

- Previous class:
 - Color vectors – RGB
 - 1-dimensional array – vector

- Now:
 - Play with sound files
 - (more vectors!)

Computing with sound requires digitization

- Sound is continuous; capture its essence by sampling
- Digitized sound is a vector of numbers

The diagram illustrates the digitization process. The top part shows 'Analog input' as a smooth green curve over 'Time'. Vertical lines mark 'Sample times'. The bottom part shows 'Digital output' as a red step function where the value is constant between sample times, representing the sampled values.

Sampling rate affects the quality

If sampling is not frequent enough, then the discretized sound will not capture the essence of the continuous sound...

The diagram shows two waveforms. The top one, labeled 'Too slow', has red dots representing samples that miss the peaks and troughs of the original wave. The bottom one, labeled 'OK', has green dots representing samples that accurately capture the shape of the wave.

Sampling Rate

Given human perception, 20000 samples/second is pretty good (20000Hz or 20kHz)

8,000 Hz	required for speech over the telephone
44,100 Hz	required for audio CD
192,400 Hz	required for HD-DVD audio tracks

Resolution also affects the quality

Typically, each sampled value is encoded as an 8-bit integer in the .wav file.

Possible values: -128, -127,...,-1,0,1,...,127

Loud: -120, 90, 122, etc.

Quiet: 3, 10, -5

16-bit used when very high quality is required.

Reading and playing .wav files

```
[y,rate,nBits] = wavread('austin.wav')
sound(y,rate)
```

A wav file is for the computer to process— software is required to play the sound.

Computing with sound in Matlab requires that we first convert the wav format data into simple numeric data—the job of `wavread`.

wavread converts the 8-bit values to floating point values between -1 and 1

```
[y,rate,nBits]= wavread('austin.wav')
```

```

0.4609
0.3516
0.2734
0.2891
0.2500
0.1484 ← y(50000:50012)
0.1094
0.1641
0.1484
0.0000
-0.1641
-0.2734
-0.3281
    
```

wavread

```
[y,rate,nBits]= wavread('austin.wav');
n = length(y);
```

```

n =
    54453
rate =
    11025
nBits =
     8
    
```

austin.wav encoded the sound with 54,453 8-bit numbers that were gathered over a span of about 54453/11025 secs

wavread

```
[data,rate,nBits]= wavread('austin.wav')
```

Name of the source file

The vector of sampled sound values is assigned to this variable

The sampling rate is assigned to this variable

The resolution is assigned to this variable

Hearing and "seeing" the sound

```
[y,rate]= wavread('austin');
sound(y, rate)
plot(1:length(y), y)
```

Usually playback at a rate equal to the sampling rate

Subvectors

- Can access just part of a vector, or subvector
- Suppose you have a vector **v**:
 - v(1)** - value in 1st cell
 - v(k)** - value in kth cell for valid k
 - v(2:5)** - the 2nd thru 5th values in **v**, as a vector
 - length(v)** - how many cells in vector **v**
 - v(1:length(v))** - all the values in **v**

Building a vector

Concatenate two vectors to make one...

```
v= ones(1,3); % a row of length 3
w= [4; 7]; % a column of length 2
x= [w; v']; % a column of length 5
```

Concatenate vectors repeatedly...

```
a= [];
for k= 1:4
    a= [a ones(1,2)];
end
% What is a?
```