



A ribbon microphone requires phantom power.	3	100.0%
Condenser microphones operate on the electrostatic principle.	6	100.0%
Directional microphones are always subject to the proximity effect.	19	84.2%
Dynamic microphones make use of the principle of electromagnetic induction.	4	100.0%
If you use a raw-frame loudspeaker without a baffle of any kind, you will experience a loss of high frequencies.	7	85.7%
Most condenser microphones have a lower moving mass than most dynamic microphones.	4	50.0%
Pure pressure gradient microphones are never subject to the proximity effect.	19	73.7%
The frequency response of a microphone is controlled primarily by the mass and stiffness of the diaphragm.	10	100.0%
The human auditory system is a transducer.	5	100.0%
Which type of connector is shown in the above image?	3	100.0%
	1	100.0%
Aliasing becomes audible as inharmonic distortion.	3	100.0%
Dither is deliberately added to a digital audio signal to trade off distortion for noise.	5	100.0%
Jitter is instability in clock speed during sampling.	6	100.0%
What type of filter does this circuit diagram show?		
A band-limited analog signal can be sampled without loss.	2	50.0%
A compressor with a ratio below 10:1 can be considered a limiter.	2	100.0%
A gate is the extreme version of an expander.	4	75.0%
A higher compressor ratio will result in larger amounts of overall compression.	7	85.7%
A higher compressor threshold will result in larger amounts of overall compression.	3	66.7%
Aliasing becomes audible as inharmonic distortion.	5	60.0%
Aliasing is caused by sample rates that exceed twice the maximum frequency in a signal.	3	66.7%
Decreasing the bit depth too much will result in aliasing.	3	66.7%
Dither is deliberately added to a digital audio signal to trade off distortion for noise.	2	50.0%
Faster compressor attack means:	1	100.0%

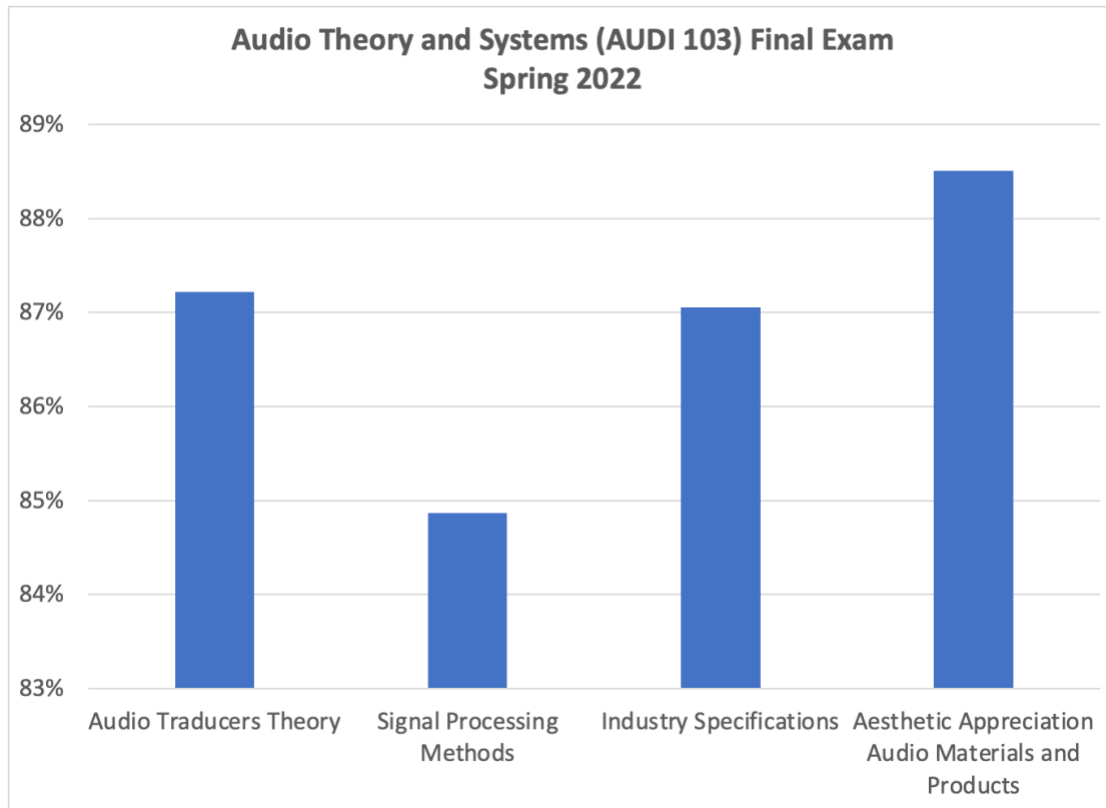
Jitter is instability in clock speed during sampling.	5	100.0%
On analog mixers, inserts are typically used for loop-in effects such as compression or distortion.	19	89.5%
Playing back a signal that has been sampled with jitter results in distortion.	2	50.0%
The output signal of a compressor will always be affected by the compressor's make-up gain, regardless of whether the input signal crosses the threshold or not.	2	100.0%
The process of sampling creates copies of the original spectrum that repeat periodically around multiples of the sample rate.	4	50.0%
The reconstruction filter in a DAC is a high-pass filter that cuts off all sidebands below the Nyquist frequency.	2	50.0%
The term dither describes irregularities of the sample clock.	3	100.0%
What type of filter does this circuit diagram show?	5	80.0%
What type of filter does this circuit diagram show?	19	89.5%
Which form of data compression does a .flac file using the FLAC coding format employ?	9	77.8%
Which form of data compression does a .opus file using the Opus coding format employ?	8	100.0%
Which form of data compression does a .wav file with PCM-encoded audio data employ?	4	100.0%
Which form of data compression does an .caf file with PCM-encoded audio data employ?	5	80.0%
Which form of data compression does an .m4a file using the AAC coding format employ?	7	85.7%
Which form of data compression does an .mp3 file using the MPEG layer III coding format employ?	5	100.0%
With a bit depth of $N = 7$ , one can express 128 different amplitude values.	6	83.3%
4, 8, and 16 are typical values for the rated nominal impedance of a loudspeaker.	3	100.0%
A bit depth of 20 gives you a dynamic range of roughly 120 dB.	6	100.0%
A loudspeaker's impedance rating is typically consistent throughout the frequency range.	6	100.0%
A loudspeakers's polar pattern typically becomes less directional for higher frequencies.	3	100.0%
A microphone's dynamic range refers to the difference between the highest and lowest frequency it can safely handle.	9	77.8%
A surface absorption coefficient, $\alpha$ , of 0.0 is referred to as the equivalent of an open window in acoustics.	5	80.0%

