

# Package ‘warbleR’

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**Type** Package

**Title** Streamline Bioacoustic Analysis

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**Description** Offers three overarching categories of functions to obtain bird vocalization recordings from the web, manage sound files, and facilitate (bio)acoustic analysis in R.

**License** GPL (>= 2)

**Imports** rjson, pbapply, RCurl, fftw, utils, stats, grDevices, foreach, doParallel, graphics, dtw, parallel, jpeg

**Depends** R (>= 3.2.1), maps, seewave (>= 2.0.1), tuneR,

**LazyData** TRUE

**NeedsCompilation** no

**Suggests** knitr, ggplot2

**VignetteBuilder** knitr

**RoxygenNote** 5.0.1

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autodetec	<i>Automatically detect vocalizations in sound files</i>
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## Description

autodetec automatically detects the start and end of vocalizations in sound files based on amplitude, duration, and frequency range attributes.

## Usage

```
autodetec(X = NULL, threshold = 15, envt = "abs", ssmooth = NULL, msmooth = NULL,
  power = 1, bp = NULL, osci = FALSE, wl = 512, xl = 1, picsize = 1, res = 100,
  flim = c(0,22), ls = FALSE, sxrow = 10, rows = 10, mindur = NULL, maxdur =
  NULL, redo = FALSE, img = TRUE, it = "jpeg", set = FALSE, flist = NULL, smadj = NULL,
  parallel = 1, path = NULL)
```

**Arguments**

X	Data frame with results from <a href="#">manualoc</a> function or any data frame with columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end).
threshold	A numeric vector of length 1 specifying the amplitude threshold for detecting signals (in %).
envt	Character vector of length 1 specifying the type of envelope to be used: "abs" for absolute amplitude envelope or "hil" for Hilbert amplitude envelope. Default is "abs".
ssmooth	A numeric vector of length 1 to smooth the amplitude envelope with a sum smooth function. Default is NULL.
msmooth	A numeric vector of length 2 to smooth the amplitude envelope with a mean sliding window. The first component is the window length and the second is the overlap between successive windows (in %). Faster than ssmooth but time detection is much less accurate. Will be deprecated in future versions. Default is NULL.
power	A numeric vector of length 1 indicating a power factor applied to the amplitude envelope. Increasing power will reduce low amplitude modulations and increase high amplitude modulations, in order to reduce background noise. Default is 1 (no change).
bp	Numeric vector of length 2 giving the lower and upper limits of a frequency bandpass filter (in kHz). Default is c(0, 22).
osci	Logical argument to add an oscillogram underneath spectrogram, as in <a href="#">spectro</a> . Default is FALSE. Not applied if ls is TRUE.
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
xl	Numeric vector of length 1, a constant by which to scale spectrogram width. Default is 1.
picsize	Numeric argument of length 1. Controls the relative size of the spectrogram. Default is 1.
res	Numeric argument of length 1 controlling resolution of images. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
flim	A numeric vector of length 2 for the frequency limit in kHz of the spectrogram, as in <a href="#">spectro</a> . Default is c(0, 22).
ls	Logical argument. If TRUE, long spectrograms as in <a href="#">lspec</a> are produced.
sxrow	A numeric vector of length 1. Specifies seconds of spectrogram per row when creating long spectrograms. Default is 10. Applied when ls = TRUE and/or when X is not provided.
rows	A numeric vector of length 1. Specifies number of rows per image file when creating long spectrograms. Default is 10. Applied when ls = TRUE and/or when X is not provided.
mindur	Numeric vector of length 1 giving the shortest duration (in seconds) of the signals to be detected. It removes signals below that threshold.

maxdur	Numeric vector of length 1 giving the longest duration (in seconds) of the signals to be detected. It removes signals above that threshold.
redo	Logical argument. If TRUE all selections will be analyzed again when code is rerun. If FALSE only the selections that do not have an 'autodetec' generated image file in the working directory will be analyzed. Default is FALSE.
img	Logical argument. If FALSE, image files are not produced. Default is TRUE.
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
set	A logical argument indicating wheter the settings of the autodetection process should be included in the image file name. If TRUE, threshold (th), envelope (envt), bandpass (bp), power (pw), smooth (smo, either mmsooth[1] or ssmooth), maxdur (mxdu), and mindur (midu) are included.
flist	character vector or factor indicating the subset of files that will be analyzed. Ignored if X is provided.
smadj	adjustment for amplitude smoothing. Character vector of length one indicating whether start end values should be adjusted. "start", "end" or "both" are the inputs admitted by this argument. Amplitude smoothing through ssmooth generates a predictable deviation from the actual start and end positions of the signals, determined by the threshold and ssmooth values. This deviation is more obvious (and problematic) when the increase and decrease in amplitude at the start and end of the signal (respectively) is not gradual. Ignored if ssmooth is NULL.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function determines the start and end of signals in the segments of the sound files listed in the input data frame. Alternatively, if no data frame is provided, the function detects signals across each entire sound file and creates long spectrograms for all sound files in the working directory. The input data frame should have the following columns: c("sound.files", "selec", "start", "end"). The output of [manualoc](#) can be used as the input data frame. This function uses a modified version of the [timer](#) function from seewave package to detect signals.

## Value

Image files with spectrograms showing the start and end of the detected signals. It also returns a data frame containing the start and end of each signal by sound file and selection number.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>). Implements a modified version of the timer function from seewave.

## Examples

```
## Not run:
# First create empty folder
setwd(tempdir())

data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "Phae.long4"))
writeWave(Phae.long1,"Phae.long1.wav")
writeWave(Phae.long2,"Phae.long2.wav")
writeWave(Phae.long3,"Phae.long3.wav")
writeWave(Phae.long4,"Phae.long4.wav")

ad <- autotetec(threshold = 5, env = "hil", ssmooth = 300, power=1,
bp=c(2,9), xl = 2, picsize = 2, res = 200, flim= c(1,11), osci = TRUE,
wl = 300, ls = FALSE, sxrow = 2, rows = 4, mindur = 0.1, maxdur = 1, set = TRUE)

#run it with different settings
ad <- autotetec(threshold = 90, env = "abs", ssmooth = 300, power = 1, redo = TRUE,
bp=c(2,9), xl = 2, picsize = 2, res = 200, flim= c(1,11), osci = TRUE,
wl = 300, ls = FALSE, sxrow = 2, rows = 4, mindur=0.1, maxdur=1, set = TRUE)

#check this folder!!
getwd()

## End(Not run)
```

---

checksel

---

*Check selection data frames*


---

## Description

checksel checks whether selections can be read by subsequent functions.

## Usage

```
checksel(X, parallel = 1, path = NULL)
```

## Arguments

X	data frame with the following columns: 1) "sound.files": name of the .wav files, 2) "sel": number of the selections, 3) "start": start time of selections, 4) "end": end time of selections. The output of <a href="#">manualoc</a> or <a href="#">autotetec</a> can be used as the input data frame.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function checks if the selections listed in the data frame correspond to .wav files in the working directory, if the sound files can be read and if so, if the start and end time of the selections are found within the duration of the sound files. Note that the sound files should be in the working directory (or the directory provided in 'path'). This is useful for avoiding errors in downstream functions (e.g. [specan](#)).

## Value

If all .wav files are ok, returns message "All files are ok!". Otherwise returns "These file(s) cannot be read" message with names of the corrupted .wav files.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## See Also

[checkwavs](#)

## Examples

```
## Not run:
# First set temporary folder
setwd(tempdir())

# save wav file examples
data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")

checksels(X = manualoc.df)

## End(Not run)
```

---

checkwavs

*Check .wav files*

---

## Description

checkwavs checks whether .wav files can be read by subsequent functions.

## Usage

```
checkwavs(X = NULL, path = NULL)
```

## Arguments

X	data frame with the following columns: 1) "sound.files": name of the .wav files, 2) "sel": number of the selections, 3) "start": start time of selections, 4) "end": end time of selections. The output of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame. If provided the function also returns the smallest number of samples from the listed selections, which limits the minimum window length (wl argument in other functions) that can be used in batch analyses. This could be useful for avoiding errors in downstream functions (e.g. <a href="#">specan</a> ).
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function checks if .wav files in the working directory can be read. Users must set the working directory where they wish to check .wav files beforehand. If X is provided it also returns the smallest number of samples from the selections listed in X (if all files can be read).

## Value

If all .wav files are ok, returns message "All files are ok!". Otherwise returns "These file(s) cannot be read" message with names of the corrupted .wav files.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## See Also

[checksels](#)

## Examples

```
## Not run:
# First set temporary folder
setwd(tempdir())

# save wav file examples
data(list = c("Phae.long1", "Phae.long2", "Phae.long3"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")

checkwavs()

## End(Not run)
```

compare.methods

*Assessing the performance of acoustic distance measurements***Description**

compare.methods creates graphs to visually assess performance of acoustic distance measurements

**Usage**

```
compare.methods(X = NULL, flim = c(0, 22), bp = c(0, 22), mar = 0.1, wl = 512, ovlp = 90,
res = 150, n = 10, length.out = 30, methods = c("XCORR",
"dfDTW", "ffDTW", "SP"), it = "jpeg", parallel = 1, path = NULL, sp = NULL)
```

**Arguments**

X	Data frame with results from <a href="#">manualoc</a> function, <a href="#">autodetec</a> function, or any data frame with columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end). Default NULL.
flim	A numeric vector of length 2 for the frequency limit in kHz of the spectrogram, as in <a href="#">spectro</a> . Default is c(0, 22).
bp	numeric vector of length 2 giving the lower and upper limits of the frequency bandpass filter (in kHz) used in the acoustic distance methods. Default is c(0, 22).
mar	Numeric vector of length 1. Specifies plot margins around selection in seconds. Default is 0.1.
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
ovlp	Numeric vector of length 1 specifying the percent overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 90.
res	Numeric argument of length 1. Controls image resolution. Default is 150.
n	Numeric argument of length 1. Defines the number of plots to be produce. Default is 10.
length.out	A character vector of length 1 giving the number of measurements of fundamental or dominant frequency desired (the length of the time series). Default is 30.
methods	A character vector of length 2 giving the names of the acoustic distance methods that would be compared. The methods available are: cross-correlation (XCORR, from <a href="#">xcorr</a> ), dynamic time warping on dominant frequency time series (dfDTW, from <a href="#">dtw</a> applied on df ts output), dynamic time warping on dominant frequency time series (ffDTW, from <a href="#">dtw</a> applied on ff ts output), spectral parameters (SP, from <a href="#">specan</a> ).
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".



parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.
sp	Data frame with acoustic parameters as the one generated by specan. Must contain 'sound.files' and 'selec' columns and the same selections as in 'X'.

