



Digital Mobile Radio A Brief Look Under The Hood

PAUL OLSON, K0WSU
K0WSU@2LAZYCATS.COM

How do we get from.....

Analog



Digital



100111001011010110
10



And Back Again.....

Analog



Digital

100111001011010110
10



The Magic Device

Analog



VOCODER

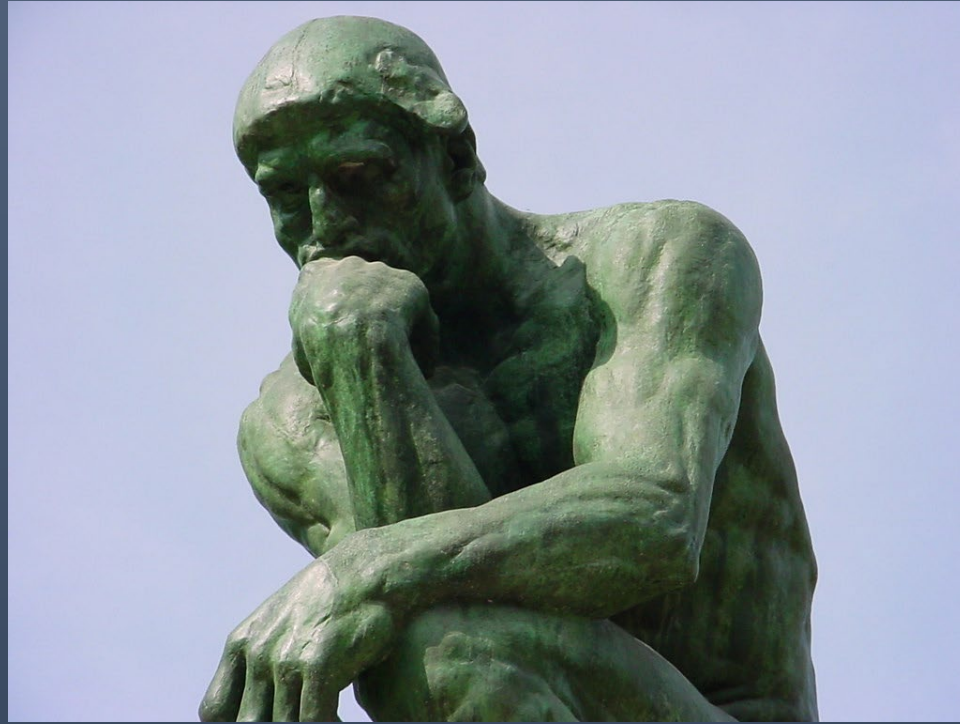
Voice Encoder

Digital

100111001011010110
10

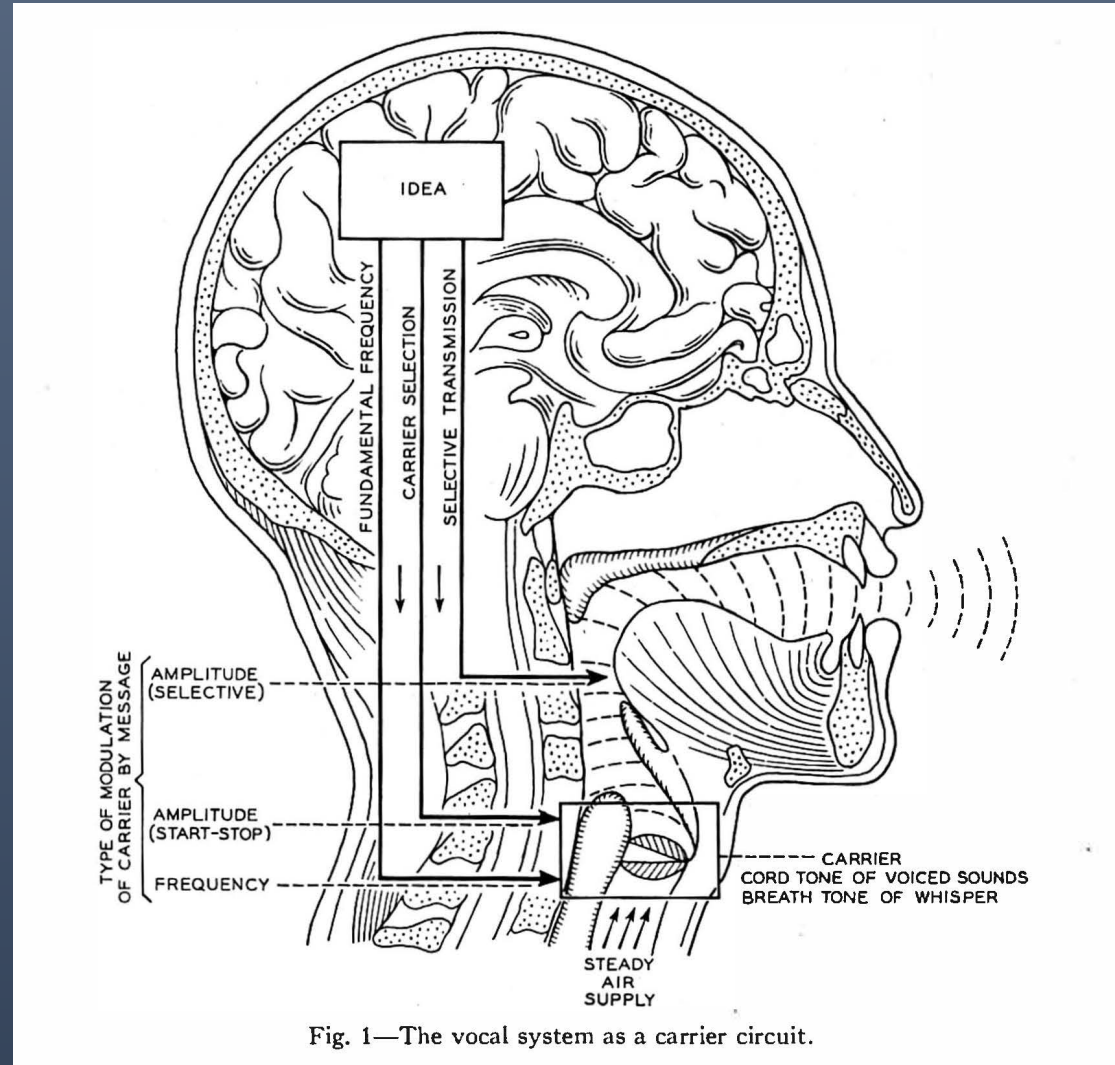


First a Bit of History

