

MATLAB Functionality for Digital Speech Processing

- MATLAB Speech Processing Code
- MATLAB GUI Implementations

Basic Functionality

- read a speech file (i.e., open a .wav speech file and read the speech sample into a MATLAB array)
- write a speech file (i.e., write a MATLAB array of speech samples into a .wav speech file)
- play a MATLAB array of speech samples as an audio file
- play a sequence of MATLAB arrays of speech samples as a sequence of audio files
- record a speech file into a MATLAB array
- plot a speech file (MATLAB array) as a waveform using a strips plot format
- plot a speech file (MATLAB array) as one or more 4-line plot(s)
- convert the sampling rate associated with a speech file (MATLAB array) to a different sampling rate
- highpass filter a speech file (MATLAB array) to eliminate hum and low frequency noise
- plot a frame of speech and its associated spectral log magnitude
- plot a spectrogram of a speech file (MATLAB array)
- plot multiple spectrograms of one or more speech files (MATLAB arrays)

Read a Speech File into a MATLAB Array

- `[xin, fs, nbits] = wavread(filename);`
- `[xin, fs] = loadwav(filename);`
 - filename is ascii text for a .wav-encoded file which contains a speech signal encoded using a 16-bit integer format
 - xin is the MATLAB array in which the speech samples are stored (in double precision format)
 - fs is the sampling rate of the input speech signal
 - nbits is the number of bits in which each speech sample is encoded (16 in most cases)
 - program `wavread` scales the speech array, xin, to range $-1 \leq xin \leq 1$, whereas `loadwav` preserves sample values of the speech file and hence array xin is scaled to range $-32767 \leq xin \leq 32767$
- `[xin1, fs, nbits] = wavread('s5.wav');`
- `[xin2, fs] = loadwav('s5.wav');`

Read a Speech File into a MATLAB Array

```

• % test_wavread.m
• % test_wavread function
• %
• % read speech samples from file 'test_16k.wav' into array x1 using wavread
• % routine
• filein='test_16k.wav';
• [x1,fs1,nbits]=wavread(filein);
•
• % print out values of fs1, nbits, wavmin1, wavmax1
• wavmin1=min(x1);
• wavmax1=max(x1);
• fprintf('file: %s, wavmin/wavmax: %6.2f %6.2f, fs1: %d, nbits: %d \n',...
• filein,wavmin1,wavmax1,fs1,nbits);
•
• % read speech samples from same file into array x2 using loadwav routine
• [x2,fs2]=loadwav(filein);
•
• % print out values of fs2, nbits, wavmin2, wavmax2
• wavmin2=min(x2);
• wavmax2=max(x2);
• fprintf('file: %s, wavmin/wavmax: %d %d, fs2: %d \n',...
• filein,wavmin2,wavmax2,fs2);

```

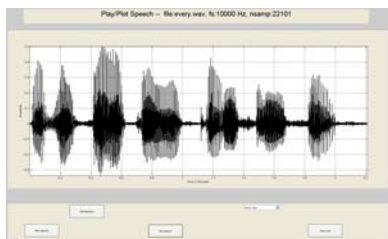
```

Terminal Display:
file: test_16k.wav, wavmin/wavmax: -1.00 1.00, fs1: 16000, nbits: 16
file: test_16k.wav, wavmin/wavmax: -32768 32767, fs2: 16000

```

Play/Plot Existing Speech File

- `Play_Plot_Speech_GUI.m`
 - MATLAB GUI for basic operations of reading in a file, playing the speech array, and plotting the speech waveform



Write a Speech Array into a Speech File

- `wavwrite(xout, fs, nbits, filename);`
- `savewav(xout, filename, fs);`
 - xout is the MATLAB array in which the speech samples are stored
 - fs is the sampling rate of the output speech signal
 - nbits is the number of bits in which each speech sample is encoded
 - filename is the ascii text for the .wav-encoded file in which the MATLAB signal array is to be stored
 - for `wavwrite` the MATLAB array xout needs to be scaled to the range $-1 \leq xin \leq 1$ whereas for `savewav` the MATLAB array xout needs to be scaled to the range $-32767 \leq xout \leq 32767$
- `wavwrite(xin1, fs, 's5out.1.wav');`
- `savewav(xin2, 's5out.2.wav', fs);`

