

CS 436

Episode 08

The Physical Layer

—

Data Encoding

Now on reserve in the library — one copy each of

“Business Data Communications” by Stallings (UWD 1609)

“Computer Networking” by Kurose & Ross (UWD 1443)

Reading from “Business Data Communications”

Chapter 8: Data Communications Fundamentals

Sections 10.1-10.4 Transmission Efficiency

Optional Reading

Cable — it’s not just for TV, IEEE Spectrum, May 1999

A Second Wind for Wiring, IEEE Spectrum, Sept 1999

Midterm scheduling survey: Administrivia → Schedule Survey

Goals for this episode: to understand

how bits are physically represented and transmitted (cont.)

esp when using analog media

multiplexing multiple signals into a single channel

Rogers and Bell ADSL at the physical layer

TA office hours

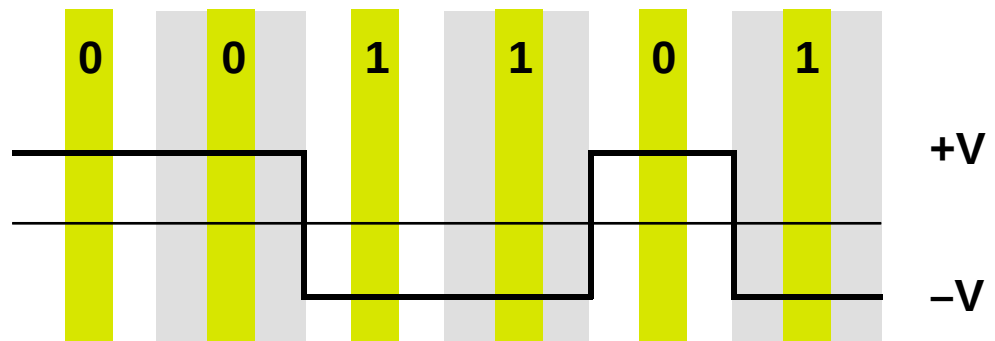
Bright: 2-4 pm Fridays in DC 3321 (the Networking Lab)

Wayne: 3-4 Tuesdays in DC 3511

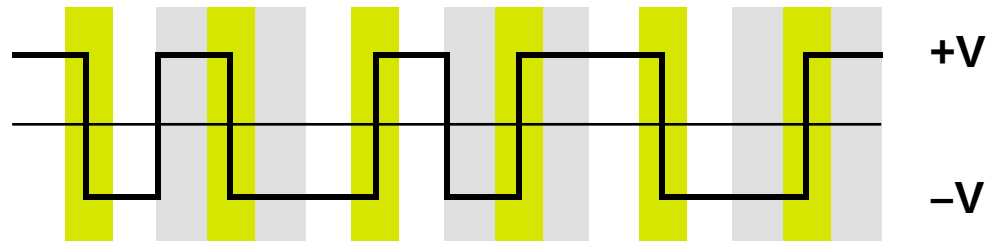
Digital Encoding of Digital Data

08

(a) NRZ-L



(c) Manchester



 bit sampling times

Non Return to Zero Level (NRZL)

common inside PC's

easy to engineer

can loose sync over long runs of 0 or 1 on a data link

Manchester

used in ethernet

always a transition at bit sampling time

— effectively a clock signal

down for 0

up for 1

transition as needed at beginning of bit interval

See text for others

Pulse Code Modulation (PCM)

periodically sample the analog signal

sampling rate $\geq$ twice max frequency in signal

EG voice data

phone network transmits below 4,000 Hz

so 8,000 samples per second, 8 bits / sample

yields 64 Kbps of data

(actually done — telephone trunks are mostly digital now)

EG a CD

44,100 samples per second

16-bit samples (65,536 levels) = two bytes / sample

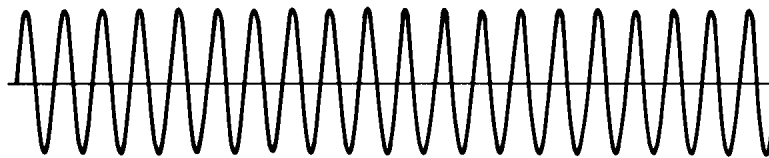
two channels for stereo

$2 \times 2 \times 44,100 = 176 \text{ KB/sec}$ — a 1x CD drive

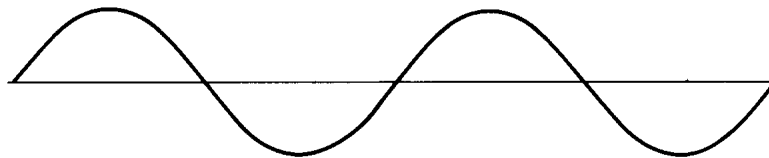
Modulating

*a relatively high-frequency “carrier signal”
by a relatively low-frequency “data signal”*

