Specification of the ENF Audio File Format

A format for ENF audio files

Version 2.0

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Abstract

The ENF Audio File Format is a tailored file format for storing and classifying ENF (Electric Network Frequency) audio data. The raw ENF audio data in ".ENF" files is uncompressed pulsecode modulation (PCM).

Keywords: ENF Audio File Format, .ENF, PCM

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Specification of the

ENF Audio File Format

1. Introduction

Electric Network Frequency (ENF) signal analysis has rapidly emerged in the forensic fields where the features of power-line-related signals can be used to identify the time and place in which audio recordings might have been made.

The ENF Audio File Format is specially designed for the efficient storage and classification of the ENF audio data. Compared with other audio file formats, e.g. WAV, AIFF, AU etc. "ENF" format has a tailored header that holds two unique fields: Location and Datetime which can be used to efficiently index ENF sample database.

The header of ENF files (filename extension ".enf") consists of 36 bytes which includes a <Classification> chunk and a <Data> chunk. The <Classification> chunk contains two subchunk: Location and Datetime. The raw sample data in ".ENF" files is uncompressed mono pulse-code modulation (PCM) which has specific sample rate, bit-width and size indicated in <Data> chunk. This illustrated in Figure 1, and Table 1.

2. Header structure of an ENF file

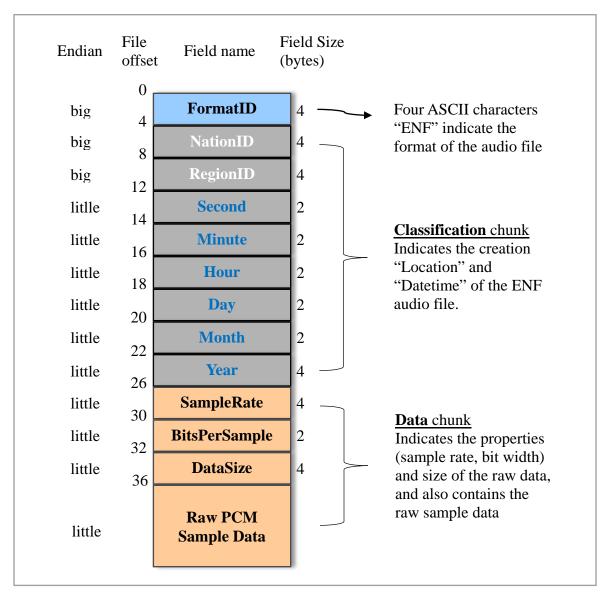


Figure 1. The "ENF" Audio File Format

2.1 The header fields description of ENF

Table 1. Fields Description

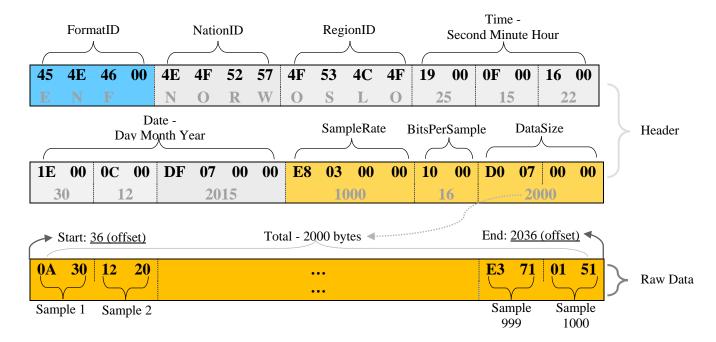
				Field	
	File	Field		Size	
Endian	Offset	Name		/bytes	Description
big	0	FormatID		4	Four ASCII "ENF" indicates the file
					format, the filename extension is ".enf"
big	4	Location	NationID	4	Up to four ASCII char, e.g. "NORW"
			RegionID	4	Up to four ASCII char, e.g. "OSLO"
little	12	DateTime	Second	2	Seconds of minutes from 0 ~ 59
			Minute	2	Minutes of hour from 0 ~ 59
			Hour	2	Hours of day from 0 ~ 24
			Day	2	Days of month from 1 ~ 31
			Month	2	Months of year from $1 \sim 12$
			Year	4	Year since 1900
little	26	SampleRate		4	The number of samples per second, e.g. 1000, which is the number of times per second that samples are taken
little	30	BitsPerSample		2	Bit depth (Sample width), e.g. 8 or 16 means 8-bit or 16-bit depth, which determines the number of possible digital values representing each sample.
little	32	DataSize		4	Indicates total sample data size in bytes. If unknown, the value of 0xffffffff should be used. Duration time can be calculated by the formula: Duration time = DataSize * 8 / (SampleRate * BitsPerSample)
	Raw PCM sample data			Pulse-code modulation (PCM) is a method used to digitally represent sampled analog signals. 8-bit samples are stored as unsigned bytes, ranging from 0 to 255. 16-bit samples are stored as 2's-complement signed integers, ranging from -32768 to 32767	

2.2 An example of ENF file

Here is an example of ENF file with bytes shown as hexadecimal numbers;



Below is the interpretation of these bytes as an ENF file



By reading the head of the ENF file, we can know that it was recorded on 2015/12/30 at 22:15:25, in Oslo of Norway with 1000Hz sample rate and 16-bit per sample. And its raw data size is total 2000 bytes, which means that its duration is only 2 seconds (1000 samples collected).

