

ECF439 Lecture

Friday March 20, 2025

New topic: speech processing

VOIP - current techniques are quite primitive

Module 4.1.1

1. Analysis - only

- a. speech recognition
- b. speaker verification
- c. speaker identification

2. Synthesis - only

- a. speech-based human-computer interaction
- b. customer support via phone
- c. text-to-speech
(useful for visually impaired individuals)

3. Analysis followed by synthesis

a. digital telephony

- 1. VOIP
- 2. mobile telephony

3. landline telephony

b. speech recording & playback

c. interactive speech systems

1. Alexa
2. Siri

Speech processing approaches

1. Waveform coding

doesn't take into account specific features of speech i.e. works for any audio waveform

examples:

a. PCM

- i) quantize
- ii) convert to binary

b. DPCM

c. ADPCM

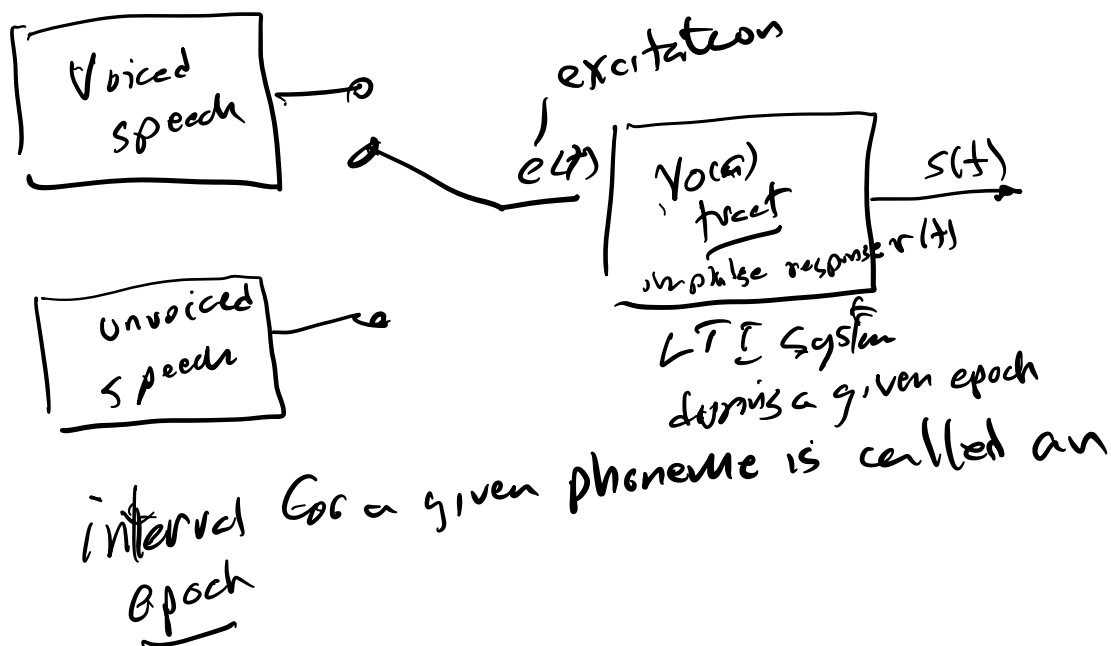
d. Filterbanks and wavelets

I will discuss this in ECE 439

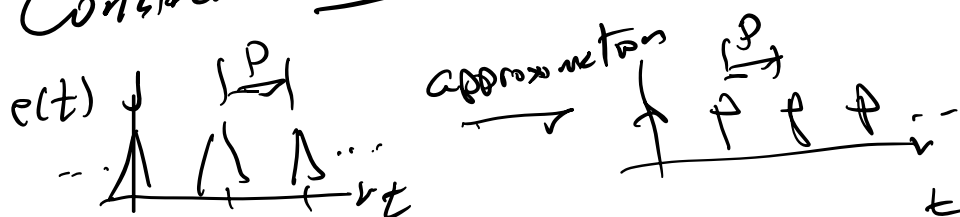
2. Speech-based models

- a. exploit specific characteristics of speech (spoken English)

Simple model for spoken languages (primarily English)



Consider a voiced phoneme



$$e(t) = \text{rep}_P[s(t)]$$

$$E(f) = \frac{1}{P} \text{comb} \frac{1}{P} [S] = \frac{1}{P} \sum_{k=-\infty}^{\infty} S(f - \frac{k}{P})$$

$$\Rightarrow s(t) = v(t) * e(t)$$

\uparrow vocal tract impulse response

$$* s(t) = \sum_{k=-\infty}^{\infty} v(t - kP)$$

1.1

$$k = \omega$$

$$S(f) = V(f)E(f)$$

$$= \frac{1}{P} \sum_{k=-\infty}^{\infty} V\left(\frac{k}{P}\right) \delta\left(f - \frac{k}{P}\right)$$

$$= \frac{1}{P} \text{comb}_{\frac{1}{P}} \left[V(f) \right]$$