## Details

This function produces graphs with spectrograms from 4 selections that allow visual inspection of the performance of acoustic distance methods at comparing those selections. The spectrograms are all plotted with the same frequency and time scales. The function compares 2 methods at a time. The methods available are: cross-correlation (XCORR, from `xcorr`), dynamic time warping on dominant frequency time series (dfDTW, from `dtw` applied on `dfts` output), dynamic time warping on dominant frequency time series (ffDTW, from `dtw` applied on `ffts` output), spectral parameters (SP, from `specan`). The graph also contains 2 scatterplots (1 for each method) of the acoustic space of all signals in the input data frame 'X'. The compared selections are randomly picked up from the pool of selections in the input data frame. The argument 'n' defines the number of comparison (e.i. graphs) to be produced. The acoustic pairwise distance between signals is shown next to the arrows linking them. The font color of a distance value correspond to the font color of the method that generated it, as shown in the scatterplots. Distances are standardized, being 0 the distance of a signal to itself and 1 the farthest pairwise distance in the pool of signals. Principal Component Analysis (`princomp`) is applied to calculate distances when using spectral parameters (SP). In that case the first 2 PC's are used. Classical Multidimensional Scalling (also known as Principal Coordinates Analysis, (`cmdscale`)) is used for all other methods. Note that SP can only be used with at least 22 selections (number of rows in input data frame) as PCA only works with more units than variables. The graphs are returned as image files in the working directory. The file name contains the methods being compared and the row number of the selections. This function uses internally a modified version of the `spectro` function from `seewave` package to create spectrograms.

## Value

Image files with 4 spectrograms of the selection being compared and scatterplots of the acoustic space of all signals in the input data frame 'X'.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>). It uses internally a modified version of the `spectro` function from `seewave` package to create spectrograms.

## Examples

```
## Not run:
# First create empty folder
setwd(tempdir())

data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "Phae.long4", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav")
```

```

writeWave(Phae.long2,"Phae.long2.wav")
writeWave(Phae.long3,"Phae.long3.wav")
writeWave(Phae.long4,"Phae.long4.wav")

compare.methods(X = manualoc.df, flim = c(0, 10), bp = c(0, 10), mar = 0.1, wl = 300,
ovlp = 90, res = 200, n = 10, length.out = 30,
methods = c("XCORR", "dfDTW"), parallel = 1, it = "jpeg")

#check this folder!
getwd()

## End(Not run)

```

---

coord.graph

*Coordinated singing graphs*


---

## Description

coord.graph creates graphs of coordinated singing and highlights the signals that overlap in time. The signals are represented by polygons of different colors.

## Usage

```
coord.graph(X, only.coor = FALSE, ovlp = TRUE, xl = 1, res= 80, it = "jpeg", img = TRUE,
tlim = NULL)
```

## Arguments

X	Data frame containing columns for singing event (sing.event), individual (indiv), and start and end time of signal (start and end).
only.coor	Logical. If TRUE only the segment in which both individuals are singing is included (solo singing is removed). Default is FALSE.
ovlp	Logical. If TRUE the vocalizations that overlap in time are highlighted. Default is TRUE.
xl	Numeric vector of length 1, a constant by which to scale spectrogram width. Default is 1.
res	Numeric argument of length 1. Controls image resolution. Default is 80.
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
img	Logical argument. If FALSE, image files are not produced. Default is TRUE. Note that images are return
tlim	Numeric vector of length 2 indicating the start and end time of the coordinated singing events to be displayed in the graphs.

## Details

This function provides visualization for coordination of acoustic signals. Signals are shown as polygon across a time axis. It also shows which signals overlap, the amount of overlap, and highlights the individual responsible for the overlap using a color code. The width of the polygons depicting the time of overlap.

## Value

The function returns a list of graphs, one for each singing event in the input data frame. The graphs can be plotted by simply calling the list. If 'img' is TRUE then the graphs are also saved in the working directory as files.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## Examples

```
## Not run:

# First set temporary folder
setwd(tempdir())

# load simulate singing events (see data documentation)
, data(sim.coord.sing)

# make coord.graphs in tiff format
coord.graph(X = sim.coord.sing, ovlp = TRUE, only.coord = FALSE, xl = 2, res = 80,
it = "tiff", img = TRUE)

#' # make coord.graphs in graphic device format
cgs <- coord.graph(X = sim.coord.sing, ovlp = TRUE, only.coord = FALSE, img = FALSE)

cgs

## End(Not run)
```

---

coord.test

*Randomization test for singing coordination*

---

## Description

Monte Carlo randomization test to assess the statistical significance of singing coordination

## Usage

```
coord.test(X, iterations = 1000, less.than.chance = TRUE, parallel = 1)
```

### Arguments

<code>x</code>	Data frame containing columns for singing event ( <code>sing.event</code> ), individual ( <code>indiv</code> ), and start and end time of signal ( <code>start</code> and <code>end</code> ).
<code>iterations</code>	number of iterations for shuffling and calculation of the expected number of overlaps. Default is 1000.
<code>less.than.chance</code>	Logical. If TRUE the test evaluates whether overlaps occur less often than expected by chance. If FALSE the opposite pattern is evaluated (whether overlaps occur more often than expected by chance). Default is TRUE.
<code>parallel</code>	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).

### Details

This function calculates the probability of finding an equal or lower number (or higher if `less.than.chance` is TRUE) of song overlaps in a coordinated singing event. The function shuffles the sequences of signals and silence-between-signals for both individuals to produce a null distribution of expected number of overlaps by chance. The observed number of overlaps is compared to this expected value. The p-values are calculated as the proportion of random expected values that were lower (or higher) than the observed value. The function runs one test for each singing event in the input data frame. The function is equivalent to the "KeepGaps" methods described in Masco et al. 2015.

### Value

A data frame with the observed number of overlaps (`obs.overlaps`), mean number of overlaps expected by chance, and p value.

### Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

### References

Masco, C., Allesina, S., Mennill, D. J., and Pruett-Jones, S. (2015). The Song Overlap Null model Generator (SONG): a new tool for distinguishing between random and non-random song overlap. *Bioacoustics*. 1-12.

### Examples

```
## Not run:
#load simulated singing data (see data documentation)
, data(sim.coord.sing)

# testing if coordination happens less than expected by chance
coord.test(sim.coord.sing, iterations = 1000, less.than.chance = TRUE)

# testing if coordination happens more than expected by chance
coord.test(sim.coord.sing, iterations = 1000, less.than.chance = FALSE)

## End(Not run)
```

---

Cryp.soui	<i>Acoustic recording of Crypturellus soui (Little Tinamou).</i>
-----------	--

---

**Description**

Acoustic recording of *Crypturellus soui* (Little Tinamou).

**Usage**

```
data(Cryp.soui)
```

**Format**

One .wav file:

**Cryp.soui** *Crypturellus soui* recording

**Source**

<http://www.xeno-canto.org/154190>

---

dfDTW	<i>Acoustic dissimilarity using dynamic time warping on dominant frequency contours</i>
-------	---

---

**Description**

dfDTW calculates acoustic dissimilarity of dominant frequency contours using dynamic time warping. Internally it applies the [dtwDist](#) from the dtw package.

**Usage**

```
dfDTW(X, wl = 512, flim = c(0, 22), length.out = 20, wn = "hanning", pal =
  reverse.gray.colors.2, ovlp = 70, inner.mar = c(5, 4, 4, 2), outer.mar =
  c(0, 0, 0, 0), picsize = 1, res = 100, cexlab = 1, title = TRUE, propwidth = FALSE,
  xl = 1, gr = FALSE, sc = FALSE, bp = c(0, 22), cex = 1,
  threshold = 15, col = "red2", pch = 16, mar = 0.05,
  lpos = "topright", it = "jpeg", img = TRUE, parallel = 1, path = NULL,
  img.suffix = "dfDTW")
```

**Arguments**

<code>x</code>	Data frame with results containing columns for sound file name ( <code>sound.files</code> ), selection number ( <code>selec</code> ), and start and end time of signal ( <code>start</code> and <code>end</code> ). The output of <code>manualoc</code> or <code>autodetec</code> can be used as the input data frame.
<code>wl</code>	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
<code>flim</code>	A numeric vector of length 2 for the frequency limit of the spectrogram (in kHz), as in <code>spectro</code> . Default is <code>c(0, 22)</code> .
<code>length.out</code>	A character vector of length 1 giving the number of measurements of dominant frequency desired (the length of the time series).
<code>wn</code>	Character vector of length 1 specifying window name. Default is "hanning". See function <code>ftwindow</code> for more options.
<code>pal</code>	A color palette function to be used to assign colors in the plot, as in <code>spectro</code> . Default is <code>reverse.gray.colors.2</code> .
<code>ovlp</code>	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <code>spectro</code> . Default is 70.
<code>inner.mar</code>	Numeric vector with 4 elements, default is <code>c(5,4,4,2)</code> . Specifies number of lines in inner plot margins where axis labels fall, with form <code>c(bottom, left, top, right)</code> . See <code>par</code> .
<code>outer.mar</code>	Numeric vector with 4 elements, default is <code>c(0,0,0,0)</code> . Specifies number of lines in outer plot margins beyond axis labels, with form <code>c(bottom, left, top, right)</code> . See <code>par</code> .
<code>picsize</code>	Numeric argument of length 1. Controls relative size of spectrogram. Default is 1.
<code>res</code>	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
<code>cexlab</code>	Numeric vector of length 1 specifying the relative size of axis labels. See <code>spectro</code> .
<code>title</code>	Logical argument to add a title to individual spectrograms. Default is TRUE.
<code>propwidth</code>	Logical argument to scale the width of spectrogram proportionally to duration of the selected call. Default is FALSE.
<code>x1</code>	Numeric vector of length 1. A constant by which to scale spectrogram width. Default is 1.
<code>gr</code>	Logical argument to add grid to spectrogram. Default is FALSE.
<code>sc</code>	Logical argument to add amplitude scale to spectrogram, default is FALSE.
<code>bp</code>	A numeric vector of length 2 for the lower and upper limits of a frequency bandpass filter (in kHz). Default is <code>c(0, 22)</code> .
<code>cex</code>	Numeric vector of length 1, specifies relative size of points plotted for frequency measurements and legend font/points, respectively. See <code>spectro</code> .
<code>threshold</code>	amplitude threshold (%) for dominant frequency detection. Default is 15.
<code>col</code>	Vector of length 1 specifying colors of points plotted to mark dominant frequency measurements. Default is "red2".