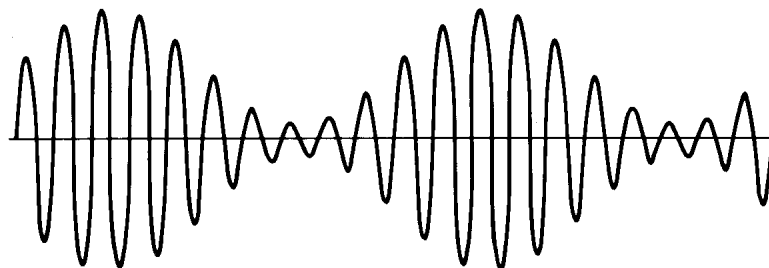
The signals

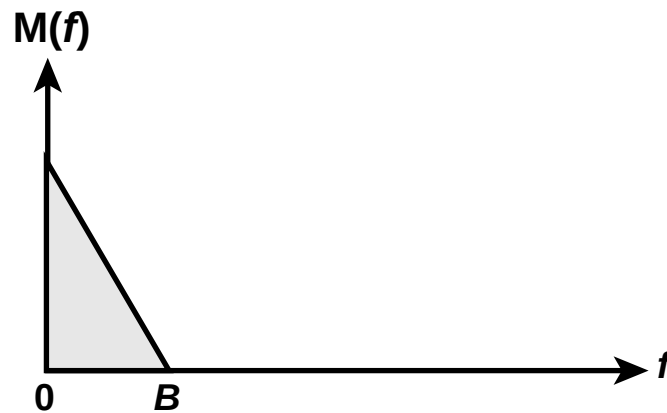


Carrier



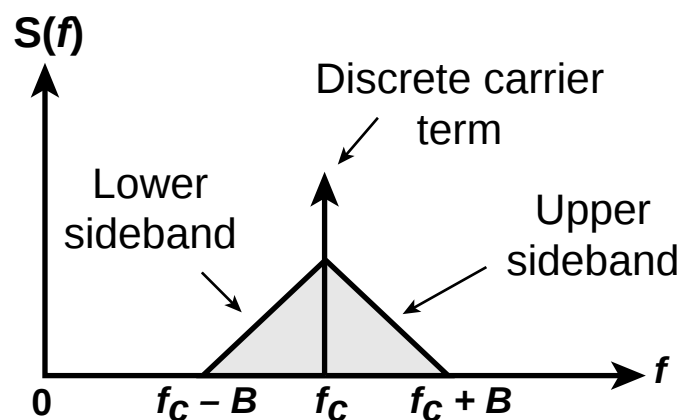
Data

 $(1 + \text{Data}) \times \text{Carrier}$



Spectrum of modulating signal

is used to modulate the carrier f_c , we get

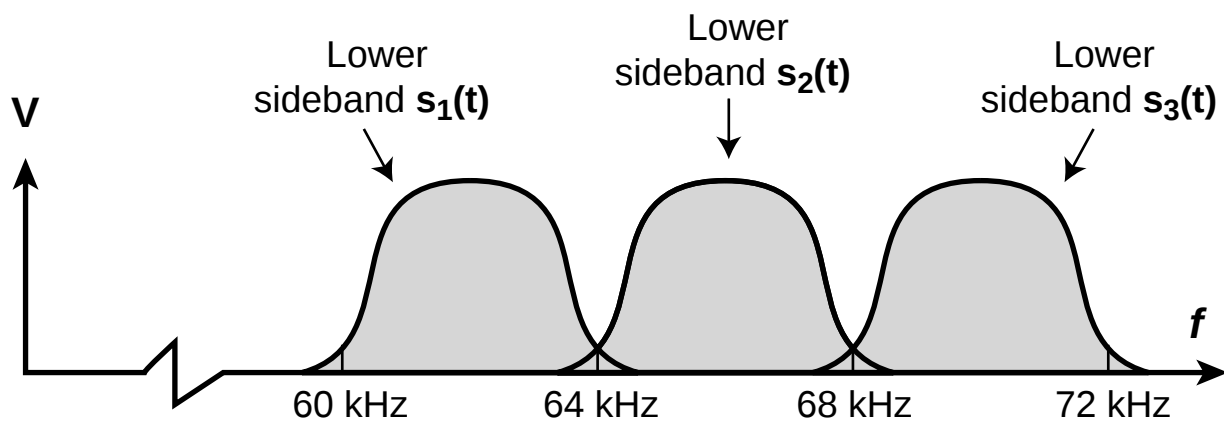
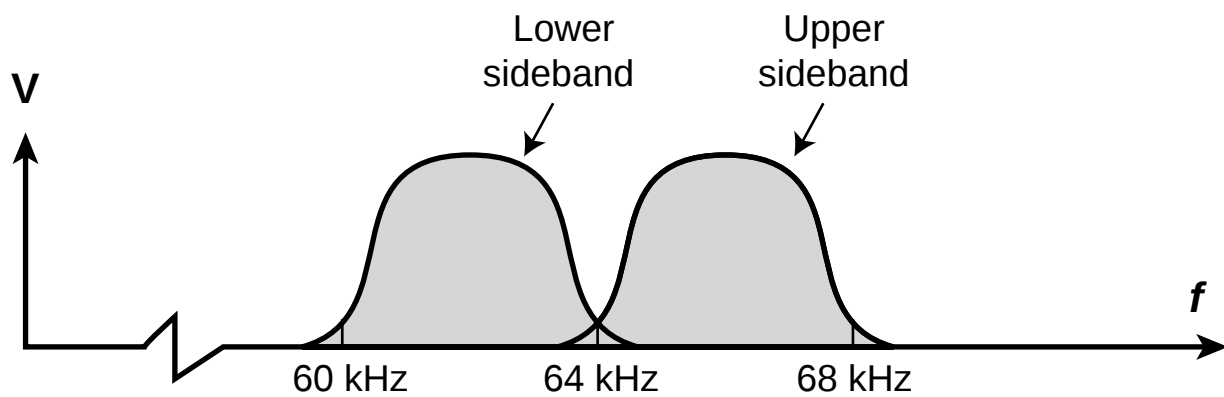
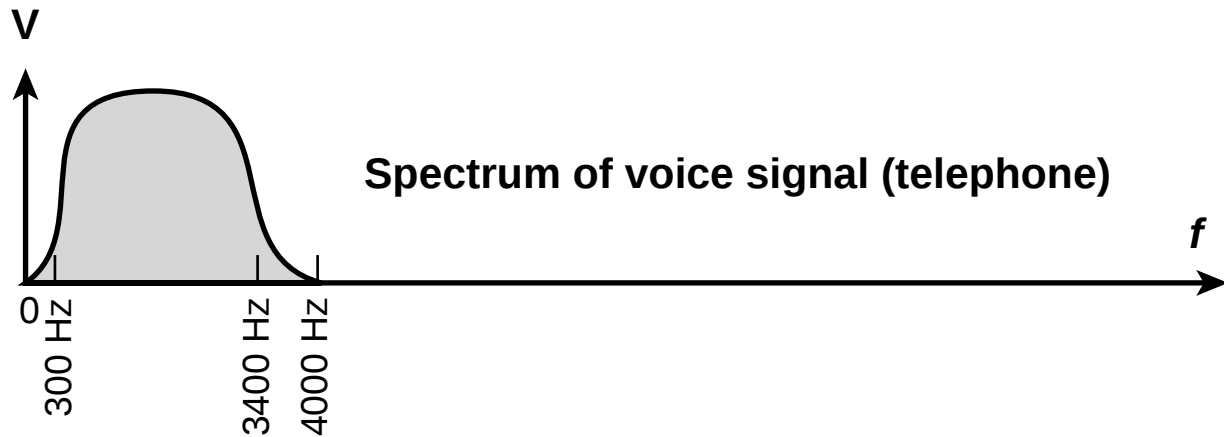


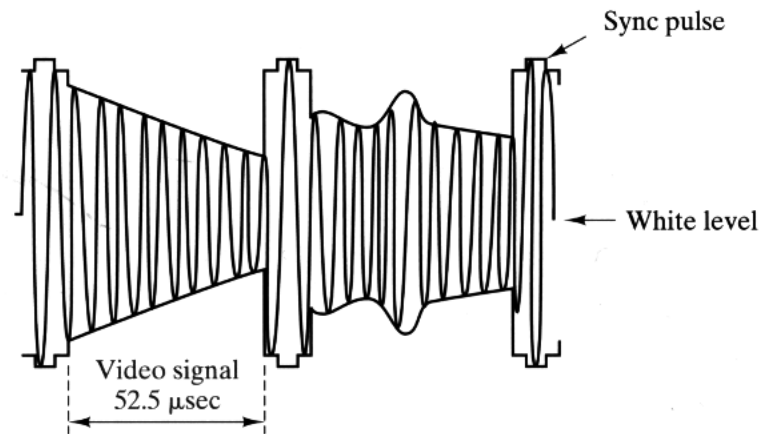
Spectrum of AM signal with carrier at f_c