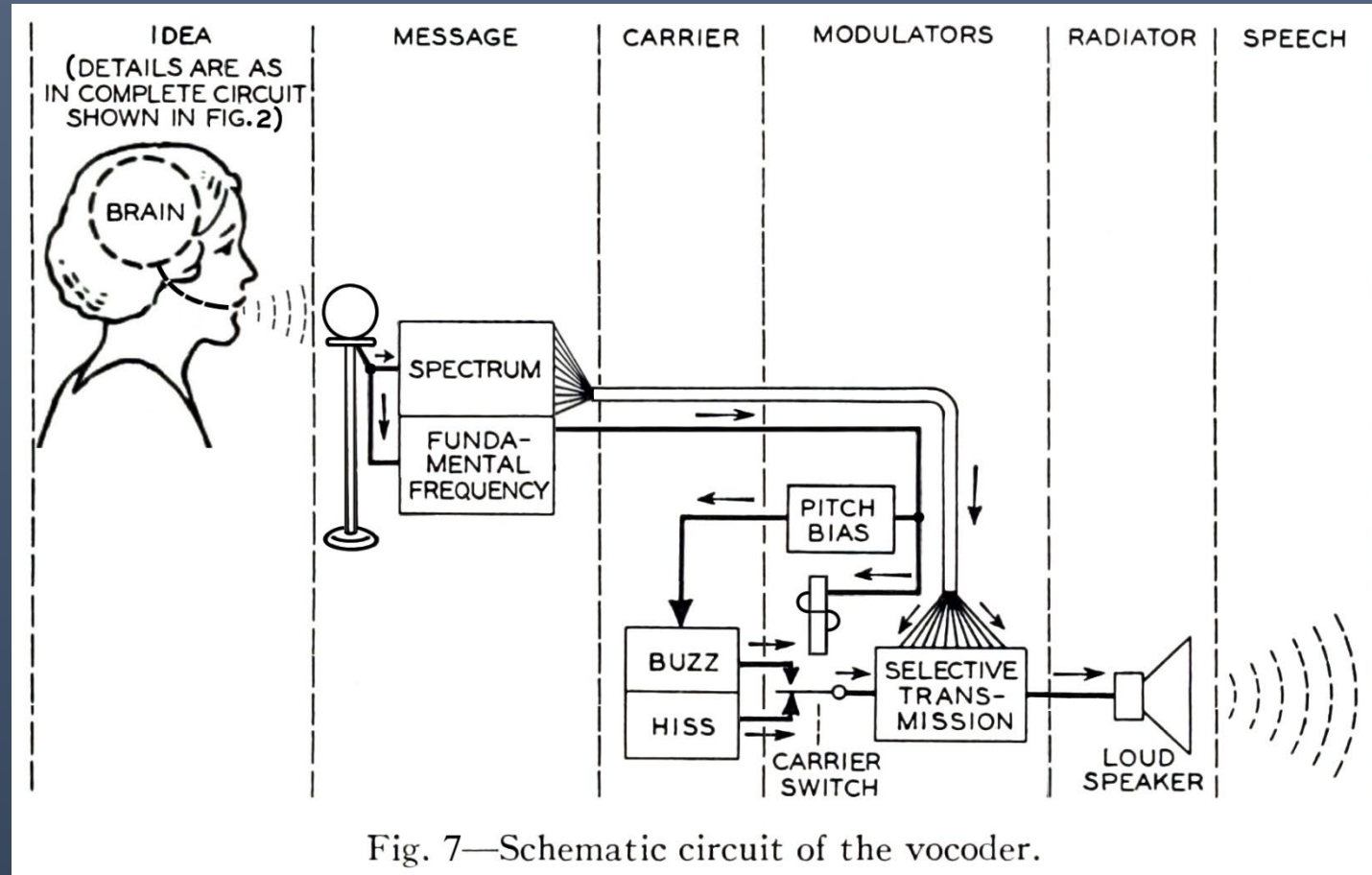


The Carrier Nature of Speech

Bell Labs Homer Dudley 1928



Bell Labs Homer Dudley Demonstrated 1939-40 at NY Worlds Fair



Worlds Fair Demonstrations

506

BELL SYSTEM TECHNICAL JOURNAL

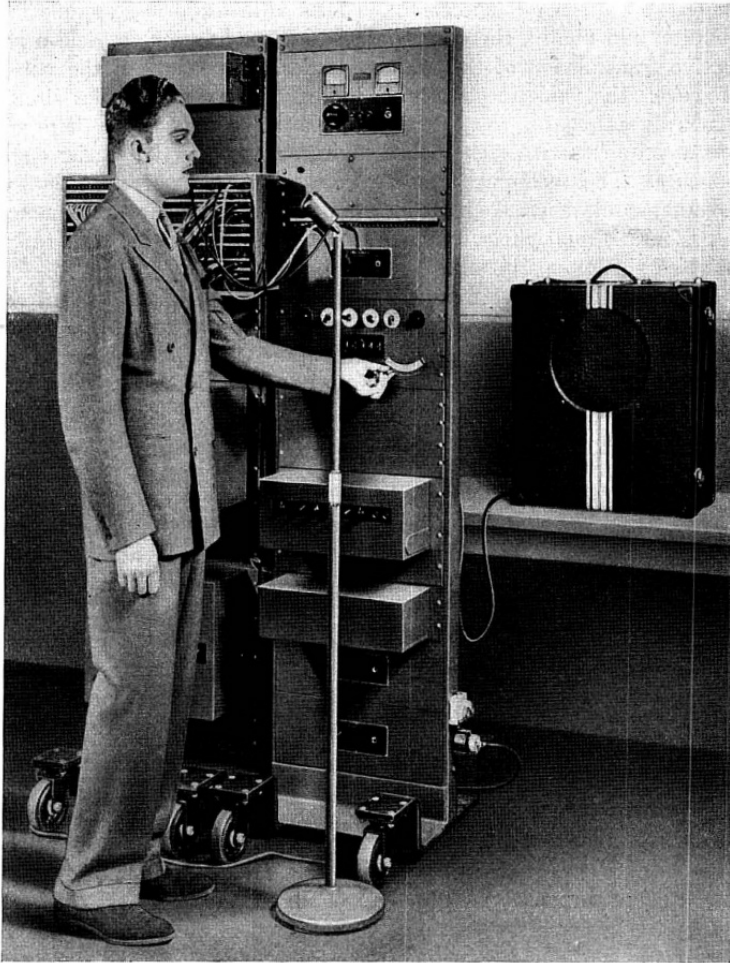


Fig. 4—The vocoder as demonstrated.

CARRIER NATURE OF SPEECH

507

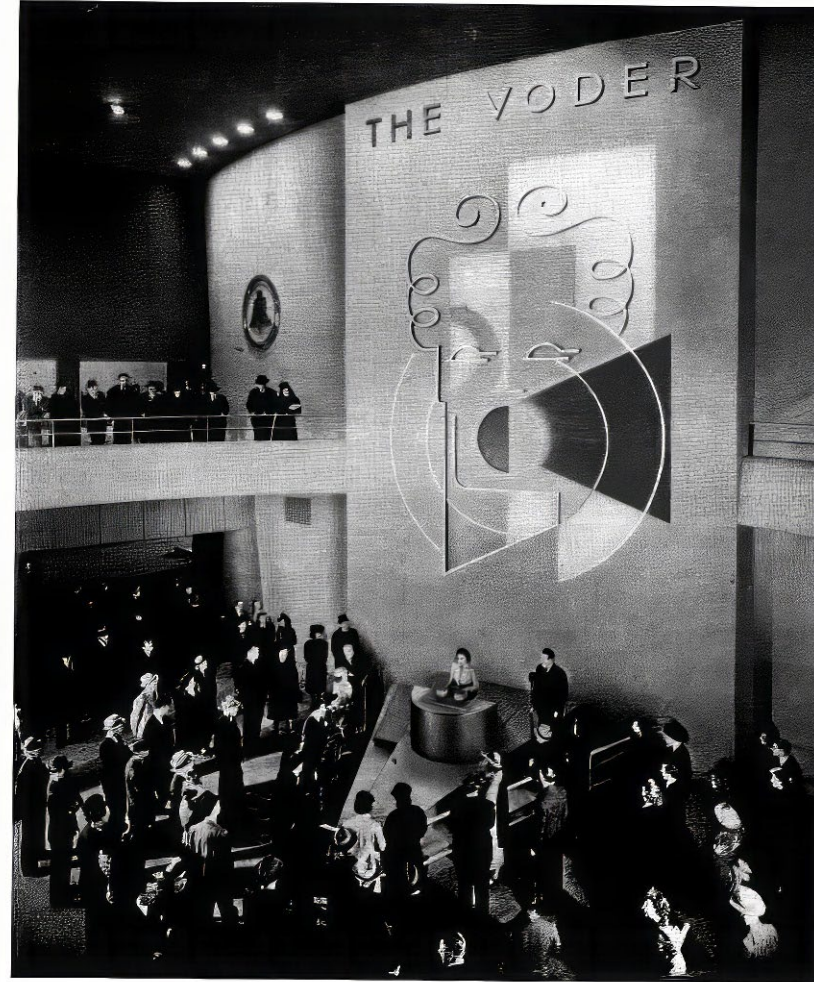


Fig. 5—The voder being demonstrated at the New York World's Fair.



