



**UNIVERSITI TUN HUSSEIN ONN MALAYSIA**

**FINAL EXAMINATION  
SEMESTER II  
SESSION 2008/2009**

**SUBJECT NAME** : DIGITAL SIGNAL PROCESSING  
**SUBJECT CODE** : BEE 3213  
**COURSE** : 3 BEE / 3 BEI  
**EXAMINATION DATE** : APRIL 2009  
**DURATION** : 3 HOURS  
**INSTRUCTION** : ANSWER FIVE (5) OUT OF SEVEN (7) QUESTIONS.

**THIS QUESTION PAPER CONTENT ELEVEN (11) PAGES**

## SOALAN DALAM BAHASA INGGERIS

- Q1** (a) Given  $x[n] = 2r[n+2] - 7\delta[n+1] + 4\text{tri}\left(\frac{n+1}{2}\right) - 4\text{rect}\left(\frac{n-1}{4}\right) + (-2)^n(u[n] - u[n-4]) - 2r[n-3] - 10u[n-4]$
- (i) Separate  $x[n]$  as its odd and even parts and sketch both of them. (7 marks)
- (ii) Find  $y[n]$  for  $y[n] = x\left[2n - \frac{7}{4}\right]$  using step interpolation (3 marks)
- (b) Classify the following as left-sided, right-sided or two-sided signals. Represent those signals in a numeric sequence to support your answer.
- (i)  $z[n] = x[n+1]y[n]$  where  $x[n] = 0.5(\delta[n] + \delta[n+1] - 2\delta[n+3])$  and  $y[n] = 3 + u[n-3]$
- (ii)  $h[n] = (n+5)u[-n-1]$
- (iii)  $z[n] = \cos\left[185n\pi k - \frac{2\pi}{3}\right]$  where  $k = \begin{cases} 1, & 0 \leq n \leq 3 \\ 0, & \text{elsewhere} \end{cases}$  (6 marks)
- (c) Given a discrete signal  $x[n] = \{1, 6, a, 7, -4, b\}$ . If  $x_{\text{av}} = 1.83$  and  $P_{\text{av}} = 19.17$ , find the value of  $a$  and  $b$  if the value  $b > 0$ . (4 marks)
- Q2** Convolution is a method of finding the zero-state response of relaxed linear time-invariant (LTI) systems, while correlation is a measure of similarity between two signals.
- (a) Given  $x[n] = \{-3, 12, 6, 0, -9\}$  for  $-2 \leq n \leq 2$  and  $h[n] = \begin{cases} (3^n - 2) & -1 \leq n \leq 3 \\ 0 & \text{elsewhere} \end{cases}$
- (i) Determine the convolution of  $x[n]$  and  $h[n]$  using sliding strip method (6 marks)
- (ii) Assume that  $x[n]$  and  $h[n]$  is periodic signal. Find the convolution using cyclic method (6 marks)

- (b) A system have an input signal of  $x[n] = 6 \text{rect}\left(\frac{2-n}{2N}\right) + \{0, -8, -5, 2, 4\}$  and impulse response  $h[n] = (2n+1)(u[n] - u[n-2]) - 3\delta[n-3]$ .

(i) Find the correlation  $r_{xh}[n]$  for the above signal for  $N=3$

(6 marks)

(ii) Based on answer from Q2(b)(i), find  $r_{xx}[n]$

(2 marks)

**Q3**

- (a) As an electrical engineer, you are required to make an analysis of digital signal. If you are given an analog signal, with aid of diagram, explain the first step before you can proceed this task.

(4 marks)

- (b) Consider a sinusoidal signal  $g(t) = \frac{5}{2} \sin(800\pi t + 60^\circ)$  V is sampled at about 25% above the Nyquist Rate. Given a dynamic range of this signal is  $\pm 3$ V and it will be encoded with 4 bits per sample. Determine for  $-1 < n < 6$ :

(i) The discrete signal  $g[n]$

(ii) The quantized signal using both truncate and rounding techniques  $gQ1[n]$  and  $gQ2[n]$

(iii) The encoded signal for both quantization techniques  $gc1[n]$  and  $gc2[n]$

(iv) The quantization error for both quantization techniques  $\epsilon1[n]$  and  $\epsilon2[n]$

