

EGT3  
ENGINEERING TRIPOS PART IIB

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Monday 24 April 2017 2 to 3.30

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**Module 4F8**

**IMAGE PROCESSING AND IMAGE CODING**

*Answer not more than **three** questions.*

*All questions carry the same number of marks.*

*The **approximate** percentage of marks allocated to each part of a question is indicated in the right margin.*

*Write your candidate number **not** your name on the cover sheet.*

**STATIONERY REQUIREMENTS**

Single-sided script paper

**SPECIAL REQUIREMENTS TO BE SUPPLIED FOR THIS EXAM**

CUED approved calculator allowed

Engineering Data Book

**10 minutes reading time is allowed for this paper.**

**You may not start to read the questions printed on the subsequent pages of this question paper until instructed to do so.**

1 (a) The Fourier transform of a 2D image gives a 2D complex function which can be used to understand the frequency structure of the image.

(i) For the image defined by  $g(u_1, u_2) = \sin(\Omega_1 u_1) \sin(\Omega_2 u_2)$ , with  $\Omega_1 = \pi/8$   $\text{rads}^{-1}$  and  $\Omega_2 = \pi/16$   $\text{rads}^{-1}$ , sketch the Fourier transform,  $G(\omega_1, \omega_2)$ , of the image. [10%]

(ii) Now suppose our image takes the form  $g(u_1, u_2) = \sin[\phi_1(u_1)] \sin[\phi_2(u_2)]$ , where  $\phi_1(u_1) = \Omega_1 u_1 + \frac{\pi}{16} u_1^2$  and  $\phi_2(u_2) = \Omega_2 u_2 + \frac{\pi}{32} u_2^2$ , with  $\Omega_1$  and  $\Omega_2$  as given in part (a)(i). Sketch the spectrum of this image and note how it varies from that in part (a)(i). [15%]

(iii) Discuss the relative importance of amplitude and phase in the Fourier transforms of 2D images. [10%]

(b) A continuous image  $g(u_1, u_2)$  is sampled on a rectangular sampling grid with spacings  $\Delta_1$  and  $\Delta_2$  in the  $u_1$  and  $u_2$  directions.

(i) If this sampled image is  $g_s(u_1, u_2)$ , write down the 2D spectrum (Fourier transform),  $G_s(\omega_1, \omega_2)$ , of  $g_s$  in terms of the Fourier transform,  $G(\omega_1, \omega_2)$ , of the original image. [10%]

(ii) Using the result in part (b)(i), explain the phenomenon of *aliasing* in images. [10%]

(iii) A continuous image is described by the function

$$g(u_1, u_2) = \cos(\alpha u_1 + \beta u_2)$$

Form the Fourier transform,  $G(\omega_1, \omega_2)$ , of  $g(u_1, u_2)$  and verify that  $g$  is bandlimited. [20%]

(iv) Find the Nyquist sampling frequencies for the image in part (b)(iii). [10%]

(v) What are the values of  $\alpha$  and  $\beta$  in part (b)(iii) if sampling spacings of  $\Delta_1 = 0.4\pi$  and  $\Delta_2 = 0.2\pi$  are the Nyquist sampling intervals? [15%]

2 (a) Filtering 2D images involves convolving the image with an *impulse response* (IR) or *point spread function* (PSF). In the majority of cases we assume that the filter is spatially invariant.

(i) Explain what is meant by a *zero-phase* filter and say why this is a desirable property for processing images. [10%]

(ii) We wish to design a filter with a desired zero-phase 2D frequency response (for example, a rectangular low-pass filter). Explain why *windowing* is necessary and describe the effect of windowing on the desired frequency response. [10%]

(iii) Describe two methods of forming 2D window functions from 1D window functions. [10%]

(b) An observed image  $\mathbf{y}$  can often be modelled as the original image  $\mathbf{x}$  convolved with a spatially invariant filter,  $\mathbf{h}$ , plus noise,  $\mathbf{n}$ :

$$\mathbf{y} = \mathbf{h} * \mathbf{x} + \mathbf{n}$$

[\* denotes convolution]

(i) Explain how, neglecting noise, one can apply *inverse filtering* to extract the original image from  $\mathbf{y}$ , given knowledge of  $\mathbf{h}$ . Comment on the performance of such inverse filtering methods in the presence of noise. [10%]

(ii) If a camera images a moving object (e.g. a chessboard pattern), by opening its shutter for  $T$  seconds, explain why the resulting image  $y(u_1, u_2)$  may be represented by

$$y(u_1, u_2) = \int_0^T x(u_1 - vt, u_2) dt$$

where we assume that  $x(u_1, u_2)$  models the object and the speed of the object is  $v$  in the  $u_1$  direction and zero in the  $u_2$  direction. [15%]