The “Receive” side

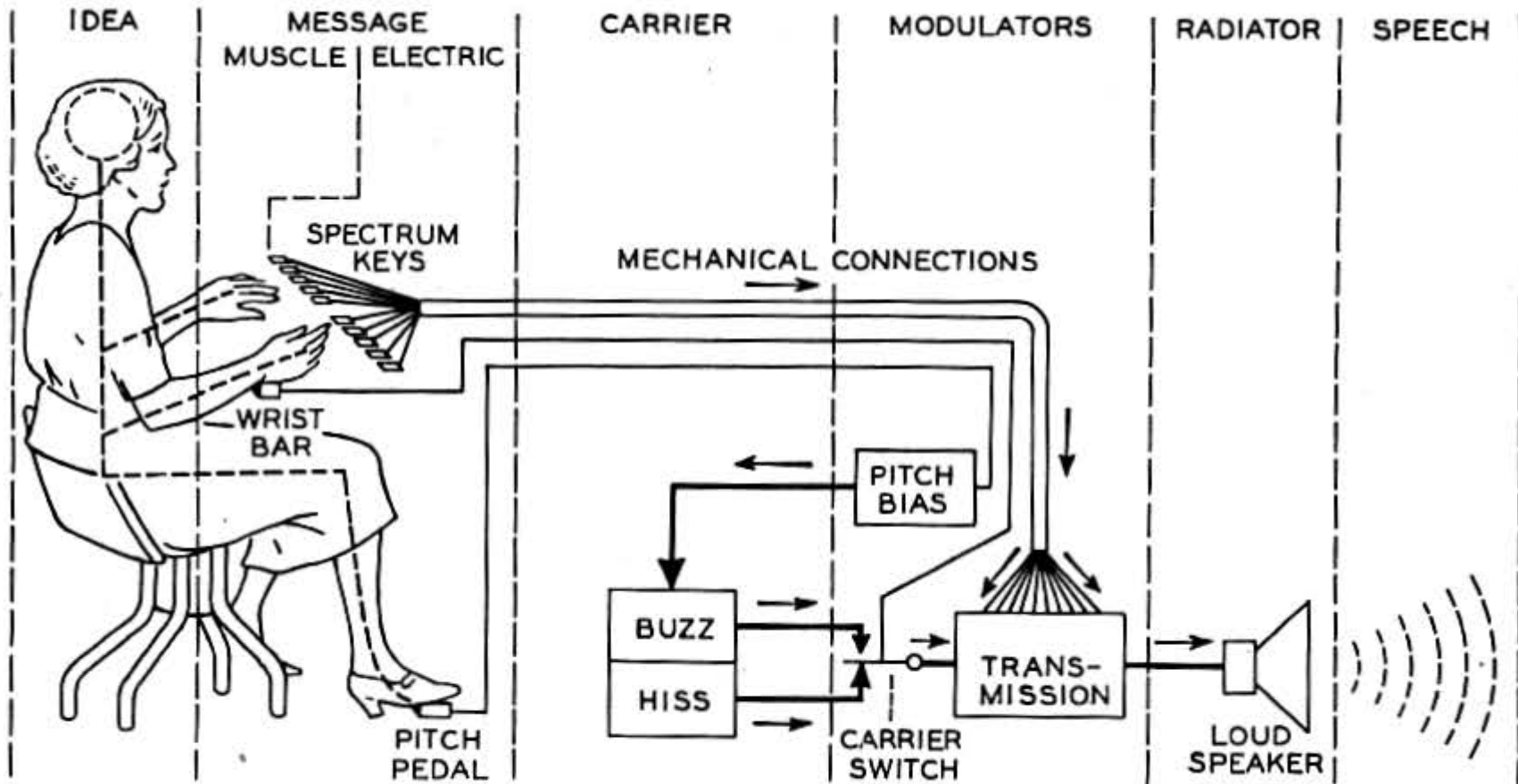


Fig. 8—Schematic circuit of the voder.

Bell Labs SIGSALY 1943

Used for encrypted Voice in WW2

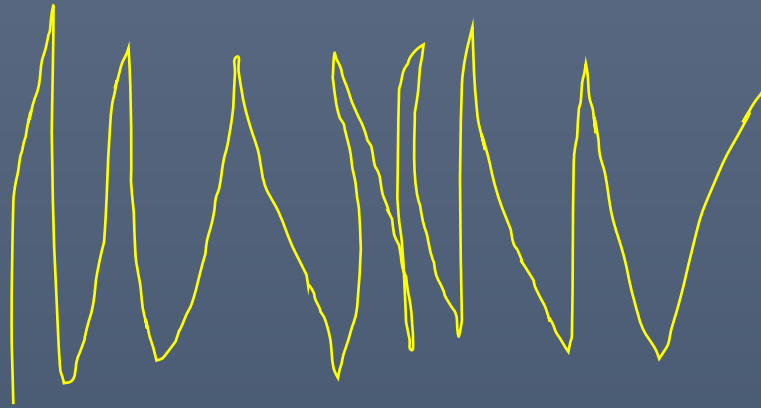


The SIGSALY terminal was massive. Consisting of 40 racks of equipment, it weighed over 50 tons, and featured two turntables which were synchronized on both the sending and the receiving end by an agreed upon timing signal from the U.S. Naval Observatory. SIGSALY used 30kW of power and required air conditioning.

This was the first use of Pulse Code Modulation and was transmitted using a form of Frequency Shift Keying.



Sampling the Analog Signal?



- Different Voices

1. Chanting Monotone
2. Whisper
3. Ventriloquistic
4. Nasal
5. Southern Drawl
6. Squeak-Drunken man's hiccup effect
7. Voice Breaking in an adolescent
8. Talker Chewing Gum
9. Crying Child
10. Reversed Inflection –Swedish Voice
11. Tremulousness of Old Man
12. Music-Singing

- How often?
- What levels?
- Fundamental Characteristics
 - Pitch
 - Resonant Frequency regions
 - Loudness
- How accurate does the reproduction need to be?



Dec. 12, 1939.

