

IMPERIAL COLLEGE LONDON

DEPARTMENT OF ELECTRICAL AND ELECTRONIC ENGINEERING  
EXAMINATIONS 2017

MSc and EEE/EIE PART IV: MEng and ACGI

**Corrected copy**

**WAVELETS AND APPLICATIONS**

Monday, 8 May 10:00 am

Time allowed: 3:00 hours

**There are FOUR questions on this paper.**

**Answer ALL questions.**

*All questions carry equal marks.*

**Any special instructions for invigilators and information for candidates are on page 1.**

Examiners responsible

First Marker(s) : P.L. Dragotti

Second Marker(s) : A. Manikas



**Special Information for the Invigilators: NONE**

**Information for Candidates:**

*Sub-sampling by an integer  $N$*

$$x_{\downarrow N}[n] \longleftrightarrow \frac{1}{N} \sum_{k=0}^{N-1} X(e^{j(\omega-2\pi k)/N}) = \frac{1}{N} \sum_{k=0}^{N-1} X(W_N^k z^{1/N}),$$

where

$$W_N^k = e^{-j2\pi k/N}.$$

*Dual Basis:*

Given a basis  $\{\varphi_i(t)\}_{i \in \mathbb{Z}}$ , the dual basis is given by the set of elements  $\{\tilde{\varphi}_i(t)\}_{i \in \mathbb{Z}}$  satisfying:

$$\langle \varphi_i(t), \tilde{\varphi}_j(t) \rangle = \delta_{i,j}.$$

*Haar scaling and wavelet functions*

The Haar scaling function is defined as:

$$\varphi(t) = \begin{cases} 1 & 0 \leq t < 1 \\ 0 & \text{otherwise} \end{cases}$$

The Haar wavelet is defined as:

$$\psi(t) = \begin{cases} 1 & 0 \leq t < 1/2 \\ -1 & 1/2 \leq t < 1 \\ 0 & \text{otherwise} \end{cases}$$

## The Questions

1. Consider the system shown in Fig. 1a. This is a two-channel filter bank where the analysis part contains also two modulators. The modulator in the upper branch modulates the input  $x[n]$  with the sequence  $p[n] = 1, \forall n$  and then filters  $p[n]x[n]$  with  $H_0(z)$ . In the lower branch, the input  $x[n]$  is modulated with the sequence  $q[n] = (-1)^n$  and the resulting sequence is then filtered with  $H_1(z)$ .

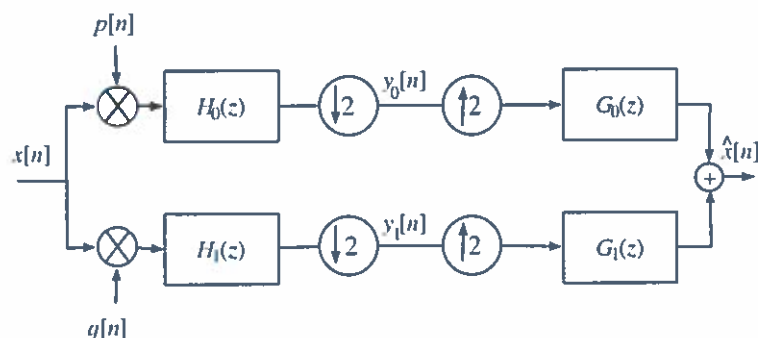


Figure 1a: Two-Channel Filter Bank with Modulation.

- (a) Express  $\hat{X}(z)$  as a function of  $X(z)$  and the filters. Then derive the two perfect reconstruction (PR) conditions the filters have to satisfy. [7]
- (b) Assume that  $G_0(z) = (1 + z^{-1})/\sqrt{2}$  and that  $H_0(z) = (1 + z)/\sqrt{2}$ , find the shortest filters  $H_1(z)$  and  $G_1(z)$  that would lead to a perfect reconstruction filter bank. [6]

Question continues on next page.

- (c) Assume now that  $G_0(z)$  and  $G_1(z)$  are half-band ideal low-pass and high-pass filters respectively as shown in Fig. 1b. Also assume that  $H_0(z) = 2G_0(z^{-1})$ . Sketch and dimension the Fourier transform of the filter  $H_1(z)$  that would lead to a perfect reconstruction filter bank.

