

VASAVI COLLEGE OF ENGINEERING (Autonomous), HYDERABAD M.E. I Year (E.C.E.) I-Semester (Make Up) Examinations, March-2016 (Communication Engineering & Signal Processing)

Multirate Signal Processing

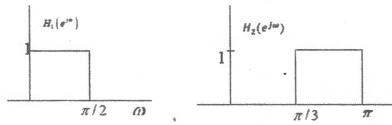
Time: 3 hours

Max. Marks: 70

Note: Answer ALL questions in Part-A and any FIVE questions from Part-B

Part-A (10 X 2=20 Marks)

- 1. Define linearity and time-invariance of a discrete time system.
- 2. The Fourier transforms of the impulse responses, $h_1(n)$ and $h_2(n)$ of two LTI systems are as shown in figure below



Find the Fourier transform of the impulse response of the overall system, when they are connected in parallel.

- 3. Determine the number of additions, multiplications and memory locations required for direct form II realization of an IIR digital filter transfer function having numerator polynomial of order M and denominator polynomial of order N.
- 4. Write the conditions on the impulse response h(n) of an FIR filter to have linear phase.
- 5. A speech signal x(t) is digitized at a sampling rate of 16 KHz. The speech signal was destroyed once the sequence x(n) was stored on a magnetic tape. Later it was required to obtain the speech signal sampled at the standard 8 KHz used in telephony. Draw the schematic diagram to do this using discrete time processing.
- 6. Determine the polyphase transfer functions $E_0(z)$, and $E_1(z)$ for two-branch polyphase realization of a length-6 FIR filter.
- 7. Check for perfect reconstruction of two-channel filter bank for the following analysis and synthesis filters: $H_0(z) = 2 z^{-1}$, $H_1(z) = 2 + 3z^{-1}$, $G_0(z) = -1 + 1.5z^{-1}$, $G_1(z) = 1 + 0.5z^{-1}$.
- 8. Obtain the perfect reconstruction condition for linear Phase FIR PR QMF Banks.
- 9. Obtain Haar wavelet and plot it for wavelet coefficients $h_1(0) = \frac{1}{\sqrt{2}}$ and $h_1(1) = -\frac{1}{\sqrt{2}}$

10. Write the dilation equations using scaling coefficients and wavelet coefficients.

Part-B (5 X 10=50 Marks)

a) An initially relaxed LTI system was tested with an input signal x(n) = 2u(n), and found to have a [4] response as shown in the following table

n	1	2	3	4	.5	 100	
y(n)	2	4	8	12	20	 20	

i) Obtain the impulse response of the system

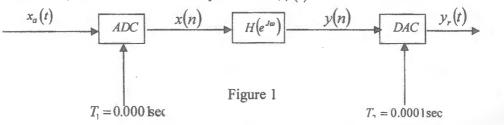
ii) Deduce the difference equation of the system

b) Consider the system shown in Figure 1, where $H(e^{j\omega})$ is an ideal LTI low pass filter with [6]

cutoff of $\pi/8$ rad/sec and the spectrum of $x_a(t)$ is shown in Figure 2.

i) What is the maximum value of T to avoid aliasing in the ADC?

ii) If 1/T=10 kHz, then what will be the spectrum of $y_r(t)$.



Contd...2

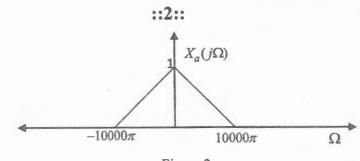
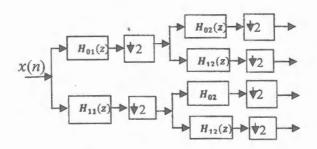


Figure 2

12. a) Explain the bilinear transformation method.

14.

- b) Design a Chebyshev IIR digital low pass filter for the following Specifications: Pass band cutoff frequency: 400Hz; Stop band cutoff frequency: 600Hz
 Pass band ripple: 1dB;Stop band ripple: 10dB
 Assuming sampling frequency of 2000Hz. Use Bilinear transformation.
- 13. a) Obtain relation between the Fourier transform of the input and output of the down sampler. [4]
 - b) Design a two stage interpolator to increase the sampling rate from 600Hz to 9 KHz.
 [6] The overall interpolator filter should satisfy the following specifications:
 passband edge : 200Hz; stopband edge : 300Hz, passband ripple:1dB; stopband ripple:60dB.



- a) Obtain non-tree equivalent for the above tree structured QMF bank and relations between the [4] analysis filters of the two structures.
- b) The analysis filters of a three channel QMF filter bank are

[7]

[5]

[5]

[3]

[7]

- H₀(z) = 1, H₁(z) = 2 + z⁻¹ + z⁻⁵, H₂(z) = 3 + z⁻¹ + 2z⁻²
 i) Can you determine the FIR synthesis filters G₀(z) and G₁(z) so that the two channel QMF bank is an alias-free and perfect reconstruction system. If so find them.
- ii) If not, find the set of stable IIR filters for an alias-free and perfect reconstruction system.
- 15. a) Obtain Daubechies wavelet filter coefficients for N=2 using coefficient domain solution.
 - b) Consider a two channel perfect reconstruction biorthogonal filter bank with the analysis [3] filters $h_0(n) = \{-1,2,6,2,-1\}/4\sqrt{2}$ and $h_1(n) = \{1,2,1\}/2\sqrt{2}$. Find the corresponding dual (synthesis) filters $\tilde{h}_0(n)$ and $\tilde{h}_1(n)$.
- 16. a) Show that the frequency spectrum $X(e^{j\omega})$ of a discrete time signal x(n) is periodic with respect [7] to ω with a period of 2π .

b) Design a optimal FIR highpass filter of mength 3 to meet the following specifications: [3]

Passband edge frequency= $f_p = 1000 Hz$; Stopband edge frequency= $f_s = 750 Hz$ Sampling frequency= 5000 Hz Tolerance ratio= $(\delta_p/\delta_s) = 2$

- 17. Write short notes on any two of the following:
 - a) Bring out the advantages of Multirate Signal Processing.
 - b) Obtain the conditions for alias free and perfect reconstruction of two channel QMF bank. [5]
 - c) Write short notes on wavelet reconstruction.