R. R. RIESZ

2,183,248

WAVE TRANSLATION

Filed Sept. 11, 1936

3 Sheets-Sheet 1

FIG. 1

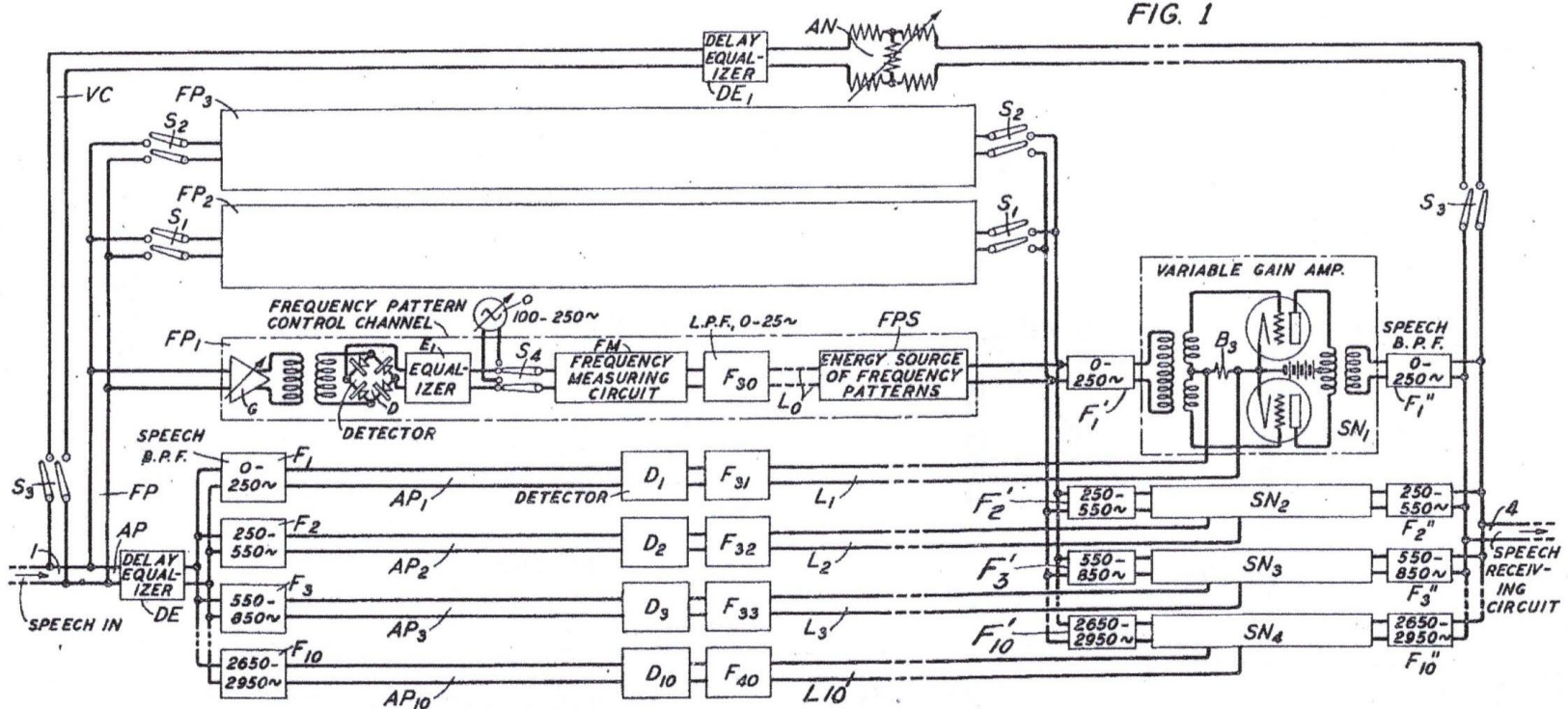


FIG. 2

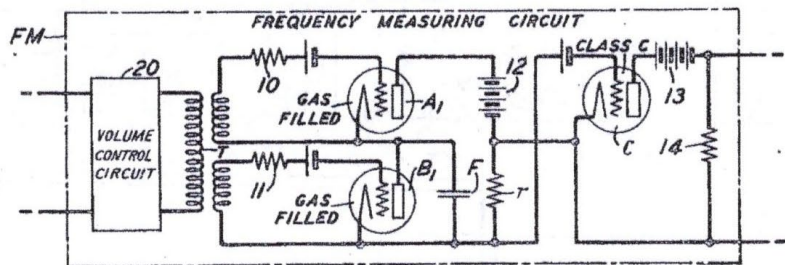
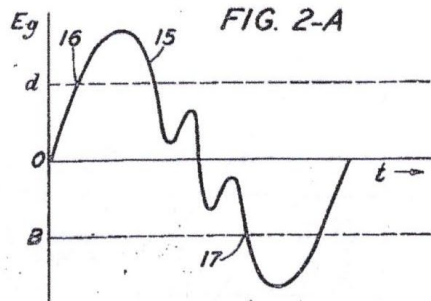
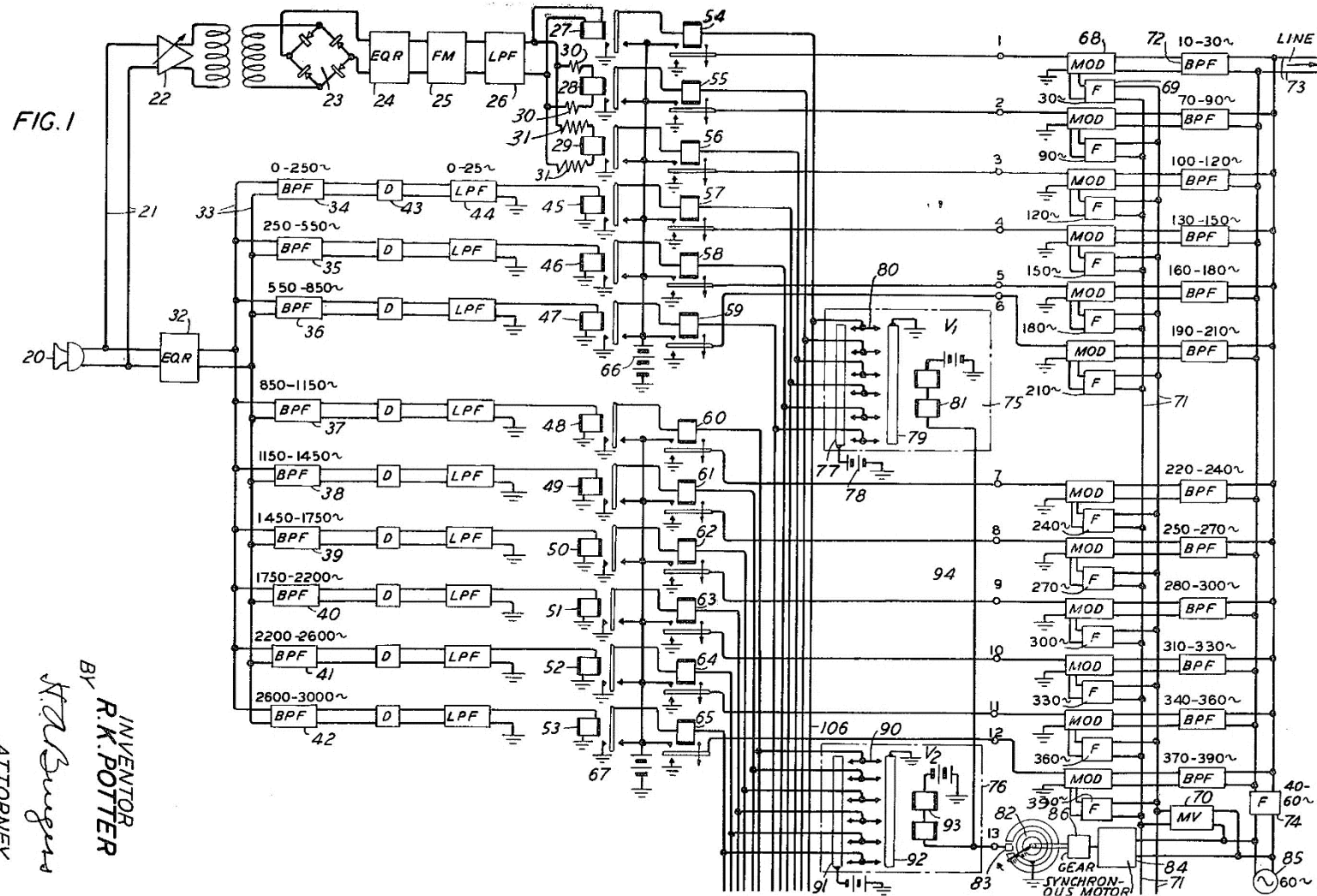


FIG. 2-A



INVENTOR
R. R. RIESZ
BY *H. A. Burgess*
ATTORNEY



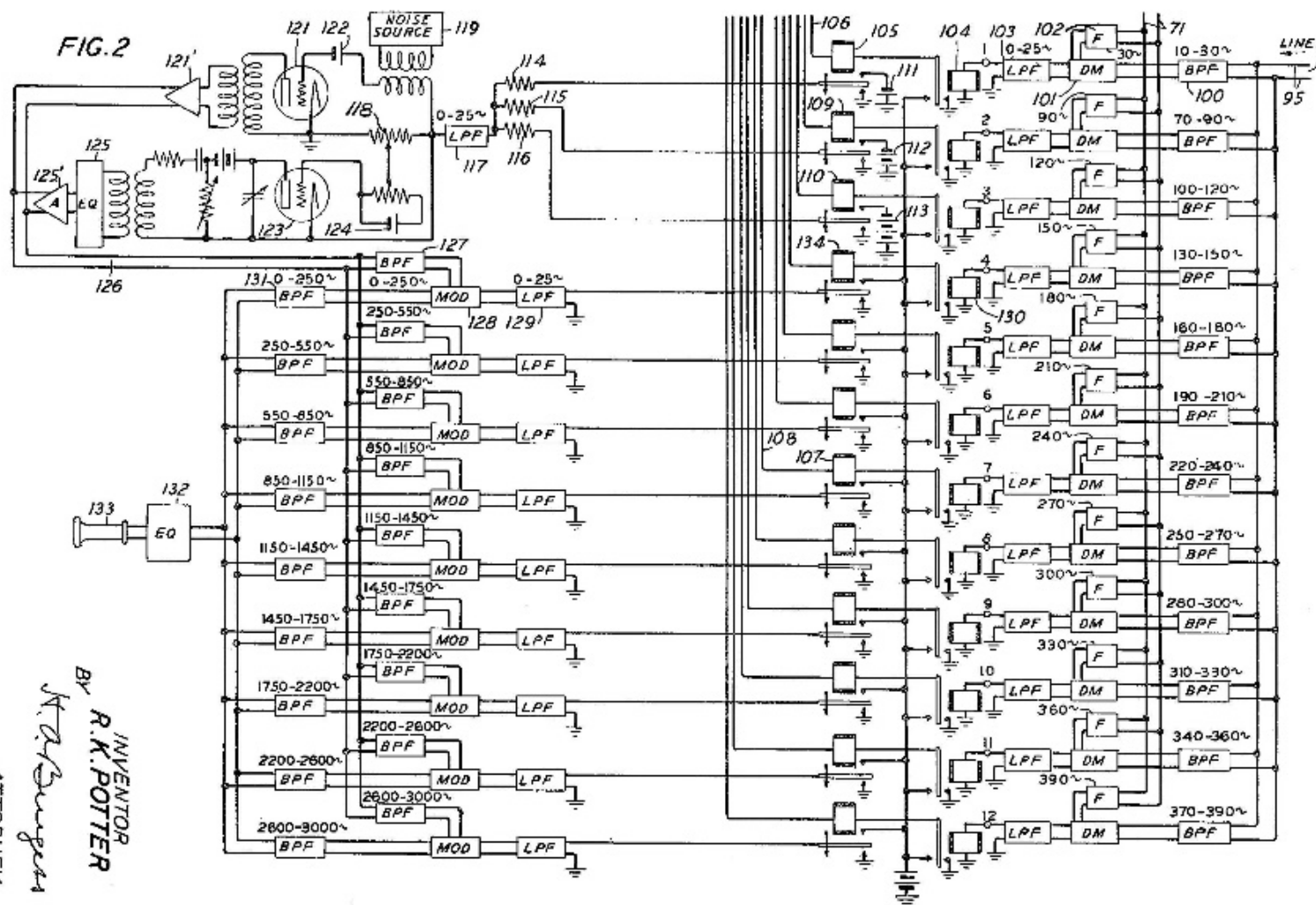


INVENTOR
R.K. POTTER
BY
K. M. Burgess
ATTORNEY

U.S. Patent June 29, 1976 Sheet 1 of 3 3,967,067

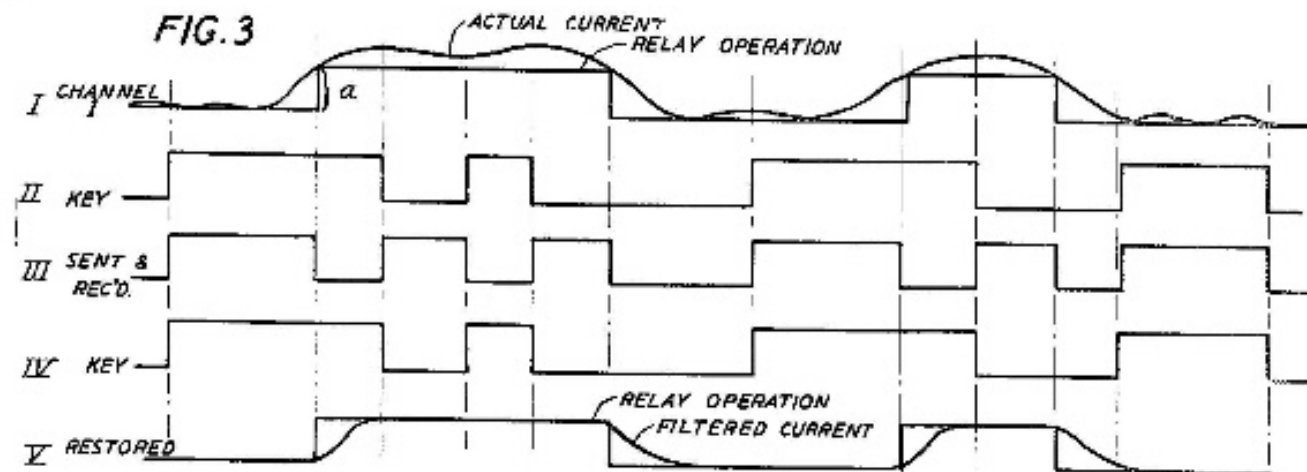
Note: 12 output signals! Each was sent over a different radio frequency.
This is the first application of Direct Sequence Spread Spectrum Radio.