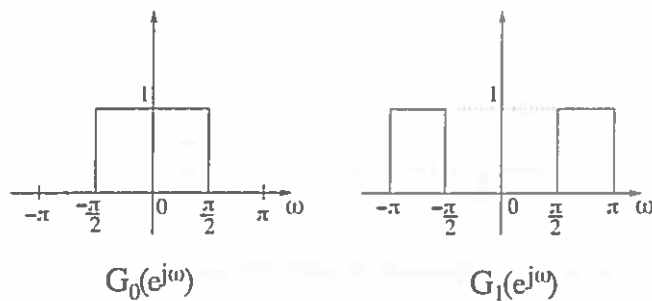


Figure 1b: Half-band synthesis filters.

[6]

- (d) Given the filters of part (c), sketch and dimension the Fourier transform of  $y_0[n]$  and  $y_1[n]$ , assuming that  $x[n]$  has the spectrum shown in Fig. 1c

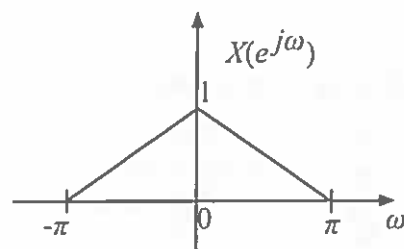


Figure 1c: Discrete-time Fourier transform of  $x[n]$ .

[6]

2. Consider the two systems shown in Fig. 2a and Fig. 2b.

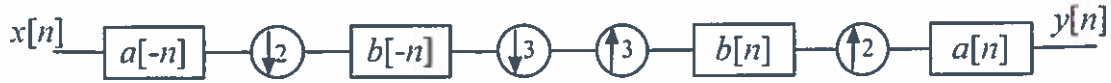


Figure 2a: First multirate system.



Figure 2b: Second multirate system.

- (a) Find the filters  $c[n]$  and  $d[n]$  and integers  $M$  and  $N$  so that the system in Fig. 2b has input-output relationship equivalent to the system in Fig. 2a. (Either a time domain or a  $z$  domain answer is acceptable.)

[5]

- (b) Given that the filter  $b[n]$  satisfies  $\langle b[n], b[n - 3k] \rangle = \delta_k$ , find (and prove) a sufficient condition on  $a[n]$  such that the mapping from  $x$  to  $y$  is idempotent. Idempotent means that if the mapping was applied again to the sequence  $y[n]$ , the output would stay  $y[n]$ .

[10]

- (c) Consider a filter bank specified by the following signal equations:

$$\begin{aligned} y_0 &= D_2 G D_2 G x \\ y_1 &= D_2 G D_2 H D_2 G x \\ y_2 &= D_2 H D_2 H D_2 G x \\ y_3 &= D_2 H x, \end{aligned}$$

where  $G$  and  $H$  are the infinite matrix representations for filtering with a lowpass filter  $g[n]$  and a highpass filter  $h[n]$ , respectively, and  $D_2$  is the matrix representation of downsampling by 2.

- i. Draw a block diagram of the system using two-channel filter banks.

[5]

- ii. Draw the equivalent single-level four-channel filter bank clearly specifying the downsampling factors and transfer functions of the filters in each branch.

[5]

3. Consider the interval  $t \in [0, 1]$  and the function

$$x(t) = \begin{cases} e^t, & \text{for } t \in [0, 1] \\ 0 & \text{otherwise.} \end{cases}$$

The aim is to find a polynomial of degree one that best approximates  $x(t)$  over the interval  $[0, 1]$ . Specifically, let

$$\varphi_1(t) = \begin{cases} 1, & \text{for } t \in [0, 1] \\ 0, & \text{otherwise} \end{cases}$$

and

$$\varphi_2(t) = \begin{cases} t, & \text{for } t \in [0, 1] \\ 0, & \text{otherwise,} \end{cases}$$

the degree-one polynomial is given by  $p(t) = \sum_{k=1}^2 a_k \varphi_k(t)$  and  $a_1, a_2$  can be found by minimizing

$$\|\epsilon(t)\|^2 = \int_0^1 \epsilon^2(t) dt$$

with  $\epsilon(t) = x(t) - p(t)$ .

- (a) Using the projection theorem, show that the first-order Taylor series of  $x(t)$  at  $t = 0$  given by:

$$p_s(t) = x(0) + x'(0)t$$

does not minimise  $\|\epsilon(t)\|^2$ .

