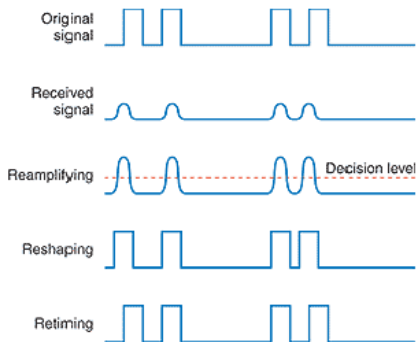


# Digital Communications

- ▶ Analog vs Digital Communication
- ▶ Pulse Code Modulation (PCM)
- ▶ Quantization
  - ▶ Uniform Quantization
  - ▶ Non-Uniform Quantization
- ▶ Quantization Error
- ▶ PCM Bandwidth
- ▶ PCM SNR

## Analog vs. Digital Communication

- ▶ Analog communication (baseband and modulated) is subject to noise.
- ▶ Pulse modulations (PAM, PWM, PPM) represent analog signals by analog variations in pulses and are also subject to noise.
- ▶ Long distance communication requires repeaters, which amplify signal and noise. Each link adds noise.
- ▶ Digital communication suppresses noise by regenerating signal.

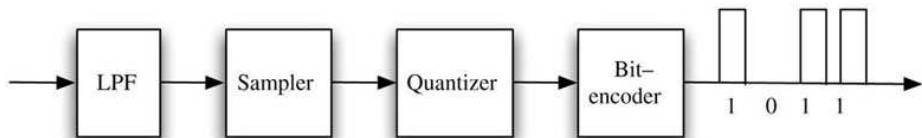


## Pulse Code Modulation (PCM)

In PCM, a signal value is represented by a sequence of pulses (digits).

Width and spacing of pulses is constant. Value of pulse is chosen from a small number of values.

Usually PCM uses only two pulse values, which represent 0 and 1.



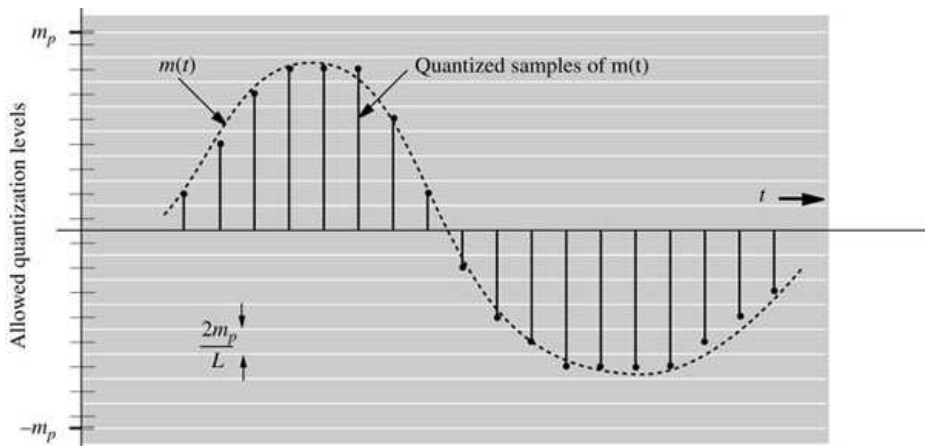
If  $m$  bits are used, then  $2^m$  signal values can be represented.

- ▶ unsigned linear:  $0, \nu, \dots, \Delta\nu(2^m - 1)$
- ▶ two's complement:  $-\Delta\nu 2^{m-1}, \dots, \Delta\nu(2^{m-1} - 1)$
- ▶ A-law and  $\mu$ -law: approximately logarithmic (more dynamic range)

## PCM and Quantization

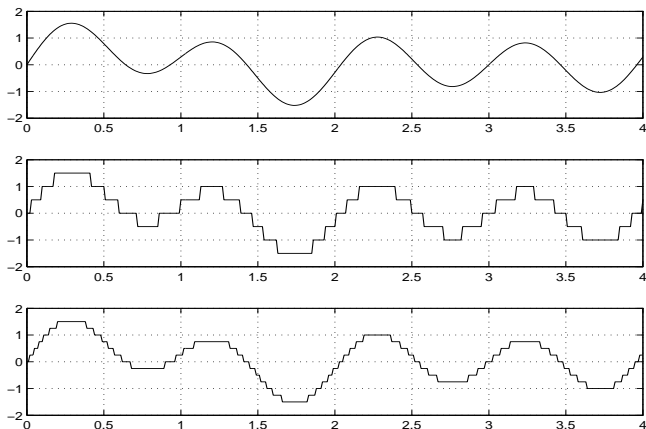
Quantization of a signal produces the closest representable value.