INVENTOR
R. K. POTTER
BY *R. A. Burgess*
ATTORNEY





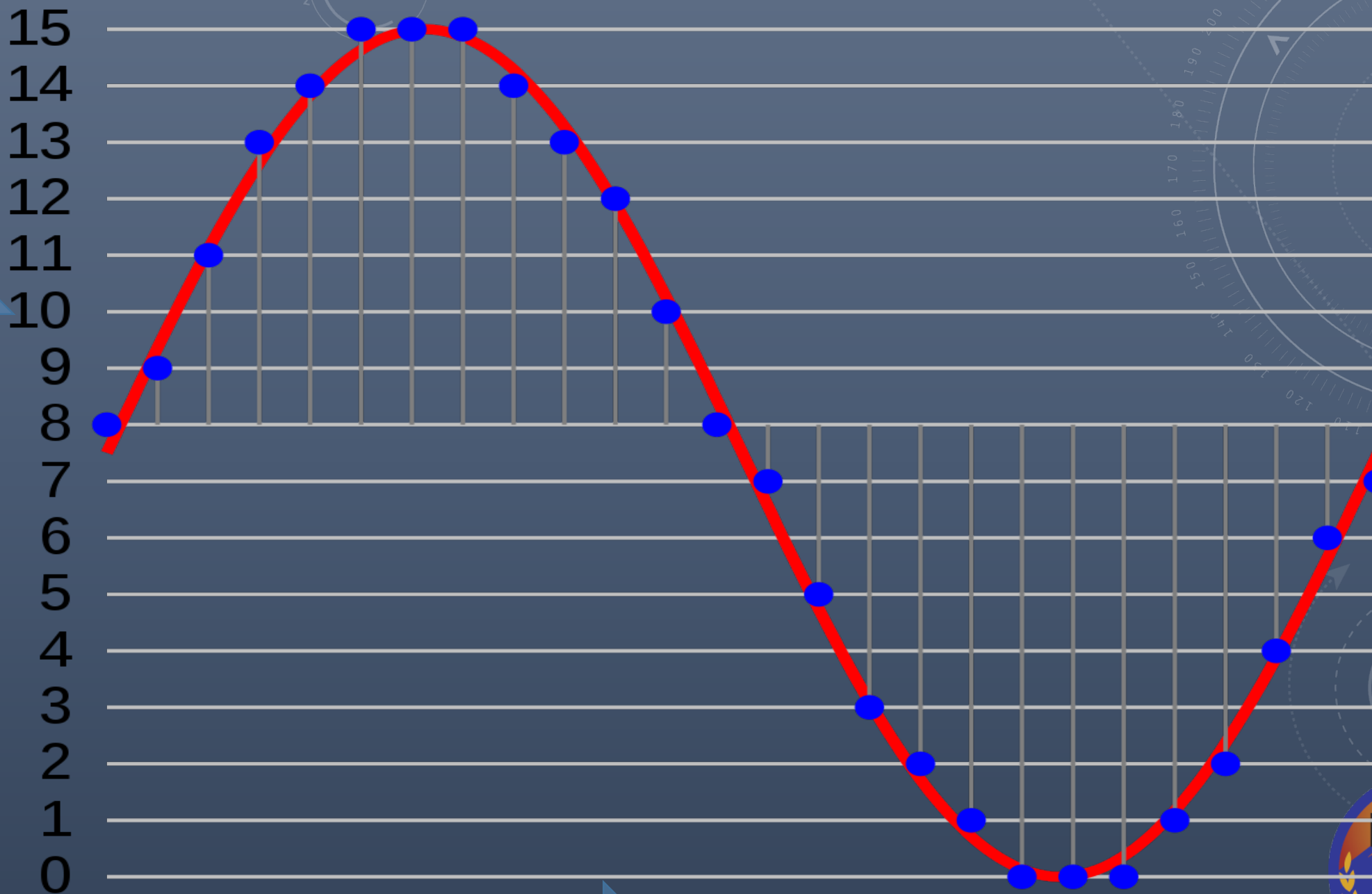
How Often to Sample?

- Nyquist-Shannon Sampling Theorem
 - Establishes a method to change a continuous-time signal to a discrete-time signal.
 - It establishes a sample rate that permits a discrete sequence of samples to capture all the information from a continuous-time signal of finite bandwidth.
- **Theorem** – If a function $x(t)$ contains no frequencies higher than B hertz, then it can be completely determined from its ordinates at a sequence of points spaced less than $1/(2B)$ seconds apart.



Example

Amplitude ↑

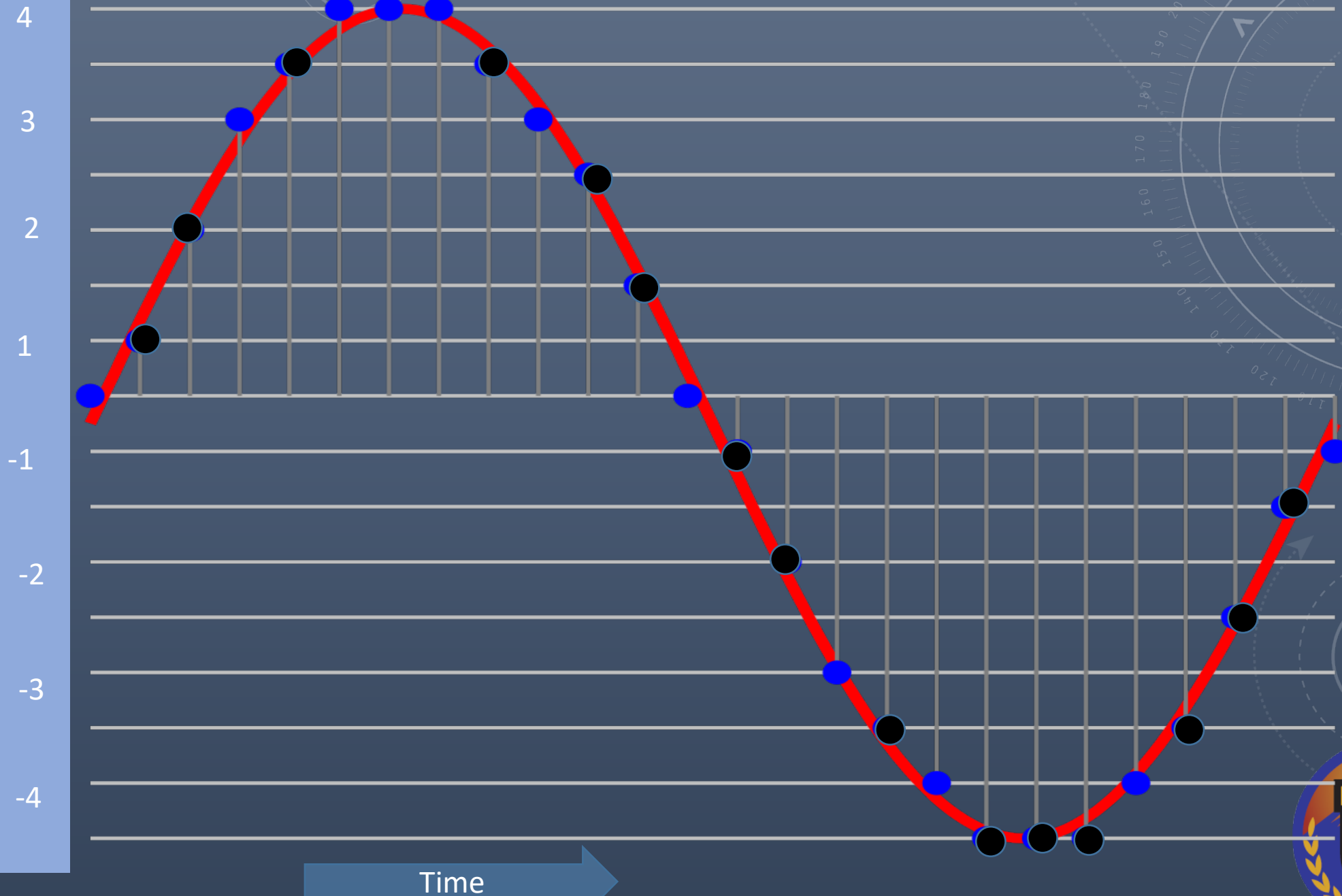


Time →



Example

Amplitude



A bit of math

- Human voice bandwidth is about 4000 Hz
- The original PCM had 8 sample points
- Applying Nyquist - $1/2B$ seconds apart. Therefore sampling would be done every $1/8000$ seconds or 8000 samples per second.
- Then with 8 sample points would result in a data stream of 8×8000 or 64,000 data points per second