Qualitatively the same story for frequency & phase modulation

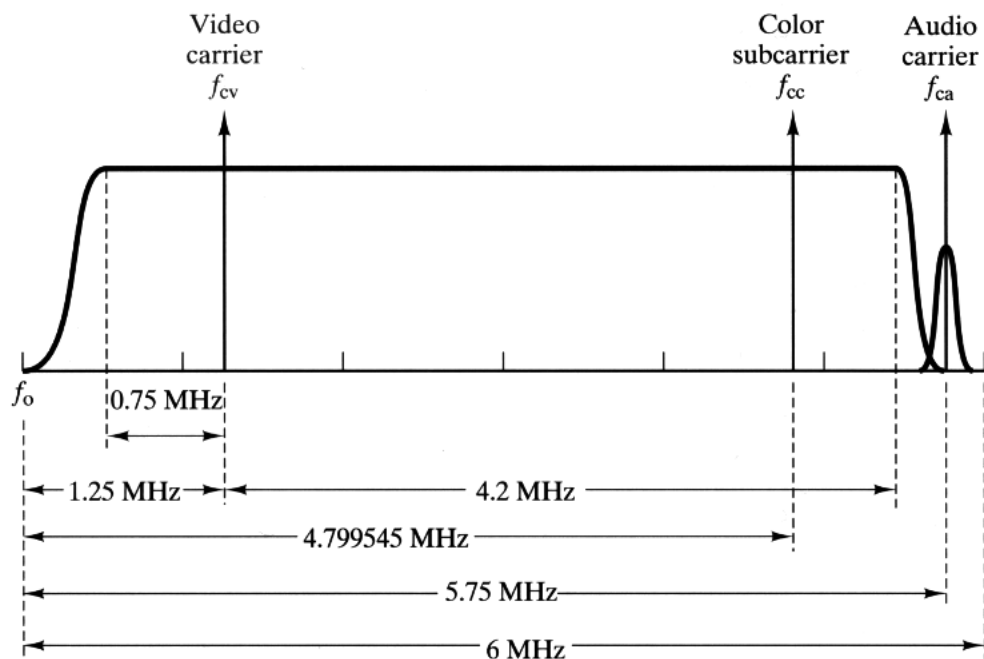
Why do this?

perhaps you simply can't transmit at the base frequency
perhaps to share a single channel among several signals
 eg transmit multiple telephone calls simultaneously
 eg AM radio, FM radio, television
 — “frequency division multiplexing” (FDM)