For fixed number of values, spacing between values increases with range.



## PCM Tradeoffs

- ▶ Signal bandwidth determines minimum sample rate
- ▶ Desired signal fidelity determines precision of reproduced signal
- ▶ Signals can be quantized using digital-to-analog converter (DAC)

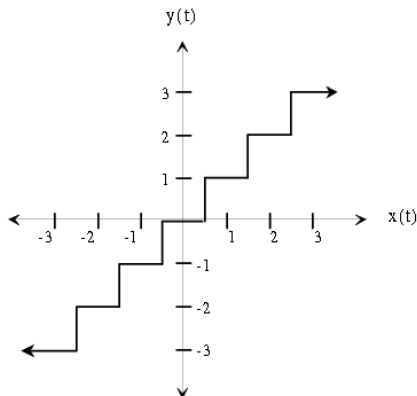


## Uniform Quantization

An ideal uniform quantizer is a nonlinear time invariant system:

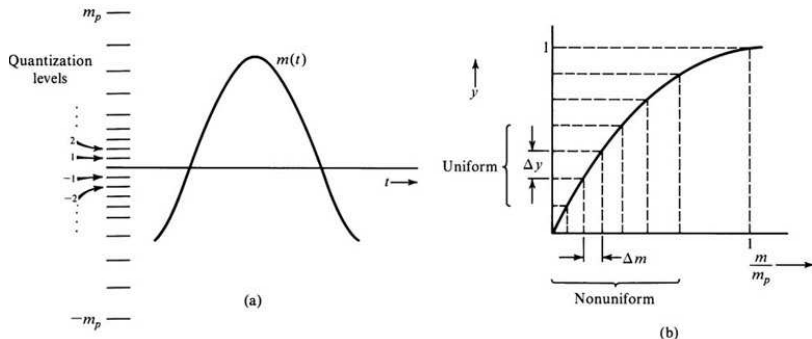
$$\tilde{g}(t) = \begin{cases} -N_l \Delta\nu & g(t) < -N_l \Delta\nu \\ n \Delta\nu & (n - \frac{1}{2}) \Delta\nu < g(t) < (n + \frac{1}{2}) \Delta\nu, -N_l < n < N_h \\ N_h \Delta\nu & g(t) > N_h \Delta\nu \end{cases}$$

$\Delta\nu$  is quantization interval.  $N_l + N_h$  is number of levels.

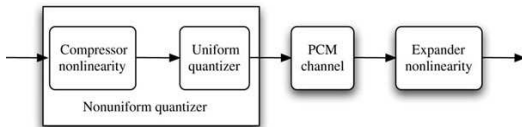


# Nonuniform Quantization

Nonuniform quantizers increase quantization intervals as magnitude of value. Interval proportional to value implies logarithmic curve.



An analog compressor (semiconductor diode) can be used.



## Nonuniform Quantization (cont.)

Logarithmic compression can be approximated by floating point, A-law, and  $\mu$ -law representations.

- ▶ Binary floating point corresponds to scientific notation:

$$\pm f \times 2^e$$

where  $f$  is *significand* or *fraction* and  $e$  is exponent. Both  $f$  and  $e$  are represented in unsigned binary.

Example: 8-bit code with one bit for sign, 4 bits for  $f$ , 3 bits for  $e$ ,

$$f = 0, 1, \dots, 15, \quad e = 0, 1, \dots, 7$$

The representable values range from  $-15 \cdot 2^7 = -1920$  to 1920.

Quantization spacing range from 1 for  $|y| < 16$  to 128 for  $|y| > 960$ .

Disadvantage of floating point is that many values have multiple representations. E.g., this scheme represents only 72 values.

