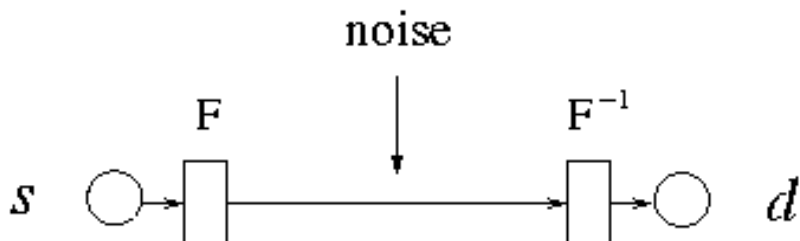


Information Transmission under Noise



Uncertainty introduced by noise:

- encoding/decoding: $a \mapsto w_a \mapsto w \mapsto [?]$
- w_a gets corrupted, i.e., becomes w
- if $w = w_b$, incorrectly conclude b as symbol
- detect w is corrupted: error detection
- correct w to w_a : error correction

Would like: If received code word $w = w_c$ for some symbol $c \in \Sigma$, then probability that actual symbol sent is indeed c is high.

$$\longrightarrow \Pr\{\text{symbol sent} = c \mid w = w_c\} \approx 1$$

\longrightarrow noiseless channel: special case (prob = 1)

In practice, w may not match any legal code word:

\longrightarrow for all $c \in \Sigma$, $w \neq w_c$

\longrightarrow then what?

Fundamental limitation to reliable data transmission:

Channel capacity C : maximum achievable reliable data transmission rate (bps) over a noisy channel (dB) with bandwidth W (Hz).

Channel Coding Theorem (Shannon): Given bandwidth W , signal power P_S , noise power P_N , channel subject to white noise,

$$C = W \log \left(1 + \frac{P_S}{P_N} \right) \text{ bps.}$$

P_S/P_N : signal-to-noise ratio (SNR)

- upper bound achieved by using longer codes
- detailed set-up/conditions omitted

Increasingly important for modern day networking:

- Power control (e.g., pocket PCs)
 - trade-off w.r.t. battery power
 - trade-off w.r.t. multi-user interference
 - signal-to-interference ratio (SIR)
- Recent trend: software radio
 - hardware-to-software migration
 - configurable

Signal-to-noise ratio (SNR) is expressed as

$$\text{dB} = 10 \log_{10}(P_S/P_N).$$

Example: Assuming a decibel level of 30, what is the channel capacity of a telephone line?

Answer: First, $W = 3000$ Hz, $P_S/P_N = 1000$. Using Channel Coding Theorem,

$$C = 3000 \log 1001 \approx 30 \text{ kbps.}$$

- compare against 28.8 kbps modems
- what about 56 kbps modems?
- DSL lines?

Digital vs. Analog Transmission

Two forms of *transmission*:

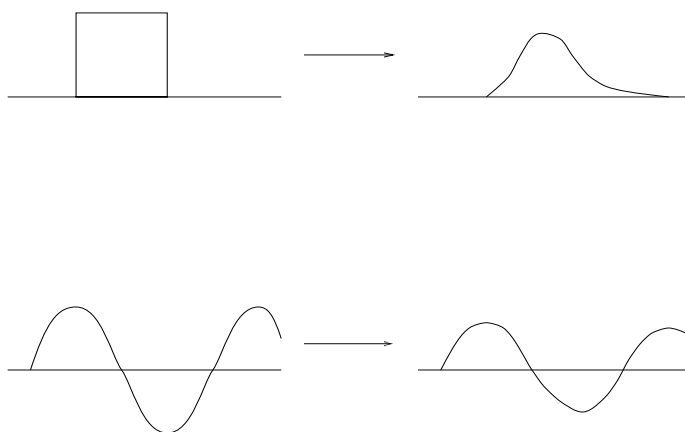
- digital transmission: data transmission using square waves
- analog transmission: data transmission using all other waves

Four possibilities to consider:

- analog data via analog transmission
→ “as is” (e.g., radio)
- analog data via digital transmission
→ sampling (e.g., voice, audio, video)
- digital data via analog transmission
→ broadband & wireless (“high-speed networks”)
- digital data via digital transmission
→ baseband (e.g., Ethernet)

Why consider digital transmission?

Common to both: problem of *attenuation*.



