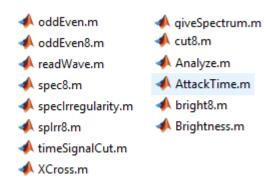
DSP - Recognition of feature of sound

Final MATLAB Project: 'Analyze'

The coded project's purpose is to compute unique parameters of sound, in order to use them to compare different sounds. The sound features included in the project are: attack time, xcrossing, brightness, spectrum irregularity, odd and even components of the spectrum. The first two features are calculated on time domain, while the others are calculated on frequency domain, on 8 different parts of the time domain.

1. The coded functions of the project:



2. The functions explained:

The main function Analyze():

This is the main function of the project, here is where are called all the functions needed to compute the values of the features. The input values needed to be insert by user are: the file, which is the audio file where the descriptors are calculated, the 'cut percent', which is the threshold of the cut which decide which part of the audio wave is interesting to be computed, and the 'resolution coefficent', which set the ratio of the resolution of the spectrum.

The output values of the function are:

- 1. attack time from channel 1,
- 2. attack time from channel 2,
- 3. x crossing from channel 1,
- 4. x crossing from channel 2,
- 5. 8 values vector of brightness from channel 1 split in 8 parts spectrum,
- 6. 8 values vector of brightness from channel 2 split in 8 parts spectrum,
- 7. 8 values vector of spectrum irregularity from channel 1 split in 8 parts spectrum,
- 8. 8 values vector of spectrum irregularity from channel 2 split in 8 parts spectrum,
- 9. 8 values vector of odd component of the spectrum from channel 1 split in 8 parts spectrum,
- 10. 8 values vector of odd component of the spectrum from channel 2 split in 8 parts spectrum,
- 11. 8 values vector of even component of the spectrum from channel 1 split in 8 parts spectrum,
- 12. 8 values vector of even component of the spectrum from channel 2 split in 8 parts spectrum.

```
function [attackTime1, attackTime2, xcross1, xcross2, vectBR1, vectBR2, vectIr1,
vectIr2, oddVector1, oddVector2, evenVector1, evenVector2] = Analyze (file,
cutPerc, coeffResol)

[ch1, ch2, fs]=readWave(file);
sig1=timeSignalCut(ch1, cutPerc);
sig2=timeSignalCut(ch2, cutPerc);
attackTime1=AttackTime(sig1, fs);
attackTime2=AttackTime(sig2, fs);
xcross1=XCross(sig1, 500);
xcross2=XCross(sig2, 500);
cutVect1=cut8(sig1);
cutVect2=cut8(sig2);
```

```
logSpectMatr1=spec8(sig1, cutVect1, fs*(coeffResol));
logSpectMatr2=spec8(sig2, cutVect2, fs*(coeffResol));
vectBR1=bright8(logSpectMatr1);
vectBR2=bright8(logSpectMatr2);
vectIr1=spIrr8(logSpectMatr1);
vectIr2=spIrr8(logSpectMatr2);
[oddVector1, evenVector1]=oddEven8(logSpectMatr1);
[oddVector2, evenVector2]=oddEven8(logSpectMatr2);
end
```

readWave():

This function called on an audio file returns the two channels(ch1, ch2) and the sampling frequency(fs) of the file.

```
function [ch1, ch2, fs] = readWave(track)
[Y, fsTMP] = audioread(track);
ch1=Y(:,1);
ch2=Y(:,2);
fs=fsTMP;
end
```

timeSignalCut():

This function from a channel(ch) and a percent value(perc) returns the cutted channel, which is the signal of interest(sig), and the two indexes where the channel had been cut, the beginning(Begin) and the ending(End) of the signal of interest on the channel.

The algorithm find these two indexes with the first two loops and set the signal as the channel between the two indexes.

```
function [sig,Begin,End] = timeSignalCut(ch, perc)
for i=1:length(ch)
    if ch(i) \ge max(ch) * (perc)
        break;
    end
end
Begin=i;
i=length(ch);
while i>2
    if ch(i) \ge max(ch) * (perc)
        break;
    end
    i=i-1;
end
End=i;
sig=ch (Begin:End);
end
```

AttackTime():