(a) Amplitude modulation with video signal



Traditional cable TV

“thick” coax cable

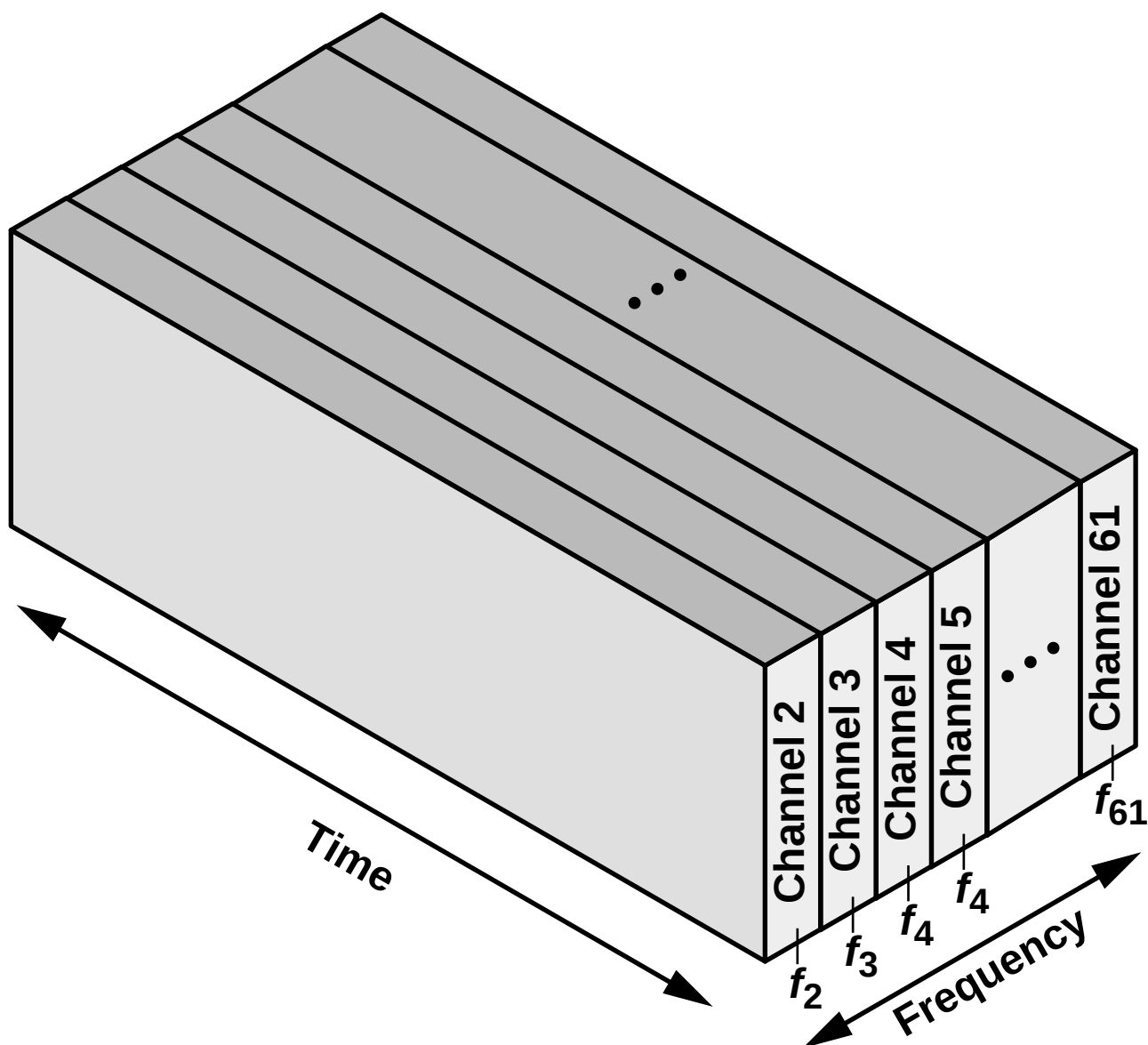
roughly 400 (500?) MHz of bandwidth

one channel requires about 6 MHz of bandwidth

so you have room for roughly 60 (75?) channels

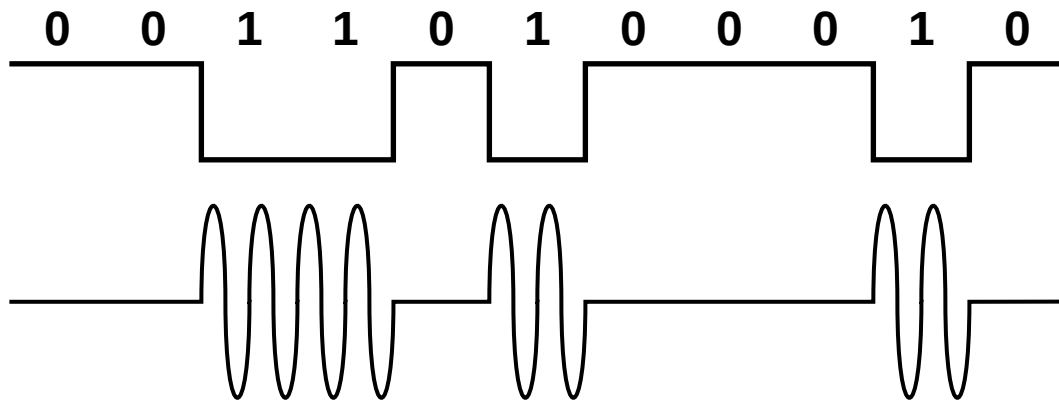
starting at 54 MHz (channel 2)

[Rogers offers channels 2 through 78]

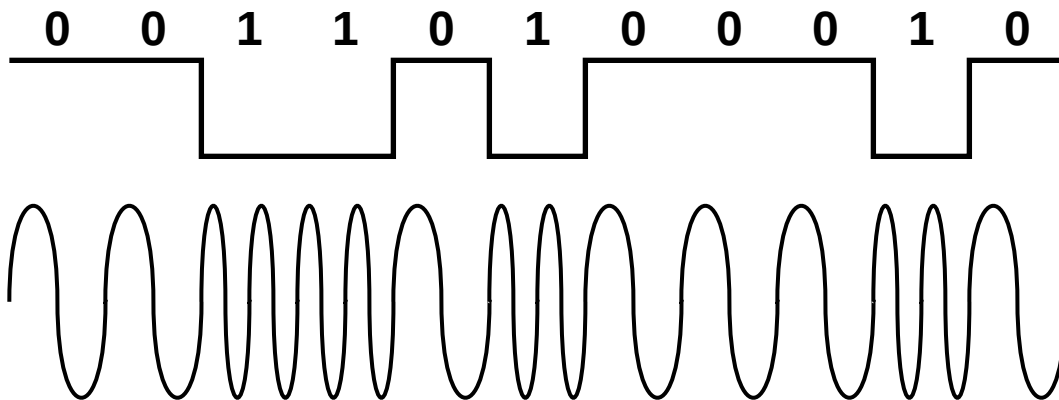


Cable TV frequency allocation

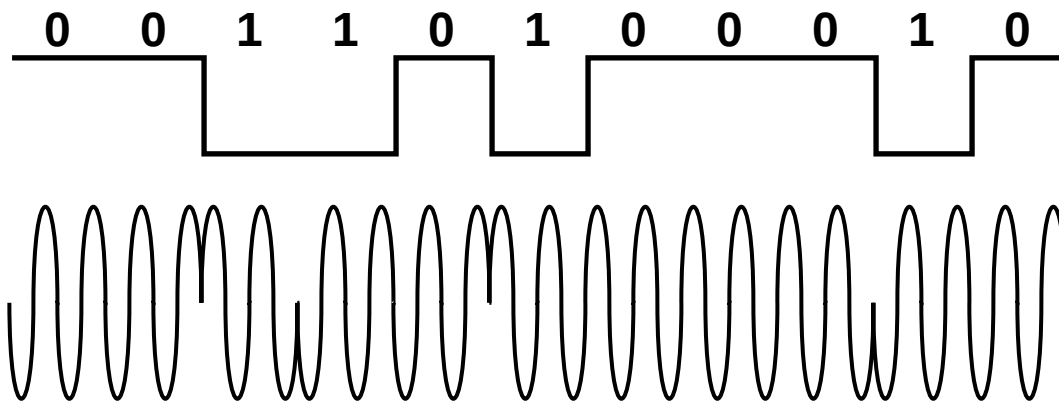
<i>Channel</i>	<i>Frequency Band</i>
2	54 - 60 MHz
3	60 - 66 MHz
...	
61	444 - 450 MHz



