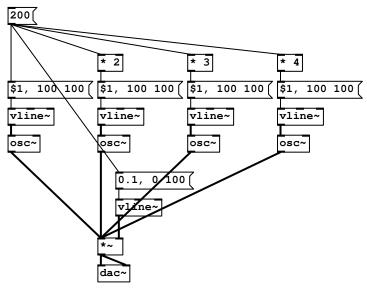
# 21M.380 Music and Technology Sound Design

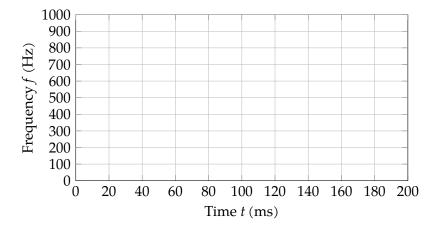
First Quiz (QZ1)
Physics & Perception of Sound, Digital Audio, Pd Basics

Monday, March 14, 2016 30 minutes, 10% of total grade

## Pd patch for questions 1.1 (p. 2) and 2.2 (p. 4)



# Diagram for question 1.1 (p. 2)



Initials:			
1	Physics of sound		
1.1	Spectrogram (15%)		

In the diagram provided for this purpose on page 1, draw a spectrogram of the sound that the Pd patch from page 1 generates when the user hits the [200( message in the top left corner at t=0 (assuming the DSP is on).<sup>1</sup>

#### 1.2 Resonating pipe (15%)

The length l of a pipe closed on one end coincides with  $\frac{\lambda_1}{4}$ , a quarter of the wavelength that corresponds to its fundamental frequency  $f_1$ . Such a pipe will produce only odd harmonics  $(f_1, f_3, f_5,$  etc.), with even harmonics  $(f_2, f_4,$  etc.) absent.

• What is the frequency  $f_5$  of the fifth harmonic created by a pipe closed at one end with a length of  $l=3.4\,\mathrm{m}$ ? Clearly show how you derive your result and quantify the result in Hz; don't leave fractions. This problem requires knowledge of the speed of sound c in air.

<sup>1</sup> Do not worry about a precise representation of amplitude, it is sufficient to indicate the correct frequencies, onset times, and stop times of all partials.

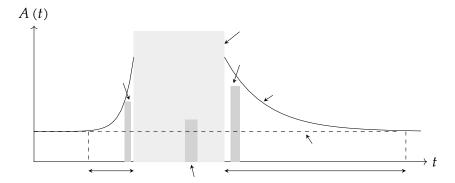
•	Describe the two aspects of the pipe's spectrum	that	would
	change if its closed end was opened.		

1. \_\_\_\_\_

2. \_\_\_\_\_

### 2 Perception of sound

#### 2.1 Psychoacoustics (15%)



- Which psychoacoustic phenomenon does the figure above illustrate? State its precise name. If there are multiple flavors of this phenomenon, indicate which of these the above figure refers to.
- Label all arrows in the above figure to support your description, using professional terminology. Provide rough estimates for the length of the time periods indicated by the two horziontal arrows.
- Describe the effect that this phenomenon has on the perception of the three sounds that are represented in the figure by the three dark gray blocks. How does this relate to the other elements in the figure?

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Initials:	21M.38
Describe to a listener who has never experienced the Pd from page 1, and who has no technical understanding o tro)acoustics whatsoever, what the patch sounds like. Embra possibilities that written text offers you in this regard. An goes, as long as your description employs text in some ca (it should not be purely pictographic) and captures the se essential perceptual characteristics.	f (elec- ace any ything apacity
3 Digital audio	
3.1 Digital audio file formats (5%)	
What is the difference between lossless and lossy comprint the context of digital audio formats?	ession
• For both lossless and lossy compression, name one exama container format (file name extension) <i>and</i> a corresponding	

4 OF 7

audio codec.

- Lossless compressed:

- Lossy compressed:

Container: \_\_\_\_\_. Codec: \_\_\_\_\_

Container: \_\_\_\_\_. Codec: \_\_\_\_\_

3.2 Sampling (5%)
• You are trying to optimize a digital audio system that plays music from before 1970, which – as we all know – only people over 40 years of age listen to. Which sample rate would you choose for this system, such that it just covers its audience's entire range of hearing? Show how you derived your result.
3.3 Digital artefacts (5%)
• You will hear a sound example, first without, then with a digital artefact. Which artefact is this an example of?
O Clipping
O Aliasing
O Click
• What has likely caused this artefact?
O A bit rate that was chosen too low
O A signal level that was chosen too high
O A sample rate that was chosen too low
• Describe the artefact, this time in purely technical terms (ideally by a single phrase). How does the original spectrum compare with the spectrum that contains the artefact?
3.4 Quantization and sampling (10%)
Check any statements that are correct (false positives are penalized)
<ul> <li>96 kHz is a sample rate widely used in professional audio production.</li> </ul>

 $\Box$  A bit depth of 16 bit is sufficient to prevent aliasing in most

situations.

Initials: \_\_\_\_ . \_\_\_\_ .

Initials:	21M.380, First quiz (Mon, 3/14/2016)
<ul> <li>Quantization creates periodically repeating copies of t spectrum (sidebands).</li> </ul>	he original
$\square$ With 9 bit, one can express 512 different amplitude v	ralues.
☐ Lossless compressed audio file formats exploit the ph of the missing fundamental in order to reduce data.	enomenon
☐ The dynamic range of a digital signal quantized at 16 bit 144 dB.	it is roughly
$\square$ 256 kbit s <sup>-1</sup> is a popular bit depth for uncompressed	audio.
One can truthfully restore signals which have been sa sample rate that complies with the sampling theorem, no quantization has occured.	•
☐ The highest frequency that can be faithfully reprodu sample rate of 8 kHz is 16 kHz.	iced with a
☐ A bit depth of 24 bit provides a dynamic range that ex of the human auditory system.	xceeds that
4 Pure Data basics	
4.1 Happy patching! (15%)	

In the provided space on page 7, implement a Pd abstraction counter.pd according to the following specifications:<sup>2</sup>

- The abstractions first and only creation argument \$1 should allow the user to set the increment, i.e., the amount by which the counter will increase upon each [bang( that it receives on its left inlet (see below). If \$1 is not provided (in which case Pd assumes it to be zero), the increment should default to 1.
- When the abstraction's left inlet receives a [bang( message, it should increment an integer counter and immediately output the new count at the abstraction's only outlet. Counting should start at the value of the increment (not zero), so the first three [bang( mesages sent to the left inlet of a [counter 2] should yield 2, 4, 6 (not 0, 2, 4).
- The abstraction's right inlet should accept a [bang( message that
  resets the counter, such that it starts the counting sequence again
  once it receives the next [bang( on its left inlet (but not earlier).
  The counting sequence should start at the value of the increment
  (not zero) also after a reset.

<sup>&</sup>lt;sup>2</sup> You can use the margins on this page for sketches (which will not be graded).

Answer to question 4.1				

Initials: \_\_\_\_ . \_\_\_\_ .

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