This function from a signal and its sampling frequency calculates the attack time. The attack time is given by the ratio between the number of samples from the start of the signal to the max value of the signal, divided by the sampling frequency.

```
attackTime[s] = \frac{\#samples\ from\ sig(1)\ to\ max(sig(:))}{sampling\ frequency[\frac{1}{S}]}
```

```
function [AtTime] = AttackTime(sig, fs)
```

```
[Ymax,i]=max(sig);
AtTime=i/fs;
end
```

XCross():

This function from a signal and the desired interval calculate the number of xcrossing of the signal after the max value. E.g. if the interval is set to 500, it calculates the xcrossing value on the 500 values of the array after the max value. Then it returns the value of the feature.

```
function [xcross] = XCross(sig, intv)
A=0;
[ymax,i]=max(sig);
if intv<=1</pre>
    intv=length(sig)*1/10;
end
if i+intv>=length(sig)
    intv=length(sig)-i;
end
for j = i : i+intv
     if sig(j)<0 && sig(j+1)>0
       A=A+1;
     end
     if sig(j)>0 && sig(j+1)<0
       A=A+1;
     end
end
xcross=A;
end
```

cut8():

This function from a signal inside and array it returns an array of indices of the array, which indices divide the signal in 8 same length parts.

```
function [cutVect] = cut8(sig)

tmp=length(sig)/8;
cutVect=[1,tmp,2*tmp,3*tmp,4*tmp,5*tmp,6*tmp,7*tmp,length(sig)];
end
```

> spec8():

This function from: a signal, the cut vector and the resolution, create a matrix with 8 rows, where in each row there is the spectrum of the correspondent part of the signal in the time domain, described by the cut vector, with the inserted resolution.

```
function [logSpec] = spec8(sig, cutVect, resolution)

logSpec(1, :)=giveSpectrum(sig(cutVect(1):cutVect(2)), resolution);
logSpec(2, :)=giveSpectrum(sig(cutVect(2):cutVect(3)), resolution);
logSpec(3, :)=giveSpectrum(sig(cutVect(3):cutVect(4)), resolution);
logSpec(4, :)=giveSpectrum(sig(cutVect(4):cutVect(5)), resolution);
logSpec(5, :)=giveSpectrum(sig(cutVect(5):cutVect(6)), resolution);
logSpec(6, :)=giveSpectrum(sig(cutVect(6):cutVect(7)), resolution);
logSpec(7, :)=giveSpectrum(sig(cutVect(7):cutVect(8)), resolution);
logSpec(8, :)=giveSpectrum(sig(cutVect(8):cutVect(9)), resolution);
```

end

giveSpectrum():

This function from a time domain signal and the resolution value, returns the frequency domain spectrum normalized to 3, and splitted in the half, with the given resolution.

```
function [logSpec]=giveSpectrum(sig,Res)
tmp=zeros(1,Res);
for i=1:length(sig)
    tmp(i)=sig(i);
end
spec=abs(fft(tmp));
logSpec=log10(1000*spec/max(spec));
logSpec=logSpec(1:floor(length(logSpec)/2));
end
```

bright8():

This function from the matrix that contains the spectrums of the parts of the signal, create a vector which contains the values of brightnesses of the spectrums(brVect).

```
function [brVect] = bright8(logSpecMatr)
brVect=[0,0,0,0,0,0,0];
[x,y]=size(logSpecMatr);
for i=1:x
    brVect(i)=Brightness(logSpecMatr(i,:));
end
end
```

BrightNess():

This function from a spectrum calculates the value of the Brightness of that. The formula of the brightness is:

```
function [BR]= Brightness(logSpec)

den=0;
num=0;
for i=1 : length(logSpec)
    add=(logSpec(i)^2)*i;
    num=num+add;
end
add=0;
for i=1 : length(logSpec)
    add=logSpec(i)^2;
    den=den+add;
end
BR=num/den;
```

$$Br = \frac{\sum_{i=0}^{n} A(i) \cdot i}{\sum_{i=0}^{n} A(i)}$$

> spIrr8():

This function from the matrix which contains the 8 spectrums of the parts of the signal returns an 8-elements vector with the values of the spectrum irregularity of each part.

```
function [IrrVect] = spIrr8(logSpecMatr)
```

```
IrrVect= [0, 0, 0, 0, 0, 0, 0];
[x, y]=size(logSpecMatr);
for i=1:x
    IrrVect(i)=specIrregularity(logSpecMatr(i, :));
end
```

specIrregularity():