Amplitude-shift keying



Frequency-shift keying



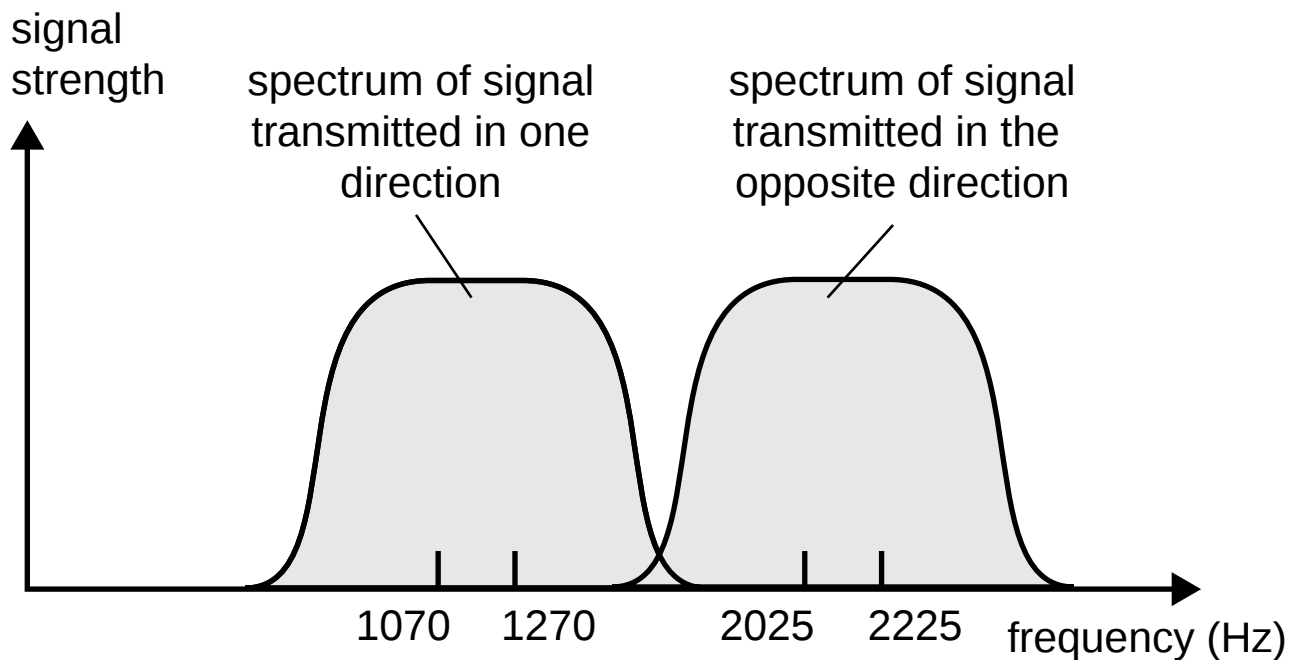
Phase-shift keying

Voice modems

designed to transmit 300–3,600 Hz
predated computer communications

Bell 108 (103?) modems (300 bps?)

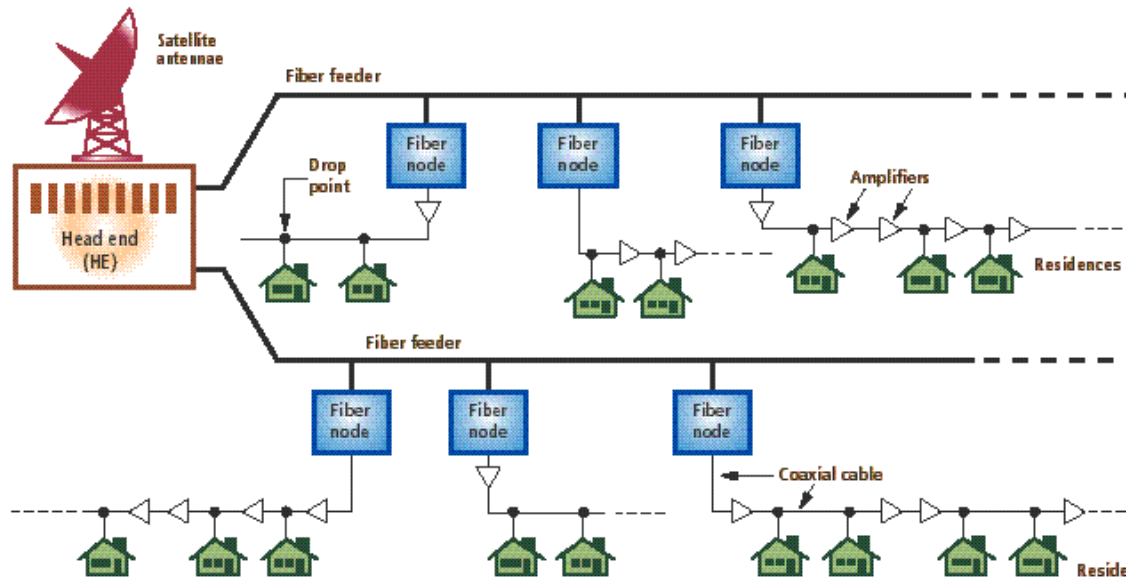
frequency shift encoding
1170 Hz $\pm$ 100 Hz for 0 and 1 in one direction
2125 Hz $\pm$ 100 Hz for 0 and 1 in the other direction



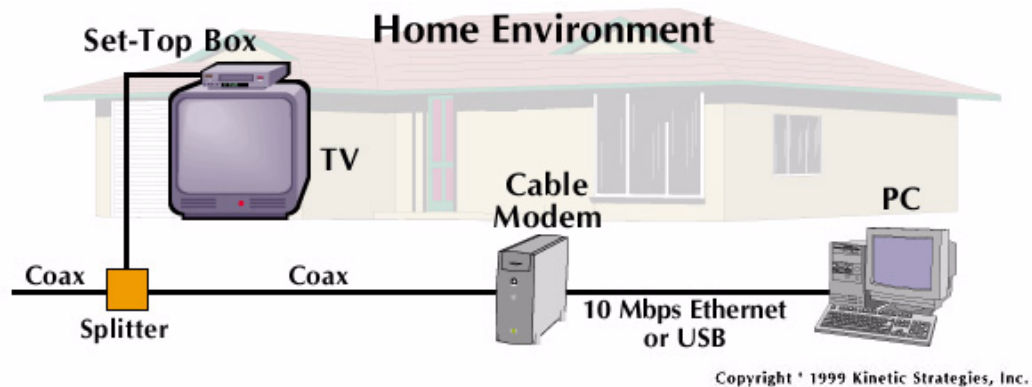
9600 bps is achieved by combining phase and amplitude encoding
data is processed 4 bits at a time
phase changes of 0°, 90°, 180°, 270° at either of 2 amplitudes
phase changes of 45°, 135°, 225°, 315° at either of 2 amplitudes
“Quadrature Amplitude Modulation” (QAM)

Higher speed modems use something called “Trellis coding”