pch	Numeric vector of length 1 specifying plotting characters for the frequency measurements. Default is 16.
mar	Numeric vector of length 1. Specifies the margins adjacent to the selections to set spectrogram limits. Default is 0.05.
lpos	Character vector of length 1 or numeric vector of length 2, specifying position of legend. If the former, any keyword accepted by xy.coords can be used (see below). If the latter, the first value will be the x coordinate and the second value the y coordinate for the legend's position. Default is "topright".
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
img	Logical argument. If FALSE, image files are not produced. Default is TRUE.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.
img.suffix	A character vector of length 1 with a suffix (label) to add at the end of the names of image files. Default is NULL.

### Details

This function extracts the dominant frequency values as a time series and then calculates the pairwise acoustic dissimilarity using dynamic time warping. The function uses the [approx](#) function to interpolate values between dominant frequency measures. If 'img' is TRUE the function also produces image files with the spectrograms of the signals listed in the input data frame showing the location of the dominant frequencies.

### Value

A matrix with the pairwise dissimilarity values. If img is FALSE it also produces image files with the spectrograms of the signals listed in the input data frame showing the location of the dominant frequencies.

### Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

### See Also

[speccreator](#) for creating spectrograms from selections, [snrspecs](#) for creating spectrograms to optimize noise margins used in [sig2noise](#) and [dfts](#), [ffts](#), [ffDTW](#) for frequency contour overlaid spectrograms.

Other spectrogram.creators: [dfts](#), [ffDTW](#), [ffts](#), [snrspecs](#), [speccreator](#), [trackfreqs](#)

## Examples

```
## Not run:
# set the temp directory
setwd(tempdir())

#load data
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long2, "Phae.long2.wav") #save sound files
writeWave(Phae.long1, "Phae.long1.wav")

# run function
dfDTW(manualoc.df, length.out = 30, flim = c(1, 12), bp = c(2, 9), wl = 300)

## End(Not run)
```

---

dfts

---

*Extract the dominant frequency values as a time series*


---

## Description

dfts extracts the dominant frequency values as a time series. of signals selected by [manualoc](#) or [autodetec](#).

## Usage

```
dfts(X, wl = 512, flim = c(0, 22), length.out = 20, wn = "hanning", pal =
  reverse.gray.colors.2, ovlp = 70, inner.mar = c(5, 4, 4, 2), outer.mar =
  c(0, 0, 0, 0), picsize = 1, res = 100, cexlab = 1, title = TRUE, propwidth = FALSE,
  xl = 1, gr = FALSE, sc = FALSE, bp = c(0, 22), cex = 1,
  threshold = 15, col = "red2", pch = 16, mar = 0.05,
  lpos = "topright", it = "jpeg", img = TRUE, parallel = 1, path = NULL,
  img.suffix = "dfts")
```

## Arguments

X	Data frame with results containing columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end). The output of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame.
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
flim	A numeric vector of length 2 for the frequency limit of the spectrogram (in kHz), as in <a href="#">spectro</a> . Default is c(0, 22).
length.out	A character vector of length 1 giving the number of measurements of dominant frequency desired (the length of the time series).
wn	Character vector of length 1 specifying window name. Default is "hanning". See function <a href="#">ftwindow</a> for more options.



pal	A color palette function to be used to assign colors in the plot, as in <a href="#">spectro</a> . Default is <code>reverse.gray.colors.2</code> .
ovlp	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 70.
inner.mar	Numeric vector with 4 elements, default is <code>c(5,4,4,2)</code> . Specifies number of lines in inner plot margins where axis labels fall, with form <code>c(bottom, left, top, right)</code> . See <a href="#">par</a> .
outer.mar	Numeric vector with 4 elements, default is <code>c(0,0,0,0)</code> . Specifies number of lines in outer plot margins beyond axis labels, with form <code>c(bottom, left, top, right)</code> . See <a href="#">par</a> .
picsize	Numeric argument of length 1. Controls relative size of spectrogram. Default is 1.
res	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
cexlab	Numeric vector of length 1 specifying the relative size of axis labels. See <a href="#">spectro</a> .
title	Logical argument to add a title to individual spectrograms. Default is TRUE.
propwidth	Logical argument to scale the width of spectrogram proportionally to duration of the selected call. Default is FALSE.
x1	Numeric vector of length 1. A constant by which to scale spectrogram width. Default is 1.
gr	Logical argument to add grid to spectrogram. Default is FALSE.
sc	Logical argument to add amplitude scale to spectrogram, default is FALSE.
bp	A numeric vector of length 2 for the lower and upper limits of a frequency bandpass filter (in kHz). Default is <code>c(0, 22)</code> .
cex	Numeric vector of length 1, specifies relative size of points plotted for frequency measurements and legend font/points, respectively. See <a href="#">spectro</a> .
threshold	amplitude threshold (%) for dominant frequency detection. Default is 15.
col	Vector of length 1 specifying colors of points plotted to mark dominant frequency measurements. Default is "red2".
pch	Numeric vector of length 1 specifying plotting characters for the frequency measurements. Default is 16.
mar	Numeric vector of length 1. Specifies the margins adjacent to the selections to set spectrogram limits. Default is 0.05.
lpos	Character vector of length 1 or numeric vector of length 2, specifying position of legend. If the former, any keyword accepted by <code>xy.coords</code> can be used (see below). If the latter, the first value will be the x coordinate and the second value the y coordinate for the legend's position. Default is "topright".
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
img	Logical argument. If FALSE, image files are not produced. Default is TRUE.

<code>parallel</code>	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
<code>path</code>	Character string containing the directory path where the sound files are located.
<code>img.suffix</code>	A character vector of length 1 with a suffix (label) to add at the end of the names of image files.

### Details

This function extracts the dominant frequency values as a time series. The function uses the [approx](#) function to interpolate values between dominant frequency measures. If there are no frequencies above the amplitude threshold at the beginning or end of the signals then NAs will be generated. On the other hand, if there are no frequencies above the amplitude threshold in between signal segments in which amplitude was detected then the values of this adjacent segments will be interpolated to fill out the missing values (e.g. no NAs in between detected amplitude segments).

### Value

A data frame with the dominant frequency values measured across the signals. If `img` is TRUE it also produces image files with the spectrograms of the signals listed in the input data frame showing the location of the dominant frequencies.

### Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

### See Also

[speccreator](#) for creating spectrograms from selections, [snrspecs](#) for creating spectrograms to optimize noise margins used in [sig2noise](#)

Other spectrogram creators: [dfDTW](#), [ffDTW](#), [ffts](#), [snrspecs](#), [speccreator](#), [trackfreqs](#)

### Examples

```
## Not run:
# set the temp directory
setwd(tempdir())

#load data
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long2, "Phae.long2.wav") #save sound files
writeWave(Phae.long1, "Phae.long1.wav")

# run function
dfts(manualoc.df, length.out = 30, flim = c(1, 12), bp = c(2, 9), wl = 300)

## End(Not run)
```

---

ffDTW	<i>Acoustic dissimilarity using dynamic time warping on fundamental frequency contours</i>
-------	--

---

## Description

ffDTW calculates acoustic dissimilarity of fundamental frequency contours using dynamic time warping. Internally it applies the [dtwDist](#) from the dtw package.

