

---

## Table of Contents

Audio Compression using DCT .....	1
Output/Input Arguments .....	1
Flags .....	1
load audio .....	1
Play input audio .....	1
Pick N samples .....	2
Selecting the frames .....	2
Pick dominant coefficients .....	2
Pick first Coefficients of DCT .....	2
Reconstruction .....	2
test the reconstructions .....	2
Moving the window over the audio to pick next samples .....	3
Calculating SNR .....	3
play the compressed audio .....	3

## Audio Compression using DCT

```
function SNR=audiocompression(N,percentCoeff,mode)
```

## Output/Input Arguments

- SNR = signal to noise ratio
- N = size of window
- percentCoeff = percentatge of DCT coefficients to be picked
- mode = operation mode(0: picks first DCT coefficients, 1: picks dominant DCT coefficients)

## Flags

```
%mode0=pick first N% DCT coefficients
%mode1=pick Dominant N% DCT coefficients
%debug1=plots window level plots

debug=0;
soundFlag=0;
```

## load audio

```
pathToAudio='/Users/talha/Desktop/DSPlabexam/sample.wav';
[audio, samplingFrequency]=audioread(pathToAudio);
```

## Play input audio

```
if soundFlag ==1
```

---

```
        sound(audio, samplingFrequency)
    end
```

## Pick N samples

```
samplesToPick=1:N;
thisWindow=0;
reconstructedAudio=[];
```

*Not enough input arguments.*

*Error in audiocompression (line 31)*  
*samplesToPick=1:N;*

## Selecting the frames

```
while samplesToPick(length(samplesToPick)) <= length(audio)

    thisWindow=audio(samplesToPick);
    currentDCT=dct(thisWindow);
```

## Pick dominant coefficients

```
if mode==1

    [sortedDCT,ind] = sort(abs(currentDCT), 'descend');
    ignoredInd=ind(floor(percentCoeff*length(sortedDCT)):end);
    currentDCT(ignoredInd) = 0;
```

## Pick first Coefficients of DCT

```
elseif mode==0
    currentDCT(floor(percentCoeff*length(currentDCT)):end)=0;

end
```

## Reconstruction

```
reconstructedWindow = idct(currentDCT);
reconstructedAudio=[reconstructedAudio;reconstructedWindow];
```

## test the reconstructions

```
if(debug==1)
    plot(thisWindow);
    hold on
    plot(reconstructedWindow);
    pause(1)
    close all
end
```

---

## Moving the window over the audio to pick next samples

```
        samplesToPick=samplesToPick+N;  
    end
```

## Calculating SNR

```
difSig=audio(1:length(reconstructedAudio))-reconstructedAudio;  
SNR = 10*log10(sum(audio.^2)/sum(difSig.^2));
```

## play the compressed audio

```
if(soundFlag==1)  
    sound(reconstructedAudio, samplingFrequency)  
end  
end
```

*Published with MATLAB® R2015b*