MULTIPLE CHOICE

- 1. The first digital code was the:
 - a. ASCII code
 - b. Baudot code

ANS: C

c. Morse coded. none of the above

2. In digital transmission, signal degradation can be removed using:

a. an amplifierc. a regenerative repeaterb. a filterd. all of the above

ANS: C

- 3. TDM stands for:
 - a. Time-Division Multiplexing
 - b. Time-Domain Multiplexing

ANS: A

- 4. Hartley's Law is:
 - a. I = ktB

b. $C = 2B \log_2 M$

ANS: A

5. The Shannon-Hartley theorem is:

a. I = ktB

b. $C = 2B \log_2 M$

ANS: B

6. The Shannon Limit is given by: a. I = ktBb. $C = 2B \log_2 M$

ANS: C

7. The Nyquist Rate can be expressed as:
a. I = ktB
b. C = 2B log₂M

ANS: D

- 8. Natural Sampling does not use:
 - a. a sample-and-hold circuit
 - b. true binary numbers

ANS: A

c. $C = B \log_2(1 + S/N)$ d. $SR = 2f_{max}$

c. Ten-Digital Manchester

d. Ten Dual-Manchester

c. $C = B \log_2(1 + S/N)$

d. $SR = 2f_{max}$

- c. $C = B \log_2(1 + S/N)$ d. $SR = 2f_{max}$
- c. $C = B \log_2(1 + S/N)$ d. $SR = 2f_{max}$
- c. a fixed sample rate
- d. an analog-to-digital converter

- 9. Which is true about aliasing and foldover distortion?
 - a. They are two types of sampling error.
 - b. You can have one or the other, but not both.
 - c. Aliasing is a technique to prevent foldover distortion.
 - d. They are the same thing.

ANS: D

- 10. Foldover distortion is caused by:
 - a. noise

- c. too few samples per second
- b. too many samples per second
- d. all of the above

ANS: C

11. The immediate result of sampling is:

a.	a sample alias	c.	PCM
b.	PAM	d.	PDM

ANS: B

- 12. Which of these is not a pulse-modulation technique:
 - a. PDM c. PPM
 - b. PWM d. PPS

ANS: D

- 13. Quantizing noise (quantization noise):
 - a. decreases as the sample rate increases
 - b. decreases as the sample rate decreases
 - c. decreases as the bits per sample increases
 - d. decreases as the bits per sample decreases

ANS: C

- 14. The dynamic range of a system is the ratio of:
 - a. the strongest transmittable signal to the weakest discernible signal
 - b. the maximum rate of conversion to the minimum rate of conversion
 - c. the maximum bits per sample to the minimum bits per sample
 - d. none of the above

ANS: A

- 15. Companding is used to:
 - a. compress the range of base-band frequencies
 - b. reduce dynamic range at higher bit-rates
 - c. preserve dynamic range while keeping bit-rate low
 - d. maximize the useable bandwidth in digital transmission

ANS: C

- 16. In North America, companding uses:
 - a. the Logarithmic Law

c. the α Law (alpha law)

b. the A Law

d. the μ Law (mu law)

ANS: D

- 17. In Europe, companding uses:
 - a. the Logarithmic Law
 - b. the A Law

ANS: B

- 18. Codec stands for:
 - a. Coder-Decoder
 - b. Coded-Carrier

ANS: A

- 19. A typical codec in a telephone system sends and receives:
 - a.4-bit numbersc.12-bit numbersb.8-bit numbersd.16-bit numbers

ANS: B

- 20. Compared to PCM, delta modulation:
 - a. transmits fewer bits per sample
 - b. requires a much higher sampling rate

ANS: D

- 21. In delta modulation, "granular noise" is produced when:
 - a. the signal changes too rapidly
 - b. the signal does not change d. the sample is too large