## Usage

```
ffDTW(X, wl = 512, flim = c(0, 22), length.out = 20, wn = "hanning", pal =
  reverse.gray.colors.2, ovlp = 70, inner.mar = c(5, 4, 4, 2), outer.mar =
  c(0, 0, 0, 0), picsize = 1, res = 100, cexlab = 1, title = TRUE, propwidth = FALSE,
  xl = 1, gr = FALSE, sc = FALSE, bp = c(0, 22), cex = 1,
  threshold = 15, col = "red2", pch = 16, mar = 0.05,
  lpos = "topright", it = "jpeg", img = TRUE, parallel = 1, path = NULL,
  img.suffix = "ffDTW")
```

## Arguments

X	Data frame with results containing columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end). The output of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame.
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
flim	A numeric vector of length 2 for the frequency limit of the spectrogram (in kHz), as in <a href="#">spectro</a> . Default is c(0, 22).
length.out	A character vector of length 1 giving the number of measurements of fundamental frequency desired (the length of the time series).
wn	Character vector of length 1 specifying window name. Default is "hanning". See function <a href="#">ftwindow</a> for more options.
pal	A color palette function to be used to assign colors in the plot, as in <a href="#">spectro</a> . Default is reverse.gray.colors.2.
ovlp	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 70.
inner.mar	Numeric vector with 4 elements, default is c(5,4,4,2). Specifies number of lines in inner plot margins where axis labels fall, with form c(bottom, left, top, right). See <a href="#">par</a> .
outer.mar	Numeric vector with 4 elements, default is c(0,0,0,0). Specifies number of lines in outer plot margins beyond axis labels, with form c(bottom, left, top, right). See <a href="#">par</a> .
picsize	Numeric argument of length 1. Controls relative size of spectrogram. Default is 1.

res	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
cexlab	Numeric vector of length 1 specifying the relative size of axis labels. See <a href="#">spectro</a> .
title	Logical argument to add a title to individual spectrograms. Default is TRUE.
propwidth	Logical argument to scale the width of spectrogram proportionally to duration of the selected call. Default is FALSE.
x1	Numeric vector of length 1. A constant by which to scale spectrogram width. Default is 1.
gr	Logical argument to add grid to spectrogram. Default is FALSE.
sc	Logical argument to add amplitude scale to spectrogram, default is FALSE.
bp	A numeric vector of length 2 for the lower and upper limits of a frequency bandpass filter (in kHz). Default is c(0, 22).
cex	Numeric vector of length 1, specifies relative size of points plotted for frequency measurements and legend font/points, respectively. See <a href="#">spectro</a> .
threshold	amplitude threshold (%) for fundamental frequency detection. Default is 15.
col	Vector of length 1 specifying colors of points plotted to mark fundamental frequency measurements. Default is "red2".
pch	Numeric vector of length 1 specifying plotting characters for the frequency measurements. Default is 16.
mar	Numeric vector of length 1. Specifies the margins adjacent to the selections to set spectrogram limits. Default is 0.05.
lpos	Character vector of length 1 or numeric vector of length 2, specifying position of legend. If the former, any keyword accepted by xy.coords can be used (see below). If the latter, the first value will be the x coordinate and the second value the y coordinate for the legend's position. Default is "topright".
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
img	Logical argument. If FALSE, image files are not produced. Default is TRUE.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.
img.suffix	A character vector of length 1 with a suffix (label) to add at the end of the names of image files. Default is NULL.

## Details

This function extracts the fundamental frequency values as a time series and then calculates the pairwise acoustic dissimilarity of the selections using dynamic time warping. The function uses the [approx](#) function to interpolate values between fundamental frequency measures. If 'img' is TRUE the function also produces image files with the spectrograms of the signals listed in the input data

frame showing the location of the fundamental frequencies. Note that if no amplitude is detected at the beginning or end of the signals then NAs will be generated. On the other hand, if amplitude is not detected in between signal segments in which amplitude was detected then the values of this adjacent segments will be interpolated to fill out the missing values (e.g. no NAs in between detected amplitude segments).

### Value

A matrix with the pairwise dissimilarity values. If `img` is `FALSE` it also produces image files with the spectrograms of the signals listed in the input data frame showing the location of the fundamental frequencies.

### Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

### See Also

[speccreator](#) for creating spectrograms from selections, [snrspecs](#) for creating spectrograms to optimize noise margins used in [sig2noise](#)

`dfDTW` [dfts](#), [ffts](#), [dfDTW](#)

Other spectrogram.creators: [dfDTW](#), [dfts](#), [ffts](#), [snrspecs](#), [speccreator](#), [trackfreqs](#)

### Examples

```
## Not run:
# set the temp directory
setwd(tempdir())

#load data
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long2, "Phae.long2.wav") #save sound files
writeWave(Phae.long1, "Phae.long1.wav")

# run function
ffDTW(manualoc.df, length.out = 30, flim = c(1, 12), img = T, bp = c(1, 9), wl = 300)

## End(Not run)
```

---

ffts

---

*Extract the fundamental frequency values as a time series*


---

### Description

`ffts` extracts the fundamental frequency values as a time series of signals selected by [manualoc](#) or [autodetec](#).

**Usage**

```
ffts(X, wl = 512, flim = c(0, 22), length.out = 20, wn = "hanning", pal =
  reverse.gray.colors.2, ovlp = 70, inner.mar = c(5, 4, 4, 2), outer.mar =
  c(0, 0, 0, 0), picsize = 1, res = 100, cexlab = 1, title = TRUE, propwidth = FALSE,
  xl = 1, gr = FALSE, sc = FALSE, bp = c(0, 22), cex = 1,
  threshold = 15, col = "red2", pch = 16, mar = 0.05,
  lpos = "topright", it = "jpeg", img = TRUE, parallel = 1, path = NULL,
  img.suffix = "ffts")
```

**Arguments**

X	Data frame with results containing columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end). The output of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame.
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
flim	A numeric vector of length 2 for the frequency limit of the spectrogram (in kHz), as in <a href="#">spectro</a> . Default is c(0, 22).
length.out	A character vector of length 1 giving the number of measurements of fundamental frequency desired (the length of the time series).
wn	Character vector of length 1 specifying window name. Default is "hanning". See function <a href="#">ftwindow</a> for more options.
pal	A color palette function to be used to assign colors in the plot, as in <a href="#">spectro</a> . Default is reverse.gray.colors.2.
ovlp	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 70.
inner.mar	Numeric vector with 4 elements, default is c(5,4,4,2). Specifies number of lines in inner plot margins where axis labels fall, with form c(bottom, left, top, right). See <a href="#">par</a> .
outer.mar	Numeric vector with 4 elements, default is c(0,0,0,0). Specifies number of lines in outer plot margins beyond axis labels, with form c(bottom, left, top, right). See <a href="#">par</a> .
picsize	Numeric argument of length 1. Controls relative size of spectrogram. Default is 1.
res	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
cexlab	Numeric vector of length 1 specifying the relative size of axis labels. See <a href="#">spectro</a> .
title	Logical argument to add a title to individual spectrograms. Default is TRUE.
propwidth	Logical argument to scale the width of spectrogram proportionally to duration of the selected call. Default is FALSE.
xl	Numeric vector of length 1. A constant by which to scale spectrogram width. Default is 1.
gr	Logical argument to add grid to spectrogram. Default is FALSE.

sc	Logical argument to add amplitude scale to spectrogram, default is FALSE.
bp	A numeric vector of length 2 for the lower and upper limits of a frequency bandpass filter (in kHz). Default is c(0, 22).
cex	Numeric vector of length 1, specifies relative size of points plotted for frequency measurements and legend font/points, respectively. See <a href="#">spectro</a> .
threshold	amplitude threshold (%) for fundamental frequency detection. Default is 15.
col	Vector of length 1 specifying colors of points plotted to mark fundamental frequency measurements. Default is "red2".
pch	Numeric vector of length 1 specifying plotting characters for the frequency measurements. Default is 16.
mar	Numeric vector of length 1. Specifies the margins adjacent to the selections to set spectrogram limits. Default is 0.05.
lpos	Character vector of length 1 or numeric vector of length 2, specifying position of legend. If the former, any keyword accepted by <code>xy.coords</code> can be used (see below). If the latter, the first value will be the x coordinate and the second value the y coordinate for the legend's position. Default is "topright".
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
img	Logical argument. If FALSE, image files are not produced. Default is TRUE.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.
img.suffix	A character vector of length 1 with a suffix (label) to add at the end of the names of image files.

## Details

This function extracts the fundamental frequency values as a time series. The function uses the [approx](#) function to interpolate values between fundamental frequency #' measures. If there are no frequencies above the amplitude threshold at the beginning or end of the signals then NAs will be generated. On the other hand, if there are no frequencies above the amplitude threshold in between signal segments in which amplitude was detected then the values of this adjacent segments will be interpolated to fill out the missing values (e.g. no NAs in between detected amplitude segments).

## Value

A data frame with the fundamental frequency values measured across the signals. If `img` is TRUE it also produces image files with the spectrograms of the signals listed in the input data frame showing the location of the fundamental frequencies.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

**See Also**

[sig2noise](#), [dfts](#), [ffDTW](#), [dfDTW](#)

Other spectrogram.creators: [dfDTW](#), [dfts](#), [ffDTW](#), [snrspecs](#), [speccreator](#), [trackfreqs](#)

**Examples**

```
## Not run:
# set the temp directory
setwd(tempdir())

#load data
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav") #save sound files
writeWave(Phae.long2, "Phae.long2.wav") #save sound files

# run function
ffts(manualoc.df, length.out = 50, flim = c(1, 12), bp = c(2, 9), wl = 300)
```

Note that fundamental frequency is not accurate for noisy signals, works better with pure tones

```
## End(Not run)
```

---

filtersels

*Subset selection data frames based on manually filtered image files*

---

**Description**

filtersels subsets selection data frames based on image files that have been manually filtered.

**Usage**

```
filtersels(X, path = NULL, lspec = FALSE, pdf = FALSE)
```

**Arguments**

X	data frame with the following columns: 1) "sound.files": name of the .wav files, 2) "sel": number of the selections. The output of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.
lspec	A logical argument indicating if the image files to be used for filtering were produced by the function <a href="#">lspec</a> . All the image files that correspond to a sound file must be deleted in order to be filtered out.
pdf	A logical argument indicating if the image files are in .pdf format. Default is FALSE. Note that pdf files can only be generated using <a href="#">lspec2pdf</a> (so they are long spectrograms). Then, if TRUE, lspec argument is ignored.



## Details

This function subsets selections (or sound files if `lspec` is `TRUE`) listed in a data frame based on the image files from spectrogram-creating functions (e.g. [speccreator](#)) in the working directory. Only the selections/sound files with and image in the working directory will remain. This is useful for excluding selections from undesired signals. Note that the image files should be in the working directory (or the directory provided in `'path'`).

