Homework 4 - Student Report CMPE 362 Spring 2016

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1 Introduction

The topic of the of homework was quite interesting and has usage in real world. So it was important for me to do this project as good as i can. It helped to have an idea about data hiding techniques in general. But in bit setting and getting parts were challenging for me.

2 Questions & Answers

2.1 Watermark on Time Domain

In the code i tried to implement given procedure. Though i did not implement same algo, my implementation was quite similar and i think i get nearly expected working script. Result of the program which is 'lsb_watermarked.wav' audio file unfortunately is "wizzy". I think it is because frequency hopping.

- I give 2 points for the MOS value.
- My SNR value was 1.454.

I think my SNR value is consistent with wizzy audio file. It should be higher to hear more clear sound. And i think I gave consistent MOS value to watermarked audio, it is also low.

```
6 file_name='son.wav';
   [cover_audio,Fs]=audioread(file_name);
  file_name='wm.bmp';
9 message=imread(file_name);
10
11
12 message=double(message);
message=round(message./256);
  message=uint8(message);
14
15
16
17 Mc=size(cover_audio,1);
18
19
   Mm=size(message, 1);
20
   Nm=size(message, 2);
21
   message1=reshape(message,1,[]);
22
23
27
28
29
30
31 for ii = 1:4:Mc
            watermark1(ii) = message1(mod(ii, 4) + 1);
32
33 end
  watermarked audio=cover audio:
34
35
```

```
for ii = 1:4:Mc
39
            watermarked_audio(ii) = bitset(cast(watermarked_audio(ii), 'uint8'), 1, watermark1(ii));
   end
40
   audiowrite('lsb_watermarked.wav', watermarked_audio,Fs);
41
42
43
44
45
46
   file_name='lsb_watermarked.wav';
47
   [watermarked_audio,Fs] = audioread(file_name);
48
49
50
51 Mw=size(watermarked_audio,1);
52
53
  for ii = 1:4:Mw
            watermark(ii) = bitget (cast (watermarked_audio(ii), 'uint8'), 1);
   end
56
57
   watermarkex= reshape(watermark, 64, 64);
62
63
64
65
66
67  s = mean(cover_audio.^2)/mean((cover_audio-watermarked_audio).^2);
  snr = 10.^log(s);
```

2.2 Watermark on Wavelet Domain

In this question i was thinking it is like first question but it was different. I searched for Haar Wavelet Transform on the net and i get some informations. But i could implement it to some point. After getting transform values i could not handle how to arrange data at the step 7, so my script implements less than half of the procedure and unfortunately gets no result. However i included my incomplete code.

2.3 Watermark on Frequency Domain

In this question i tried to implement my own algorithm that looks for peak values of fft transform and puts bits accordingly. For the algorithm we need to have bins(blocksize) so as to determine bit embed cycle, also together with this we should select peak frequency. I tried determine these as looking equal loudness curve but i could not understand form the figure and it happened to be different values. As too look in algorithm, first, I determined my bins as 8820 which is 0.2 second because frequency of the wav file is 44100 Hz. I chose peak frequency of watermark embedding as 20 Hz('wm_peak'), 'amply' is the amplification of the watermark.

After preprocessing i scaled '.bmp' file as '.wav' file and put in a array. Then in the for loop as increasingly size of the block determined, i first take fft of sound then i look different peak values to encode the figure bits. Bit is embedded if fft peaks is lower than the peak i determined and with amplification 'amply' determined. After that i take inverse fft of it and write the file.

I get quite good sound with very little noise because i got fairly good SNR value.

- I give 4 points for the MOS value.
- My SNR value was 15.07.

```
audio_input='son.wav';
   watermark_input='wm.bmp';
6
   output_marked='freqwatermarked.wav';
7
   wm_output='wm_ex.bmp';
  blocksize=8820; % For a track with 44100 samples per second
10
11
12
   first_block=1; % to leave about 6 s empty
13
                    % corresponds to a low frequency peak for the embedding;
14
   wm_peak=20:
15
16
  amply=7;
17
18
19 [data,Fs]=audioread(audio_input);
20 data_orig=data; %hold value to calculate snr value
21 M=size(data,1);
22 N=floor(M/blocksize); % both channels are used for watermark embedding
```

```
24
25
26 wml=imread(watermark_input); % read watermark image
27 z1=size(wm1,1);
28 z2=size(wm1,2);
29
   scale=1:
   if(N<(z1*z2))
30
        scale=N/(z1*z2);
31
32
   wm_temp=mat2gray(wm1);
33
   if(scale~=1)
34
35
        wm_temp=imresize(wm_temp, scale);
36
37
38
  s1=floor(size(wm_temp, 1)/2)*2;
40 s2=floor(size(wm_temp, 2)/2)*2;
41 wm=imresize(wm_temp,[s1 s2]);
42 wm_length=s1*s2;
43 wm_vec=round(reshape(wm, wm_length, 1)./256);
44 if (wm_length>(N-first_block+1))
        error('Error: Watermark to long');
45
46 end
   wm_pad(1:length(wm_vec))=wm_vec; % watermark bytes in a vector
47
48
49
   1 = -1:
50
    \textbf{for } k=\texttt{first\_block:wm\_length/2+first\_block-1} \\  \  \, \textit{\$ k runs over the blocks} 
51
52
        x=data((k-1)*blocksize+1:k*blocksize,1);
53
        y=fft(x);
54
        mark=0;
55
56
        n=wm_peak;
57
          while (abs (y(n)) == abs(y(n+1)) \&\& mark == 0)
             n=n+1;
             if(n==floor(blocksize/2))
60
                 mark=1;
61
62
             end
          end
63
        if (n~=floor(blocksize/2))
64
65
66
67
          if(wm_pad(1) == 0)
68
69
                 if (abs(y(n)) < abs(y(n+1))) % change the values, if necessary
70
71
                     temp1=y(n+1);
                     y(n+1) = y(n);
72
                     y(n) = temp1;
                 end
75
                 y(n) = y(n) *amply;
                 y(blocksize-n+2)=conj(y(n));
76
77
                 y(blocksize-n+1)=conj(y(n+1));
78
79
          end
          elseif(wm_pad(1)==1)
80
81
              if (abs(y(n))>abs(y(n+1)))
                     temp1=y(n+1);
                                               % fft peaks have equal
82
                     y(n+1) = y(n);
83
                     y(n) = temp1;
84
```

```
end
86
            y(n+1)=y(n+1)*amply;
             89
       end
90
      \mbox{\ensuremath{\$}} inverse transform of the watermarked fft block
91
      data((k-1)*blocksize+1:k*blocksize,1)=ifft(y);
92
  end
93
94
95
96
97 audiowrite(output_marked,data,Fs);
103 s = mean(data_orig.^2)/mean((data_orig-data).^2);
104 snr = 10.^{\log(s)};
```