

FACULTY OF ENGINEERING, MATHEMATICS & SCIENCE SCHOOL OF ENGINEERING

Electronic & Electrical Engineering

Engineering
Senior Sophister
Annual Examinations

Hilary Term, 2016

Digital Signal Processing (4C5)

5th January 2016

Venue: Exam Hall

Time: 14.00 - 16.00

Dr. W. Dowling

Instructions to Candidates:

Answer THREE questions. All questions carry equal marks.

Materials permitted for this examination:

Mathematical Tables are available from the invigilators, if required.

Non-programmable calculators are permitted for this examination.

Please indicate the make and model of your calculator on each answer book used.

Q.1 (a) A continuous-time signal $x_a(t)$ has the Fourier transform $X_a(j\omega)$. The discrete-time signal x[n] is derived from $x_a(t)$ by periodic sampling:

$$x[n] = x_a(nT)$$
, where T is a constant.

Let $X(e^{j\Omega})$ denote the discrete-time Fourier transform of x[n]. Show that

$$X(e^{j\Omega}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X_a \left(j \left(\frac{\Omega}{T} - \frac{2\pi k}{T} \right) \right).$$

[11 marks]

(b) A system for sampling rate reduction by a factor of 1.5 is shown in Fig. Q1-1.

$$x[n] \longrightarrow \boxed{\uparrow 2} \longrightarrow r[n] \longrightarrow \boxed{h[n]} \longrightarrow w[n] \longrightarrow \sqrt{3} \longrightarrow y[n]$$
 Fig. Q1-1
$$r[n] = \begin{cases} x[n/2], & n = 0, \pm 2, \pm 4, \dots, \\ 0, & \text{otherwise} \end{cases}$$

$$w[n] = \sum_{k=-\infty}^{\infty} r[k]h[n-k]$$

$$y[n] = w[3n]$$

The ideal discrete-time low-pass filter has a unit sample response, h[n], and a frequency response, $H(e^{j\Omega})$, given by

$$H(e^{j\Omega}) = \begin{cases} 2, & |\Omega| < \frac{\pi}{3} \\ 0, & \frac{\pi}{3} < |\Omega| \le \pi \end{cases}$$

Let $R(e^{j\Omega})$ and $Y(e^{j\theta})$ denote the discrete-time Fourier transforms of the sequences r[n] and y[n] respectively. A continuous-time signal $x_a(t)$ has the Fourier transform $X_a(j\omega)$ shown in Fig. Q1-2.

continued ...

[Q.1 ctd.]

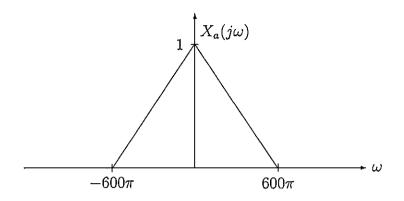


Fig. Q1-2

If $x[n] = x_a(nT)$, and the sampling period T = 1 millisecond,

(i) sketch $R(e^{j\Omega})$ for $-\pi \leq \Omega \leq \pi$, and

[5 marks]

(ii) sketch $Y(e^{j\theta})$ for $-\pi \le \theta \le \pi$.

[4 marks]

Q.2 (a) The sequence x[n] is zero for n < 0 and for n > N-1. Assume that $N = 2^M$, where M is a positive integer. Let g[n] = x[2n], and h[n] = x[2n+1]. Show that the N-point discrete Fourier transform (DFT) of the sequence x[n] can be obtained by appropriately combining the N/2-point DFTs of the sequences g[n] and h[n].