## Value

If all `.wav` files are ok, returns message "All files are ok!". Otherwise returns "These file(s) cannot be read" message with names of the corrupted `.wav` files.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## Examples

```
## Not run:
# First set temporary folder
setwd(tempdir())

# save wav file examples
data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")

speccreator(manualoc.df, flim = c(0, 11), inner.mar = c(4, 4.5, 2, 1), outer.mar = c(4, 2, 2, 1),
  picsize = 2, res = 300, cexlab = 2, mar = 0.05, wl = 300)

#go to the working directory and delete some images

#filter selection data frame

#this data frame does not have the selections corresponding to the images that were deleted
fmloc

#now using lspec images
lspec(sxrow = 2, rows = 8, pal = reverse.heat.colors, wl = 300, ovlp = 10)

#go to the working directory and delete lspec images (the ones with several rows of spectrograms)

#filter selection data frame

## End(Not run)
```

---

imp.raven

---

*Import Raven selections*


---

## Description

imp.raven imports Raven selection data from many files simultaneously. Files must be in .txt format.

## Usage

```
imp.raven(path = NULL, sound.file.col = NULL, all.data = FALSE, recursive = FALSE)
```

## Arguments

path	A character string indicating the path of the directory in which to look for the text files. If not provided (default) the function searches into the current working directory. Default is NULL).
sound.file.col	A character string with the name of the column listing the sound files in the selection text files. Default is NULL).
all.data	Logical. If TRUE) all columns in text files are returned. Default is FALSE). Note that all files should contain exactly the same columns in the same order.
recursive	Logical. If TRUE) the listing recurse into sub-directories.

## Details

The function import raven selection data from many files simultaneously. Files must be in .txt format. Note that selection files including data from multiple recordings cannot be imported.

## Value

A single data frame with information of the selection files. If all.data argument is set to FALSE) the data frame contains the following columns: selec, start, end, and selec.file. If sound.file.col is provided the data frame will also contain a 'sound.files' column. In addition, all rows with duplicated data are removed. This is useful when both spectrogram and waveform views are included in the Raven selection files. If all.data is set to TRUE then all columns in selection files are returned.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## See Also

[imp.syrinx](#)

**Examples**

```
## Not run:
# First set temporary folder
setwd(tempdir())

data(selection.files)

write.table(selection.files[[1]],file = "100889-Garrulax monileger.selections.txt",
row.names = FALSE, sep= "\t")

write.table(selection.files[[2]],file = "1023-Arremonops rufivirgatus.selections.txt",
row.names = FALSE, sep= "\t")

#providing the name of the column with the sound file names
rav.dat<-imp.raven(sound.file.col = "End.File", all.data = FALSE)

View(rav.dat)

#getting all the data
rav.dat<-imp.raven(all.data = TRUE)
View(rav.dat)

## End(Not run)
```

---

imp.syrinx

---

*Import Syrinx selections*


---

**Description**

imp.syrinx imports Syrinx selection data from many files simultaneously. All files must have the same columns.

**Usage**

```
imp.syrinx(path = NULL, all.data = FALSE, recursive = FALSE)
```

**Arguments**

path	A character string indicating the path of the directory in which to look for the text files. If not provided (default) the function searches into the current working directory. Default is NULL).
all.data	Logical. If TRUE) all columns in text files are returned. Default is FALSE). Note that all files should contain exactly the same columns in the same order.
recursive	Logical. If TRUE) the listing recurse into sub-directories.

**Value**

A single data frame with information of the selection files. If `all.data` argument is set to `FALSE` the data frame contains the following columns: `selec`, `start`, `end`, and `selec.file`. If `sound.file.col` is provided the data frame will also contain a `'sound.files'` column. In addition, all rows with duplicated data are removed. This is useful when both spectrogram and waveform views are included in the Syrinx selection files. If `all.data` is set to `TRUE` then all columns in selection files are returned.

**Author(s)**

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

**See Also**

[imp.raven](#)

**Examples**

```
## Not run:
# First set temporary folder
setwd(tempdir())

#load data
data(selection.files)

write.table(selection.files[[3]],file = "harpyeagle.wav.txt",row.names = FALSE,
  col.names = FALSE, sep= "\t")

write.table(selection.files[[4]],file = "Phae.long4.wav.txt",row.names = FALSE,
  col.names = FALSE, sep= "\t")

syr.dat<-imp.syrinx(all.data = FALSE)

View(syr.dat)

#getting all the data
syr.dat<-imp.syrinx(all.data = TRUE)

View(syr.dat)

## End(Not run)
```

---

lspec

---

*Create long spectrograms of whole sound files*


---

**Description**

`lspec` produces image files with spectrograms of whole sound files split into multiple rows.

**Usage**

```
lspec(X = NULL, flim = c(0,22), sxrow = 5, rows = 10, collev = seq(-40, 0, 1),
      ovlp = 50, parallel = 1, wl = 512, gr = FALSE, pal = reverse.gray.colors.2,
      cex = 1, it = "jpeg", flist = NULL, redo = TRUE, path = NULL)
```

**Arguments**

<code>X</code>	Data frame with results from <a href="#">manualoc</a> or any data frame with columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end). If given, two red dotted lines are plotted at the start and end of a selection and the selections are labeled with the selection number (and selection comment, if available). Default is NULL.
<code>flim</code>	A numeric vector of length 2 indicating the highest and lowest frequency limits (kHz) of the spectrogram, as in <a href="#">spectro</a> . Default is c(0,22).
<code>sxrow</code>	A numeric vector of length 1. Specifies seconds of spectrogram per row. Default is 5.
<code>rows</code>	A numeric vector of length 1. Specifies number of rows per image file. Default is 10.
<code>collev</code>	A numeric vector of length 3. Specifies levels to partition the amplitude range of the spectrogram (in dB). The more levels the higher the resolution of the spectrogram. Default is seq(-40, 0, 1).
<code>ovlp</code>	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 50. High values of ovlp slow down the function but produce more accurate selection limits (when X is provided).
<code>parallel</code>	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).
<code>wl</code>	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
<code>gr</code>	Logical argument to add grid to spectrogram. Default is FALSE.
<code>pal</code>	Color palette function for spectrogram. Default is reverse.gray.colors.2. See <a href="#">spectro</a> for more palettes.
<code>cex</code>	A numeric vector of length 1 giving the amount by which text (including sound file and page number) should be magnified. Default is 1.
<code>it</code>	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
<code>flist</code>	character vector or factor indicating the subset of files that will be analyzed. Ignored if X is provided.
<code>redo</code>	Logical argument. If TRUE all selections will be analyzed again when code is rerun. If FALSE only the selections that do not have a image file in the working directory will be analyzed. Default is FALSE.
<code>path</code>	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

The function creates spectrograms for complete sound files, printing the name of the sound files and the "page" number (p1-p2...) at the upper right corner of the image files. If results from [manualoc](#) are supplied (or an equivalent data frame), the function delimits and labels the selections. This function aims to facilitate visual inspection of multiple files as well as visual classification of vocalization units and the analysis of animal vocal sequences.

## Value

image files with spectrograms of whole sound files in the working directory. Multiple pages can be returned, depending on the length of each sound file.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## Examples

```
## Not run:
# First create empty folder
setwd(tempdir())
# save sound file examples
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")

lspec(sxrow = 2, rows = 8, pal = reverse.heat.colors, wl = 300)

# including selections
lspec(sxrow = 2, rows = 8, X = manualoc.df, pal = reverse.heat.colors, redo = TRUE, wl = 300)

check this folder
getwd()

## End(Not run)
```

---

lspec2pdf

lspec2pdf *combines lspec images in .jpeg format to a single pdf file.*


---

## Description

lspec2pdf combines lspec images in .jpeg format to a single pdf file.

## Usage

```
lspec2pdf(keep.jpeg = TRUE, overwrite = FALSE, parallel = 1, path = NULL)
```

## Arguments

keep.jpeg	Logical argument. Indicates whether jpeg files should be kept (default) or remove. (including sound file and page number) should be magnified. Default is 1.
overwrite	Logical argument. If TRUE all jpeg pdf will be produced again when code is rerun. If FALSE only the ones missing will be produced. Default is FALSE.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

The function combines spectrograms for complete sound files from the [lspec](#) function into a single pdf (for each sound file).

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## Examples

```
## Not run:
# First create empty folder
setwd(tempdir())

# save sound file examples
data(list = c("Phae.long1", "Phae.long2"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")

lspec(sxrow = 2, rows = 8, pal = reverse.heat.colors, wl = 300, it = "jpeg")

#now create single pdf removing jpeg
lspec2pdf(keep.jpeg = FALSE)

check this floder
getwd()

## End(Not run)
```

## Description

manualoc produces an interactive spectrographic view in which the start and end times of acoustic signals can be measured.

**Usage**

```
manualoc(wl = 512, flim = c(0,12), seltime = 1, tdisp = NULL, reccomm =
  FALSE, wn = "hanning", title = TRUE, selcomm = FALSE, osci = FALSE, player =
  NULL, pal = reverse.gray.colors.2, path = NULL, flist = NULL)
```

**Arguments**

<code>wl</code>	A numeric vector of length 1 specifying the spectrogram window length. Default is 512.
<code>flim</code>	A numeric vector of length 2 specifying the frequency limit (in kHz) of the spectrogram, as in the function <a href="#">spectro</a> . Default is <code>c(0,12)</code> .
<code>seltime</code>	A numeric vector of length 1 indicating the time interval in seconds at which the spectrograms are produced with higher resolution ( <code>ovlp = 70</code> ) and oscilograms (if <code>osci = TRUE</code> ). Default is 1 second.
<code>tdisp</code>	A numeric vector of length 1 specifying the length in seconds of the total sound file to be displayed. Default is <code>NULL</code> which displays the full sound file.
<code>reccomm</code>	Logical argument. If <code>TRUE</code> pops up a comment window at the end of each sound file. The comment needs to be quoted. Default is <code>FALSE</code> .
<code>wn</code>	A character vector of length 1 specifying the window function (by default "hanning"). See function <a href="#">ftwindow</a> for more options.
<code>title</code>	Logical argument. If <code>TRUE</code> the name of the sound file will be printed as the main title of the spectrogram window. Default is <code>TRUE</code>
<code>selcomm</code>	Logical argument. If <code>TRUE</code> pops up a comment window after each selection. The comment is printed as a label on the selected unit. The comment must be quoted. Default is <code>FALSE</code>
<code>osci</code>	Logical argument. If <code>TRUE</code> adds a oscillogram whenever the spectrograms are produced with higher resolution (see <code>seltime</code> ). Default is <code>FALSE</code> .
<code>player</code>	Path to or name of a program capable of playing a wave file by invocation from the command line. If under Windows and no player is given, windows player will be chosen as the default. "vlc" works in Linux if vlc player is installed. The external program must be closed before resuming analysis. Default is <code>NULL</code> .
<code>pal</code>	A color palette function to be used to assign colors in the plot, as in <a href="#">spectro</a> . Default is <code>reverse.gray.colors.2</code> . See Details.
<code>path</code>	Character string containing the directory path where the sound files are located. If <code>NULL</code> (default) then the current working directory is used.
<code>flist</code>	character vector or factor indicating the subset of files that will be analyzed. Ignored if <code>X</code> is provided.