(v) Find the Signal-to-Noise Ratio (SNR) for both quantized signals

(vi) Based on Q3(b) (v), comment the performance of both quantized signals and which quantization technique give a minimum error

(vii) Suggest another way to improve the Signal-to-Noise Ratio (SNR)

(16 marks)

- Q4** (a) Let  $X_{DFT}[k] = \{2, 0.5 + 0.866j, 0.5 - 0.866j\}$ . Compute its Inverse Discrete Fourier Transform (IDFT) (2 marks)
- (b) Given the Discrete Fourier Transform (DFT) pair of a discrete signal,  
 $z[n] = \{1, -2, 3, 0, -1, 1, 2\} \longleftrightarrow$   
 $Z_{DFT}[k] = \{4, \boxed{A}, -3.23 + 5.55j, 3.72 + 2.32j, \boxed{B}, \boxed{C}, 1.01 - 0.74j\}$ .  
 By using the properties of DFT;
- (i) Find A, B and C  
 (ii) Determine the sequence  $w[n] = z[-n]$  and its DFT (3 marks)
- (c) The Discrete Fourier Transform (DFT) of a discrete signal,  $c[n]$  is given by  $C_{DFT}[k] = \{6, 3 - 1j, -8, 3 + 1j\}$ . Apply Decimation in Time (DIT) Fast Fourier Transform (FFT) algorithm to get its discrete signal,  $c[n]$ . (15 marks)
- Q5** (a) Differentiate the region of convergence (ROC) of the following signals:
- (i) Causal and anti-causal finite signal.  
 (ii) Causal and anti-causal infinite signal. (4 marks)
- (b) (i) Given  $u[n] \longleftrightarrow \frac{z}{z-1}$  with its ROC:  $|z| > 1$ . Modify the signal  $u[n]$  so that its ROC:  $|z| < 1$  and support your answer using defining relation. (4 marks)
- (ii) By using the properties of z-transform, determine the z-transform and the ROC of the signal  $x[n] = \left[ 2\left(\frac{2}{3}\right)^n + \frac{5}{6} \right] u[n]$  (4 marks)
- (c) Let  $X(z) = \frac{1 + 3z^{-1}}{1 - 3z^{-1} + z^{-2}}$ . Calculate the inverse z-transform of  $X(z)$  by using long division method for the following cases:
- (i) Right-sided signal  
 (ii) Left-sided signal  
 (Notes : Show the first 5 terms only) (8 marks)

- Q6** (a) List down four applications area of the DSP processor in modern life. (2 marks)
- (b) A lowpass filter  $\frac{3(z+1)^2}{31z^2 - 26z + 7}$  operates at  $S = 10$  kHz, and its cutoff frequency is  $f_c = 2$  kHz. Use this filter to design a lowpass filter with a cutoff frequency of 1 kHz. (10 marks)
- (c) Design a bandpass filter with a 3-dB bandwidth of 8 kHz and a center frequency of 9 kHz. The sampling frequency is 35 kHz. Plot the response of  $H(z)$  in the range  $0 \leq f \leq 20$  kHz with step size of 2 kHz. (8 marks)
- (Hint:  $H(s) = \frac{1}{(s+1)}$ , cutoff frequency is 1 rad/s)

**Q7** Design an FIR filter using Hamming window to meet the following specifications:

Passband: [1, 10] kHz

Stopband: [5, 6] kHz

$A_p = 2$  dB

$A_s = 40$  dB

The sampling frequency,  $S = 25$  kHz.

(20 marks)

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### Euler's Identity

$$e^{\pm j\theta} = \cos\theta \pm j\sin\theta$$

$$\cos\theta = \frac{1}{2}(e^{j\theta} + e^{-j\theta})$$

$$\sin\theta = \frac{1}{j2}(e^{j\theta} - e^{-j\theta})$$

### Finite Summation Formula

$$\sum_{k=0}^n \alpha^k = \frac{1 - \alpha^{n+1}}{1 - \alpha}, \quad \alpha \neq 1$$

### Infinite Summation Formula

$$\sum_{k=0}^{\infty} \alpha^k = \frac{1}{1 - \alpha}, \quad |\alpha| < 1$$

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**Table 1 : Properties of the Discrete Fourier Transform (DFT)**