- decrease in signal strength as a function of distance
- increase in attenuation as a function of frequency

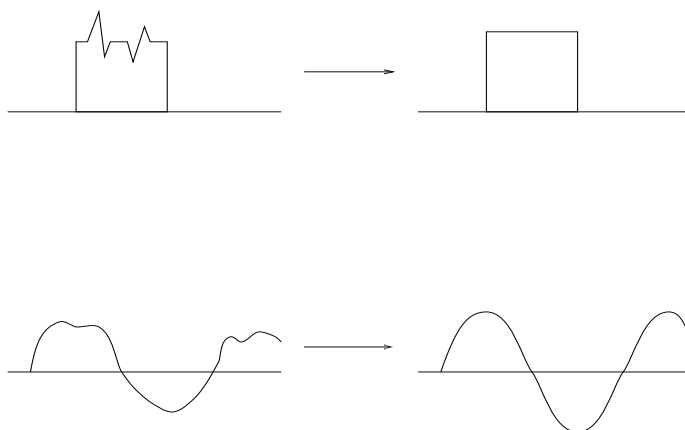
Rejuvenation of signal via amplifiers (analog) and repeaters (digital).

Delay distortion: different frequency components travel at different speeds.

Most problematic: effect of noise

→ thermal, interference, ...

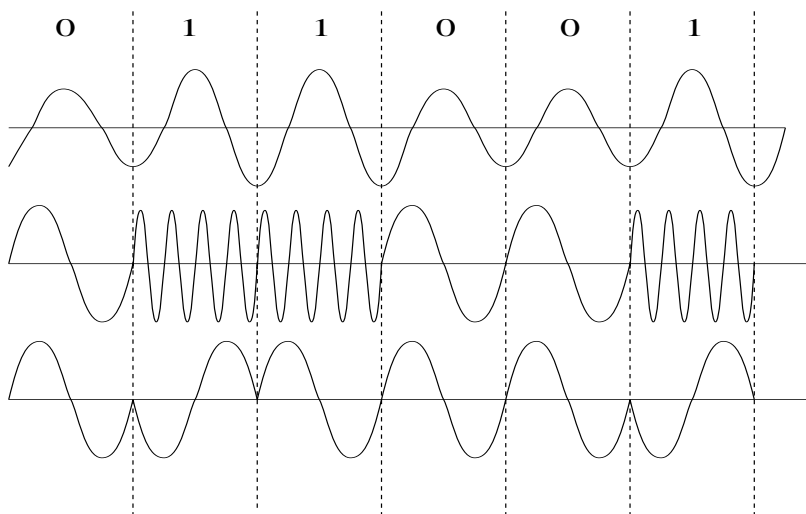
- Analog: Amplification also amplifies noise—filtering out just noise, in general, is a complex problem.
- Digital: Repeater just generates a new square wave; more resilient against ambiguity.



Analog Transmission of Digital Data

Three pieces of information to manipulate: amplitude, frequency, phase.

- Amplitude modulation (AM): encode bits using amplitude levels.
- Frequency modulation (FM): encode bits using frequency differences.
- Phase modulation (PM): encode bits using phase shifts.



FM radio uses ... FM!

AM radio uses ... AM!

iPod & radio experiment uses ... ?

Why is FM radio clearer (“high fidelity”) than AM radio?

Broadband uses ... ?

Baud Rate vs. Bit Rate

Baud rate: Unit of time within which carrier wave can be altered for AM, FM, or PM.