**Details**

Users can zoom-in a specific sound file segment by clicking at the start and end (left side and right side) of the segment. To select the start and end of a vocalization unit the users need to click at the end and then at the start (right side and left side) of the unit. In addition, 6 "buttons" are provided at the upper right side of the spectrogram that allow to display a full view of the spectrogram ("Full view"), go back to the previous view ("Previous view"), stop the analysis ("Stop"), go to the



next sound file ("Next rec"), play the current view using external software ("Play", see "player" argument), or delete the last manual selection in the current sound file ("Del-sel"). When a unit has been selected, the function plots a red circle with the selection number in the middle point of the selection in the spectrogram. It also plots vertical dotted lines at the start and end of the selection. The circle and lines "disappear" when the selection is deleted ("Del-sel" button). Only the last selection can be deleted.

The function produces a .csv file (manualoc\_output.csv) with information about the .wav file name, selection number, start and end time, selection comment (selcomm), and sound file comment (rec-comm). The file is saved in the working directory and is updated every time the user moves into the next sound file (Next rec "button") or stop the process (Stop "button"). When resuming the process (after "stop" and re-running the function in the same working directory), the function will keep the previous selections and will only pick up .wav files that are not present in the .csv file (not previously analyzed). When users go to the next sound file (Next rec "button") without making any selection the file is still included in the .csv file, with NA's in the "end", "time" and "selec" field.

Windows length (wl) controls the temporal and frequency precision of the spectrogram. A high "wl" value increases the frequency resolution but reduces the temporal resolution, and vice versa. Any color palette that comes with the seewave package can be used: temp.colors, reverse.gray.colors.1, reverse.gray.colors.2, reverse.heat.colors, reverse.terrain.colors, reverse.topo.colors, reverse.cm.colors, heat.colors, terrain.colors, topo.colors, cm.colors. The function is slow when working on files of length > 5min. In most cases other sound analysis softwares for manually selecting acoustic signals (e.g. Raven, Syrinx) should be preferred.

### Value

.csv file saved in the working directory with start and end time of selections.

### Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

### See Also

[seltailor](#)

### Examples

```
## Not run:
#First create empty folder
setwd(tempdir())

# save wav file examples
data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "Phae.long4"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")
writeWave(Phae.long4, "Phae.long4.wav")

manualoc(wl = 300)
# need to use the buttoms to manipulate function
# check working directory for .csv file after stopping function
```

```
#check here:
getwd()

## End(Not run)
```

---

manualoc.df	<i>Data frame of manualoc() selections.</i>
-------------	---

---

### Description

A data frame containing information for calls selected using [manualoc](#).

### Usage

```
data(manualoc.df)
```

### Format

A data frame with 11 rows and 6 variables:

**sound.files** recording names  
**selec** selection numbers within recording  
**start** start times of selected call or element  
**end** end times of selected call or element  
**sel.comment** selection comments  
**rec.comment** recording comments

### Source

Marcelo Araya Salas, warbleR

---

mp32wav	<i>Convert .mp3 files to .wav</i>
---------	-----------------------------------

---

### Description

mp32wav converts several .mp3 files in working directory to .wav format

### Usage

```
mp32wav(samp.rate = 44.1, parallel = 1, from.path = NULL, to.path = NULL,
normalize = NULL)
```

**Arguments**

samp.rate	Sampling rate at which the .wav files should be written. The maximum permitted is 44.1 kHz (default). Units should be kHz.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
from.path	Character string containing the directory path where the .mp3 files are located. If NULL (default) then the current working directory is used.
to.path	Character string containing the directory path where the .wav files will be saved. If NULL (default) then the current working directory is used.
normalize	Character string containing the units to be used for amplitude normalization. Check ( <a href="#">normalize</a> ) for details. If NULL (default) no normalization is carried out.

**Details**

convert all .mp3 files in working directory to .wav format. Function used internally to read .mp3 files ([readMP3](#)) sometimes crashes.

**Value**

.wav files saved in the working directory with same name as original mp3 files.

**Author(s)**

Marcelo Araya-Salas (<araya-salas@cornell.edu>) and Grace Smith Vidaurre

**Examples**

```
## Not run:
# First set temporary folder
setwd(tempdir())

#Then download mp3 files from xeno-canto
querxc(qword = "Phaethornis aethopygus", download = TRUE)

# Convert all files to .wav format
mp32wav()

#check this folder!!
getwd()

## End(Not run)
```

---

Phae.long1

*Acoustic recording #1 of Phaethornis longirostris*

---

**Description**

Acoustic recording #1 of *Phaethornis longirostris* (Long-billed Hermit).

**Usage**

data(Phae.long1)

**Format**

One .wav file:

**Phae.long1** *Phaethornis longirostris* #1 recording

**Source**

<http://www.xeno-canto.org/contributor/EMCWQLLKEW>

---

Phae.long2

*Acoustic recording #2 of Phaethornis longirostris*

---

**Description**

Acoustic recording #2 of *Phaethornis longirostris* (Long-billed Hermit).

**Usage**

data(Phae.long2)

**Format**

One .wav file:

**Phae.long2** *Phaethornis longirostris* #2 recording

**Source**

<http://www.xeno-canto.org/contributor/EMCWQLLKEW>

---

Phae.long3Acoustic recording #3 of *Phaethornis longirostris*

---

**Description**

Acoustic recording #3 of *Phaethornis longirostris* (Long-billed Hermit).

**Usage**

data(Phae.long3)

**Format**

One .wav file:

**Phae.long3** *Phaethornis longirostris* #3 recording

**Source**

<http://www.xeno-canto.org/contributor/EMCWQLLKEW>

---

Phae.long4Acoustic recording #1 of *Phaethornis longirostris*

---

**Description**

Acoustic recording #4 of *Phaethornis longirostris* (Long-billed Hermit).

**Usage**

data(Phae.long4)

**Format**

One .wav file:

**Phae.long4** *Phaethornis longirostris* #4 recording

**Source**

<http://www.xeno-canto.org/contributor/EMCWQLLKEW>

querxc

*Access Xeno-Canto recordings and metadata*

## Description

querxc downloads recordings and metadata from Xeno-Canto (<http://www.xeno-canto.org/>).

## Usage

```
querxc(qword, download = FALSE, X = NULL, file.name = c("Genus", "Specific_epithet"),
parallel = 1, path = NULL)
```

## Arguments

qword	Character vector of length one indicating the genus, or genus and species, to query Xeno-Canto database. For example, <i>Phaethornis</i> or <i>Phaethornis longirostris</i> . ( <a href="http://www.xeno-canto.org/">http://www.xeno-canto.org/</a> ).
download	Logical argument. If FALSE only the recording file names and associated metadata are downloaded. If TRUE, recordings are also downloaded to the working directory as .mp3 files. Default is FALSE. Note that if the recording is already in the working directory (as when the downloading process has been interrupted) it will be skipped. Hence, resuming downloading processes will not start from scratch.
X	Data frame with a 'Recording_ID' column and any other column listed in the file.name argument. Only the recordings listed in the data frame will be download (download argument is automatically set to TRUE). This can be used to select the recordings to be downloaded based on their attributes.
file.name	Character vector indicating the tags (or column names) to be included in the sound file names (if download = TRUE). Several tags can be included. If NULL only the Xeno-Canto recording identification number ("Recording_ID") is used. Default is c("Genus", "Specific_epithet"). Note that recording id is always used (whether or not is listed by users) to avoid duplicated names.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function queries for avian vocalization recordings in the open-access online repository Xeno-Canto (<http://www.xeno-canto.org/>). It can return recordings metadata or download the associated sound files.

**Value**

If X is not provided the function returns a data frame with the following recording information: recording ID, Genus, Specific epithet, Subspecies, English name, Recordist, Country, Locality, Latitude, Longitude, Vocalization type, Audio file, License, URL, Quality, Time, Date. Sound files in .mp3 format are downloaded into the working directory if download = TRUE or if X is provided; a column indicating the names of the downloaded files is included in the output data frame.

**Author(s)**

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

**Examples**

```
## Not run:
# First create empty folder
setwd(tempdir())

#search without downloading
df1 <- querxc(qword = "Phaethornis anthophilus", download = FALSE)
View(df1)

#downloading files
querxc(qword = "Phaethornis anthophilus", download = TRUE)
#check this folder!!
getwd()

## End(Not run)
```

---

selection.files	<i>Selections files from Raven and Syrinx.</i>
-----------------	--

---

**Description**

Selections files from Raven and Syrinx.

**Usage**

```
data(selection.files)
```

**Format**

**selection.files** Selections from the commercial software ‘Raven’ and ‘Syrinx’

seltailor

*Interactive view of spectrograms to tailor start and end of selections*

## Description

seltailor produces an interactive spectrographic view (similar to [manualoc](#)) in which the start and end times of acoustic signals listed in a data frame can be adjusted.