Some writers use mantissa instead of significand.



## Nonuniform Quantization (cont.)

Telephone systems use ITU standardized compression formula.

- ▶  $\mu$ -law: North America and Japan. For  $\mu = 255$  (for 8-bit codes),

$$y = \operatorname{sgn}(x) \frac{1}{\ln(1 + \mu)} \ln(1 + \mu|x|), \quad (0 < x < 1)$$

- ▶ A-law: Europe, rest of world.

$$y = \begin{cases} \operatorname{sgn}(x) \frac{A|x|}{1 + \ln(A)} & |x| < \frac{1}{A} \\ \operatorname{sgn}(x) \frac{1 + \ln(A|x|)}{1 + \ln(A)} & \frac{1}{A} < |x| < 1 \end{cases}$$

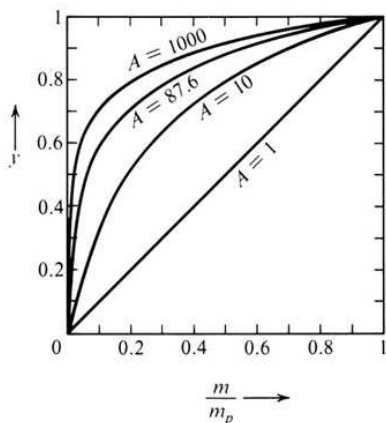
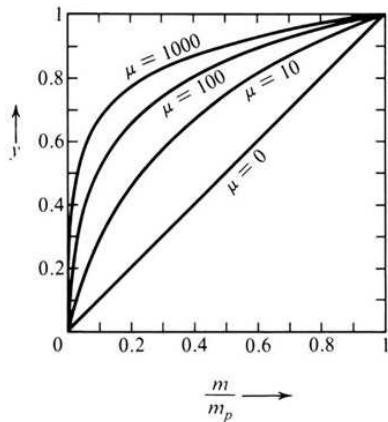
The standard value is  $A = 87.7$ .

For both laws, the input to the compressor is

$$x = \frac{m(t)}{m_p}$$

where  $-m_p \leq m(t) \leq m_p$ .

## Comparison of $\mu$ -Law and A-Law



- ▶  $\mu$ -law provides slightly larger dynamic range than A-law.
- ▶ A-law has smaller proportional distortion for small signals.
- ▶ A-law is used for international connections if at least one country uses it.

## $\mu$ -Law Implementation

Both  $\mu$ -law and A-law expanders are piecewise linear.

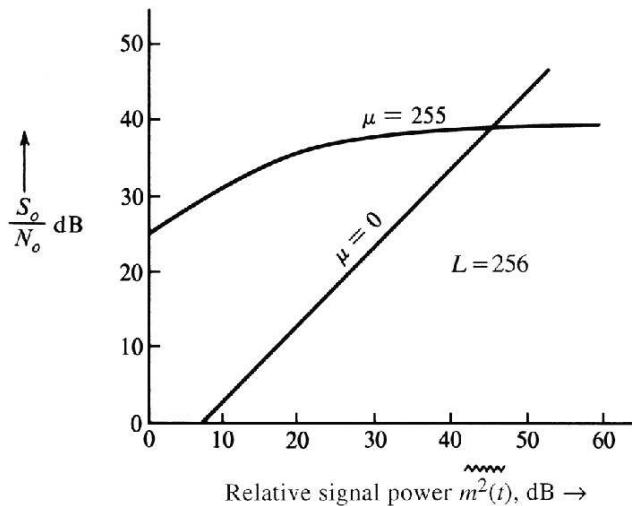
This table shows how 7 bits are expanded.

Numerical Value Range	Number Of Intervals	Interval Size	Total Intervals
0	1	1	1
1 - 16	15	2	30
17 - 32	16	4	64
33 - 48	16	8	128
49 - 64	16	16	256
65 - 80	16	32	512
81 - 96	16	64	1024
97 - 112	16	128	2048
113 - 127	16	256	4096
			<hr/>
			8159

In practice this table is used by the  $\mu$ -law encoder.

## $\mu$ -Law Signal-to-Noise Ratio

The average power of a compressed signal is closer to the peak power.



## Quantization Error

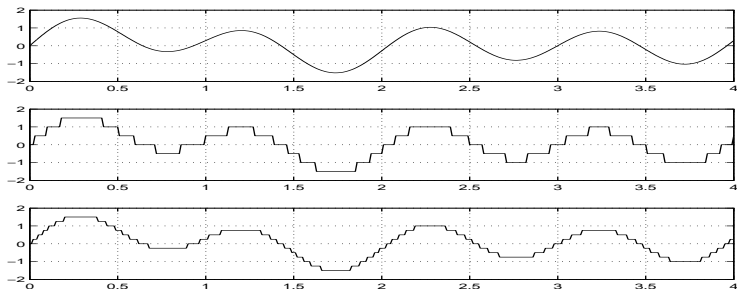
Uniform quantization with  $L$  levels of a signal with peak amplitude  $m_p$  has maximum quantization error

$$\max \text{ error} = \frac{m_p}{L},$$

and mean square error

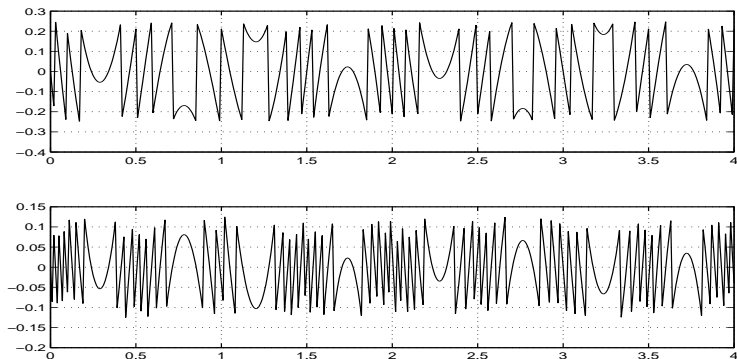
$$\text{average square error} = \frac{m_p^2}{3L^2}$$

Example: signal quantized to 4 and 16 levels.



## Quantization Error (cont.)

Quantization error for quantizing to 4 and 16 levels.



Power of quantization error for above example is

$$0.0863 \approx \frac{1}{3} 2^{-2} \quad (L = 4) \quad \text{and} \quad 0.0180 \approx \frac{1}{3} 2^{-4} \quad (L = 16)$$

Fact: for  $\mu$ -law, SNR is  $\frac{3L^2}{(\ln(1 + \mu))^2}$  if  $\mu \gg m_p/\text{rms}(m(t))$ .

## Bandwidth vs. Quantization Error

What bandwidth is needed to transmit a PCM encoded signal?

Example: suppose that we want maximum error  $0.5\% m_p$  for a 3 kHz signal.

$$\frac{\Delta\nu}{2} = \frac{m_p}{L} = \frac{0.5}{100} m_p \implies L = 200 < 2^8$$

At Nyquist sample rate

$$R_N = 2 \cdot 3000 = 6000 \text{ Hz}$$

we need  $6000 \cdot 8 = 48000$  bits/sec.

Fact: a bandlimited signal can convey two symbols per Hz.

For binary PCM, we need  $48000/2 = 24000$  Hz.

For practical reasons, we sample faster than the Nyquist rate.

E.g., at rate 4000 Hz, the required bandwidth is 32 kHz.

## PCM SNR

The signal-to-noise ratio is

$$SNR = \frac{\text{average signal power}}{\text{average noise power}}$$

For uniform quantization noise,

$$\text{average signal power} \approx am_p^2 \quad (a \approx \frac{1}{2})$$

$$\text{quantization error} \approx \frac{1}{3}(m_p/L)^2$$

$$SNR \approx cL^2 = c2^{2m}$$

where  $m$  is the number of bits in the PCM sample, so  $L = 2^m$ .  $c$  is a constant.

SNR grows exponentially with the number of bits.

If we measure SNR in dB,

$$SNR_{dB} = 10 \log_{10}(c2^{2m}) = 10 \log_{10}(c) + 2m \log_{10}2 = (\alpha + 6m) \text{dB}$$

where  $\alpha = 10 \log_{10} c$ .

Increasing  $n$  by one bit improves SNR by 6 dB! One bit quadruples SNR.