→ signalling rate

→ e.g., clock

Not synonymous with bit rate: e.g., AM with 8 levels, PM with 8 phases

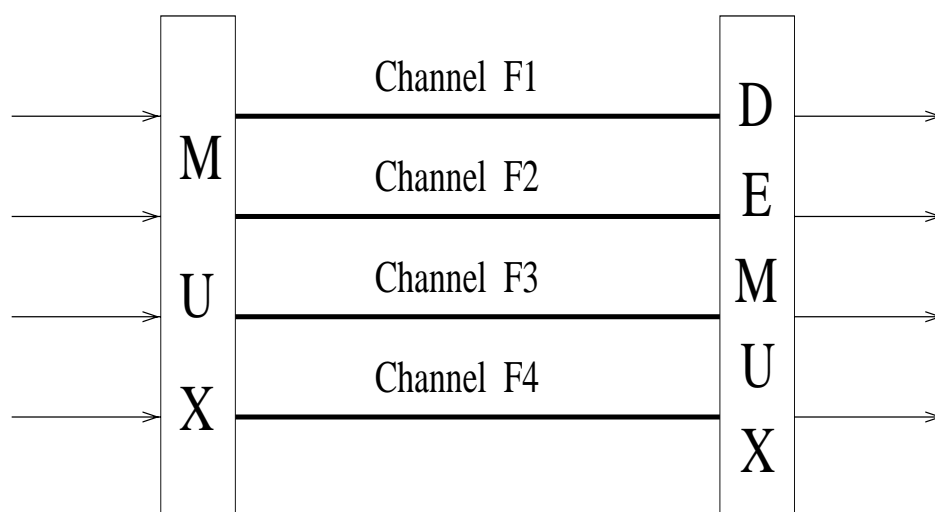
→ bit rate (bps) = 3 × baud rate

... less than one bit per baud?

Broadband vs. Baseband

Presence or absence of carrier wave: allows many channels to co-exist at the same time

→ frequency division multiplexing (FDM)



Ex.: AM radio (535 kHz–1705 kHz)

→ tuning to specific frequency: Fourier transform

→ coefficient (magnitude) carries bit information

Ex.: FM radio

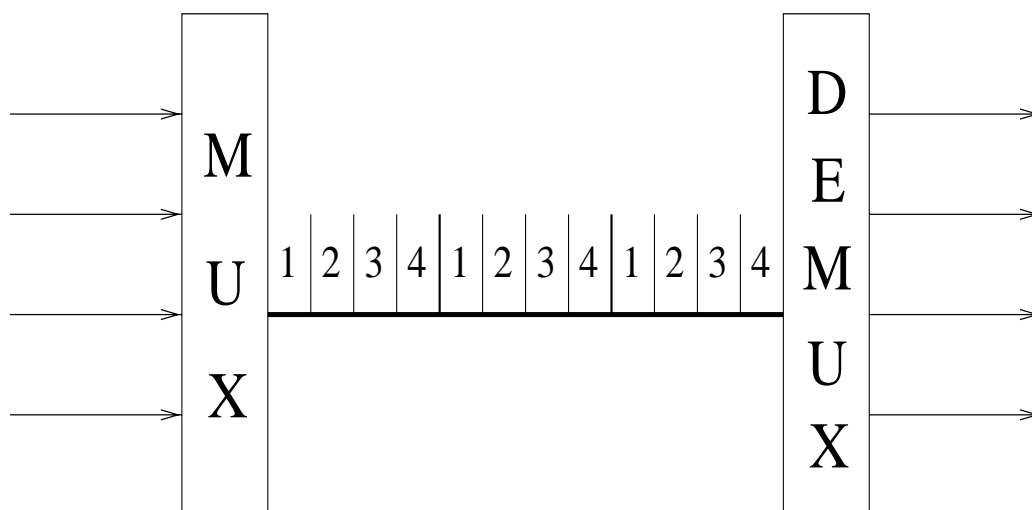
- 88 MHz–108 MHz
- 200 kHz slices
- how does it work?
- better or worse than AM?

Ex.: Digital radio

- digital audio radio service
- GEO satellites (a.k.a. satellite radio)
- uses 2.3 GHz spectrum (a.k.a. S-band)
- e.g., XM, Sirius