Write a Speech Array into a Speech File

- % write out array x1 into speech file using wavwrite routine
- wavwrite(x1,fs1,nbits,'file1out.wav');
- % write out array x2 into speech file using savewav routine
- savewav(x2,'file2out.wav',fs2);



Play a Speech File

- sound(x, fs);
- soundsc(x, fs);
 - for sound the speech array, x, must be scaled to the range $-1 \leq x \leq 1$
 - for soundsc any scaling of the speech array can be used
 - fs is the sampling rate f the speech signal
- [xin, fs] = loadwav('s5.wav'); % load speech from s5.wav;
- xinn = xin/abs(max(xin)); % normalize to range of - 1 to 1;
- sound(xinn, fs); % play out normalized speech file;
- soundsc(xin, fs); % play out unnormalized speech file;

Play Multiple Speech Files

- play_multiple_files.m;
 - sequence of filenames read in via filelist, keyboard or file search
- Example of usage to play out 3 speech files in sequence:
 - kbe=filename entry via filelist(2), keyboard(1), or file search(0):1; % keyboard chosen
 - N=number of files to be played in a group:3; % play out 3 files
 - i=1; filename: s1.wav;
 - i=2; filename: s2.wav;
 - i=3; filename: s3.wav

Play Multiple Speech Files

- test_play_files.m
 - play the following sequence of files:

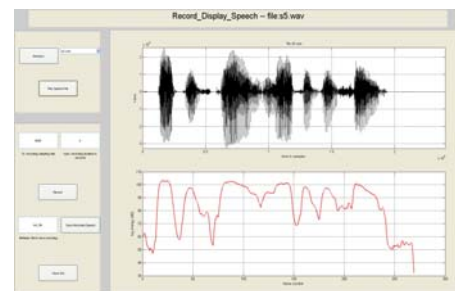
Maple_short.wav
s1.wav
beep.wav
test_16k.wav
beep.wav
s2.wav

Record Speech into MATLAB Array

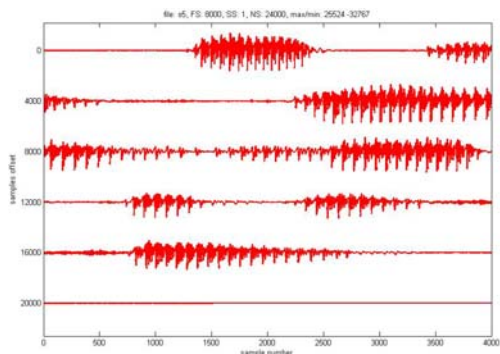
- record_speech.m (calls MATLAB function wavrecord.m)
- function y=record_speech(fs, nsec);
 - fs: sampling frequency
 - nsec: number of seconds of recording
 - y: speech samples array normalized to peak of 32767

