F74056247 曾大瑋

程式語言使用 python

程式檔案是使用 jupyter-notebook 所寫,建議用 jupyter-notebook 開啟

作業使用 window 為 rectangular window

Code:

讀入所需的 library 及檔案

```
import matplotlib.pyplot as plt
import scipy.fftpack as sf
from scipy.io import wavfile
import os
import numpy as np

files = os.listdir("./")
samplerate, data = wavfile.read("./sample.wav")
print("Sample rate at "+str(samplerate)+" Hz")
print("Have "+str(len(data))+" samples")
```

音訊倍率調整

把音訊的範圍調整到+1.0~-1.0

並且畫出波型圖

```
samples=[]
temp=[]
for item in range(len(data)):
    samples.append(data[item])

for item in range(len(samples)):
    temp.append(abs(samples[item]))

maxium=max(samples)

for item in range(len(samples)):
    samples[item] = samples[item]/maxium

temp=np.sum(samples)/len(samples)
for item in range(len(samples)):
    samples[item]-=temp
```

```
plt.figure(figsize=[20,6])
plt.plot(samples)
plt.title('Waveform')
plt.xlabel('Sample')
plt.show()
```

計算 short Term energy

先計算出有幾個 frame 數

chunk 則是儲存 frame 用,每 chunk 儲存 160samples= 1 frame

再畫出 short time energy

```
sampsPerMilli = int(samplerate / 1000)
msFrame = 20
samFrame = sampsPerMilli * msFrame
frameNumber = (len(samples)-len(samples)%samFrame) / samFrame
drop=len(samples)%samFrame
# drop last frame part to prevent overlap
if drop>0:
    samples=samples[:-1*drop]
print ('Have ' + str(sampsPerMilli) + ' samples per ms')
print ('Therefore '+ str(samFrame) + ' samples in 1 frame(1 frame= '+
str(msFrame) +' ms )')
print ('Have '+str(int(frameNumber)) + ' number of frames ')
Have 8 samples per ms
Therefore 160 samples in 1 frame(1 frame= 20 ms)
Have 168 number of frames
#function to split chunks
def chunks(l, n):
    for i in range(0, len(l), n):
        yield l[i:i + n]
# calculat short term energy
def shortTermEnergy(frame):
    return sum( [ abs(x)^{**2} for x in frame ] ) / len(frame)
x=list(chunks(samples, 160))
STE = [] # list of short-time energies
for item in range(len(x)):
    STE.append(shortTermEnergy(x[item]))
plt.figure(figsize=[20,6])
plt.plot(STE)
plt.title('Short-Term Energy')
plt.xlabel('frame')
plt.show()
```

計算 Zero crossing rate

使用 numpy 的 np.sign 取出當前 frame 內每個 sample 是正值或負值

總和出當前 frame 的 zero crossing rate

計算 end point

先算出每個 frame 的 volume

用當前音量的大小判定 end point

取當前平均差 0.1 標準差當作是 start point 及 end point

紅色線為 start point 綠色為 end point

```
volume=[]
def calculateVolume(frame):
    result=frame-np.median(frame)
    return np.sum(np.abs(result))
for item in range(len(x)):
    volume.append(calculateVolume(x[item]))
startXposList=[]
endXposList=[]
isStart=False
avgVol=sum(volume)/len(volume)
stdVol=np.std(volume)
for item in range(len(volume)-1):
    if volume[item+1]>avgVol+stdVol*0.1 and isStart==False:
        isStart=True
        startXposList.append((item))
    if volume[item+1] < avgVol-stdVol*0.1 and isStart == True:</pre>
        isStart=False
```

```
endXposList.append((item))
```

```
plt.figure(figsize=[20,6])
plt.title('Volume for endPoint detection')
plt.xlabel('frame')
plt.plot(volume)

for item in range(len(startXposList)):
    #print(startXposList[item])
    plt.plot([startXposList[item], startXposList[item]],[0,20],'r', lw=1)

for item in range(len(endXposList)):
    #print(endXposList[item])
    plt.plot([endXposList[item], endXposList[item]],[0,20],'g', lw=1)
```

Pitch

對每個 frame 做 fourier transform 取得 Spectrum

取出其中最大值作為此 frame 的 pitch

```
voiceVector = []
y=np.array(x)
# low pass and high pass filter
Low_cutoff=80
High_cutoff= 300
F_sample=20
for window in range(len(y)):
    temp=np.array(y[window])
    M = temp.size
    Spectrum = sf.rfft(temp, n=M)
    [Low_cutoff, High_cutoff, F_sample] = map(float, [Low_cutoff, High_cutoff,
F_sample])
    #Convert cutoff frequencies into points on spectrum
    [Low_point, High_point] = map(lambda F: F//F_sample * M, [Low_cutoff,
High_cutoff])
    voiceVector.append(max(Spectrum))
plt.figure(figsize=[20,6])
plt.plot(voiceVector)
plt.title('Pitch')
plt.xlabel('frame')
plt.show()
```

Result:

