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Turn in Assignment for Basic Audio Engineering -Chapter #20 - Final

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Basic Audio Engineering - Chapter #20 - Final -Quiz

1. Electronic circuits can be broken down into two types.



- C. Feeder D. Electronic 4. To cleanly capture the detail of a hi-hat, a _____ microphone should be used. A. Dynamic B. Condenser C. Ribbon D. Boundary
- 5. Electricity loses energy over long distance so low voltage transmissions help to extend voltage over a long distance.



6. _____ mics are generally speaking the most rugged type of microphone design.



7. Miking an acoustic guitar near the _____ may result in too much low end.



- C. Pick Guard D. String Pegs
- 8. The ______ configuration uses two figure 8 microphones, with their heads close together, at 90 degree angles.



9. What type of microphone circuit is pictured here?



10. _____ microphones use a diaphragm attached to a moving coil that surrounds a magnet. Voltage is created through induction.



11. ______ refers to the difference between what frequencies a performance is actually transmitting, to what the microphone is capturing.



12. ______ is credited as the inventor of the microphone.



13. A ______ works by storing recorded sounds in memory, allowing the individual sounds to be played back as they are triggered.

14. The ______ refers to the amount of possible amplitude values present in the digital audio signal.

O A. Sample Rate
o B. Bit Depth
C . Pulse Code
O D. Amplitude Level

15. A ______ was one of the first digital audio devices. It is a device that keeps track of the order trigger events are played in.



16. _____ remains the most widely used digital audio format today, however is a lossy compressed format that is inferior to other formats.

A. wav B. jpg C. mp3 D. flac

17. The Nyquist Theorem states that:

A. The highest frequency able to be sampled is double the sample rate.
B. The highest amplitude able to be sampled is half the sample rate.
C. The highest frequency able to be sampled is half the sample rate.
D. The highest frequency able to be sampled is 3 times the sample rate.

18. Electricity is basically the flow of protons from one atom to the next.



19. How much headroom should you leave for the mastering engineer?



20. The master is:

A. the head engineer at the studio B. the 1st copy of a song or album that all duplications are based off of. C. the degree that a producer needs to work for Warner-Chapel records. D. the first reproduction of a song or album.

21. Checking your mix in mono:

A. enables you to see potential problems that may occur if someone hears your song on a radio station.

B. can help you examine potential phasing issues resulting from the left and right channels being mixed together.

C. Both A and B **D.** None of the Above 22. Bouncing takes the output of the master (analog or digital) and records it on one stereo track.



23. Audiosuite plugins save processing power by:



24. You must have a clip selected to use Audiosuite plugins



25. TDM stands for:

A. Time Division Multiplexing B. Timed Decent Multiprocessing C. Total Dimension Multiprocessing D. Too Dope Man (as in that beat is TDM!!!)

26. The VST (Virtual Studio Technology) Plugin Format was developed by _____

A. Avid

- B. Cubase C. Steinberg D. Antares
- 27. The render button on an Audiosuite plugin:

• A. Renders the changes you have made to the audio file itself (and creates a new file in the clip list)

- B. Renders video projects
 C. Renders at a different sample rate
 D. None of the Above
- 28. The ______tab allows for detailed routing of your inputs and outputs for Pro Tools and has configurations for your interface.
 - A. Bus Setup B. I/O Setup C. Playback Engine D. Disk Allocation
- 29. The ______ has three variations in its drop down menu. Time, separation, and, object.

A. Grabber Tool B. Scrubber Tool C. Trimmer Tool D. Zoomer Tool 30. ______ introduces a predetermined amount of delay into your signal path for each channel so that all channels play back in sync.



31. Key Command: New Track



32. Inserts process ______ of the signal and are inserted directly into your signal flow.



33. _____ is an edit mode that conforms clips to predetermined time values in your edit window.



- **D.** Spot Mode
- - A. Digital Audio Workspace
 B. Dynamic Audio Workbrowser
 C. Digi Automated Workstation
 D. Digital Audio Workstation
- 35. The first version of Pro Tools was released in _____.



