

LINEAR PREDICTIVE CODING for speech (vocoders)

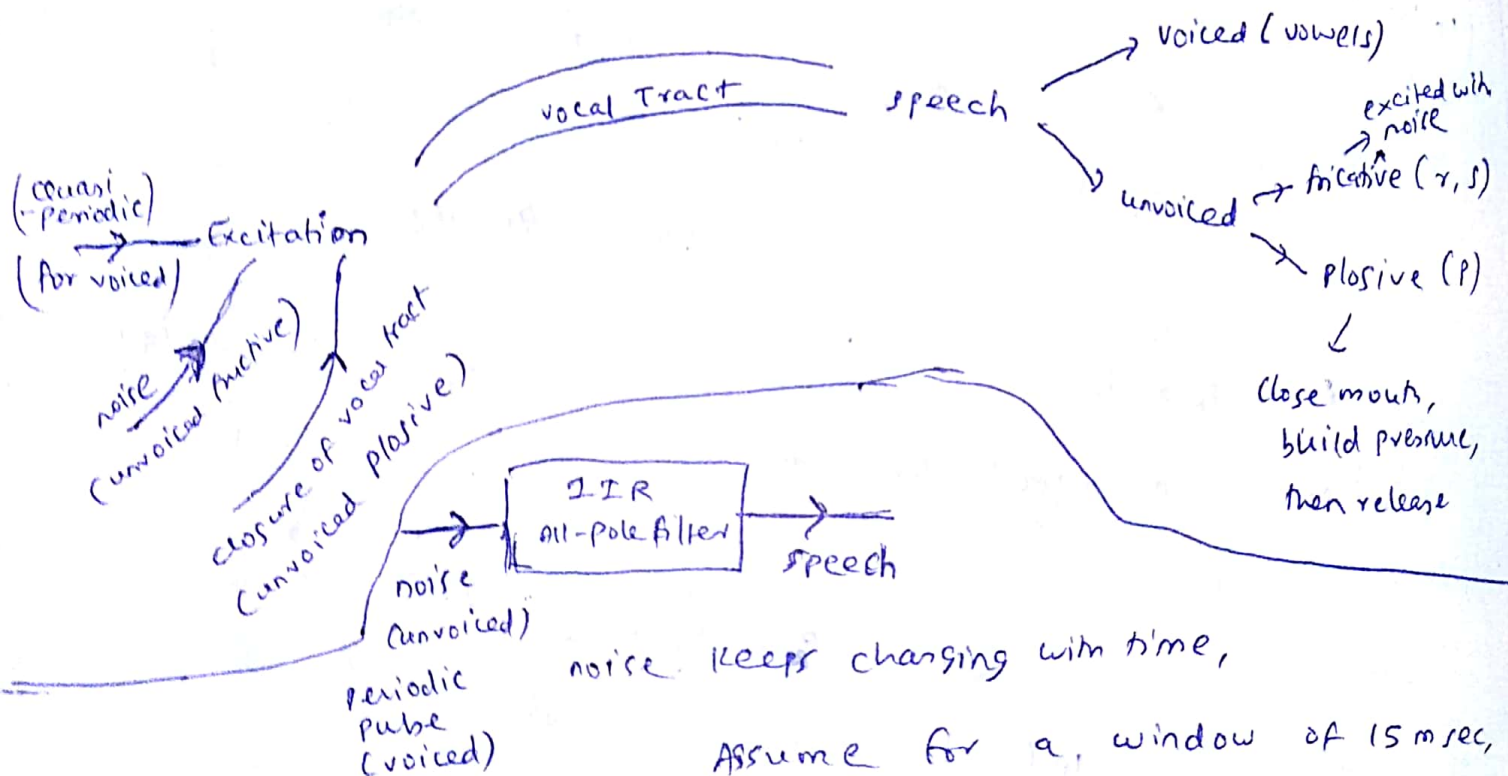
LPC ^{used} → specific for speech
 PCM → can be used for anything other than speech also
 because it models human vocal tract

Type of Source Coding

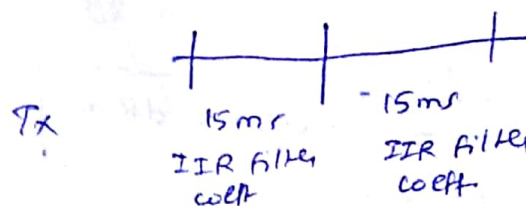
CELP

Code ~~Excited~~ Excited LPC

IS 95 CDMA



from Tx, IIR filter coefficients are only transmitted



~~$x(n) = a_1 y(n-1)$~~

for PCM, we directly see encoded bits

for DPCM, we transmit no prediction error.