3. References

[1] EBU-TECH 3285: Specification of the Broadcast Wave Format (BWF)

[2] WAVE PCM sound file format: http://soundfile.sapp.org/doc/WaveFormat/

[3] Pulse-code modulation: https://en.wikipedia.org/wiki/Pulse-code_modulation

[4] Audio file format: https://en.wikipedia.org/wiki/Audio_file_format

Appendix

Appendix A: ENF Audio Recorder Application

ENF Audio Recorder is a handy tool for collecting the variation in frequency in the electric network via the audio card with an AC adapter with the correct voltage. It can assist in forensic research for determining the time of the recording.

ENF Audio Recorder is developed in Qt C++ programming language and can be run on Windows systems. Its GUI looks like as below Fig 4.1.

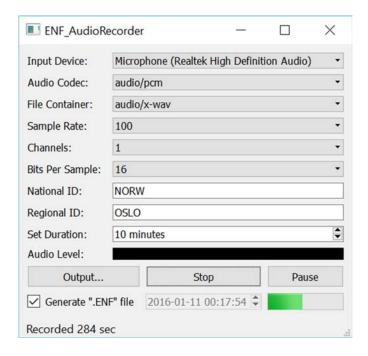


Fig 4.1 ENF Audio Recorder GUI

Users can select Input Device, File Container (-way, or -raw), Sample Rate (100, 500, 1000, 8000, 11025, 16000...up to 192000), Channels (1/2), BitsPerSample (16), and specify the recording place (National ID, and Regional ID). And also this application support manual stop record or automatically stop by specify duration – changing the value of "Set Duration".

The record start time shall be automatically generated once users start record (by press Record button after finish settings.). And then the green bar will appear to indicate the progress of record if user specified a duration. After finish record (by Stop button or reach Duration setting), if user clicked the option "Generate .ENF file", the application will generate a proper ENF file with a suffix of ".enf"

The latest version download link as below:

https://www.dropbox.com/s/ugxucvc9ouel65j/ENF_Recorder_v3.rar?dl=0

Appendix B: ENF Audio File Format definition with C++

The below 2 figs show ENF Audio Format definitions with C++.

```
▼ struct Location{
      uint32 t NationID; //indicates National ID of an audio like "NORW"
      uint32_t RegionID; //indicates Regional ID of an audio like "OSLO"

▼ struct DateTime{
                     //seconds of minutes from 0 to 59
      uint16_t sec;
      uint16 t min;
                      //minutes of hour from 0 to 59
      uint16 t hour; //hours of day from 0 to 24
      uint16_t day; //day of month from 1 to 31
      uint16_t mon; //month of year from 1 to 12
uint32_t year; //year since 1900
  };
 typedef struct ENF {
      uint32 t formatID;//indicates audio format here is "ENF"
      Location locationID; //indicates record location
      DateTime creationTime; //indicates recording time
      uint32_t sampleRate; // the sample rate of audio in HZ
      uint16 t bitsperSample; //number of bits per sample: 8/16 or 32.
      uint32 t dataSize; //audio data size in bytes
  } ENF Header;
class AudioENF
    FILE *fENF;
   ENF_Header enfHeader;
 public:
     AudioENF();
      ~AudioENF();
      void Open(const char* filename, OpenMode OM);
     void setFormat(string formt="ENF");
     void setLocation();
                                                 //set by enter location
      void setLocation(string sNat, string sReg);//set loc directly
     void setTime();
                                                 //set by enter datetime
     void setLocalTime();
     void setTime(struct tm *ctm);
                                                 //set datetime
     void setRate(uint32_t rate=8000);
     void setBits(uint16_t bits=16);
     void setRateBits();
                                                 //set by enter rate and bits
     void setRateBits(uint32 t rate, uint16 t bits);
     void setSize(uint32 t newSize) {enfHeader.dataSize = newSize; }
      void copyHeader(WaveFile& wav);
     void copyHeader(ENF Header& eh);
      void setDefaultHeader();
     void Close();
     void WriteHeader();
      void WriteHeader (ENF Header& eh); // write header by copy exist header
      void WriteHeader(WaveFile& wav); // write header from wavfile
      void WriteData(const int16_t* data, uint32_t num);
      void getHeader(ENF Header& eh);
      uint32 t getDataSize() {return enfHeader.dataSize;}
      uint32 t getfileSize() {return fileSize(*fENF);}
      void PrintHeader(); // print out header info for debug
      friend int fileSize(FILE& inf);
};
```