## Sound analysis and synthesis with the package Seewave

## Jérôme Sueur<sup>1,\*</sup>, Thierry Aubin<sup>2</sup>, Caroline Simonis<sup>3</sup>

1. Muséum national d'Histoire naturelle, Département Systématique et Évolution, CNRS UMR 7205, CP 50, 45 rue Buffon, 75005 Paris, France

2. Neurobiologie de l'Apprentissage, de la Mémoire et de la Communication, CNRS UMR 8620, Bât. 446, Université Paris-Sud, 91405 Orsay Cedex, France

3. Muséum national d'Histoire naturelle, Département Écologie et Gestion de la Biodiversité, MNHN USM 301 & CNRS UMR 7179, CP 55, 57 rue Buffon, 75005 Paris, France

\* Contact author: sueur@mnhn.fr

Keywords: time series, sound, Fourier, amplitude, frequency

Seewave is a package for sound analysis and synthesis. It has been first submitted to CRAN in 2006, a new version having been released about all semester. The package has been initially written to help analysing sound produced by animals. However, it is now used in different contexts such as telemetry, optical signals, high frequency vibrations, meteor signal shape and others.

Sound input can be achieved using different object classes : (*i*) usual classes (numeric vector, numeric matrix) if the sampling frequency is provided, (*ii*) time series classes (ts, mts), and (*iii*) sound-specific classes (Wave of the package tuneR and Sample of the package Sound).

Seewave currently includes more than 70 functions. Sounds are edited as oscillogram or amplitude envelope in single or multi-framed windows. An option can be set to move along the signal using a time slider. Signals can be modified with cutting, inserting, pasting, muting, fading, and repeating functions.

In the time/amplitude domain, signal and silence durations can be automatically measured. Amplitude filters can help reducing a background noise. Amplitude modulations can be automatically removed or quickly modified interactively and echoes can be generated through Doppler effect.

In the frequency domain, 15 statistical descriptive parameters (dominant peak, quality factor, entropy, spectral flatness, etc) are extracted from a frequency spectrum by calling a single function. The fundamental frequency of harmonic series is detected by the autocorrelation or cepstral method, while the instantaneous frequency is obtained by the zero-crossing method or Hilbert transform. Seewave provides a short-term Fourier transform to return mean spectra, 2D and 3D spectrograms. Fourier window size, overlap and zero-padding options allow the user to improve graphical representation, and to reduce the uncertainty principle.

To test for the temporal and frequency similarity of two sounds, cross-correlations, surface computation and coherence can be computed.

New sound is designed with sinusoidal amplitude modulations and linear and/or sinusoidal frequency modulations. Simple additions together with requency filters and linear frequency shifts ensure the modification or generation of complex sound.

## References

Sueur J, Aubin T, Simonis C (2008). Seewave: a free modular tool for sound analysis and synthesis. Bioacoustics, 18: 213–226.

Sueur J (2009). Seewave,

http://sueur.jerome.perso.neuf.fr/seewave.html.