SGN-14007 Introduction to Audio Processing

Exercise 5 Week 48

In this exercise, you will implement one of the commonly used features in audio signal processing - **MFCC**. There are two questions with 1 point for each summing up to 2 points. Bonus question is optional worthing 0.5 points. The submission should consist of a **report** of your observations and the python **code**. Alternatively, both can be combined in a Jupyter notebook as well.

Question 1: Mel filterbank (1 point)

- (a) Load the given audio file (audio.wav) to get the audio signal and its sampling rate. Hint: librosa.load('audio.wav')
- (b) Create a mel filterbank with the following settings:
 - Sampling rate obtained from (a)
 - N fft = 512
 - N mel = 40

Hint: librosa.filters.mel(sr,n fft,n mel)

(c) Plot the mel filterbank and report your observations. Hint: librosa.display.specshow

Question 2: MFCC (1 point)

- (a) Pre-emphasis the audio signal (audio.wav) with the given Equation $y(t) = x(t) \alpha x(t-1)$, $\alpha = 0.97$. Hint: $np.append(x[0], x[1:] - \alpha * x[:-1])$
- (b) In the stft loop you implemented earlier, for each frame, do the following:
 - Window each frame (using hamming window) Hint: signal.hamming
 - Calculate the fft
 - Collect the power spectrum (you will get power spectrum)
 - Multiply it with mel filterbank you created in Question 1 (you will get mel spectrum)
 - Take log operation (you will get logarithmic mel spectrum) Hint: 20 * np.log10
 - DCT (Finally, you will get MFCC)
- (c) Plot logarithmic power spectrums, mel spectrums, logarithmic mel spectrums, MFCC.
- (d) Implement MFCC using librosa and compare with yours. Report your observations.

Bonus Question: Implement your own mel filterbank with the same setting as Question 1 (0.5 point)

In Question 1, you used librosa library to get mel filterbank. Now try to implement by yourself!

Hint:

- convert hz to mel using Equation $F_{mel} = 2595 log_{10} (1 + f/700)$
- uniformly distributed on the mel scale
- convert mel to hz back using the inverse of Equation above
- Mel filter of band m:
 - 1. starts at 0 amplitude at $F_{mel,c}$ (m-1)
 - 2. has maximum amplitude 1 at $F_{mel,c}$ (m)
 - 3. decays to zero at $F_{mel,c}$ (m+1)

You can follow the equation below.

$$H_m(k) = \begin{cases} 0 & k < f(m-1) \\ \frac{k - f(m-1)}{f(m) - f(m-1)} & f(m-1) \le k < f(m) \end{cases}$$

$$1 & k = f(m)$$

$$\frac{f(m+1) - k}{f(m+1) - f(m)} & f(m) < k \le f(m+1)$$

$$0 & k > f(m+1)$$

- to have a flat spectrum in mel domain for a DFT magnitude spectrum, the mel bands need to be scaled

Plot it and compare with librosa implementation. Report your observations.