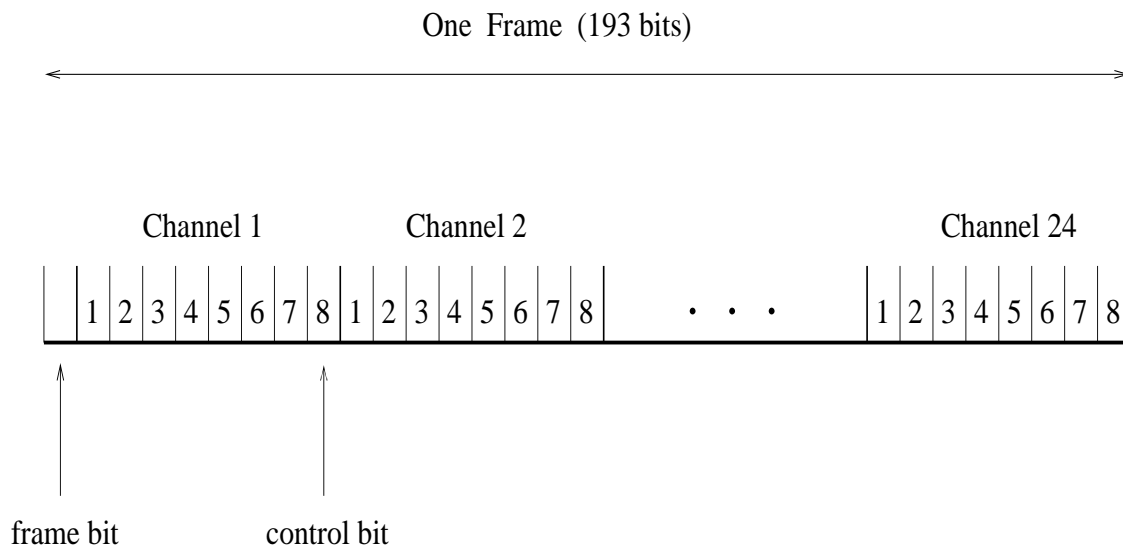
In the absence of carrier wave, can still use multiplexing:

→ time-division multiplexing (TDM)



- digital transmission of analog data
 - first digitize
 - PCM (e.g., PC sound cards), modem
- digital transmission of digital data
 - e.g., telephony backbone network

Example: T1 carrier (1.544 Mbps)



- 24 simultaneous users
- 7 bit quantization

Assuming 4 kHz telephone channel bandwidth, Sampling Theorem dictates 8000 samples per second.

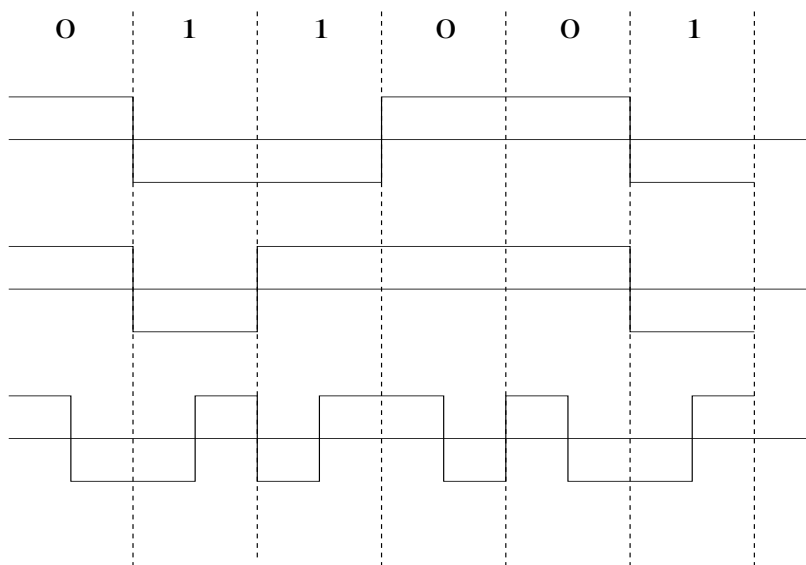
→ 125 μ sec inter-sample interval

Bandwidth = $8000 \times 193 = 1.544$ Mbps

Digital transmission of digital data

Direct encoding of square waves using voltage differentials; e.g., -15V–+15V for RS-232-C.

- NRZ-L (non-return to zero, level)
- NRZI (NRZ invert on ones)
- Manchester (biphase or self-clocking codes)



→ baud rate vs. bit rate of Manchester?

Trade-offs:

- NRZ codes—long sequences of 0's (or 1's) causes synchronization problem; need extra control line (clock) or sensitive signalling equipment.
- Manchester codes—synchronization achieved through self-clocking; however, achieves only 50% efficiency vis-à-vis NRZ codes.

4B/5B code

Encode 4 bits of data using 5 bit code where the code word has at most one leading 0 and two trailing 0's.

0000 \leftrightarrow 11110, 0001 \leftrightarrow 01001, etc.

→ at most three consecutive 0's

→ efficiency: 80%

Multiplexing techniques:

- TDM
- FDM
- mixture (FDM + TDM); e.g., TDMA
- CDMA (code division multiple access) or spread spectrum
 - wireless communication
 - competing scheme with TDMA