

- 1) What is the Spectral spacing between the spectral samples?
- 2) To What analog frequency does the index $K=150$ corresponds?
- 3) To What analog frequency does the index $K=800$ corresponds?

Q.4 A The linear time invariant system is characterized by the system function 3 marks

$$H(Z) = \frac{3 - 4z^{-1}}{1 - 3.5z^{-1} + 1.5z^{-2}}$$

Specify the ROC of $H(Z)$ and determine $h(n)$ for the following conditions.

- 1) The system is stable
- 2) The system is causal
- 3) The system is purely anticausal

B Find the linear convolution through circular convolution of $x(n)$ and $y(n)$. 2 mark

$$x(n) = \delta(n) + \delta(n-1) + \delta(n-2)$$

$$y(n) = 2\delta(n) - \delta(n-1) + 2\delta(n-2)$$

C Consider the LTI system which is stable and for which $H(Z)$, the Z transform of the impulse response is given by 5 marks

$$H(Z) = \frac{3 - 7z^{-1} + 5z^{-2}}{1 - 2.5z^{-1} + z^{-2}}$$

Suppose $X(n)$, the input to the system is a unit step sequence. Determine the output

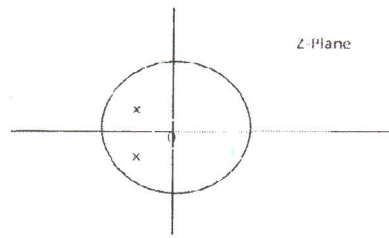
Q.5 A Select proper option 3 mark

- 1) Which type of filter gives linear phase
 - a) FIR filter
 - b) IIR filter
 - c) both FIR and IIR
 - d) depends on design
- 2) In filter design windowing technique, increasing the length of window function, -----the width of main lobe, which in turn----- transition width
 - a) Increases, increases
 - b) Increases, decreases
 - c) Decreases, increases
 - d) Decreases, Decreases

- 3) Increasing the length of filter, ----- number of ripples in the passband and stopband
 - a) Increases
 - b) Decreases
 - c) Does not change
 - d) Can't predict

- 4) The impulse response of ideal low pass filter is -----
 - a) causal and infinite
 - b) Noncausal and Finite
 - c) Noncausal and infinite
 - d) bidirectional and infinite

- 5) The following Pole zero diagram corresponds to



- a) High pass filter
 b) Bandpass filter
 c) Bandstop filter
 d) Low pass filter
- 6) Which type of filters are stable
 a) FIR
 b) IIR
 c) Both
 d) Can not predict

- B The desired response of a low-pass filter is

5 mark

$$H_d(\omega) = e^{-j2\omega}, \quad -\pi/4 \leq \omega \leq \pi/4$$

$$= 0, \quad \pi/4 < \omega \leq \pi$$

Design FIR filter to meet above response with hamming window function
 Of length seven.

Use following formula for Hamming window

$$W(n) = 0.54 - 0.46 \cos 2\pi n/M - 1 \quad 0 \leq n \leq M-1$$

$$= 0 \text{ otherwise}$$

- C What is Gibb's phenomenon? How it is alleviated 2 mark
- Q.6 A Compare FIR and IIR filter. In which applications one type will be preferred over the other? 3mark
- B What are the requirements of a Digital Signal Processor?
 How DSP processor differs from Microprocessor 3 marks
- C Determine the direct form I and II realization for a third transfer function 4 marks

$$H(Z) = \frac{0.28Z^2 + 0.319Z + 0.04}{0.5z^3 + 0.3Z^2 + 0.17Z - 0.2}$$

-----GOOD LUCK-----