Hybrid fibre-coax distribution¹



At each house²



1. IEEE Spectrum, May 1999, *Cable—It's not just for TV*

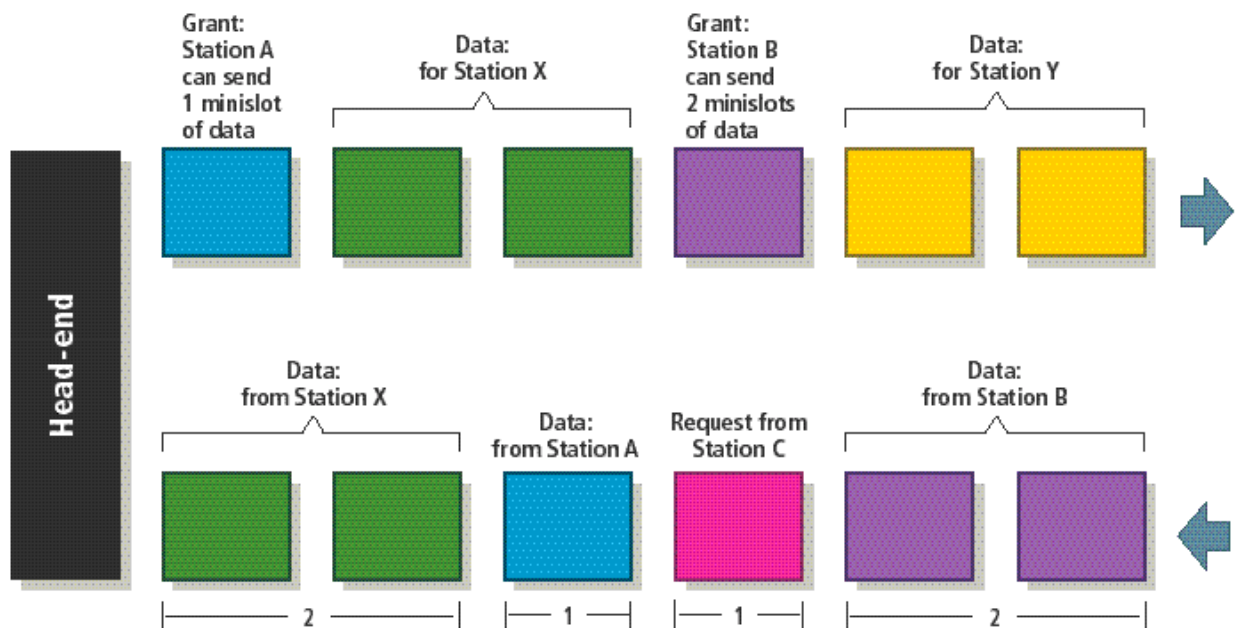
2. <http://www.cabledatacomnews.com/cm/cmic1.html>

One channel is reserved for “downstream” (incoming) data
at speeds up to 30 Mbps using 64 QAM
ie 6 bits of data per signal change (baud)
encoded by various combinations of amplitude and phase

“Upstream” (outgoing) data
in a separate frequency band
somewhere between 4 and 42 MHz
at speeds between 500 Kbps and 10 Mbps
using QPSK (phase shift keying)

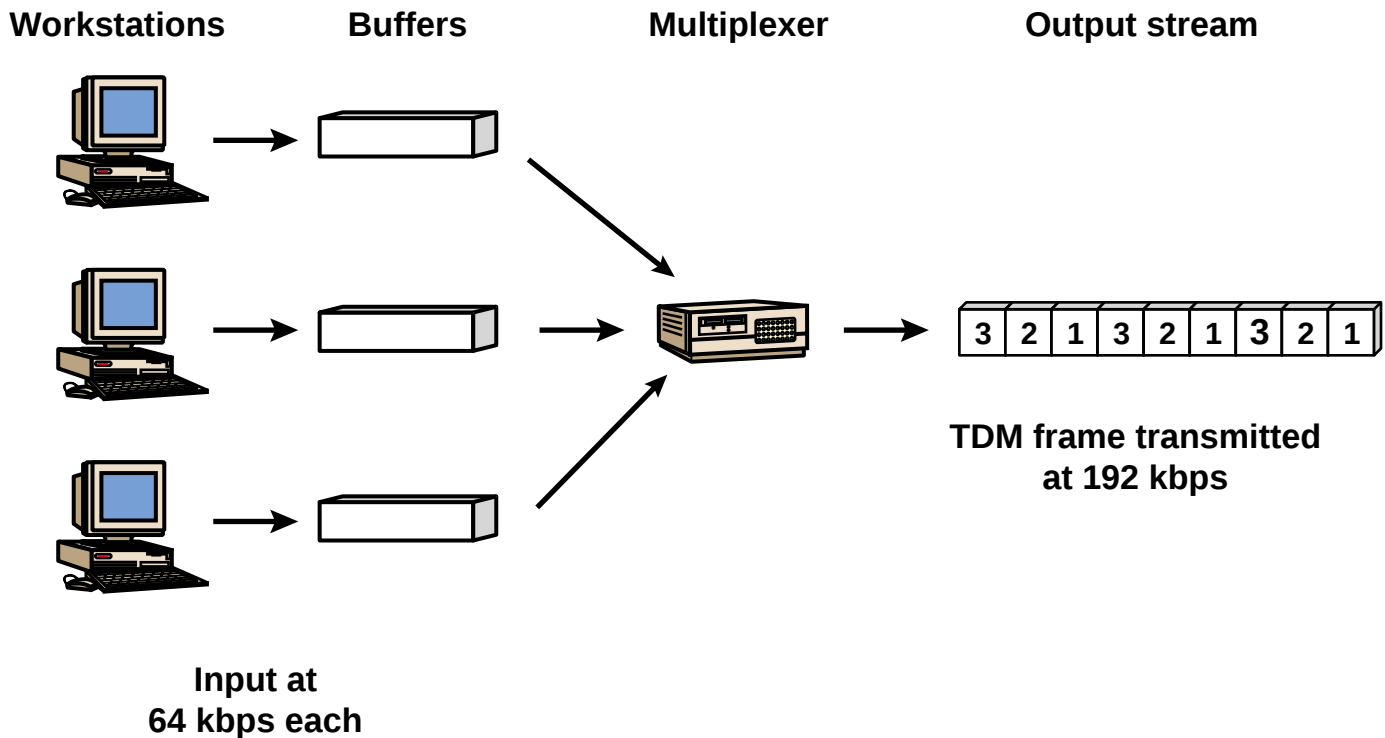
Headend is responsible for multiplexing / demultiplexing
internet traffic
cable TV
internet telephony (not much used ... yet)

Access is shared using “Time Division Multiplexing” (TDM)



“Pure” TDM

each node takes its turn in a regular, repeating sequence



eg telephone trunk lines

*the DS-? hierarchy of standard data rates / frame structure
which multiplex “circuit-switched” phone calls*

Possible inefficiency

*one workstation may have nothing to transmit
but a slot / frequency band is reserved for it anyway*

Possible solutions

statistical multiplexing

CSMA/CD

“Carrier Sense Multiple Access with Collision Detection”

need for buffering, danger of overflow

adds overhead of identifying whose data is whose

Although the cable modems we currently use are capable of speeds up to 10 Mbps we have throttled the cable modems back to 3 Mbps downstream (to the home) and 400 Kbps upstream (to the Internet). This allows us to maintain a more consistent level of network performance.