ANS: B

- 22. Compared to PCM, adaptive delta modulation can transmit voice:
 - a. with a lower bit rate but reduced quality
 - b. with a lower bit rate but the same quality
- c. only over shorter distances

c. can suffer slope overload

c. the bit rate is too high

d. all of the above

d. only if the voice is band-limited

ANS: B

- 23. Which coding scheme requires DC continuity:
 - a. AMI
 - b. Manchester

c. unipolar NRZd. bipolar RZ

ANS: C

- 24. Manchester coding:
 - a. is a biphase code
 - b. has a level transition in the middle of every bit period
 - c. provides strong timing information
 - d. all of the above

ANS: D

25. The number of framing bits in DS-1 is:

- c. the α Law (alpha law)
- d. the μ Law (mu law)
- c. Code-Compression
- d. none of the above

	a. 1 b. 2	c. d.	4 8
	ANS: A		
26.	Framing bits in DS-1 are used to:a. detect errorsb. carry signalingANS: C	c. d.	synchronize the transmitter and receiver all of the above
27.	So-called "stolen" bits in DS-1 are used to:a. detect errorsb. carry signalingANS: B	c. d.	synchronize the transmitter and receiver all of the above
28.	The number of bits per sample in DS-1 is: a. 1 b. 2 ANS: D	c. d.	4 8
29.	The number of samples per second in DS-1 is: a. 8 k b. 56 k	c. d.	64 k 1.544×10^{6}
30.	The bit rate for each channel in DS-1 is: a. 1.544 Mb/s b. 64 kb/s ANS: B	c. d.	56 kb/s 8 kb/s
31.	In DS-1, bits are transmitted over a T-1 cable a a. 1.544 MB/s b. 64 kb/s ANS: A	t: c. d.	56 kb/s 8 kb/s
32.	 A T-1 cable uses: a. Manchester coding b. bipolar RZ AMI coding ANS: B 	c. d.	NRZ coding pulse-width coding
33.	The number of frames in a superframe is: a. 6 b. 12 ANS: B	c. d.	24 48
34.	A typical T-1 line uses:		

- a. twisted-pair wire
- b. coaxial cable

ANS: A

- 35. "Signaling" is used to indicate:
 - a. on-hook/off-hook condition
 - b. busy signal

ANS: D

- 36. A vocoder implements compression by:
 - a. constructing a model of the transmission medium
 - b. constructing a model of the human vocal system
 - c. finding redundancies in the digitized data
 - d. using lossless techniques

ANS: B

- 37. Compared to standard PCM systems, the quality of the output of a vocoder is:
 - a. much better
 - b. somewhat better

d. not as good

c. about the same

ANS: D

COMPLETION

- 1. Digitizing a signal often results in ______ transmission quality.
 - ANS: improved better

2. To send it over an analog channel, a digital signal must be ______ onto a carrier.

ANS: modulated

3. To send it over a digital channel, an analog signal must first be ______.

ANS: digitized

4. In analog channels, the signal-to-noise ratio of an analog signal gradually ______ as the length of the channel increases.

ANS: decreases gets worse

5. The ______ value of a pulse is the only information it carries on a digital channel.

ANS: binary

- c. fiber-optic cable
- d. microwave

c. ringing

d. all of the above

6.	Α	A repeater is used to restore the shape of pulses on a digital cable.		
	ANS:	regenerative		
7.	There a	re techniques to detect and some errors in digital transmission.		
	ANS:	correct		
8.	Conver transm	ting an analog signal to digital form is another source of in digital ission systems.		
	ANS: error noise			
9.		-division multiplexing is easily done in digital transmission.		
	ANS:	Time		
10.	All pra	ctical communications channels are band		
	ANS:	limited		
11.		Law gives the relationship between time, information capacity, and bandwidth.		
	ANS:	Hartley's		
12.	Ignorir for a gi	g noise, the theorem gives the maximum rate of data transmission ven bandwidth.		
	ANS:	Shannon-Hartley		
13.	The and a g	limit gives the maximum rate of data transmission for a given bandwidth iven signal-to-noise ratio.		
	ANS:	Shannon		
14.		sampling is done without a sample-and-hold circuit.		
	ANS:	Natural		
15.	The format	Rate is the minimum sampling rate for converting analog signals to digital		
	ANS:	Nyquist		
16.		distortion occurs when an analog signal is sampled at too slow a rate.		
	ANS:	Foldover		

