

UNIVERSITY OF SWAZILAND
MAIN EXAMINATION, SECOND SEMESTER MAY 2012

FACULTY OF SCIENCE

DEPARTMENT OF ELECTRICAL AND ELECTRONIC ENGINEERING

**TITLE OF PAPER: INTRODUCTION TO DIGITAL SIGNAL
PROCESSING**

COURSE CODE: EE443

TIME ALLOWED: THREE HOURS

INSTRUCTIONS:

1. Answer any **FOUR (4)** of the following five questions.
2. Each question carries 25 marks.
3. Tables of selected window functions and selected Z-transform pairs are attached at the end.

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HAS BEEN GIVEN BY THE INVIGILATOR**

THIS PAPER CONTAINS EIGHT (8) PAGES INCLUDING THIS PAGE

QUESTION ONE (25 marks)

- (a) A signal $x(t) = 5 \sin \omega t$ is digitized using 10 bits.
- (i) Calculate the magnitude of the amplitude quantization level. (3 marks)
 - (ii) Calculate the signal-to-quantization-noise ratio (SQNR). (2 marks)
- (b) For CD quality music assume that $SQNR = 6.02B - 1.6$ dB. It is required that the SQNR be at least 96 dB. What should be the minimum number of bits used in the digitization? (3 marks)
- (c) A sinusoidal signal of frequency 2 kHz is passed through an 8th order Butterworth filter of cut-off frequency 5 kHz. It is then digitized using 10-bits. Find the minimum sampling frequency if the aliased signal amplitude at 2 kHz should not exceed the rms value of the quantization noise. (10 marks)
- (d) A digital communications link carries binary coded words representing samples of an input signal $x(t) = 3 \cos 600\pi t + 2 \cos 1800\pi t$ volts.
- (i) Find the Nyquist sampling rate employed. (2 marks)
 - (ii) Find an expression for the discrete-time signal if the signal is sampled at 5 kHz. (5 marks)

QUESTION TWO (25 marks)

- (a) A linear system with impulse response $h[n] = [1, -2, -3, 4]$ has an input $x[n] = [1, 1, 0, 1, 1]$. Use the convolution to find its output sequence. (5 marks)

- (b) A system is described by the difference equation

$$y(n) = \frac{5}{6}y(n-1) - \frac{1}{6}y(n-2) + x(n) - \frac{1}{2}x(n-1)$$

Find a closed form expression for impulse response of the system. Hint: use the z-transform. (10 marks)

- (c) A system is described by $y(n) = \frac{1}{4}y(n-1) + \frac{1}{8}y(n-2) + x(n) + 3x(n-1) + 2x(n-2)$.

- (i) Express its z-transform as a sum of partial fractions. (7marks)
- (ii) Systems in the form of a sum of partial fractions can be realized as parallel subsystems where each partial fraction is regarded as a subsystem. Draw the realization of this system in parallel form. (3 marks)

QUESTION THREE (25 marks)

- (a) An IIR filter is described by $y(n) = 0.7y(n-1) + x(n) - 2x(n-1)$. If the sampling frequency is 800 Hz, find
- (i) the magnitude, (9 marks)
 - (ii) the phase, (2 marks)
- of its frequency response at a frequency of 1200 rad/s.
- (b) Obtain expressions for the magnitude and phase response of the FIR filter whose impulse response is given by $h[n] = [0.25, 0.4, 0.8, 0.4, 0.25]$. (8 marks)
- (c) An FIR filter has a transfer function $H(z) = 1 + 0.765z^{-1} + z^{-2} = \frac{z^2 + 0.765z + 1}{z^2}$.
- (i) Sketch the pole-zero diagram of the filter. (2 marks)
 - (ii) Given that the sampling rate is 8 kHz, determine the input frequency that will be maximally attenuated. (4 marks)

QUESTION FOUR (25 marks)

- (a) A signal is to be digitally filtered using an IIR filter based on an analogue second-order

Butterworth filter with a normalized transfer function $H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$.

Using the Bilinear Transformation, obtain the transfer function of the equivalent low pass digital filter, given a sampling rate of 120 Hz and a cut-off frequency of 20 Hz. Your calculations should be accurate to 4 decimal places. (15 marks)

- (b) Obtain the DFT of the sequence $x(n) = [1, -1, 2, 2]$. (10 marks)

QUESTION FIVE (25 marks)

(a) Using the windowed sinc method with a Hamming window, determine the coefficients of a 7-tap linear-phase FIR low pass filter. The filter is to have a cut-off frequency of 3 kHz and sampling frequency of 16 kHz. (15 marks)

(b) A discrete-time system has the following pole-zero description:

Zeros: $z = +j1.5$ and $z = -j1.5$

Poles: $z = 0.8$, $z = 0.5 + j0.5$ and $z = 0.5 - j0.5$

(i) Is the system stable? Give a reason for your answer. (2 marks)

(ii) Derive the difference equation of the system. (8 marks)

TABLE OF Z-TRANSFORMS OF SOME COMMON SEQUENCES

Discrete-time sequence $x(n), n \geq 0$	Z-transform $H(z)$
$k\delta(n)$	k
k	$\frac{kz}{z-1}$
$ke^{-\alpha n}$	$\frac{kz}{z-e^{-\alpha}}$
$k\alpha^n$	$\frac{kz}{z-\alpha}$
kn	$\frac{kz}{(z-1)^2}$
kn^2	$\frac{kz(z+1)}{(z-1)^3}$
$k n \alpha^n$	$\frac{k\alpha z}{(z-\alpha)^2}$

QUANTIZATION

For a sine wave $SQNR = 6.02B + 1.76$ dB.

LOW PASS TO LOW PASS TRANSFORMATION

$$s = \frac{s}{\omega_p} \text{ where pre-warped frequency } \omega_p' = \tan\left(\frac{\pi f_c}{f_s}\right)$$

SUMMARY OF IMPORTANT FEATURES OF SELECTED WINDOW FUNCTIONS

Name of Window	Normalized Transition Width	Passband Ripple (dB)	Main lobe relative to Sidelobe (dB)	Max. Stopband attenuation (dB)	6 dB normalized bandwidth (bins)	Window Function $w(n), n \leq (N-1)/2$
Rectangular	$0.9/N$	0.7416	13	21	1.21	1
Hanning	$3.1/N$	0.0546	31	44	2.00	$0.5 + 0.5 \cos\left(\frac{2\pi n}{N}\right)$
Hamming	$3.3/N$	0.0194	41	53	1.81	$0.54 + 0.46 \cos\left(\frac{2\pi n}{N}\right)$
Blackman	$5.5/N$	0.0017	57	74	2.35	$0.42 + 0.5 \cos\left(\frac{2\pi n}{N-1}\right) + 0.08 \cos\left(\frac{4\pi n}{N-1}\right)$
Kaiser	$2.93/N$ ($\beta=4.54$)	0.0274		50		$\frac{I_0\left(\beta \left\{1 - \left[\frac{2n}{N-1}\right]^2\right\}^{\frac{1}{2}}\right)}{I_0(\beta)}$
	$4.32/N$ ($\beta=6.76$)	0.00275		70		
	$5.71/N$ ($\beta=8.96$)	0.000275		90		

$$\text{Bin width} = \frac{f_s}{N} \text{ Hz}$$