(iii) Show that the motion blur in part (b)(ii) can be modelled by  $\mathbf{y} = \mathbf{h} * \mathbf{x} + \mathbf{n}$ , and find the filter  $\mathbf{h}$ . [20%]

(iv) Neglecting noise, construct a simple deblurring filter for this motion blur and discuss the performance of this filter. [15%]

(v) Discuss (qualitatively) why the Wiener filter should produce better results than inverse filtering, and how one can do even better by non-linear deconvolution techniques. [10%]

3 (a) A coding system for achieving image compression comprises an encoder and a decoder, each of which may be split into three main processing blocks. Draw a block diagram of the system and briefly explain the function of each block. [20%]

(b) In a particular coding system, a 4-point discrete cosine transform (DCT) is used to analyse the input image and reconstruct the output image. Explain why it is desirable for the one-dimensional transform matrix  $T$  to be orthonormal (unitary). Show how  $T$  may be used to calculate the two-dimensional (2D) transform  $Y$  of a  $4 \times 4$  block of image pixels  $X$ , and also how  $X$  may be reconstructed from any given  $4 \times 4$  block of coefficients  $Y$ . [20%]

(c) The matrix  $T$  may be expressed as

$$T = \begin{bmatrix} a & a & a & a \\ b & c & -c & -b \\ a & -a & -a & a \\ c & -b & b & -c \end{bmatrix} \quad \text{where} \quad \begin{cases} a = 1/2 \\ b = \cos(\pi/8)/\sqrt{2} \\ c = \cos(3\pi/8)/\sqrt{2} \end{cases}$$

Show that  $T$  is orthonormal and obtain expressions, in terms of  $a$ ,  $b$  and  $c$ , for the basis functions of the 2D DCT corresponding to coefficients  $y_{1,1}$ ,  $y_{1,2}$ ,  $y_{2,1}$  and  $y_{2,2}$  in the upper left quarter of  $Y$ . (Hint: you may obtain the basis functions by calculating the pixel matrices  $X$ , which correspond to coefficient matrices  $Y$  in which just a single coefficient is unity and the rest are zero.) [20%]

(d) Without further calculation, describe the form of the remaining 12 of the 16 basis functions of the  $4 \times 4$ -point 2D DCT. Suggest reasons why, for images of the real world, the entropies of the 16 resulting DCT subbands, indexed by  $(i, j)$ , tend to decrease as  $(i + j)$  increases from 2 to 8. [20%]

(e) Further reductions in the number of bits of the encoded images may be achieved by choosing the quantiser step-sizes independently for each DCT subband. Explain what property of the human visual system allows this to give improved tradeoffs of perceived image quality versus coded bit-rate (compared with a uniform step-size for all subbands), and how this relates to the 16 DCT basis functions. [20%]

4 (a) A two-band filter bank for analysing one-dimensional (1D) signals comprises filters  $H_0(z)$  and  $H_1(z)$ . The corresponding filter bank for reconstructing signals comprises filters  $G_0(z)$  and  $G_1(z)$ . The analysis filter outputs are down-sampled by 2 by discarding alternate samples, and the inputs to the reconstruction filters are up-sampled by 2 by insertion of zeros in the alternate sample positions. Sketch a block diagram of this filter-bank system and explain why the down and up samplers are used for applications involving signal compression. [20%]

(b) Show that a down-sampler by 2, followed by an up-sampler by 2, converts a signal with z-transform  $Y(z)$  into a signal with z-transform  $[Y(z) + Y(-z)]/2$ . [15%]

(c) Hence, for the system described in part (a), obtain an expression for the output  $\hat{X}(z)$  of the reconstruction filter bank in terms of the four filters,  $H_0$ ,  $H_1$ ,  $G_0$  and  $G_1$ , and the input  $X$  to the analysis filter bank. From this, obtain the two conditions on the four filters to achieve perfect reconstruction of  $X$  at the output  $\hat{X}$ . [20%]

(d) Show how the block diagram of the two-band analysis filter bank may be extended to achieve a 1D wavelet transform over 2 scales (or levels), and then sketch the block diagram for a standard two-dimensional (2D) wavelet transform over the same number of scales. (Assume that  $H_0$  is a lowpass filter and that  $H_1$  is highpass; and similarly for  $G_0$  and  $G_1$ .) Describe briefly what simple features of an input image are picked out by each 2D wavelet subband. [25%]

(e) Explain how the perfect reconstruction property for a simple filter-bank system, as derived in part (c), may be applied to the 2D wavelet transform system of part (d) to ensure perfect image reconstruction in the absence of quantising effects; and also briefly discuss desirable properties for  $G_0$  and  $H_0$  if quantising artefacts are to have low human visibility. [20%]

**END OF PAPER**

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