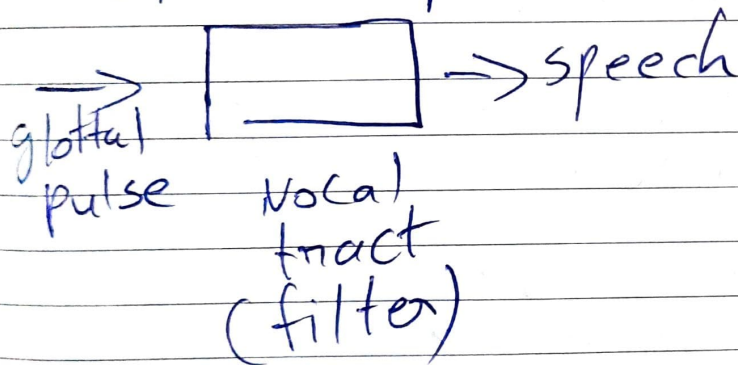
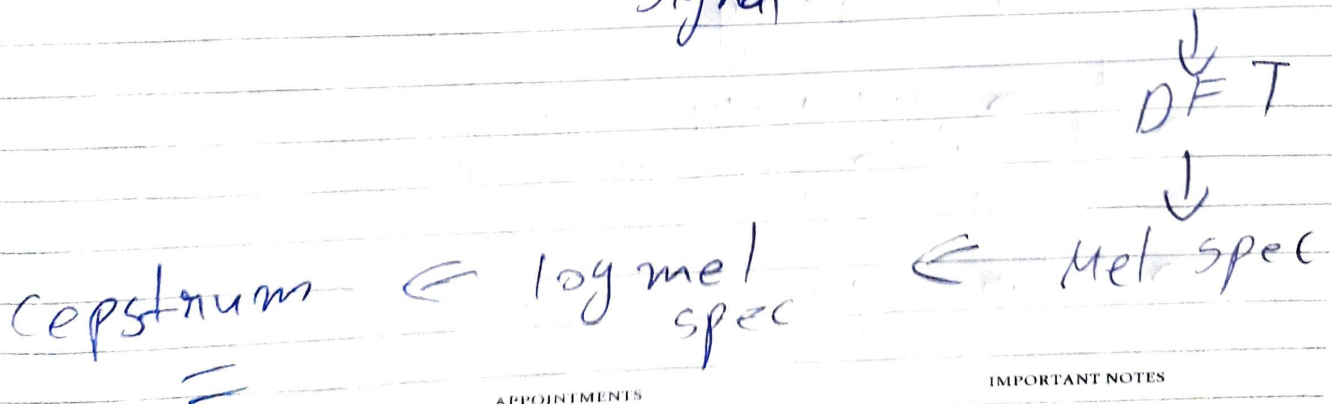


ASR:- (Automatic speech recognition)

- The goal of ASR is to go through a audio file and extract text out of it.
- To detect this we need to understand how speech is produced.



- As speech is a non-stationary signal we can extract using cepstral analysis.
- segment of speech → preemphasized signal → windowed signal



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MARCH

Tuesday

2021

14th Wk • 089-276

FEBRUARY

2021 MARCH

WK	S	M	T	W	T	F	S	WK	S	M	T	W	T	F	S
5		1	2	3	4	5	6	9		1	2	3	4	5	6
6	7	8	9	10	11	12	13	10	7	8	9	10	11	12	13
7	14	15	16	17	18	19	20	11	14	15	16	17	18	19	20
8	21	22	23	24	25	26	27	12	21	22	23	24	25	26	27
9	28							13	28	29	30	31			

The problem with cepstrum is that we can get complex numbers out of it.

so we use MFCC features to get the low frequency region we want

It uses discrete cosine transform at the end of analysis instead of inverse Fourier transform

13 → dimension MFCC can capture most of the timbre.

Δ MFCC → 13 dimension

$\Delta\Delta$ MFCC → 13 dimension

Total MFCC → 39 dimension

Hidden Markov model:-

→ Markov property satisfaction:-

→ It is memory less.

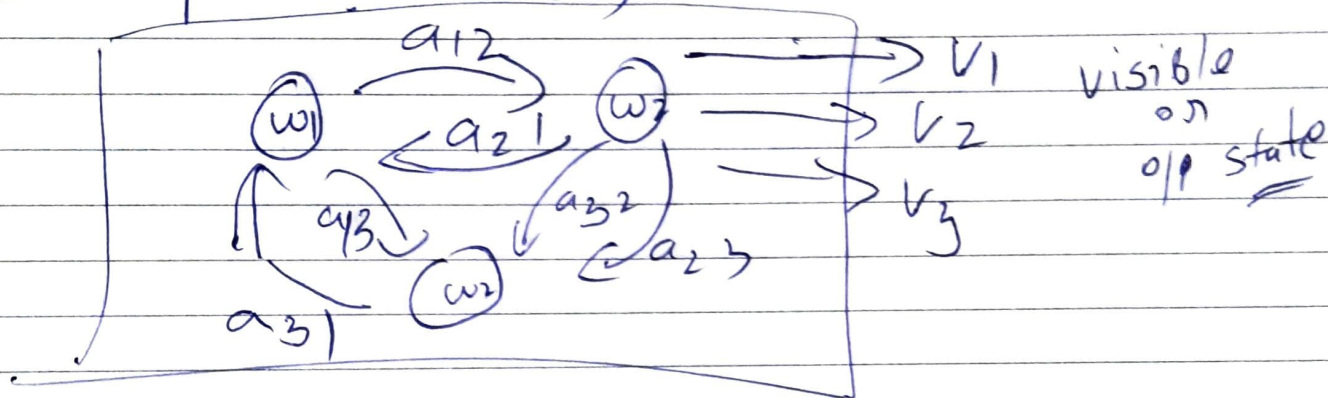
→ It takes only one state in consideration.

→ future prediction relies on present

State.

→ It is like finite state machine.

- ① Hidden states:- (w)
- ② output states (v)



Input → d

Phoneimes → /t/ /θ/ /d/

↳ In my model.

so it most closely
related to /d/

then it checks by ~~sim~~ using this state
what will be the next probable sound
It is most likely going to be a vowel
like dog, disk, dank etc.