Record Speech into MATLAB Array

- record_display_speech_GUI.m

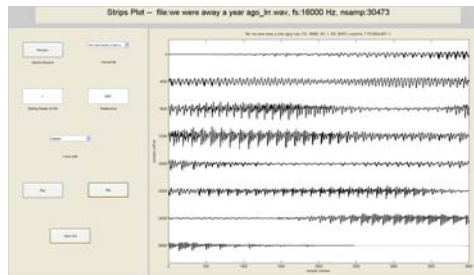


Plot Speech Using Strips Plot

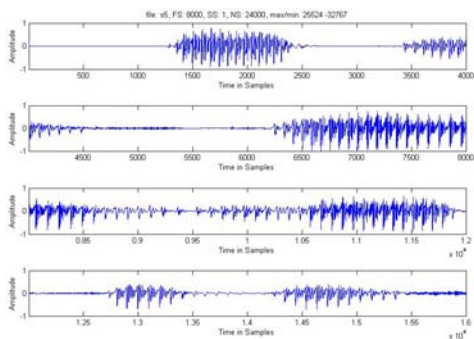


Plot Speech Using Strips Plot

- strips_plot_GUI.m



Plot Speech Using 4-Line Plot

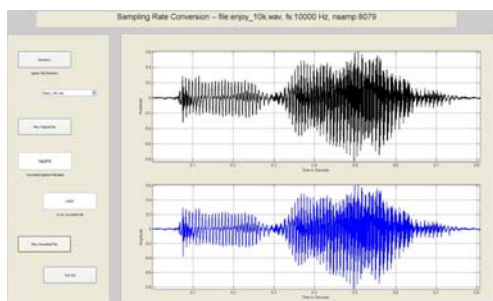


Sample Rate Conversion

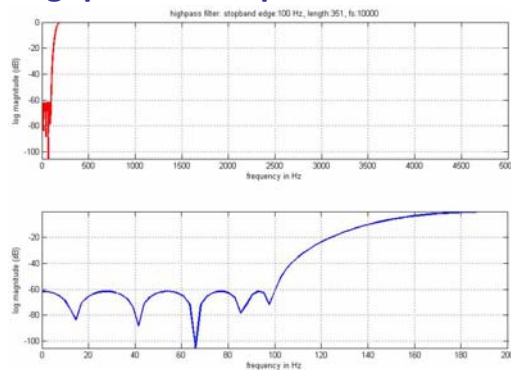
- $y = \text{srconv}(x, \text{fsin}, \text{fsout});$
 - x : input speech array;
 - fsin : input speech sampling rate;
 - fsout : desired speech sampling rate;
- Example:
 - $[\text{xin}, \text{fsin}] = \text{loadwav}('s5.wav');$ % $\text{fsin}=8000$;
 - $\text{fsout} = 10000;$ % desired sampling rate;
 - $y = \text{srconv}(\text{xin}, \text{fsin}, \text{fsout});$

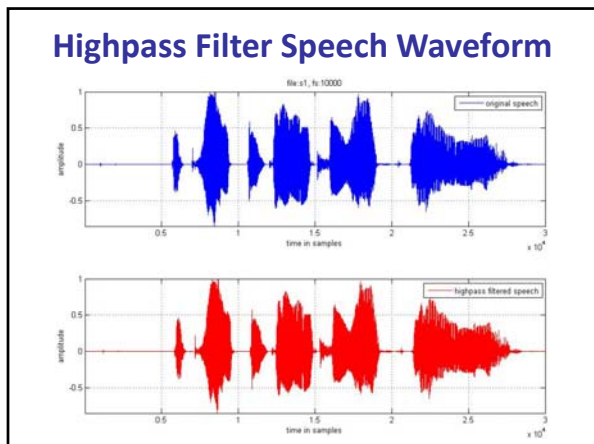
Sample Rate Conversion

- SRC_GUI.m



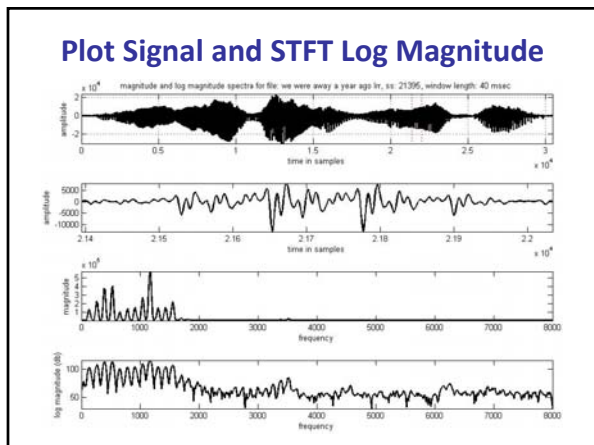
Highpass Filter Speech Waveform





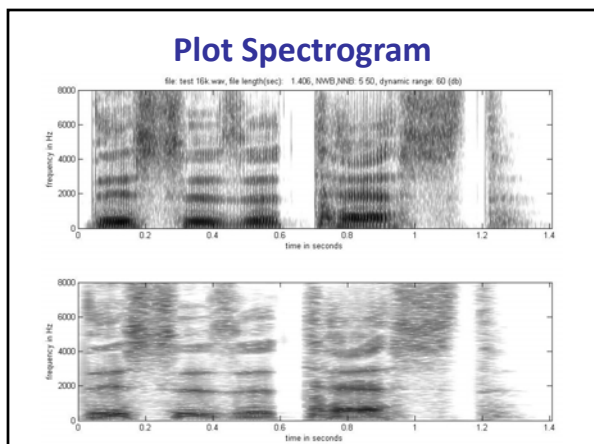
Highpass Filter Speech Waveform

- highpass_filter_GUI.m



Multiple Spectra GUI

- multiple_spectra_GUI.m



Plot Spectrogram

- spectrogram_GUI.m