A surface absorption coefficient, $\alpha$ , of 0.0 means all the sound falling on that surface is reflected back into the room.	3	100.0%
T60 is measured in a given room as the time difference between the moment of arrival of the direct sound, and the moment of arrival of its first reflection.	3	66.7%
The reverberation time, T60, is the time it takes the residual sound-pressure level in a room to drop by 60 dB.	3	100.0%
The reverberation time, T60, can vary with frequency.	5	100.0%
The surface absorption coefficient, $\alpha$ , is always a constant with regards to frequency.	1	100.0%
What is 0.0024 U, in engineering notation?	2	100.0%
What is 0.024 U, in engineering notation?	3	66.7%
What is 0.0829 U, in engineering notation?	5	100.0%
What is 0.829 U, in engineering notation?	4	75.0%
What is 1.776 kU, in regular decimal notation?	3	100.0%
What is 1.776 MU, in regular decimal notation?	6	83.3%
What is 250 mU, in regular decimal notation?	1	100.0%
What is 250 $\mu$ U, in regular decimal notation?	3	100.0%
What is 34,567.9 U, in engineering notation?	2	50.0%
What is 345,679 U, in engineering notation?	3	33.3%
What is 6.03 mU, in regular decimal notation?	3	66.7%
What is 60.3 mU, in regular decimal notation?	3	100.0%
	10	100.0%
Which of the following visualizations of sound is the above image an example of?		
	4	50.0%
Which of the following visualizations of sound is the upper part of the above image an example of?		
A loudspeaker with an efficiency of 1% will convert 99% of the electrical energy coming into it to heat.	6	100.0%
A room's critical distance will (everything else being equal) increase if its T60 time increases.	3	100.0%
A room's reverberation time, T60, is proportional to its volume.	1	100.0%
Dynamic loudspeakers tend to exhibit distortion at the second and third harmonic.	8	75.0%
One characteristic of the reverberant field is that its sound level varies randomly across the room.	5	80.0%
The critical distance is the distance from a sound source in a room at which direct and reverberant sound levels are exactly equal.	5	80.0%

The high-frequency response of dynamic microphones tends to be superior to that of condenser microphones.	4	100.0%
The time interval between the arrival of the direct sound and the first major reflection has little or no influence on the auditory perception of a given space's size.	4	100.0%
When using auxiliary sends to feed on-stage monitor loudspeakers, they are usually set to post-fader.	19	89.5%
Which of the following visualizations of sound is the above image an example of?	8	100.0%
Which of the following visualizations of sound is the upper part of the above image an example of?	5	60.0%
Which of the following visualizations of sound is the upper part of the above image an example of?	5	100.0%

In terms performance per category, the results were:

Category	% Correct
Audio Traducers Theory	87.2%
Signal Processing Methods	84.9%
Industry Specifications	87.1%
Aesthetic Appreciation Audio Materials and Products	88.5%



The results suggest...