Property	Signal	DFT	Remarks
Shift	$x[n - n_0]$	$X_{DFT}[k]e^{-j2\pi kn_0/N}$	No change in magnitude.
Shift	$x[n - 0.5N]$	$(-1)^k X_{DFT}[k]$	Half-period shift for even $N$ .
Modulation	$x[n]e^{j2\pi kn_0/N}$	$X_{DFT}[k - k_0]$	
Modulation	$(-1)^n x[n]$	$X_{DFT}[k - 0.5N]$	Half-period shift for even $N$ .
Folding	$x[-n]$	$X_{DFT}[-k]$	This is circular folding.
Product	$x[n]y[n]$	$\frac{1}{N} X_{DFT}[k] \otimes Y_{DFT}[k]$	The convolution is periodic.
Convolution	$x[n] \otimes y[n]$	$X_{DFT}[k] Y_{DFT}[k]$	The convolution is periodic.
Correlation	$x[n] \otimes \otimes y[n]$	$X_{DFT}[k] Y_{DFT}^*[k]$	The correlation is periodic.
Central Ordinates	$x[0] = \frac{1}{N} \sum_{k=0}^{N-1} X_{DFT}[k], \quad X_{DFT}[0] = \sum_{n=0}^{N-1} x[n]$		
Central Ordinates	$x\left[\frac{N}{2}\right] = \frac{1}{N} \sum_{k=0}^{N-1} (-1)^k X_{DFT}[k] \quad (N \text{ even}),$ $X_{DFT}\left[\frac{N}{2}\right] = \sum_{n=0}^{N-1} (-1)^n x[n] \quad (N \text{ even})$		
Parseval's Relation	$\sum_{n=0}^{N-1}  x[n] ^2 = \frac{1}{N} \sum_{k=0}^{N-1}  X_{DFT}[k] ^2$		

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**Table 2: Properties of z-transform.**

Property	Signal	z-transform
Linearity	$a_1x_1[n] + a_2x_2[n]$	$a_1X_1(z) + a_2X_2(z)$
Time reversal	$x[-n]$	$X(z^{-1})$
Time shifting	i) $x(n - k)$ ii) $x(n + k)$	i) $z^{-k}X(z)$ ii) $z^kX(z)$
Convolution	$x_1(n) * x_2(n)$	$X_1(z)X_2(z)$
Correlation	$r_{x_1x_2}(l) = \sum_{n=-\infty}^{\infty} x_1(n)x_2(n-l)$	$R_{x_1x_2}(z) = X_1(z)X_2(z^{-1})$
Scaling	$a^n x(n)$	$X(az^{-1})$
Differentiation	$nx[n]$	$z^{-1} \frac{dX(z)}{dz^{-1}}$ or $-z \frac{dX(z)}{dz}$
Time differentiation	$x[n] - x[n-1]$	$X(z)(1 - z^{-1})$
Time integration	$\sum_{k=0}^{\infty} X(k)$	$X(z) = \left( \frac{z}{z-1} \right)$
Initial value theorem	$\lim_{n \rightarrow 0} x(n)$	$\lim_{ z  \rightarrow \infty} X(z)$
Final value theorem	$\lim_{n \rightarrow \infty} x(n)$	$\lim_{ z  \rightarrow 1} \left( \frac{z-1}{z} \right) X(z)$



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**Table 3: Digital-to-Digital Transformations**