This is what the phone company standardized on.
64Kbps per voice channel

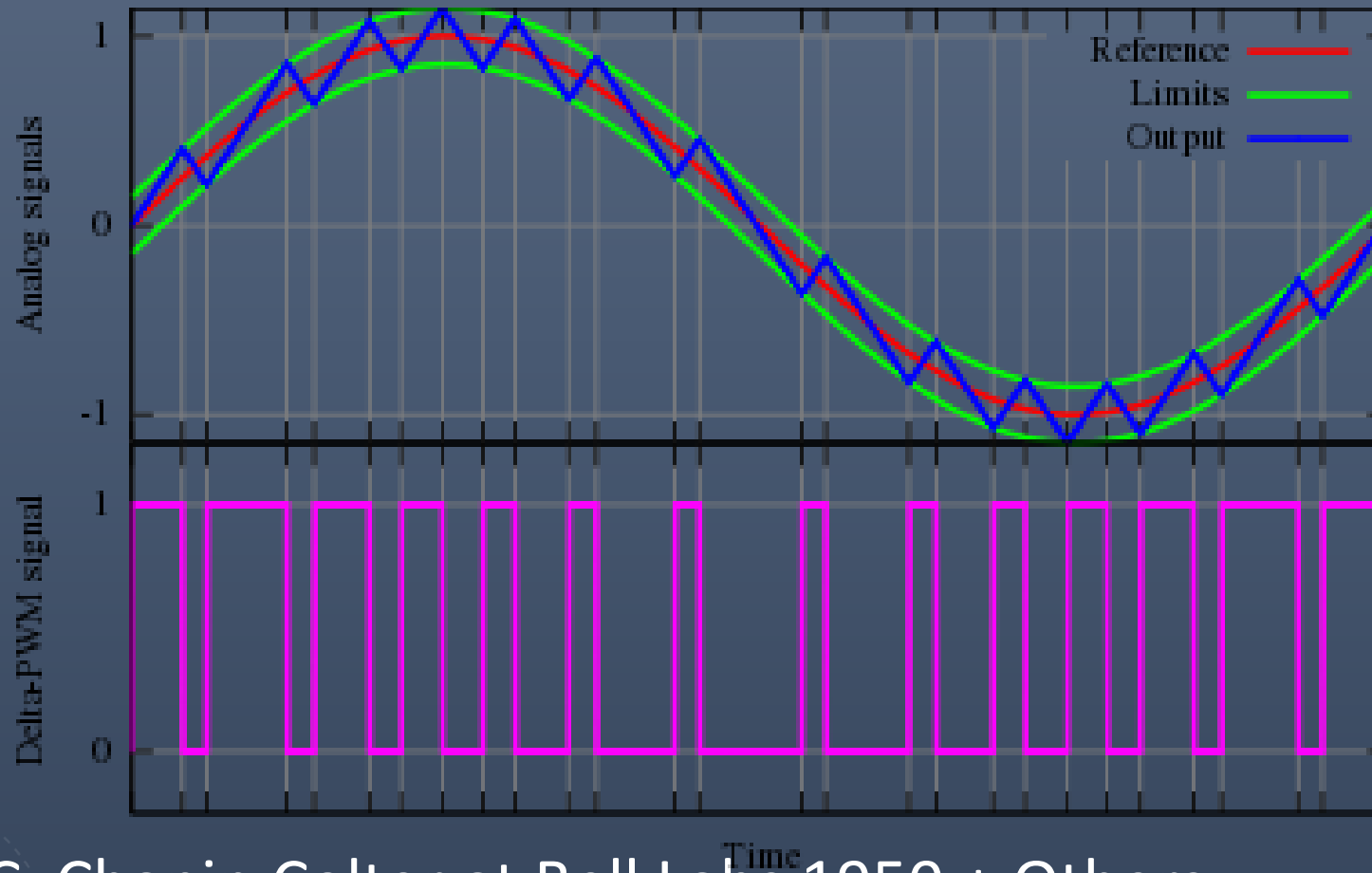


This is a big bandwidth HOG!!!!

- Nyquist- Shannon has saddled us with a more or less fixed sampling frequency (8000 samples/second).
- That leaves only one parameter we can alter.
- Do we really need 8 amplitude sample points?
- What can we do to reduce the number of sample points?



Differential Pulse-Code Modulation



- C. Chapin Colter at Bell Labs 1950 + Others



Key tenants of DPCM

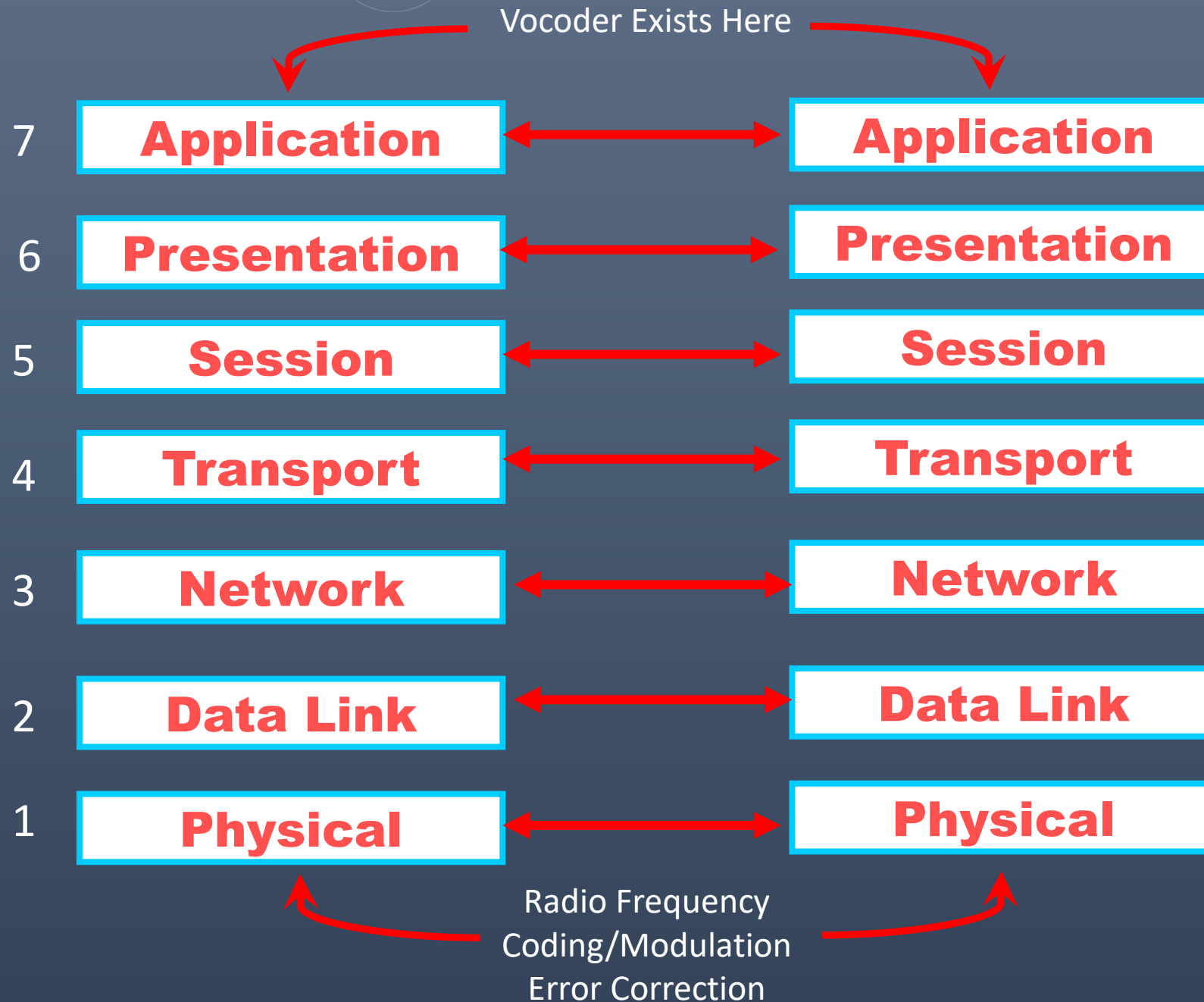
- The analog Signal is approximated with a series of segments
- Each segment of the approximated signal is compared to the previous bits and the successive bits are determined by this comparison
- Only the change of information is sent, either an increase or decrease in value
- However, it requires a higher sampling rate.



Adaptive Delta Modulation (1968)

- Also called Continuously Variable Slope Delta Modulation.
- The step size is not fixed.
- Selected by NASA for mission control communications.





Rules Define a Layer

Before we get into the individual layers we need to present a definition of how they function and relate to one another.

- A layer should be defined for every distinct level of functionality.
- The function of each should be well defined and limited in its scope.
- International standardized protocols layer should be used wherever possible.
- The interface between the layers should minimize the interaction between them
- The interface between layers is well defined.