[8 marks]

(b) Draw the complete signal flow graph for an 8-point decimation-in-time fast Fourier transform (FFT) algorithm. [12 marks]

Q.3 (a) Let $H_c(s)$ denote the transfer function of a continuous-time filter. The transfer function of a discrete-time filter, H(z), is obtained by applying the bilinear transformation to $H_c(s)$:

$$H(z) = H_c(s) \bigg|_{s = (1 - z^{-1})/(1 + z^{-1})}$$

Show that the frequency responses of the discrete-time and continuous-time filters are related by

$$H\left(e^{j\Omega}\right) = H_c(j\omega)\bigg|_{\omega = \tan(\Omega/2)}$$

[8 marks]

(b) A discrete-time low-pass filter with frequency response, $H\left(e^{j\Omega}\right)$, is to be designed to meet the following specifications:

$$\begin{split} 0.89 & \leq \left| H\left(e^{j\Omega}\right) \right| \leq 1, \qquad |\Omega| \leq 0.2\pi \\ \left| H\left(e^{j\Omega}\right) \right| \leq 0.18, \qquad \quad 0.6\pi \leq |\Omega| \leq \pi \end{split}$$

The filter is to be designed by applying the bilinear transformation to the transfer function of an appropriate Butterworth continuous-time filter.

Verify that a second order filter is sufficient to meet the specifications.

Determine the transfer function, H(z), of the discrete-time filter.

Note that the transfer function of a second order Butterworth low-pass prototype filter is

$$H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

[12 marks]

Q.4 (a) Using a rectangular window sequence, design a causal, 15-point, discrete-time generalised linear phase filter with a magnitude response which approximates the ideal band-pass response, $|H_{id}(e^{j\Omega})|$, given by

$$\left| H_{id} \left(e^{j\Omega} \right) \right| = \begin{cases} 0, & |\Omega| < \frac{\pi}{3} \\ 1, & \frac{\pi}{3} < |\Omega| < \frac{2\pi}{3} \\ 0, & \frac{2\pi}{3} < |\Omega| \le \pi \end{cases}$$

[12 marks]

(b) Let $X\left(e^{j\Omega}\right)$ denote the discrete-time Fourier transform of the real sequence x[n]. If r[n]=x[-n], show that $R\left(e^{j\Omega}\right)$, the discrete-time Fourier transform of r[n], is given by

$$R\left(e^{j\Omega}\right) = X^*(e^{j\Omega})$$

where * denotes complex conjugation.

[2 marks]

(c) Let h[n] be the unit-sample response of a causal filter with an arbitrary phase characteristic. Assume that h[n] is real and denote its Fourier transform by $H\left(e^{j\Omega}\right)$. Let x[n] be a real finite duration sequence. The sequence x[n] is first filtered to get g[n]:

$$g[n] = \sum_{k=-\infty}^{\infty} h[k]x[n-k]$$

The sequence r[n] = g[-n] is then filtered to get w[n]:

$$w[n] = \sum_{k=-\infty}^{\infty} h[k]r[n-k]$$

The sequence y[n]=w[-n]. Let $X\left(e^{j\Omega}\right)$ and $Y\left(e^{j\Omega}\right)$ denote the discrete-time Fourier transform of x[n] and y[n] respectively. Show that

$$Y(e^{j\Omega}) = X(e^{j\Omega}) |H(e^{j\Omega})|^2$$
 [6 marks]

Q.5 (a) Let x[n] denote a finite-duration sequence of length M such that x[n]=0 for n<0 and $n\geq M$. Let X(z) denote the z-transform of x[n]. If we sample X(z) at $z=e^{j(2\pi/N)k},\quad k=0,1,2,\ldots,N-1$, we obtain

$$X_1[k] = X(z)|_{z=\rho j(2\pi/N)k}$$
, $k = 0, 1, 2, ..., N-1$.

The number of samples N is *less than* the duration of the sequence M; i.e. N < M. The sequence $x_1[n]$ is obtained as the inverse DFT of $X_1[k]$. Determine the relation between $x_1[n]$ and x[n]. [12 marks]

(b) Consider a finite-duration sequence x[n] of length M such that x[n] = 0 for n < 0 and $n \ge M$. We want to compute samples of the discrete-time Fourier transform of x[n] at the N equally spaced frequencies

$$\Omega_k = \frac{2\pi k}{N}, \quad k = 0, 1, \dots, N - 1.$$

Determine and justify procedures for computing the N samples of the discrete-time Fourier transform using only one N-point DFT for the following cases:

(i) N > M; and (ii) N < M.

[8 marks]