## Usage

```
seltailor(X = NULL, wl = 512, flim = c(0,22), wn = "hanning", mar = 0.5,
osci = FALSE, pal = reverse.gray.colors.2, ovlp = 70, auto.next = FALSE, pause = 1,
comments = TRUE, path = NULL)
```

## Arguments

X	data frame with the following columns: 1) "sound.files": name of the .wav files, 2) "selec": number of the selections, 3) "start": start time of selections, 4) "end": end time of selections. The output of <a href="#">seltailor</a> or <a href="#">autodetec</a> can be used as the input data frame. Other data frames can be used as input, but must have at least the 4 columns mentioned above. Required. Notice that, if an output file ("seltailor_output.csv") is found in the working directory it will be given priority over an input data frame.
wl	A numeric vector of length 1 specifying the spectrogram window length. Default is 512.
flim	A numeric vector of length 2 specifying the frequency limit (in kHz) of the spectrogram, as in the function <a href="#">spectro</a> . Default is c(0,22).
wn	A character vector of length 1 specifying the window function (by default "hanning"). See function <a href="#">ftwindow</a> for more options.
mar	Numeric vector of length 1. Specifies the margins adjacent to the start and end points of the selections to define spectrogram limits. Default is 0.5.
osci	Logical argument. If TRUE adds a oscillogram whenever the spectrograms are produced with higher resolution (see seltime). Default is FALSE. The external program must be closed before resuming analysis. Default is NULL.
pal	A color palette function to be used to assign colors in the plot, as in <a href="#">spectro</a> . Default is reverse.gray.colors.2. See Details.
ovlp	Numeric vector of length 1 specifying the percent overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 70.
auto.next	Logical argument to control whether the functions moves automatically to the next selection. The time interval before moving to the next selection is controled by the 'pause' argument.
pause	Numeric vector of length 1. Controls the duration of the waiting period before moving to the next selection (in seconds). Default is 1.



comments	Logical argument specifying if 'sel.comment' (when in data frame) should be included in the title of the spectrograms. Default is TRUE.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function produces an interactive spectrographic view (similar to [manualoc](#)) in which users can select a new start and end of a vocalization unit (e.g. elements) by clicking at the end and then at the start of the signal (in any order). In addition, 2 "buttons" are provided at the upper right side of the spectrogram that allow to stop the analysis ("Stop") or go to the next sound file ("next sel"). When a unit has been selected, the function plots red dotted lines in the start and end of the selection in the spectrogram. The lines "disappear" when a new selections is made. Only the last selection is kept for each selection that is adjusted. The function produces a .csv file (seltailor\_output.csv) with the same information than the input data frame, except for the new time coordinates, plus a new column (X\$tailored) indicating if the selection has been tailored. The file is saved in the working directory and is updated every time the user moves into the next sound file (next sel "button") or stop the process (Stop "button"). If no selection (by clicking on the 'next' button) the original time coordinates are kept. When resuming the process (after "stop" and re-running the function in the same working directory), the function will continue working on the selections that have not been analyzed.

Windows length (wl) controls the temporal and frequency precision of the spectrogram. A high "wl" value increases the frequency resolution but reduces the temporal resolution, and vice versa. Any color palette that comes with the seewave package can be used: temp.colors, reverse.gray.colors.1, reverse.gray.colors.2, reverse.heat.colors, reverse.terrain.colors, reverse.topo.colors, reverse.cm.colors, heat.colors, terrain.colors, topo.colors, cm.colors. Note that, unlike [manualoc](#), you cannot zoom in the spectrogram [seltailor](#). The zoom can be adjusted by setting the mar argument.

## Value

.csv file saved in the working directory with start and end time of selections.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## See Also

[manualoc](#)

## Examples

```
## Not run:
#First create empty folder
setwd(tempdir())

data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "Phae.long4", "manualoc.df"))
writeWave(Phae.long1,"Phae.long1.wav")
writeWave(Phae.long2,"Phae.long2.wav")
writeWave(Phae.long3,"Phae.long3.wav")
```

```

writeWave(Phae.long4,"Phae.long4.wav")

seltailor(X = manualoc.df, flim = c(1,12), wl = 300, auto.next = FALSE)

# need to use the buttoms to manipulate function

# Read output .csv file
seltailor.df <- read.csv("seltailor_output.csv")
seltailor.df

# check this directory for .csv file after stopping function
getwd()

## End(Not run)

```

---

sig2noise	<i>Measure signal-to-noise ratio</i>
-----------	--------------------------------------

---

## Description

sig2noise measures signal-to-noise ratio across multiple files.

## Usage

```
sig2noise(X, mar, parallel = 1, path = NULL)
```

## Arguments

X	Data frame with results from <a href="#">manualoc</a> or any data frame with columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end).
mar	numeric vector of length 1. Specifies the margins adjacent to the start and end points of selection over which to measure noise.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

Signal-to-noise ratio (SNR) is a measure of the level of a desired signal compared to background noise. The function divides the mean amplitude of the signal by the mean amplitude of the background noise adjacent to the signal. A general margin to apply before and after the acoustic signal must be specified. Setting margins for individual signals that have been previously clipped from larger files may take some optimization, as for calls within a larger file that are irregularly separated. When margins overlap with another acoustic signal nearby, the signal-to-noise ratio (SNR) will be inaccurate. Any SNR less than or equal to one suggests background noise is equal to or overpowering the acoustic signal. [snrspecs](#) can be used to troubleshoot different noise margins.

**Value**

Data frame similar to `autodetec` output, but also includes a new variable with the signal-to-noise values.

**Author(s)**

Marcelo Araya-Salas (<araya-salas@cornell.edu>) and Grace Smith Vidaurre

**Source**

[https://en.wikipedia.org/wiki/Signal-to-noise\\_ratio](https://en.wikipedia.org/wiki/Signal-to-noise_ratio)

**Examples**

```
## Not run:
# First set temporary folder
setwd(tempdir())

data(list = c("Phae.long1", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav") #save sound files

# specifying the correct margin is important
# use snrspecs to troubleshoot margins for sound files
sig2noise(manualoc.df[grep("Phae.long1", manualoc.df$sound.files), ], mar = 0.2)

# this smaller margin doesn't overlap neighboring signals
sig2noise(manualoc.df[grep("Phae.long1", manualoc.df$sound.files), ], mar = 0.1)

## End(Not run)
```

---

sim.coor.sing

*Simulated coordinated singing events.*

---

**Description**

`sim.coor.sing` Selections of simulated interactive singing events. The simulated events use the mean and standard deviation of real lekking *Phaethornis longirostris* (Long-billed Hermit hummingbird) songs and intervals between songs (e.i gaps). Three events are simulated: overlapping signals (ovlp), alternating signals (altern) and non-synchronized signals (uncoor).

**Usage**

```
data(sim.coor.sing)
```

**Format**

**sim.coor.sing** Simulated coordinated singing events that overlap and do not overlap most of the time, for use with `coor.test`

snrspecs

*Spectrograms with background noise margins***Description**

snrspecs creates spectrograms to visualize margins over which background noise will be measured by [sig2noise](#).

**Usage**

```
snrspecs(X, wl = 512, flim = c(0, 22), wn = "hanning", ovlp = 70,
inner.mar = c(5, 4, 4, 2), outer.mar = c(0, 0, 0, 0), picsize = 1,
res = 100, cexlab = 1, title = TRUE,
propwidth= FALSE, xl=1, osci = FALSE, gr = FALSE, sc = FALSE, mar = 0.2,
snrmar = 0.1, it = "jpeg", parallel = 1, path = NULL)
```

**Arguments**

X	Data frame with results from <a href="#">manualoc</a> or any data frame with columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end).
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
flim	A numeric vector of length 2 for the frequency limit in kHz of the spectrogram, as in <a href="#">spectro</a> . Default is c(0, 22).
wn	Character vector of length 1 specifying window name. Default is "hanning". See function <a href="#">ftwindow</a> for more options.
ovlp	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 70.
inner.mar	Numeric vector with 4 elements, default is c(5,4,4,2). Specifies number of lines in inner plot margins where axis labels fall, with form c(bottom, left, top, right). See <a href="#">par</a> .
outer.mar	Numeric vector with 4 elements, default is c(0,0,0,0). Specifies number of lines in outer plot margins beyond axis labels, with form c(bottom, left, top, right). See <a href="#">par</a> .
picsize	Numeric argument of length 1, controls relative size of spectrogram. Default is 1.
res	Numeric argument of length 1 that controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
cexlab	Numeric vector of length 1 specifying relative size of axis labels. See <a href="#">spectro</a> .
title	Logical argument to add a title to individual spectrograms. Default is TRUE.
propwidth	Logical argument to scale the width of spectrogram proportionally to duration of the selected call. Default is FALSE.

x1	Numeric vector of length 1, a constant by which to scale spectrogram width if propwidth = TRUE. Default is 1.
osci	Logical argument to add an oscillogram underneath spectrogram, as in <a href="#">spectro</a> . Default is FALSE.
gr	Logical argument to add grid to spectrogram. Default is FALSE.
sc	Logical argument to add amplitude scale to spectrogram, default is FALSE.
mar	Numeric vector of length 1. Specifies the margins adjacent to the start and end points of the selections to define spectrogram limits. Default is 0.2.
snrmar	Numeric vector of length 1. Specifies the margins adjacent to the start and end points of the selections where noise will be measured. Default is 0.1.
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function can be used to test different margins to facilitate accurate SNR measurements when using [sig2noise](#) down the line. Setting margins for individual calls that have been previously clipped from larger files may take some optimization, as for calls within a larger file that are irregularly separated. Setting inner.mar to c(4,4.5,2,1) and outer.mar to c(4,2,2,1) works well when picsize = 2 or 3. Title font size, inner.mar and outer.mar (from mar and oma in par) don't work well when osci or sc = TRUE, this may take some optimization by the user.