Padlipsky's Rule

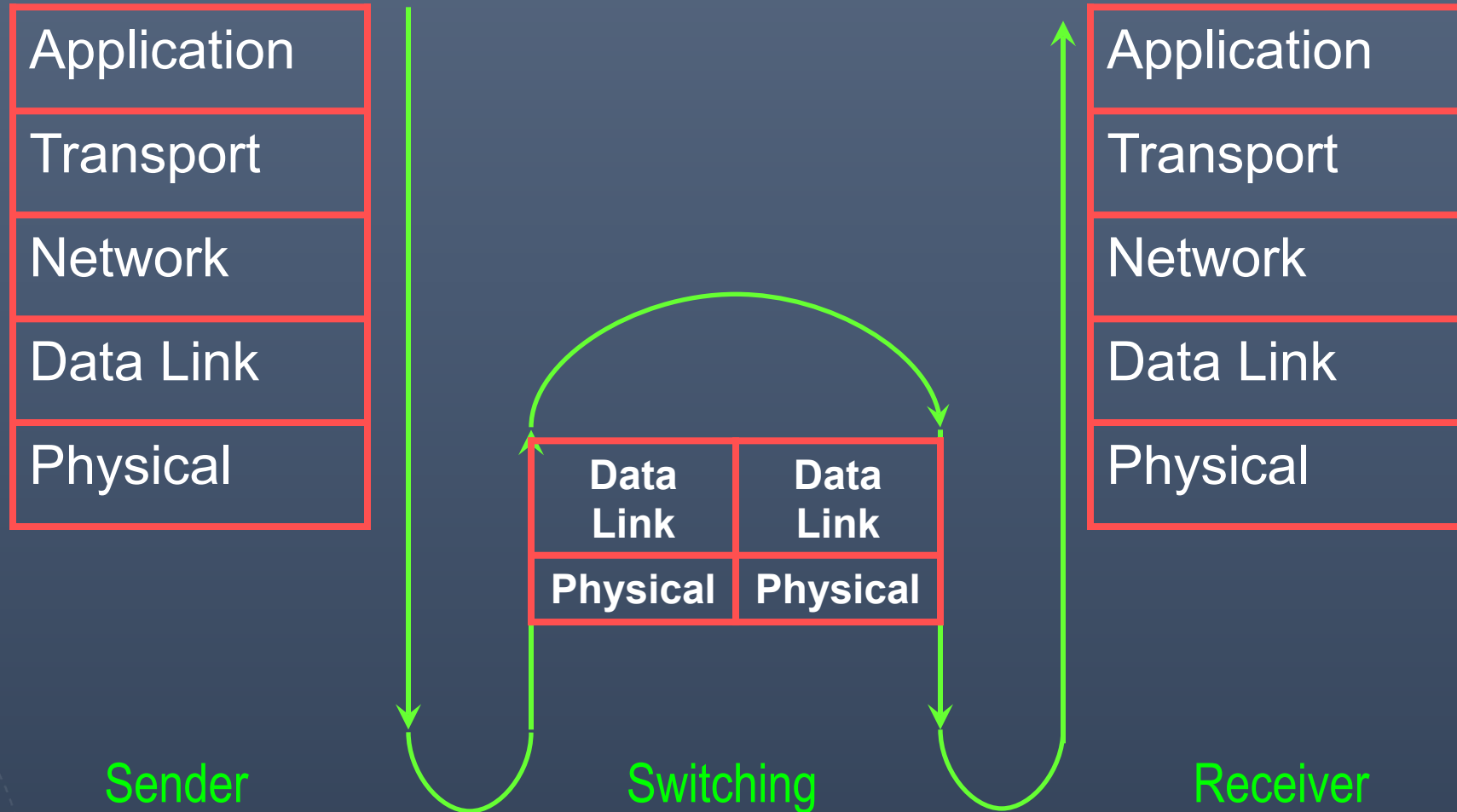
If you know what you're doing, three layers is enough.

If you don't, even seventeen won't help.

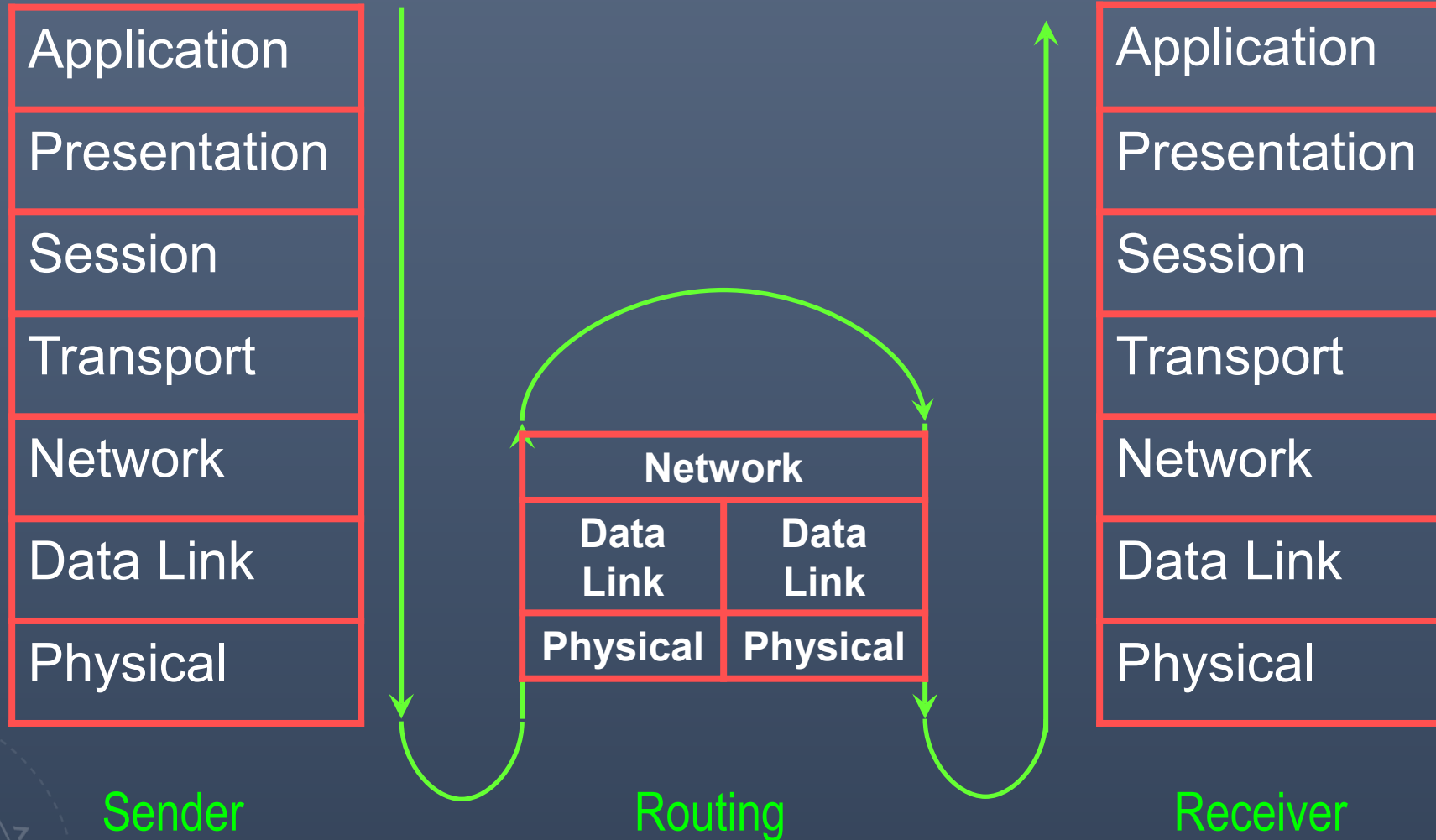


Ethernet Switching

(Station to Station)



IP Routing (End to End)



ETSI TR 102 398 V1.4.1 (2018-11)