17.	means that higher frequency baseband signals from the transmitter "assume the identity" of low frequency baseband signals at the receiver when cont divitally	
	identity of low-inequency baseband signals at the receiver when sent digitally.	
	ANS: Aliasing	
18.	The output of a sample-and-hold circuit is a pulse modulated signal.	
	ANS: amplitude	
19.	modulation is the most commonly used digital modulation scheme.	
	ANS: Pulse-code	
20.	noise results from the process of converting an analog signal into digital format.	
	ANS: Quantizing	
21.	is used to preserve dynamic range using a reasonable bandwidth.	
	ANS: Companding	
22.	In North America, compression is done using thelaw equation.	
	ANS:	
	mu	
23.	In Europe, compression is done using thelaw equation.	
	ANS: A	
24.	A is an IC that converts a voice signal to PCM and vice versa.	
	ANS: codec	
25.	In a PCM system, the samples of the analog signal are first converted to bits before being compressed to 8 bits.	
	ANS: 12	
26.	The number of bits per sample transmitted in delta modulation is	
	ANS:	
	one	
27.	Delta modulation requires a sampling rate than PCM for the same quality of reproduction.	
	ANS: higher	

28.	noise is produced by a delta modulator if the analog signal doesn't change.
	ANS: Granular
29.	In delta modulation, overload can occur if the analog signal changes too fast.
	ANS: slope
30.	The size varies in adaptive delta modulation.
	ANS: step
31.	Adaptive delta modulation can transmit PCM-quality voice at about the bit rate of PCM.
	ANS: half
32.	Unipolar NRZ is not practical because most channels do not have continuity.
	ANS: DC
33.	In AMI, binary ones are represented by a voltage that alternates in
	ANS: polarity
34.	Long strings of should be avoided in AMI.
	ANS: zeros
35.	Manchester code has a level in the center of each bit period.
	ANS: transition
36.	Manchester coding provides information regardless of the pattern of ones and zeros.
	ANS: timing
37.	There are channels in a DS-1 frame.
	ANS: 24
38.	DS-1 uses a bit to synchronize the transmitter and receiver.
	ANS: framing
39.	In DS-1, each channel is sampled times per second.
	ANS: 8000
40.	Data is carried over a T-1 line at a rate of bits per second.

ANS: 1.544×10^{6}

41. A group of 12 DS-1 frames is called a ______.
ANS: superframe
42. From a group of twelve frames, signaling bits are "stolen" from every ______ frame.
ANS: sixth
43. ______ compression transmits all the data in the original signal but uses fewer bits to do it.

ANS: Lossless

SHORT ANSWER

1. Use Hartley's Law to find how much time it would take to send 100,000 bits over a channel with a bandwidth of 2,000 hertz and a channel constant of k = 10.

ANS: 5 seconds

2. Use the Shannon-Hartley theorem to find the bandwidth required to send 12,000 bits per second if the number of levels transmitted is 8.

ANS: 2000 hertz

3. What is the Shannon Limit of a channel that has a bandwidth of 4000 hertz and a signal-to-noise ratio of 15?

ANS: 16 kbps

4. What is the minimum required number of samples per second to digitize an analog signal with frequency components ranging from 300 hertz to 3300 hertz?

ANS: 6600 samples/second

5. What is the approximate dynamic range, in dB, of a linear PCM system that uses 12 bits per sample?

ANS: 74 dB

6. What is the approximate data rate for a system using 8 bits per sample and running at 8000 samples per second?

ANS: 64 kbps

7. If bits were "stolen" from every DS-1 frame, what would the useable data-rate be for each channel in the frame?

ANS: 56 kbps

8. Assuming maximum input and output voltages of 1 volt, what is the output voltage of a μ-law compressor if the input voltage is 0.388 volt?

ANS: 0.833 volt