But here, " " " " IIR filter coeff.

$$y(n) = a_1 y(n-1) + a_2 y(n-2) + \dots + a_m y(n-m) + g x(n)$$

$$\frac{Y(z)}{X(z)} = H(z) = \frac{G}{1 + a_1 z^{-1} + \dots + a_m z^{-m}}$$

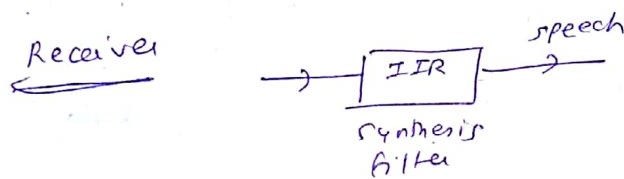
SYNTHESIS FILTER

IIR filter $H(z) = \frac{G}{1 + \sum_{k=1}^m a_k z^{-k}}$ $z = e^{j\omega}$

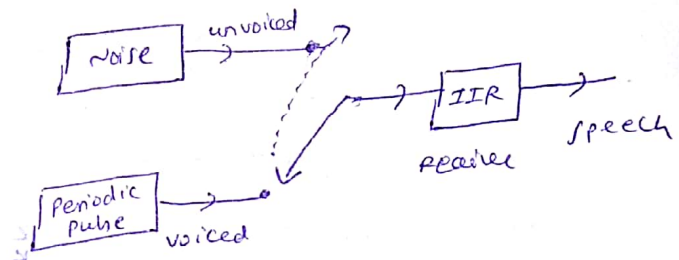
we have to transmit $T_x(\text{Gain}, \underbrace{a_1, a_2, \dots, a_m}_{\text{IIR filter coeff.}})$

suppose order $M = 10$

$T_x(G, a_1, a_2, \dots, a_{10})$



Voiced / Unvoiced \Rightarrow
pitch \Rightarrow



Each filter coeff \Rightarrow 6 bits / coeff.

order 10 \Rightarrow 10 coeffs \Rightarrow $6 \times 10 = 60$ bits for IIR filter

Gain = 6 bits

pitch = 5 bits

voiced/unvoiced = 1 bit

So, for every 15 msec, we are transmitting

$$60 + 6 + 5 + 1 = 72 \text{ bits for 15 msec}$$

Bit rate = 4.8 Kbps (much lower than PCM (64 Kbps) and DPCM (48 Kbps))

but provides same quality.

Computation at the ~~Receiver~~ Transmitter:

Determine voiced/unvoiced

(Energy of voiced) \gg Energy of unvoiced

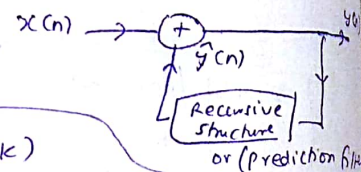
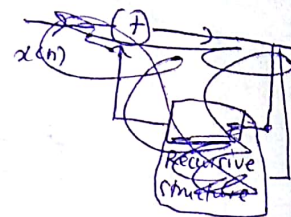
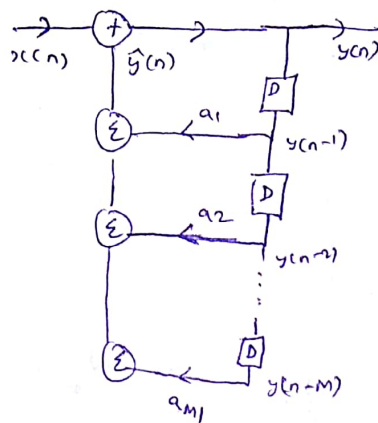
(zero crossing of unvoiced) \gg (zero crossing of voiced)

(Auto-correlation coefficient) voiced \gg Auto-correlation coefficient of unvoiced

This is how Tx decides whether it is voiced or unvoiced.

occurs at Tx

Determination of IIR filter coefficient (Symmetric filter)



$$y(n) = x(n) + \sum_k a_k y(n-k)$$

$$y_n = x(n) + \sum_k a_k y(n-k)$$

$$-x(n) = \underbrace{\sum_k a_k y(n-k)}_{\text{prediction of } y(n) \text{ based on past sample}} - y(n) = \hat{y}(n) - y(n) = e_n$$

$$e_n = \text{Prediction Error}$$

Job of Tx:

For every 15 msec, find a_1, a_2, \dots, a_m that minimizes e_n^2 (mean squared error)

Assume in 15 msec, we get N samples.

$$E = \sum_{n=1}^N e_n^2 \quad \text{Find } \{a_1, a_2, \dots, a_m\} \text{ min}(E)$$

Minimize E:

$$\frac{\partial E}{\partial a_1} = 0$$

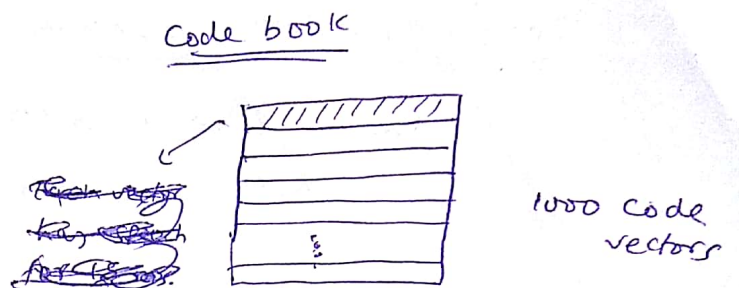
$$\frac{\partial E}{\partial a_2} = 0$$

and so on...

CELP



For every 15 ms,
the ~~speech~~ ~~best~~ best
code word is chosen
and that code vector
is transmitted.



So, Tx transmits $\rightarrow 10 \times 6 = 60$ coeff. bits
6 bits gain
code vector