Chris W., Rogers@Home Electronic Support Group, Mar 99

Everyone has a phone

*it only uses 4 KHz of bandwidth for voice (0 - 4000 Hz)
but the wire has ~ 1 MHz of usable bandwidth*

So the phone company

*multiplexes/demultiplexes voice and data on your phone line
at the central office
and rents you a modem that does the same thing
at your house*

Downstream

*occupies the 250 KHz to 1 MHz frequency band
1.5 - 9 Mbps, depending on line quality
up to 5.5 km*

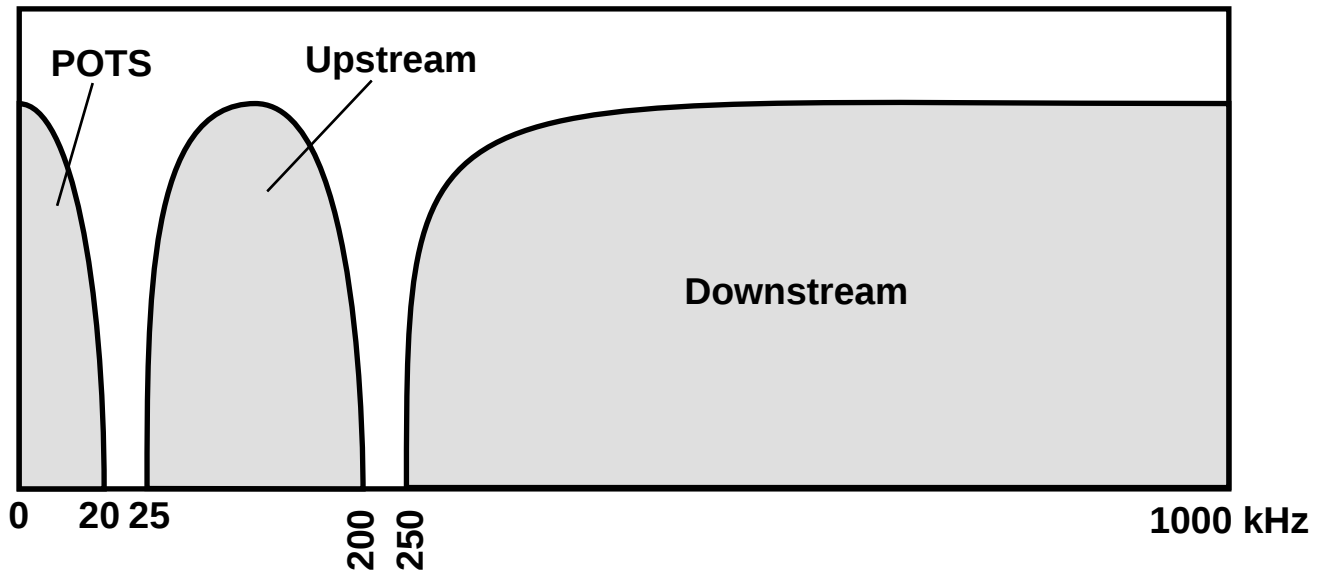
Upstream

*occupies the 25 KHz to 200 KHz frequency band
16 Kbps to 1 Mbps*

FDM is used WITHIN each stream: “discrete multitone”

*subdivide 250 KHz - 1 MHz into 4 KHz subchannels
test on initialization, decide bit rate per subchannel (0-60 Kbps)*

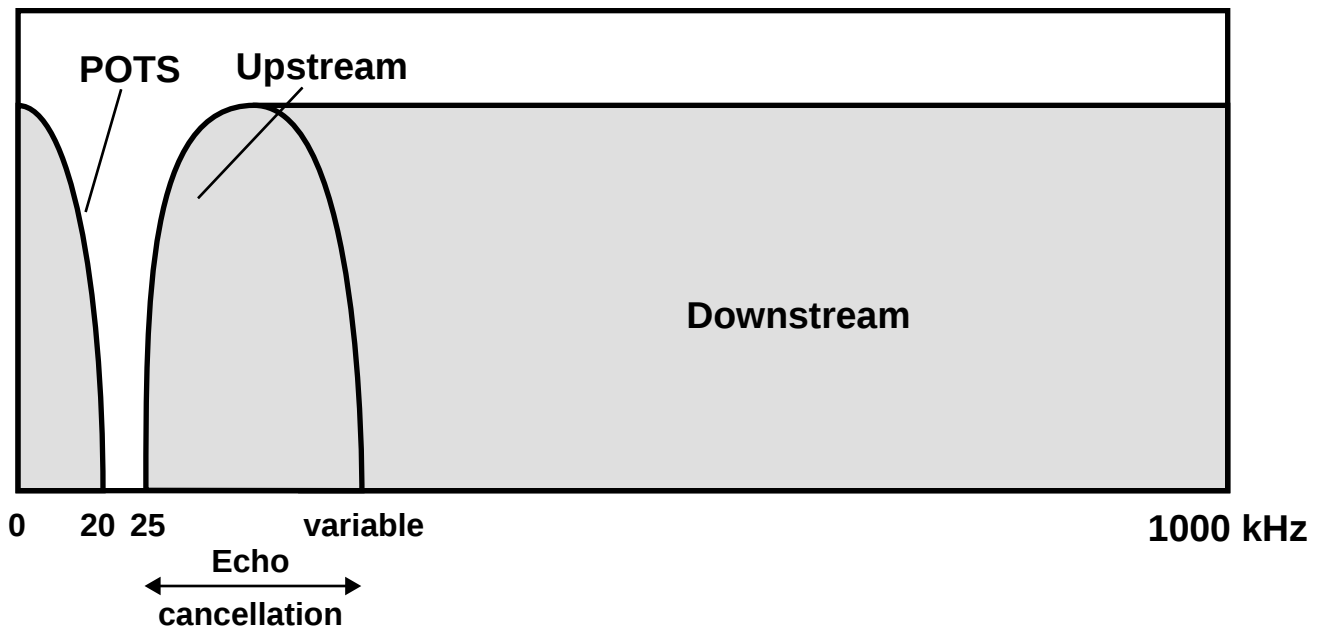
1. http://www.adsl.com/adsl_tutorial.html. See also *A Second Wind for Wiring*, IEEE Spectrum, September 1999.



POTS

Plain Old Telephone Service

Adding “echo cancellation” to increase downstream bandwidth



Echo cancellation

*sender subtracts its own signal from what's on the wire
yielding the other party's signal (!)*

Is the medium in place?

*stringing the wire or cable is expensive
(that's why we don't all have fibre...)*

Bandwidth

coax wins

— or does it? cable access is shared...

Security

ADSL wins

— it's not shared, so packets can't be sniffed

— can be tapped, but that's much more work

Nominal asymmetry is roughly the same (8 to 1)

both assume you're surfing

not hosting a web site

Measured at about 6 am

Time to ping my home from a university machine (average of 50 trials)

Rogers: 56 ms

Bell: 64 ms

Download speed (Netscape installer from Netscape)

Rogers: 305 KB / sec

Bell: 96 KB / sec

Upload speed (to a university machine, namely Oscar.math.uwaterloo.ca)

Rogers: 44 KB / sec

Bell: 13 KB / sec

Measured at about 7:30 pm

Time to ping my home from a university machine (average of 50 trials)

Rogers: 214 ms

Bell: 69 ms

Download speed (Netscape installer from Netscape)

Rogers: 224 KB / sec

Bell: 96 KB / sec

Upload speed (to a university machine, namely Oscar.math.uwaterloo.ca)

Rogers: 40 KB / sec

Bell: 13 KB / sec

Cable is a “shared access medium”
ADSL is not (at least to the local office)

Latency

comparable when lightly loaded
cable latency deteriorates more under (presumed) load
ADSL should not deteriorate at all (on the subscriber loop)

Bandwidth

cable bandwidth deteriorates during high-usage times of day,
but is “always” much superiour to ADSL’s.