This function calculates the spectrum irregularity of the given spectrum. The spectrum irregularity formula is:

oddEven8():

This function from the matrix which contains the 8 spectrums of the parts of the signal returns two 8-elements vectors. The first "oddVector" contains the values of the ratio of the odd components of each of the 8 spectrums. The second "evVector" contains the values of the ratio of the even components of each of the 8 spectrums.

```
function [oddVector, evVector] = oddEven8(logSpecMatr)

oddVector=[0, 0, 0, 0, 0, 0, 0, 0];
evVector=[0, 0, 0, 0, 0, 0, 0];
[x,y] = size(logSpecMatr);
for i = 1:x
        [oddVector(i), evVector(i)] = oddEven(logSpecMatr(i, :));
end
```

oddEven():

end

This function calculates two values: 'odd'(the ratio of the odd components of the spectrum) and 'ev'(the ratio of the even components of the spectrum). These are the formulas of the two descriptors:

```
function [odd, ev]= oddEven(logSpec) Ev = \frac{\sqrt{\sum_{i=1}^{N} A_{2i}^{2}(i)}}{\sqrt{\sum_{j=1}^{N} A_{j}^{2}(j)}} den=0; for j=1:length(logSpec) den=(logSpec(j)^{2})+den; end den=den^{(1/2)}; evNum=0; for i=2:2:length(logSpec) evNum=evNum+(logSpec(i)^{2}); end evNum=evNum^{(1/2)}; oddNum=0; for i=1:2:length(logSpec) oddNum=oddNum+(logSpec(i)^{2});
```

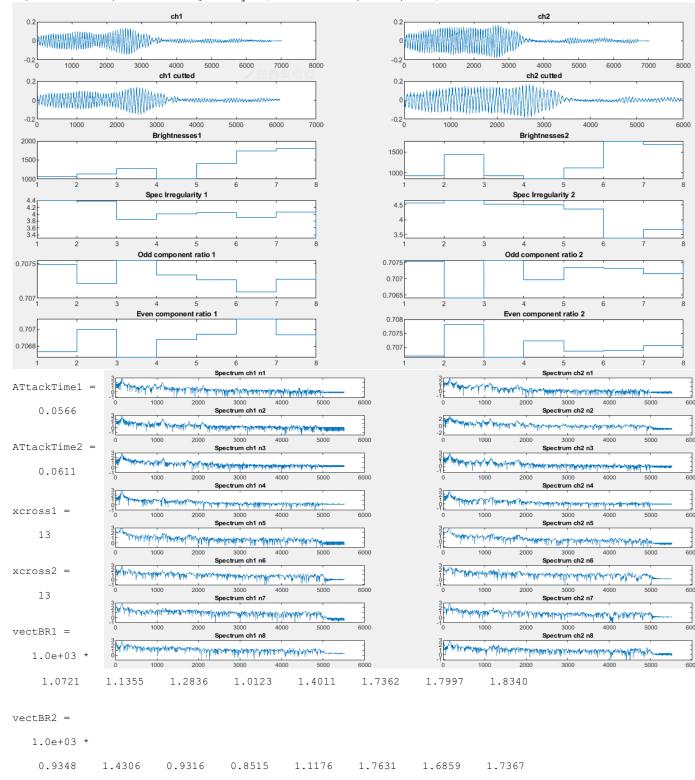
oddNum=oddNum^(1/2);
ev=evNum/den;
odd=oddNum/den;

end

3. Results with the given sound waves:

On cuckoo.wav:

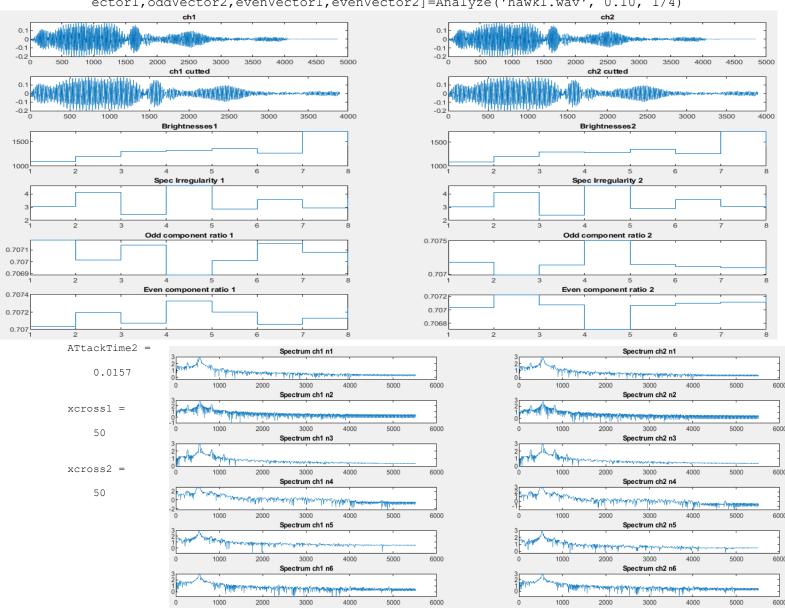
[ATtackTime1,ATtackTime2,xcross1,xcross2,vectBR1,vectBR2,vectIr1,vectIr2,oddVector1,oddVector2,evenVector1,evenVector2]=Analyze('cuckoo.wav', 0.10, 1/4)



<pre>vectIr1 =</pre>											
	4.4047	4.3767	3.8537	4.0038	4.0367	3.8975	4.0640	3.2969			
<pre>vectIr2 =</pre>											
	4.5792	4.6618	4.5282	4.5160	4.3688	3.3756	3.6698	3.8491			
oddVector1 =											
	0.7075	0.7072	0.7075	0.7073	0.7073	0.7071	0.7073	0.7073			
oddVector2 =											
	0.7075	0.7064	0.7076	0.7070	0.7073	0.7073	0.7072	0.7074			
evenVector1 =											
	0.7067	0.7070	0.7067	0.7069	0.7069	0.7071	0.7069	0.7069			
evenVector2 =											
	0.7067	0.7078	0.7066	0.7072	0.7069	0.7069	0.7071	0.7068			

On hawk1.wav:

[ATtackTime1, ATtackTime2, xcross1, xcross2, vectBR1, vectBR2, vectIr1, vectIr2, oddV ector1, oddVector2, evenVector1, evenVector2] = Analyze('hawk1.wav', 0.10, 1/4)



	Spectrum ch1 n7					Spectrum ch2 n7					
	Eldin, Mil	الالمعتقد أرمانه بليمي	للمناه بمراطاته التنط		.]	i ilin	ملخيالهم المهمور وال	واساداسار وبيدار	والمناهب للبيالي	spel-	
	0 1	000 2000	3000 Spectrum ch1 n8	4000 5000	6000	0	1000	2000 3000 Spectrum o	4000 h2 n8	5000	60
	والملهمان	فأسارناه اللخرموس		andkaditeritari		2	hermanhly		dingh kelasa sebi	Mulas	-
vectBR1 =	0 1	000 2000	3000	4000 5000	6000	0	1000	2000 3000	4000	5000	60
1.0e+03 *											
1.0955	1.2018	1.2931	1.3242	1.3582	1.2635	1.7099	1.4152				
vectBR2 =											
1.0e+03 *											
1.0867	1.2000	1.2929	1.2893	1.3588	1.2642	1.7313	1.4115				
vectIr1 =											
3.0745	4.1076	2.4237	4.6369	2.8372	3.5720	2.9555	3.7048				
vectIr2 =											
3.0142	4.1043	2.4148	4.6384	2.8991	3.5661	3.0524	3.7274				
oddVector1 =											
0.7072	0.7070	0.7071	0.7069	0.7070	0.7072	0.7071	0.7070				
oddVector2 =											
0.7072	0.7070	0.7071	0.7075	0.7071	0.7071	0.7071	0.7070				
evenVector1 =	=										
0.7070	0.7072	0.7071	0.7073	0.7072	0.7071	0.7071	0.7072				
evenVector2 =	=										
0.7070	0.7072	0.7071	0.7067	0.7071	0.7071	0.7071	0.7072				
N 0 = 1 = 1 = 1 = 1 = 1 = 1 = 1 = 1 = 1 =	4 -										

On hawk1.wav:

[AT tack Time 1, AT tack Time 2, xcross 1, xcross 2, vect BR 1, vect BR 2, vect Ir 1, vect Ir 2, odd Vector 1, odd Vector 2, even Vector 1, even Vector 2] = Analyze ('hawk 2. wav', 0.10, 1/4)