## PCM SNR

Consider two cases for a 4 kHz bandwidth signal

- ▶  $L = 64$ ,  $m = 6$  bits

$$SNR_{dB} = \alpha + 36 \text{ dB}$$

- ▶  $L = 256$ , and  $m = 8$  bits

$$SNR_{dB} = \alpha + 48 \text{ dB}$$

We've gained 12 dB in SNR. However, the PCM bandwidth has increased only from

$$(2 \times 4 \text{ kHz})(6 \text{ bits})/2 = 24 \text{ kbits/sec}$$

to

$$(2 \times 4 \text{ kHz})(8 \text{ bits})/2 = 32 \text{ kbits/sec}$$

We only need 1/3 greater bandwidth for a 12 dB improvements in SNR. The value of  $\alpha$  (and  $c$ ) depend on the quantization method, but are constants given that.

## Logarithmic Units

In communications we often measure ratios using logarithms.

The *bel* (B) is the  $\log_{10}$  of a ratio. More useful is the *decibel* (dB):

$$\frac{a}{b} \text{ in dB is } 10 \log_{10} \frac{a}{b}$$

Examples:  $2 \Leftrightarrow 3.01 \approx 3$  dB,  $5 \Leftrightarrow 4.77 \approx 5$  dB

Why measure in dB?

- ▶ Some sensors (human eyes, ears) respond to logarithm of signal power.
- ▶ Many transmission media have attenuation that is exponential in length. Thus the signal loss in dB is proportional to length.
- ▶ Calculating how much power is needed in a communications system requires a *link budget*, which is additive in dB.

$$\text{rcv power (dBm)} = \text{xmit power (dBm)} + \text{gains (dB)} - \text{losses (dB)}$$

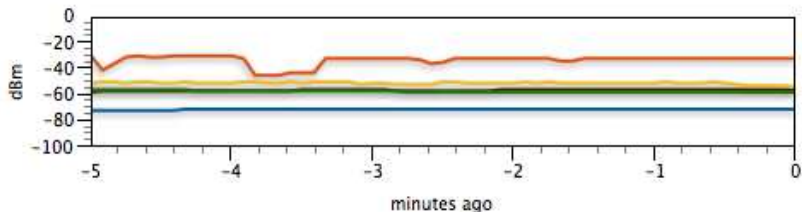
- ▶ Since dB measures ratio, we must specify a reference value for 0 dB.
  - ▶ dBW: 0 dB = 1 W
  - ▶ dBm: 0 dB = 1 mW

## Audio Volume (Sound Pressure)

- ▶ The SI unit of pressure is the pascal (Pa):  $1 \text{ N/m}^2$
- ▶ Atmospheric pressure at sea level is  $14.7 \text{ lb/in}^2$  or  $101325 \text{ Pa}$
- ▶ Audio reference level:  $20 \mu\text{Pa}$ , threshold of human hearing. Another reference level is  $1 \text{ pW} = 10^{-12} \text{ W}$ .
- ▶ Example audio levels (rms of sound pressure):
  - ▶ 20–30 dB: quiet room
  - ▶ 60 dB: TV at normal volume
  - ▶ 85 dB: hearing damage (long term)
  - ▶ 100 dB: jack hammer
  - ▶ 140 dB: aircraft carrier deck
  - ▶ 120 dB: vuvuzuela, thunderclap, chain saw
  - ▶ 175 dB: stun grenade
  - ▶ 194 dB: atmospheric pressure (shock wave)

# Wireless Receive Power and SNR

Data from John Gill.



Client	Signal	Noise	Rate	Type
D8:30:62:5A:79:1C	-32	-79	104	802.11b/g/n
B8:F6:B1:19:51:63	-54	-79	216	802.11b/g/n
4C:B1:99:AE:C4:89	-58	-79	65	802.11b/g/n
9C:20:7B:92:C3:02	-71	-79	65	802.11b/g/n
78:E4:00:BE:AF:B3	-56	-79	54	802.11b/g

Receive power ranges from  $-32$  to  $-71$  dB ( $6.3 \times 10^{-7}$  to  $7.9 \times 10^{-11}$  W).

The most powerful transmitter was 4 feet from the receiver.