Advertised performance

Bell quotes 120 KB/sec incoming and 15 KB/sec outgoing
Rogers quotes 384 KB/sec incoming and 50 KB/sec outgoing
Nobody guarantees anything

Prices are comparable if you have Bell long distance service

Bell provides a DYNAMIC IP address

Rogers provides a STATIC IP address (changing?)

Both disable Windows file sharing, but not Mac file sharing

Both appear to make about the same number of people unhappy

How does the receiver know when to sample the signal?

circuits inside a PC share a common timing signal

but what about circuits at two ends of a comm link?

how do they synchronize their clocks?

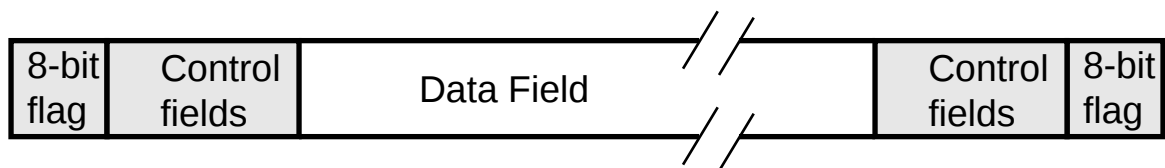
how do they KEEP them synchronized?

Synchronous communication

embed a clock signal in the data — eg Manchester encoding

special leading bits to identify the beginning of a data block

generic example “frame”

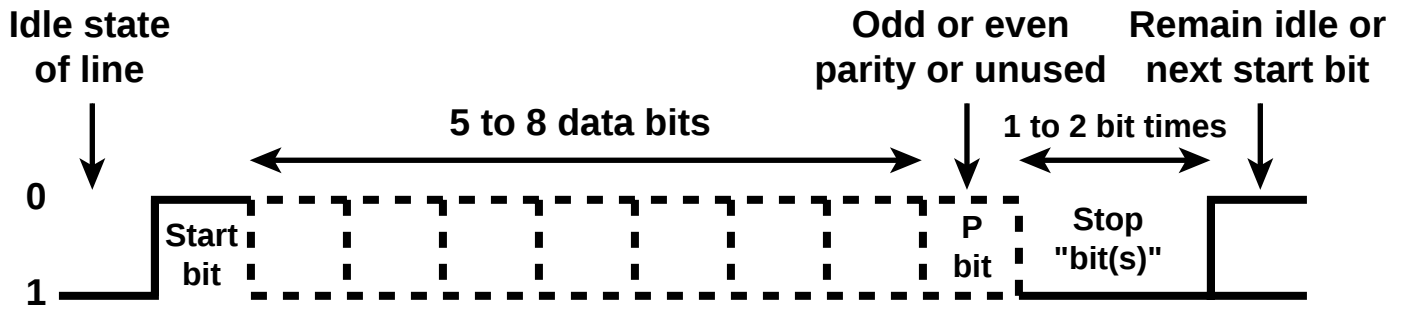


Asynchronous communication

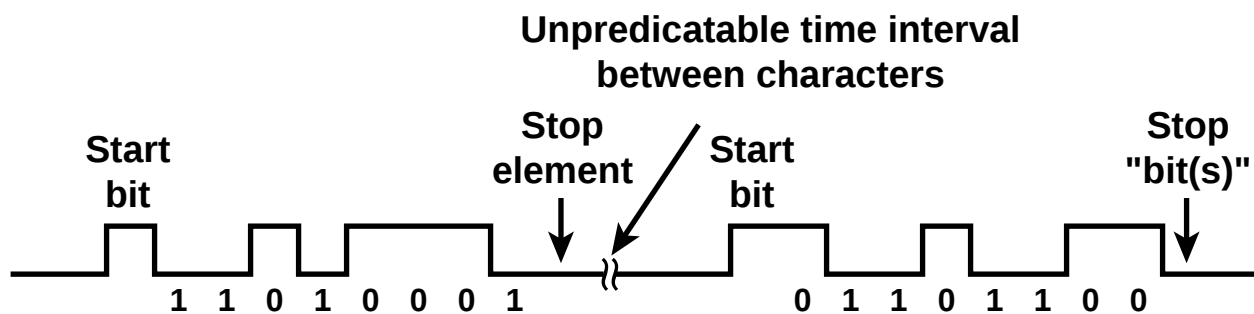
don't try for long-term / constant synchronization

instead, synch at the beginning of a data block

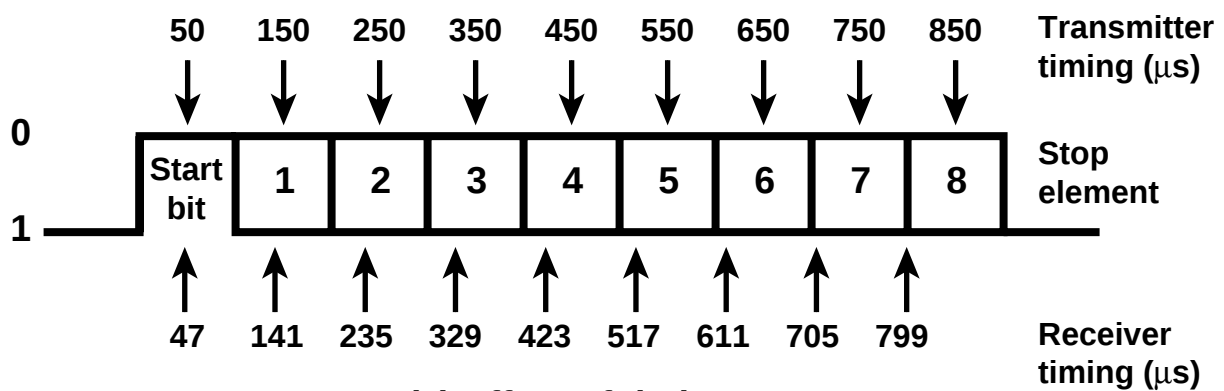
keep data blocks small so clock drift doesn't matter



(a) Character format



(b) 8-bit asynchronous character stream



(c) Effect of timing error