DSP LAB 9

1.Abstract

End-Point Detection, EPD (端點檢測)是用來取得信號語音開始以及結束的位置,所以又可以稱為Speech Detection或Voice Activity Detection(VAD)。常見端點檢測的方法大多是使用音量以及time domain的方式檢測,但各有利弊。

2.Goal of this Lab:

本次 Lab 的目的是要detect出笑聲的波段或是特別音訊的波段並且將各個部份擷取下來成各個音檔,藉由選擇不同的閾值,可以將笑聲/數字的頻段擷取下來,並將笑聲/數字的頻段完整的擷取下來。

3.Method

以這次實驗來說,使用音量大小來當作評估依據,透過Energy大小以及閾值的選擇來濾掉非笑聲的區段,再透過index相距等於一的區塊,把這些區塊合併成一塊,以及分開那些index相距大於1的區塊,將相鄰笑聲的區塊合併並且把不相鄰的笑聲區塊切割出來,最後透過各區間內點的數量大於10認定它為笑聲並存下來。

4. Pseudo code

取得 audio的所有資料點, audio 的sample rate

frameSize = 2048 overLap = 1024

use CALenergy split data in multiple frame and generate Energy list

CAlEnergy function

Step = frameSize - overLap
Framenum = ceil((len(data)) /Step)
for i in Framenum:
 extract frame which is in ith block on data
 E[i]=sum(frame * frame)
Return E

Filter_array = energy內所有array大於threshold的index Split_array = consecutive(Filter_array)

threshold1 = max(Audio_Energy)*0.1 threshold2 = min(Audio_Energy)*5000.0 threshold3 = min(Audio_Energy)+max(Audio_Energy)*0.0005 threshold4 = np.sqrt(max(Audio_Energy))*450

Consecutive function

```
if |Filter_array[a+1]-Filter_array[a] | !=1
add the (a+1) to the variable temp
```

Use np.split to generate every block in data by the temp

#target is to merge the consecutive indexes to be a one block and split nonconsecutive index to become two different array.

```
plot(audio_data)
for i in split_array:
    if len(i) >10:
        I[0] -> start line index,i[-1] -> end line index.
        plot(start line),
        plot(end line)
        store(i)
```

A is the length of energy_array

zeros = initialize a array have length A and element are all 0

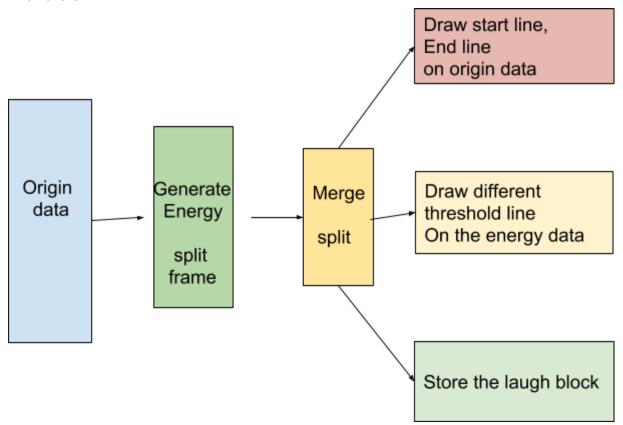
Plot (theshold1) use zeros

Plot (theshold2) use zeros

Plot (theshold3) use zeros

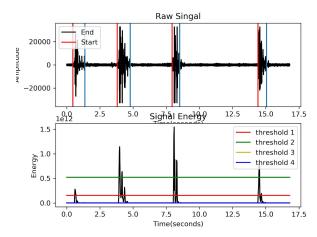
Plot (theshold4) use zeros

4.flowchart

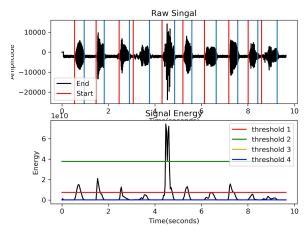


5.Result

Laughter(demo)



數字1-9(sample_audio.wav)



數字0-9(sample_audio_1.wav)

