# 2007 Assessment Report



### 2007 VET Music Industry (Tech Prod) GA 2: Aural and written exam

#### **SPECIFIC INFORMATION**

For each question, an outline answer (or answers) is provided. In some cases the answer given is not the only answer that could have been awarded marks.

#### **Section A**

Question 1

Marks	0	1	Average
%	27	73	0.8

Reverb added

**Question 2** 

Marks	0	1	2	Average
%	15	55	30	1.2

2a.

Delay echo

#### 2b.

Any one of:

- regeneration
- feedback
- repeats.

**Question 3** 

Marks	0	1	2	3	4	Average
%	19	12	10	25	34	2.5

3a.

1 kHz

3b.

125 Hz

3c.

8 kHz

3d.

 $4 \, \mathrm{kHz}$ 

#### **Question 4**

Marks	0	1	2	Average
%	84	9	7	0.3

Any one of:

- high shelving boost
- more treble
- high frequencies added.

**Question 5** 

Question e								
Marks	0	1	2	3	4	Average		
%	44	18	24	8	6	1.2		

#### 5ai.

The vocal has had compression added.

#### 5aii.

The dynamic range has been reduced.

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#### 5bi.

The drums have been time compressed.

#### 5bii.

The audio plays back more quickly (but the pitch remains the same).

#### **Ouestion 6**

£				
Marks	0	1	2	Average
%	15	26	60	1.5

#### Problem

- popping
- plosives

#### Solution

- use a pop shield, popper stopper, etc.
- use a HPF, bass roll off, move microphone

#### **Question 7**

Marks	0	1	2	3	4	Average
%	11	31	31	22	5	1.8

#### 7a.

A harmony part was added – doubled not double tracked.

#### 7b.

#### Either of:

- reverb was taken off snare
- gate was added to snare.

#### 7c.

Delay was added to the guitar.

#### 7d.

Band pass filter was added.

#### **Question 8**

Marks	0	1	2	Average
%	17	8	75	1.6

#### Problem

- guitar lead has an intermittent connection
- faulty guitar lead
- faulty guitar socket

#### Solution

- replace guitar lead
- replace guitar

#### **Question 9**

Marks	0	1	Average
%	44	56	0.6

#### Either of:

- crossfade
- fades.

#### **Question 10**

Marks 0		1	2	3	Average
%	28	27	32	13	1.3

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- Error: click
- Solution: change edit point to zero crossing and adjust edit points such that the tempo is maintained

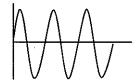
#### **Section B**

#### **Question 1**

Marks	0	1	2	3	4	Average
%	1	3	26	30	41	3.1

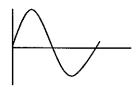
1a.

ii.



1b.

i.



1c.

- X = time
- Y = amplitude, volume or voltage

#### **Question 2**

- 4							
	Marks	0	1	2	3	4	Average
	%	22	40	28	9	2	1.3

2a.

10 dB/decibels

2bi.

Threshold of hearing: 0 dB

2bii

Threshold of pain: 120-140 dB

2c.

8 hours

#### **Question 3**

Marks	0	1	2	Average
%	89	1	10	0.2

343 m/s or ms<sup>-1</sup> or meters per second

#### **Question 4**

Question -						
Marks	0	1	2	3	4	Average
%	22	1	1	4	72	3.1

- A = attack
- D = decay
- S = sustain
- R = release

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#### **Question 5**

Marks	0	1	2	3	Average
%	19	18	18	45	1.9

#### 5a.

20-20 kHz

#### 5b.

150 Hz

#### **Question 6**

Question c	¿ destion o							
Marks	0	1	2	3	4	Average		
%	7	28	31	10	25	2.2		

#### 6a.

96 kHz

#### 6bi.

20-22.05 kHz

#### 6bii.

48 kHz

#### 6c.

ii. 1 minute @ 96 kHz, 16 bit

#### **Question 7**

Z			
Marks	0	1	Average
%	25	75	0.8

Audio CD

#### **Question 8**

Question				
Marks	0	1	2	Average
%	54	23	22	0.7

#### 8a.

0:0.114

#### **8b.**

Click, pop, etc.

#### **Question 9**

£					
Marks	0	1	2	3	Average
%	64	11	12	13	0.8

#### 9a.

#### Either of:

- incorrect sample rate
- the imported file has a sample rate that is lower than the DAW.

#### 9b.

#### Both of:

- convert the sample rate of the imported file to match the DAW
- change the sample rate of the DAW to match the imported file.

#### **Question 10**

	£				
I	Marks	0	1	2	Average
ĺ	%	62	15	23	0.6

Normalising is a process that involves finding the highest peak (loudest part) of a waveform and then increasing the volume (gain) of the whole waveform until the peak is at its allowable maximum before clipping occurs.

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#### **Question 11**

Marks	0	1	2	3	4	5	Average
%	4	26	18	22	18	13	2.6

- 1. phase reverse, invert polarity
- 2. high mid frequency select
- 3. high mid cut/boost control
- 4. pan control sends signal between left and right or odd and even buses
- 5. PFL pre fader listen to monitor the channel through headphones or on meter without affecting the mix

#### **Question 12**

Marks	0	1	2	Average
%	20	29	51	1.3

- 1. mix control allows the balance between wet/dry signal to be set
- 2. decay controls the time taken for the reverb to decay away (RT<sub>60</sub>)

#### **Question 13**

C						
Marks	0	1	2	3	Average	
%	39	8	23	29	1.4	

#### 13a.

Any one of:

- phase cancellation
- snare gets softer
- body of snare sound is decreased.

#### 13b.

Both of:

- reverse the phase of one microphone
- move one microphone further away/closer.

#### **Question 14**

ĺ	Marks	0	1	2	3	Average
ĺ	%	12	29	33	26	1.7

#### All of:

- only the PA mix is recorded, not the stage sound
- softer instruments come out loudest
- no crowd noise is recorded.

#### **Ouestion 15**

Z					
Marks	0	1	2	3	Average
%	5	12	34	50	2.3

#### Any three of:

- reduce the LF of the acoustic guitar channel, feedback stopper
- reduce the LF of the foldback send
- reduce the level of the guitar
- reverse the phase of the channel.

#### **Question 16**

Marks	0	1	2	3	4	Average
%	17	10	19	21	33	2.4

#### 16a.

Either of:

- 48 v that can be sent down a microphone cable/balanced line to any device that requires it
- the power supply for condenser microphones or active DIs that does not interfere with the audio signal.

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#### 16b.

Both of:

- condenser microphone
- active DI.

**Question 17** 

Marks	0	1	2	3	Average
%	23	24	29	24	1.5

#### 17ai.

Ohms

#### 17aii.

Resistance or impedance

#### 17b.

 $4 \Omega$ 

#### **Question 18**

Marks	0	1	2	3	Average
%	24	32	31	13	1.3

#### 18a.

Any two of:

- · recording out
- drum (or other instrument, vox, CD, fx, etc.) master
- delay feed
- sidefill/foldback/drumfill.

#### 18b.

Any of one:

- unity gain
- 1:1
- no attenuation no boost.

#### **Question 19**

Marks	0	1	2	3	Average
%	43	4	9	44	1.6

It splits the incoming audio into three frequency bands – lo, mid and hi.

One mark was awarded for identifying the split, and the further three marks for naming each of the frequency bands.

#### **Ouestion 20**

1	£						
	Marks	0	1	2	3	4	Average
	%	58	17	13	5	8	0.9

It allows the individual cut/boost of frequencies in one third of an octave bands across the 20–20 kHz spectrum.

#### **Question 21**

Marks	0	1	2	Average
%	72	3	24	0.5

#### 21a.

post (fader) aux

#### 21b.

pre (fader) aux

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#### **Question 22**

Marks	0	1	2	Average
%	8	32	60	1.5

Any two of:

- microphone/volume/gain too loud
- microphone too close to speaker
- (directional) microphone pointing at speaker PA.

#### **Question 23**

Marks	0	1	2	3	Average
%	63	22	6	9	0.6

#### 23a.

It is an increased low frequency response/output when the microphone is very close (closer than a few centimetres) to the sound source.

#### 23b.

Cardioid/unidirectional microphones

#### **Question 24**

l	Marks	0	1	2	3	Average
	%	22	54	15	9	1.1

#### 24a.

To avoid audible hum loops.

#### 24b.

Both of:

- it might draw too much current/power
- it could create audible buzz.

#### **Question 25**

Question 25								
Marks	0	1	2	3	Average			
%	21	24	22	32	1.7			

#### 25a.

Omni-directional

#### 25b.

Cardioid/unidirectional

#### 25c.

Figure 8/bi-directional