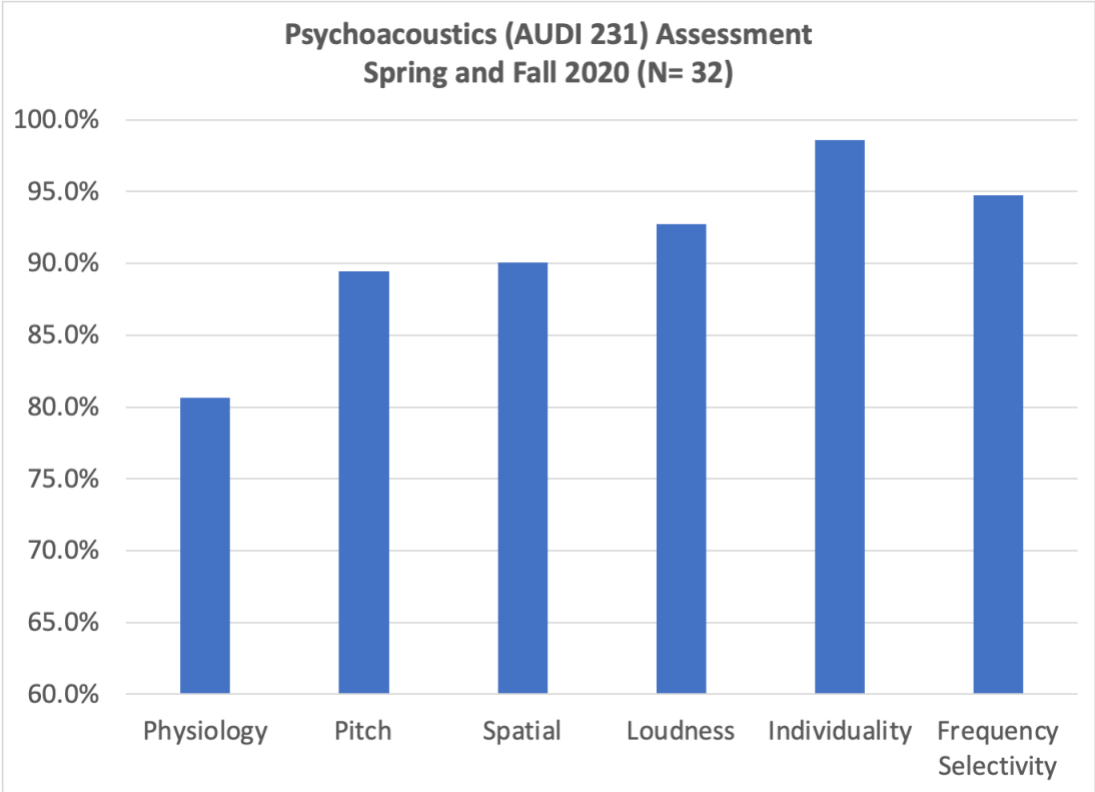
### Psychoacoustics

The department examined performance on the 16 question Final Exam in the over two semesters – Spring 2020 and Fall 2020. (Because the 16<sup>th</sup> question covered multiple categories, it was not included in the analysis.)

Question	Category	% Correct
Q. 1: Why does the ear exhibit increased sensitivity at around 1kHz to 6kHz (i.e. how does the ear accomplish this and what value is it for human perception)?	Physiology	80.6%
Q.2 When multiple notes are played together, we have a chord. A chord can be either consonant (harmonious) or dissonant (inharmonic). What quantity makes a chord consonant or dissonant? (You may use examples to explain it.) Does different temperaments play a role in the harmonicity of the chords?	Pitch	81.3%
Q.3 What is temperament? How does it make a difference in music performance? Compare the three major temperaments introduced in class?	Pitch	97.7%

Q. 4 For two sound sources in front and back, what are the ILDs and ITDs?	Spatial	92.2%
Q. 5 Playing the same source in front and back, do they sound the same to an average listener	Spatial	75.0%
Q.6 Why can a listener turn his/her head to resolve the ambiguity in discriminating front and back sources?	Spatial	87.5%
Q. 7 Using the equal loudness contours graph, show as follows,	Loudness	92.7%
Q.8 Describe ILD and its role in sound source localization.	Spatial	98.8%
Q.9 Describe ITD and its role in sound source localization.	Spatial	96.9%
Q.10 Describe HRTF and its role in sound source localization.	Individuality	99.4%
Q.11 When listening to a classical opera, do you prefer headphones or loudspeakers?	Individuality	100.0%
Q.12 Explain in detail the “precedence effect”	Spatial	96.1%
Q.13 You are hired to create a new sound file format capable of reducing file size while maintaining sound quality.	Frequency Selectivity	93.8%
Q.14 What is masking effect?	Frequency Selectivity	95.7%
Q.15 individuality (the difference among different listeners) in spatial hearing.	Individuality	96.4%

<b>Category</b>	<b>N</b>	<b>Mean</b>
Physiology	1	80.6%
Pitch	2	89.5%
Spatial	6	90.1%
Loudness	1	92.7%
Individuality	3	98.6%
Frequency Selectivity	2	94.7%



Note: We still need Audio faculty to comment on this draft.