[5]

- (b) Denote with  $V = \text{span}\{\varphi_1(t), \varphi_2(t)\}$  the sub-space generated by  $\varphi_i(t)$  with  $i = 1, 2$  over the interval  $t \in [0, 1]$ , the minimization of the error is achieved by computing the orthogonal projection of  $x(t)$  onto  $V$ . Recall that this is given by:

$$p_v(t) = \sum_{i=1}^2 \langle x(t), \tilde{\varphi}_i(t) \rangle \varphi_i(t)$$

where  $\{\tilde{\varphi}_i(t)\}_{i=1}^2$  are the two dual-basis functions.

- i. Since  $\tilde{\varphi}_i(t) \in V$  we can write  $\tilde{\varphi}_i(t) = \sum_{k=1}^2 \alpha_{i,k} \varphi_k(t)$ . Using this fact, determine the two dual-basis functions  $\tilde{\varphi}_i(t)$ ,  $i = 1, 2$ . That is, find the coefficients  $\alpha_{i,k}$ ,  $i = 1, 2$ ;  $k = 1, 2$ .

[5]

Question continues on next page.

ii. Given the dual basis

A. Compute the inner products  $\langle x(t), \tilde{\varphi}_i(t) \rangle$ ,  $i = 1, 2$ . [5]

B. Write the exact expression for  $p_v(t) = \sum_{i=1}^2 \langle x(t), \tilde{\varphi}_i(t) \rangle \varphi_i(t)$ . [5]

C. Verify that the error  $\epsilon_v(t) = x(t) - p_v(t)$  is orthogonal to  $V$ . [5]



4. Let  $\varphi(t)$  be the Haar scaling function and  $\psi(t)$  be the Haar wavelet function. Let  $V_j$  be the subspace generated by  $\varphi_{j,n}(t) = \sqrt{2^{-j}}\varphi(2^{-j}t - n)$ ,  $n \in \mathbb{Z}$  and let  $W_j$  be the subspace generated by  $\psi_{j,n}(t) = \sqrt{2^{-j}}\psi(2^{-j}t - n)$ ,  $n \in \mathbb{Z}$ . Consider the function defined on  $0 \leq t < 1$  given by

$$f(t) = \begin{cases} 0 & 0 \leq t < 1/4 \\ 1 & 1/4 \leq t < 3/8 \\ 0 & 3/8 \leq t < 1. \end{cases}$$

- (a) Express  $f(t)$  in terms of the basis elements of  $V_k$ . In other words, find the coefficients  $c_{k,n}$ ,  $n \in \mathbb{Z}$  that lead to the decomposition

$$f(t) = \sum_{n \in \mathbb{Z}} c_{k,n} \varphi_{k,n}(t). \quad (1)$$

Pick a negative  $k$  so that only one coefficient in Eq. (1) is non-zero.

[6]

- (b) Given the  $k$  chosen in part (a), decompose  $f(t)$  into its component parts that belong to the subspace  $W_{k+1}, W_{k+2}, \dots, W_{-1}, W_0$ , and  $V_0$ . In other words, find the coefficients  $d_{k+1,n}, d_{k+2,n}, \dots, d_{0,n}$  and  $c_{0,n}$ ,  $n \in \mathbb{Z}$  that lead to the following decomposition

$$f(t) = \sum_{n \in \mathbb{Z}} c_{0,n} \varphi_{0,n}(t) + \sum_{j=k+1}^0 \sum_{n \in \mathbb{Z}} d_{j,n} \psi_{j,n}(t). \quad (6)$$

- (c) Sketch and dimension each of the decompositions of part (b). That is, sketch

$$f_{V_0}(t) = \sum_{n \in \mathbb{Z}} c_{0,n} \varphi_{0,n}(t)$$

and the components

$$f_{W_j}(t) = \sum_{n \in \mathbb{Z}} d_{j,n} \psi_{j,n}(t) \quad \text{for } j = k, k+1, \dots, 0. \quad (7)$$

- (d) Verify the Parseval equality. That is, verify that:

$$\|f(t)\|^2 = \sum_n |c_{0,n}|^2 + \sum_{j=k+1}^0 \sum_n |d_{j,n}|^2. \quad (6)$$