**Electromagnetic compatibility
and Radio spectrum Matters (ERM);
Digital Mobile Radio (DMR)
General System Design**



ETSI TR 102 398

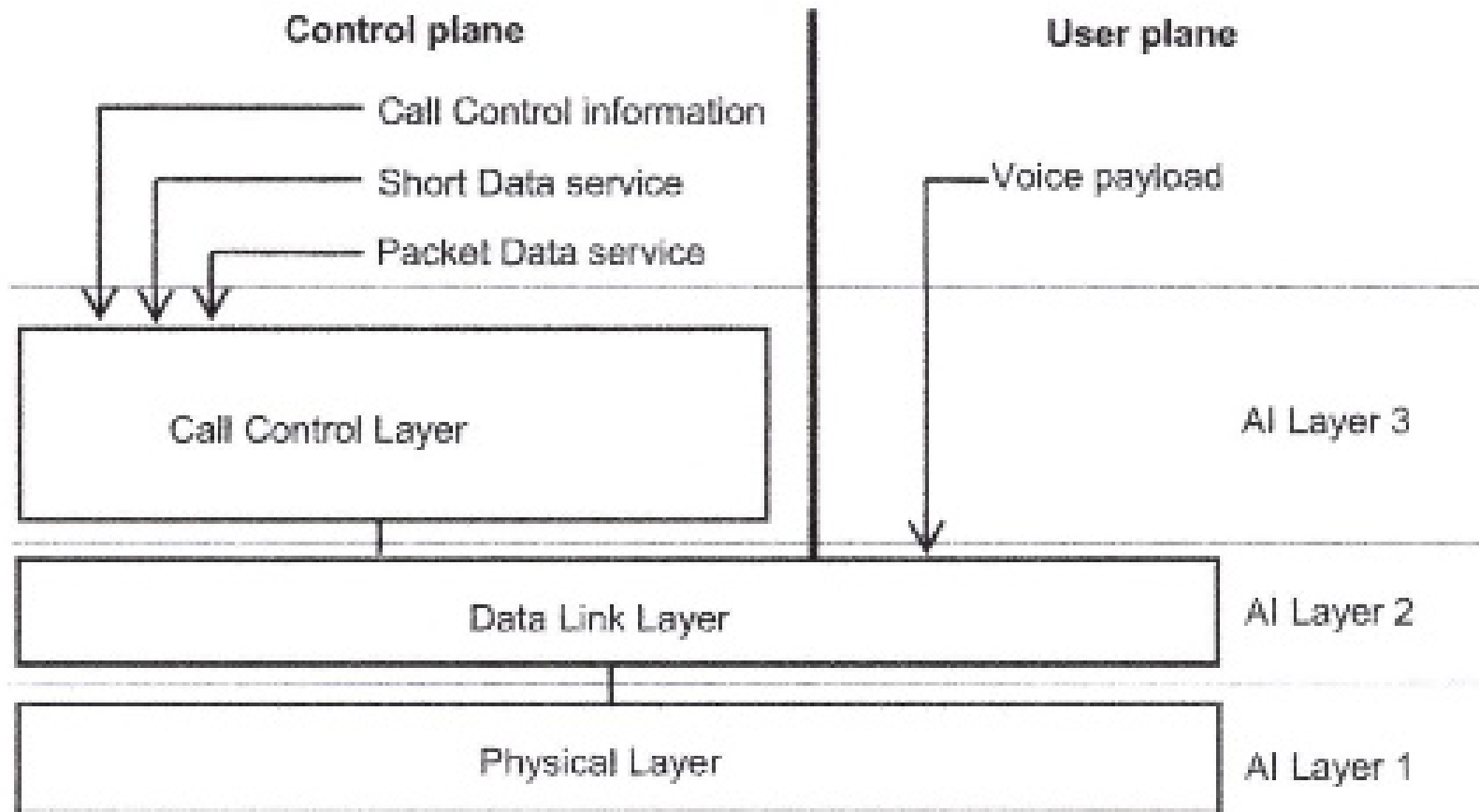


Figure 4.1: DMR protocol stack



The Vocoder is not included!

- 5.1.3 The vocoder In order to achieve interoperability between units from different suppliers, the same vocoder or a completely compatible vocoder will have to be used. In order to avoid undue restrictions being placed on suppliers and thus limiting the markets that they may choose to address, it has been agreed not to specify any particular vocoder in the standard. There is a vocoder socket specified in the standard and any chosen vocoder should be compatible with the present document.

However:

Each end of the conversation has to employ the same vocoder.



Which one do we use?



INTERNATIONAL TELECOMMUNICATION UNION

CCITT

THE INTERNATIONAL
TELEGRAPH AND TELEPHONE
CONSULTATIVE COMMITTEE

G.726

GENERAL ASPECTS OF DIGITAL
TRANSMISSION SYSTEMS;
TERMINAL EQUIPMENTS

40, 32, 24, 16 kbit/s ADAPTIVE
DIFFERENTIAL PULSE CODE
MODULATION (ADPCM)

Recommendation G.726



Geneva, 1990



INTERNATIONAL TELECOMMUNICATION UNION

ITU-T

TELECOMMUNICATION
STANDARDIZATION SECTOR
OF ITU

G.711

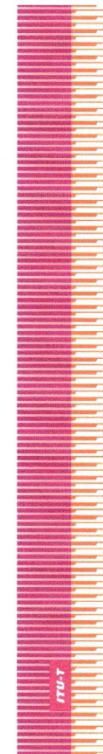
GENERAL ASPECTS OF DIGITAL TRANSMISSION
SYSTEMS

TERMINAL EQUIPMENTS

PULSE CODE MODULATION (PCM) OF
VOICE FREQUENCIES

ITU-T Recommendation G.711

(Extract from the Blue Book)



International Telecommunication Union

ITU-T

TELECOMMUNICATION
STANDARDIZATION SECTOR
OF ITU

G.722

(09/2012)

SERIES G: TRANSMISSION SYSTEMS AND MEDIA,
DIGITAL SYSTEMS AND NETWORKS

Digital terminal equipments – Coding of voice and audio
signals

7 kHz audio-coding within 64 kbit/s

Recommendation ITU-T G.722



ANSI/TIA-102.BABA 2003
Approved: December 23, 2003

TIA STANDARD

Project 25

Vocoder Description

TIA-102.BABA

DECEMBER 2003

TELECOMMUNICATIONS INDUSTRY ASSOCIATION



Representing the telecommunications industry in
cooperation with the Electronic Industries Alliance



Electronic Industries Alliance



What Vocoder is being used?

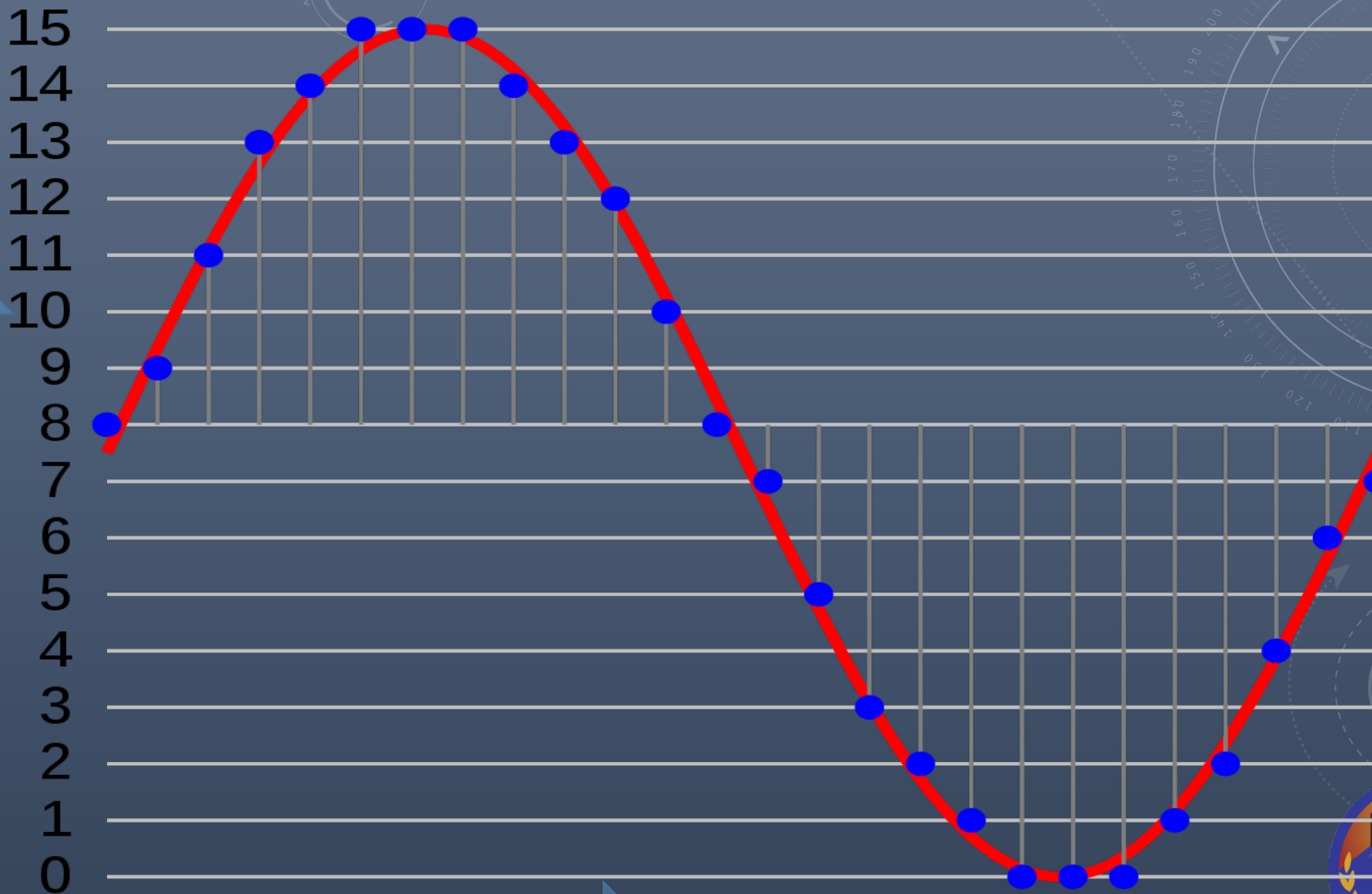
Mostly Proprietary Chips from
DSP Innovations© with the AMBE2+ Vocoder
(Advanced Multi Band Excitement)

While the Vocoder is not in the ETSI standard!
Project 25 specifies this chip.



Example

Amplitude



Time



Conclusion

We talked About

- The historical beginnings of the Vocoder
- A ground breaking early application of a Vocoder
- Several of the underlying methods of digitizing speech
- The OSI 7 layer model
- Applicable Standards or not!



Any
Questions



