THIS PAPER IS NOT TO BE REMOVED FROM THE EXAMINATION HALL



CO3346 ZB

BSc EXAMINATION

Creative Computing

Sound and music

Wednesday 8 May 2019: 14.30 - 16.45

Time allowed: 2 hours and 15 minutes

DO NOT TURN OVER UNTIL TOLD TO BEGIN

There are FOUR questions in this paper. Candidates should answer **THREE** questions. All questions carry equal marks, and full marks can be obtained for complete answers to a total of **THREE** questions. The marks for each part of a question are indicated at the end of the part in [.] brackets.

Only your first **THREE** answers, in the order that they appear in your answer book, will be marked.

There are 75 marks available on this paper.

A handheld calculator may be used when answering questions on this paper but it must not be pre-programmed or able to display graphics text or algebraic equations. The make and type of machine must be stated clearly on the front cover of the answer book.

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Question 1 Computational models of music cognition

(a)	What is cognitive science? What is the role of computational modelling in cognitive science?	[3]
(b)	Name and describe TWO application areas for computational models which relate to musical cognition.	[4]
(c)	Music can be represented as a sequence of notes or as an audio recording. Describe the characteristics of these two representations and name file formats that could be used to store them in a computer.	
		[4]
(d)	You have some music represented as a sequence of notes. Name TWO analyses that you can do if you are able to group the notes somehow.	[2]
(e)	Name the set of principles that can be used to group information. Name TWO principles and how they can be used to support the two tasks you named in the previous answer.	
		[4]
(f)	In science, it is common to create models that allow us to make predictions about events or phenomena in the real world. Provide an example of a musical model that can predict something, stating what it aims to predict.	
		[2]
(g)	Explain how an n-gram model can be built out of a sequence of musical notes. Write out examples of the n-grams.	[3]
(h)	Name a key-finding algorithm. What are its inputs and outputs? Does it use an n-gram model?	[3]

Question 2 Computer Music and Pure Data

(a)	The Pure Data environment can be described as a kind of integrated development environment (IDE). During your studies you will have used other IDEs to develop text based languages. Describe THREE ways in which the Pure Data IDE and a text language IDE differ.	[9]
(b)	Write step by step instructions explaining how to build a Pure Data patch that loads and plays back an audio file. Draw diagrams to illustrate your steps, showing how the patch is built. There should be at least FIVE steps.	[10]
(c)	What is synthesized sound? How is it different from sampled sound?	[2]
(d)	Draw a Pure Data patch which implements sound synthesis, using at least two audio generators or processors, including some GUI controls. Annotate the patch, explaining how it works.	[4]

Question 3 Algorithmic Composition

(a)	A composer rolls two six-sided dice and uses the sequence of numbers, which are in the range 2-12, to select notes for a melody. Is this algorithmic composition or normal composition? Why?	[3]
(b)	Name a composer who has used algorithmic composition.	[1]
(c)	Name and briefly describe a non-random technique that can be used for algorithmic composition.	[2]
(d)	Name a computer music performance technique that involves writing code in front of an audience, where the audience can see the code for a musical algorithm being written by the performer.	[1]
(e)	Why do you think people use the technique you mentioned above to perform music? Give TWO reasons.	[2]
(f)	Live algorithms are discussed in the subject guide. They are musical algorithms that can improvise music along with a human performer. Name and describe the FOUR desirable characteristics of these systems.	[8]
(g)	Consider a jazz saxophone player improvising with a human band. For each of the desirable characteristics of a live algorithm you mentioned above, state if the saxophonist has these characteristics.	[4]
(h)	Explain the key elements of a live algorithm based on a swarm model.	[4]

Question 4 Music Information Retrieval

(a) Name a music information retrieval task that has high specificity and one that has low specificity. [2] (b) Explain how a recommender system works. What is the input? What is the output? How is the data structured? Draw a diagram if that helps your explanation. [6] (c) Cepstrum coefficients describe the timbre of the music they are extracted from. Do you think that timbre is an appropriate feature to build a recommender system upon? Justify your answer. [2] (d) Name the two main types of data that are used in music information retrieval. [2] (e) Which type of data would be more appropriate for use in a system which aims to automatically assign a genre to pieces of music? Why did you choose that type of data and why did you not choose the other one? [4] (f) Consider a music recommender system. Given the three inputs below. state how a recommender system might generate a list of recommendations. What kind of data would it need to achieve this? What kind of results might it provide? i. The user has been listening to Mozart's compositions. [3] ii. The user has been searching for acoustic guitars on the Amazon online store. [3] iii. The user has only just signed up for the service, so there is no information about them. [3]

END OF PAPER