Subject: Re: reading wav files Posted by hahn on Tue, 16 Jan 2001 10:29:24 GMT

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jnettle1@utk.edu (jnettle1) wrote:

> Does anyone have a routine/set of routines to read *.wav files in IDL? And if > so, do you feel like sharing it?? :)

I process wav files randomly, so I have some read/write routines written on my own. However, these routines work on stereo encoded 16 bits per sample files only and ignore anything but the data chunk. I can send you the routines by mail if needed.

The following structe defines the header. It is 44 bytes long. It assumes that the data chunk follows the format descriptor block.

```
WAV_HDR = { WAVE_HEADER, $
       ID1:'****'.
                              ID2:'******
                   LoF:0L.
                                              LoH:0L, $
       FMT:0,
                    NChan:0,
                                 SpSc:0L,
                                                 DpSc:0L, $
                                ID3:'****',
       BypSa:0,
                     BipSa:0,
                                               LoD:0L }
        the letters "RIFF"
; ID1
; LOF
         bytes from here (i.e. from byte 8) to end of file
        the letters "WAVE" followed by "fmt "
ID2
       "fmt" is the beginning of the format descriptor block
         Length of Header (here 16 Bytes format descriptor) that follows
 LoH
       ---- start of format header ----
 FMT
          format tag, 1 = PCM, 257 = IBM mu Law, 258 = IBM a Law,
                    259 = IBM Adaptive PCM (adpcm)
           channels (1 = Mono, 2 = Stereo ...)
 NChan
          samples per second
 SpSc
 DpSc
          data bytes per second = samples per second * bytes per sample
 BypSa
          bytes per sample = channels * (bits per sample / 8)
          bits per sample
 BipSa
       ---- end of format header ----
       Now watch for "data": Other blocks will be skipped.
        the letters "data". This is the beginning of the data block
 ID3
         data length, i.e. the number of bytes of the actual data.
 LoD
       ---- start of sampled data ----
       for multi-channel samples 1st byte is channel 1, 2nd byte is
       channel 2, 3rd byte channel 3 (or if 2 channels, channel 1 again)
       Amplitude (y-values) from -127 to 127 or -32767 to 32767.
       ---- end of sampled data ----
       Comments may follow. Don't play 'em
```

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