From	Band Edges	Mapping $z \rightarrow$	Parameters
Lowpass to lowpass	$\Omega_c$	$\frac{z - \alpha}{1 - \alpha z}$	$\alpha = \sin[0.5(\Omega_D - \Omega_C)] / \sin[0.5(\Omega_D + \Omega_C)]$
Lowpass to highpass	$\Omega_c$	$\frac{C(z+1)}{z-1}$	$\alpha = -\cos[0.5(\Omega_D + \Omega_C)] / \cos[0.5(\Omega_D - \Omega_C)]$
Lowpass to bandpass	$[\Omega_1, \Omega_2]$	$\frac{-(z^2 + A_1 z + A_2)}{A_1 z^2 + A_2 z + 1}$	$K = \tan(0.5\Omega_D) / \tan[0.5(\Omega_2 - \Omega_1)]$ $\alpha = -\cos[0.5(\Omega_2 + \Omega_1)] / \cos[0.5(\Omega_2 - \Omega_1)]$ $A_1 = 2\alpha K / (K + 1)$ $A_2 = (K - 1) / (K + 1)$
Lowpass to bandstop	$[\Omega_1, \Omega_2]$	$\frac{(z^2 + A_1 z + A_2)}{A_1 z^2 + A_2 z + 1}$	$K = \tan(0.5\Omega_D) / \tan[0.5(\Omega_2 - \Omega_1)]$ $\alpha = -\cos[0.5(\Omega_2 + \Omega_1)] / \cos[0.5(\Omega_2 - \Omega_1)]$ $A_1 = 2\alpha / (K + 1)$ $A_2 = -(K - 1) / (K + 1)$

**Table 4: Direct Analog-to-Digital Transformations for Bilinear Design**

From	Band Edges	Mapping $s \rightarrow$	Parameters
Lowpass to lowpass	$\Omega_c$	$\frac{z-1}{C(z+1)}$	$C = \tan(0.5\Omega_c)$
Lowpass to highpass	$\Omega_c$	$\frac{C(z+1)}{z-1}$	$C = \tan(0.5\Omega_c)$
Lowpass to bandpass	$\Omega_1 < \Omega_0 < \Omega_2$	$\frac{z^2 - 2\beta z + 1}{C(z^2 - 1)}$	$C = \tan[0.5(\Omega_2 - \Omega_1)]$ , $\beta = \cos \Omega_0$ or $\beta = \frac{\cos[0.5(\Omega_2 + \Omega_1)]}{\cos[0.5(\Omega_2 - \Omega_1)]}$
Lowpass to bandstop	$\Omega_1 < \Omega_0 < \Omega_2$	$\frac{C(z^2 - 1)}{z^2 - 2\beta z + 1}$	$C = \tan[0.5(\Omega_2 - \Omega_1)]$ , $\beta = \cos \Omega_0$ or $\beta = \frac{\cos[0.5(\Omega_2 + \Omega_1)]}{\cos[0.5(\Omega_2 - \Omega_1)]}$

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**Table 5 : Windows for FIR Filter Design**

Note: $I_0(x)$ is the modified Bessel Function of order zero.	
Window	Expression $w[n]$ , $-0.5(N-1) \leq n \leq 0.5(N-1)$
Boxcar	1
Cosine	$\cos\left(\frac{n\pi}{N-1}\right)$
Riemann	$\text{sinc}^2\left(\frac{2n}{N-1}\right)$ , $L > 0$
Bartlett	$1 - \frac{2 n }{N-1}$
von Hann (Hanning)	$0.5 + 0.5 \cos\left(\frac{2n\pi}{N-1}\right)$
Hamming	$0.54 + 0.46 \cos\left(\frac{2n\pi}{N-1}\right)$
Blackman	$0.42 + 0.5 \cos\left(\frac{2n\pi}{N-1}\right) + 0.08 \cos\left(\frac{4n\pi}{N-1}\right)$
Kaiser	$\frac{I_0\left(\pi\beta\sqrt{1-4\left[n/(N-1)\right]^2}\right)}{I_0(\pi\beta)}$

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**Table 6 : Characteristics of the windowed spectrum for various windows.**

<b>Window</b>	<b>Peak Ripple <math>\delta_p = \delta_s</math></b>	<b>Passband Attenuation AWP (dB)</b>	<b>Peak Sidelobe Attenuation AWS (dB)</b>	<b>Transition Width <math>FWS \approx C/N</math></b>
Boxcar	0.0897	1.5618	21.7	$C = 0.92$
Cosine	0.0207	0.3600	33.8	$C = 2.10$
Riemann	0.0120	0.2087	38.5	$C = 2.50$
von Hann (Hanning)	0.0063	0.1103	44.0	$C = 3.21$
Hamming	0.0022	0.0384	53.0	$C = 3.47$
Blackman	$1.71 \times 10^{-4}$	$2.97 \times 10^{-3}$	75.3	$C = 5.71$