

# **GLASGOW COLLEGE UESTC**

**Exam paper**

## **Advanced Digital Communication (UESTC4028)**

**Date: 2<sup>nd</sup> Jan. 2020**

**Time: 14:30-16:30pm**

**Attempt all PARTS. Total 100 marks**

**Use one answer sheet for each of the questions in this exam.  
Show all work on the answer sheet.**

**Make sure that your University of Glasgow and UESTC Student Identification  
Numbers are on all answer sheets.**

**An electronic calculator may be used provided that it does not allow text storage  
or display, or graphical display.**

**All graphs should be clearly labelled and sufficiently large so that all elements  
are easy to read.**

**The numbers in square brackets in the right-hand margin indicate the marks  
allotted to the part of the question against which the mark is shown. These  
marks are for guidance only.**

- Q1
- (a) Draw a constellation diagram of 16QAM modulation, labelling the constellation points in binary. [5]
  - (b) Calculate how many times faster 16QAM modulation can transmit than BPSK modulation in terms bits per symbol. [5]
  - (c) Consider a BPSK system with a bit rate of 1 Mbit/s. The received waveform  $s_1(t) = A\cos(w_o t)$  and  $s_2(t) = -A\cos(w_o t)$ , are coherently detected with a matched filter. The value of  $A$  is 10 mA, and you may assume that the single-side noise power spectrum density is  $N_o = 10^{-11}$  W/Hz and the signal power and energy per bit are normalized relative to a 1  $\Omega$  load.
    - (i) Calculate the noise power. [5]
    - (ii) Find the energy per bit  $E_b$ . [5]
    - (iii) Find the bit error probability  $P_b$  using the following reference equation  $Q(x) = \frac{1}{x\sqrt{2\pi}} \exp(-\frac{x^2}{2})$ . [5]
- Q2
- (a) Explain the advantages and disadvantages of zero-forcing (ZF) equalization, comparing it with the minimum mean square error (MMSE) equalization. [5]
  - (b) Design a 3-tap ZF equalizer for input  $x(n) = \{0, 0.1, -0.15, -0.87, -0.12, 0.2, 0\}$  in which  $x(0) = -0.87$ . [10]
  - (c) Describe the impulse response of a matched filter for detecting the discrete signal shown in Figure Q2. [5]

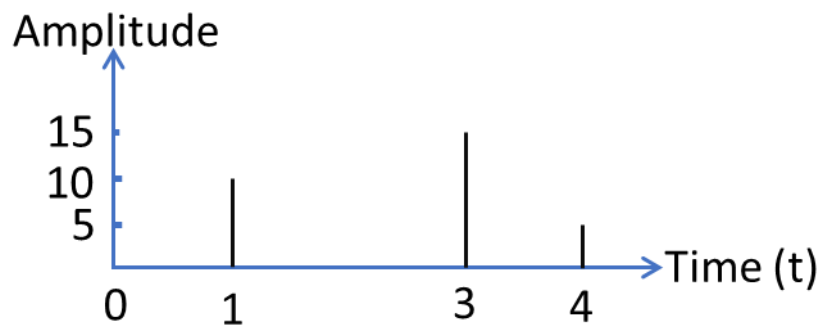


Figure Q2

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- (d) With the signal of part (c) in Figure Q2 at the input to the filter, show the output as a function of time. [5]

Q3 (a) Consider wireless communication systems,

- (i) Explain why channel coding is important. [3]  
(ii) Explain the cost of using channel coding. [3]

(b) Considering a (7,4) coding whose parity check matrix  $\mathbf{H}$  is

$$\mathbf{H} = \begin{bmatrix} 1 & 0 & 0 & 1 & 0 & 1 & 1 \\ 0 & 1 & 0 & 1 & 1 & 1 & 0 \\ 0 & 0 & 1 & 0 & 1 & 1 & 1 \end{bmatrix}$$

Assuming a received codeword  $\mathbf{r} = [1 \ 0 \ 0 \ 1 \ 0 \ 0 \ 1]$ , calculate the syndrome vector  $\mathbf{s}$ . [7]

(c) Given the look up table of Table Q3, decode the transmitted codeword. [12]

Table Q3

Syndrome	Error Pattern
100	1000000
010	0100000
001	0010000
110	0001000
011	0000100
111	0000010
101	0000001

Q4 Considering a communication system with bandwidth of 45 kHz. The (time domain) channel impulse response  $\mathbf{h}[n] = [1, 0.5]$ . The transmitted signal vector in an OFDM symbol is  $\mathbf{x}[n] = [1, -1, -1]$ . Assume we design an inter-symbol-interference (ISI) free cyclic prefix based orthogonal frequency division multiplexing (CP-OFDM) system with subcarrier spacing 15 kHz to

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transmit the signal  $\mathbf{x}[n]$  over the multi-path channel  $\mathbf{h}[n]$ , by using the OFDM system sketched in Figure Q4.

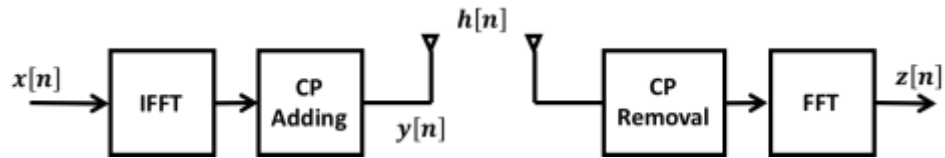


Figure Q4

- (a) Explain the role of CP in an OFDM system. [5]
- (b) Design the minimum length of CP in samples and second, respectively. [5]
- (c) Write down the transmitted signal  $\mathbf{y}[n]$  with a CP. [5]
- (d) Calculate the channel frequency response  $\mathbf{H}$  corresponding to the modulated signals at the receiver side after FFT. [5]
- (e) Assume the system is noise free (i.e., no noise in the system), calculate the received signal  $\mathbf{z}[n]$  at the receive side. [5]

End of question paper