36. A(n) ______ patch bay connection is broken when a patch cable is inserted into either the top or the bottom jack.



37. What connector is this?



38. What connector is this?



D. ELCO

39. What connector is this?



A. RCA B. TRS C. DB25 D. XLR

40. Microphone level is_____.



41. Before recording an instrument it is usually a good idea to______.

A. Check the tuning B. Listen to the instrument in the room C. Find the best place to mic D. All of the above

42. Placing ______ in a kick drum will help to deaden it.



43. What type of reverb is created within a small metal coil?



44. A song can be made 100% ready for duplication, including, metadata such as ISRC codes and SKU numbers using Pro Tools.



45. Dither is used to:

A. Fix frequency recreation errors.
B. Adjust the relative volume of a mix.
C. Fix quantization errors due to lowering sample rate and bit depth.
D. Hone in on frequency problems during the mastering process.

46. The period of time from the early 90s to early 2000s where we saw less and less dynamic range in our music and more and more relative loudness was referred to as the:

0
A. Loudness Match
0
B. Loud Chase
0
C. Volume Wars
0
D. Loudness Wars

47. ______ is the Red Book CD Standard for sample rate and bit depth.



48. A ______ is really a series of delays.



49. ______ is measured in Hertz abbreviated Hz: 1 Hz = 1 cycle per second. A. Rarefaction B. Diffusion C. Frequency

D. Wave Period

50. ______ are any materials or structures that is naturally or designed to be acoustically unreflective.

A. Diffusers
• B. Absorbers
C. Traps
D. Phasers

51. The ________is the point space where an audio engineer (or other listener) is most likely to be sitting for monitoring and mixing.



52. A resistor:



53. An EQ is usually a combination of filter circuits.



54. A low pass filter:

A. Allows frequencies below a user defined center frequency to pass unaffected.

B. Allows frequencies above a user defined center frequency to pass unaffected.

C. Allows frequencies in a certain bandwidth to pass unaffected.

D. None of the above.

55. What is MIDI an acronym for?

A. Musical Interface Digital Instructions
 B. Music Inside Digital Information
 C. Musical Instrument Digital Interface
 D. Musical Instrument Diode Isolator

56. A Parametric EQ gives the user:

A. All three parameters of EQ, bandwidth, frequency, and amplitude.
B. All three parameters of EQ, bandwidth, amplitude, and threshold.
C. Both parameters of EQ, bandwidth and frequency.
D. None of the above

57. A ______ EQ has predefined center frequencies and a predetermined bandwidth amount. All the user has control over is the amount of boost and cut.

A. Fully Parametric B. Program EQ C. Graphic EQ (Fixed Frequency)

- **D.** Multiband Compressor
- 58. Which tools would we use to alter the presence and spatial characteristics of the delay or reverb signal within our mix?

O A. Volume
O B. Panning
C. EQ
• D. All of the Above

59. Which two synthesizers were involved at the first public MIDI connection at the 1983 Winter NAMM Show?:

A. Sequential Circuits Prophet-600 and MiniMoog Model D
B. Sequential Circuits Prophet-600 and Roland Jupiter-6
C. Dave Smith Instruments Poly Evolver and Moog Slim Phatty
D. Roland Juno-6 and Oberheim SEM

60. The ______ determines the amount of gain reduction of a compressor.

A. Ratio B. Attack C. Threshold D. Release

61. Velocity refers to:

A. The dynamic range of a given sound

- B. The number of layers in a given synth patch
- **C.** The force by which a given note is struck
- **D.** A feature introduced in 1982 and still found on most digital instruments
- 62. Modern ______ recording is a method of recording where signals are stored as a continual wave in or on the media, usually magnetic tape.



63. A ______ reduces dynamic range.



64. The ______ determines how quickly gain reduction is reset after the signal drops back below the threshold.



65. Polyphony refers to:

- A. The number of harmonics in a given sound
- **B.** The ability to play more than one note at a time
- **C.** The number of notes in a melody
- **D.** The number of tracks in your sequencer

66. Time Based Effects can add:



67. Mastering is:

A. The art of getting music ready for duplication.

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B. The process of making sure the relative loudness of the final product is competitive in the genre of music.

- **C.** The final polish music production.
- •
- **D.** . All of the above
- 68. _____ implies that a signal is being split, and one half of the split signal is compressed separately using an aux send, and is then recombined with the original signal.



69. One of the earliest uses of something resembling ______ was during WWII when it was found that steady vibrations reduced the errors of trajectory calculation computers.

A. ISRC
O B PQ Codes
 Dither
O Dither
D. Noise Cancellation

70. Low frequency buildup describes a particular type of constructive ______in the low frequency spectrum.



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