## Value

Spectrograms per selection marked with margins where background noise will be measured.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>) and Grace Smith Vidaurre

## Source

[https://en.wikipedia.org/wiki/Signal-to-noise\\_ratio](https://en.wikipedia.org/wiki/Signal-to-noise_ratio)

## See Also

[trackfreqs](#) for creating spectrograms to visualize frequency measurements by [specan](#), [speccreator](#) for creating spectrograms after using [manualoc](#)

Other spectrogram creators: [dfDTW](#), [dfts](#), [ffDTW](#), [ffts](#), [speccreator](#), [trackfreqs](#)

## Examples

```
## Not run:
# First create empty folder
setwd(tempdir())

data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav") #save sound.files
writeWave(Phae.long2, "Phae.long2.wav")

# make Phae.long1 and Phae.long2 spectrograms
# snrmar needs to be smaller before moving on to sig2noise()

snrspecs(manualoc.df, flim = c(0, 14), inner.mar = c(4,4.5,2,1), outer.mar = c(4,2,2,1),
picsize = 2, res = 300, cexlab = 2, mar = 0.2, snrmar = 0.1, it = "jpeg", wl = 300)

# make only Phae.long1 spectrograms
# snrmar now doesn't overlap neighboring signals

snrspecs(manualoc.df[grepl(c("Phae.long1"), manualoc.df$sound.files), ], flim = c(3, 14),
inner.mar = c(4,4.5,2,1), outer.mar = c(4,2,2,1), picsize = 2, res = 300, cexlab = 2,
mar = 0.2, snrmar = 0.01, wl = 300)

#check this folder!!
getwd()

## End(Not run)
```

---

specan

---

*Measure acoustic parameters in batches of sound files*


---

## Description

specan measures acoustic parameters on acoustic signals for which the start and end times are provided.

## Usage

```
specan(X, bp = c(0,22), wl = 512, threshold = 15, parallel = 1, fast = TRUE, path = NULL)
```

## Arguments

X	Data frame with the following columns: 1) "sound.files": name of the .wav files, 2) "sel": number of the selections, 3) "start": start time of selections, 4) "end": end time of selections. The output of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame.
bp	Numeric vector of length 2 giving the lower and upper limits of the frequency bandpass filter (in kHz). Default is c(0, 22).

wl	A numeric vector of length 1 specifying the spectrogram window length. Default is 512.
threshold	amplitude threshold (%) for fundamental frequency and dominant frequency detection. Default is 15.
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). For windows OS the <code>parallelsugar</code> package should be installed.
fast	Logical. If TRUE (default) then the <code>peakf</code> acoustic parameter (see below) is not computed, which substantially increases performance (~9 times faster).
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

### Details

The output of `manualoc` or `autodetec` can be used directly without any additional modification. The function measures 25 acoustic parameters (if `fast = TRUE`) on each selection in the data frame. Most parameters are produced internally by `specprop`, `fpeaks`, `fund`, and `dfreq` from the package `seewave`.

### Value

Data frame with the following acoustic parameters:

- `duration`: length of signal
- `meanfreq`: mean frequency (in kHz)
- `sd`: standard deviation of frequency
- `median`: median frequency (in kHz)
- `Q25`: first quantile (in kHz)
- `Q75`: third quantile (in kHz)
- `IQR`: interquantile range (in kHz)
- `skew`: skewness (see note in `specprop` description)
- `kurt`: kurtosis (see note in `specprop` description)
- `sp.ent`: spectral entropy
- `sfm`: spectral flatness
- `mode`: mode frequency
- `centroid`: frequency centroid (see `specprop`)
- `peakf`: peak frequency (frequency with highest energy)
- `meanfun`: average of fundamental frequency measured across acoustic signal
- `minfun`: minimum fundamental frequency measured across acoustic signal
- `maxfun`: maximum fundamental frequency measured across acoustic signal
- `meandom`: average of dominant frequency measured across acoustic signal
- `mindom`: minimum of dominant frequency measured across acoustic signal

- maxdom: maximum of dominant frequency measured across acoustic signal
- dfrange: range of dominant frequency measured across acoustic signal
- modindx: modulation index. Calculated as the accumulated absolute difference between adjacent measurements of dominant frequencies divided by the dominant frequency range
- startdom: dominant frequency measurement at the start of the signal
- enddom: dominant frequency measurement at the end of the signal
- dfslope: slope of the change in dominant through time  $([enddom - startdom] / duration)$

### Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>) and Grace Smith Vidaurre

### Examples

```
## Not run:
# First set temporary folder
setwd(tempdir())

data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "Phae.long4", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")
writeWave(Phae.long4, "Phae.long4.wav")

a <- specan(X = manualoc.df, bp = c(0, 22))

# using a diferent threshold
a <- specan(X = manualoc.df, bp = c(0, 22), threshold = 20)
# View(a)

## End(Not run)
```

---

speccreator

*Spectrograms of selected signals*

---

### Description

speccreator creates spectrograms of signals selected by [manualoc](#) or [autodetec](#).

### Usage

```
speccreator(X, wl = 512, flim = c(0, 22), wn = "hanning", pal
= reverse.gray.colors.2, ovlp = 70, inner.mar = c(5, 4, 4, 2), outer.mar =
c(0, 0, 0, 0), picsize = 1, res = 100, cexlab = 1, title = TRUE,
propwidth = FALSE, xl = 1, osci = FALSE, gr = FALSE, sc = FALSE, line = TRUE,
mar = 0.05, it = "jpeg", parallel = 1, path = NULL)
```



**Arguments**

<code>x</code>	Data frame with results containing columns for sound file name ( <code>sound.files</code> ), selection number ( <code>selec</code> ), and start and end time of signals ( <code>start</code> and <code>end</code> ). The output of <code>manualoc</code> or <code>autodetec</code> can be used as the input data frame.
<code>wl</code>	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
<code>flim</code>	A numeric vector of length 2 for the frequency limit (in kHz) of the spectrogram, as in <code>spectro</code> . Default is <code>c(0, 22)</code> .
<code>wn</code>	Character vector of length 1 specifying window name. Default is "hanning". See function <code>ftwindow</code> for more options.
<code>pal</code>	A color palette function to be used to assign colors in the plot, as in <code>spectro</code> . Default is <code>reverse.gray.colors.2</code> .
<code>ovlp</code>	Numeric vector of length 1 specifying the percent overlap between two consecutive windows, as in <code>spectro</code> . Default is 70.
<code>inner.mar</code>	Numeric vector with 4 elements, default is <code>c(5,4,4,2)</code> . Specifies number of lines in inner plot margins where axis labels fall, with form <code>c(bottom, left, top, right)</code> . See <code>par</code> .
<code>outer.mar</code>	Numeric vector with 4 elements, default is <code>c(0,0,0,0)</code> . Specifies number of lines in outer plot margins beyond axis labels, with form <code>c(bottom, left, top, right)</code> . See <code>par</code> .
<code>picsize</code>	Numeric argument of length 1. Controls relative size of spectrogram. Default is 1. Ignored when <code>propwidth</code> is TRUE.
<code>res</code>	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
<code>cexlab</code>	Numeric vector of length 1 specifying the relative size of axis labels. See <code>spectro</code> .
<code>title</code>	Logical argument to add a title to individual spectrograms. Default is TRUE.
<code>propwidth</code>	Logical argument to scale the width of spectrogram proportionally to duration of the selection. Default is FALSE.
<code>xl</code>	Numeric vector of length 1. A constant by which to scale spectrogram width if <code>propwidth</code> = TRUE. Default is 1.
<code>osci</code>	Logical argument to add an oscillogram underneath spectrogram, as in <code>spectro</code> . Default is FALSE.
<code>gr</code>	Logical argument to add grid to spectrogram. Default is FALSE.
<code>sc</code>	Logical argument to add amplitude scale to spectrogram, default is FALSE.
<code>line</code>	Logical argument to add red lines at start and end times of selection. Default is TRUE.
<code>mar</code>	Numeric vector of length 1. Specifies the margins adjacent to the start and end points of selections, dealineating spectrogram limits. Default is 0.05.
<code>it</code>	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".

parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

## Details

This function provides access to bath process of the [spectro](#) function from the 'seewave' package. The function creates spectrograms for visualization of vocalizations. Setting inner.mar to c(4,4.5,2,1) and outer.mar to c(4,2,2,1) works well when picsize = 2 or 3. Title font size, inner.mar and outer.mar (from mar and oma) don't work well when osci or sc = TRUE, this may take some optimization by the user.

## Value

Image files containing spectrograms of the signals listed in the input data frame.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>) and Grace Smith Vidaurre

## See Also

[trackfreqs](#) for creating spectrograms to visualize frequency measurements by [specan](#), [snrspecs](#) for creating spectrograms to optimize noise margins used in [sig2noise](#)

Other spectrogram.creators: [dfDTW](#), [dfts](#), [ffDTW](#), [ffts](#), [snrspecs](#), [trackfreqs](#)

## Examples

```
## Not run:
# First set empty folder
setwd(tempdir())
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav") #save sound files
writeWave(Phae.long2, "Phae.long2.wav")

# make spectrograms

speccreator(manualoc.df, flim = c(0, 11), res = 300, mar = 0.05, wl = 300)

#' #check this folder!!
getwd()

## End(Not run)
```

**Description**

trackfreqs creates spectrograms to visualize dominant and fundametal frequency measurements (contours) of signals selected by [manualoc](#) or [autodetec](#).

**Usage**

```
trackfreqs(X, wl = 512, flim = c(0, 22), wn = "hanning", pal =
  reverse.gray.colors.2, ovlp = 70, inner.mar = c(5, 4, 4, 2), outer.mar =
  c(0, 0, 0, 0), picsize = 1, res = 100, cexlab = 1, title = TRUE, propwidth = FALSE,
  xl = 1, osci = FALSE, gr = FALSE, sc = FALSE, bp = c(0, 22), cex = c(0.6, 1),
  threshold = 15, contour = "both", col = c("skyblue", "red2"),
  pch = c(21, 24), mar = 0.05, lpos = "topright", it = "jpeg", parallel = 1,
  path = NULL, img.suffix = NULL, custom.contour = NULL, pb = TRUE)
```

**Arguments**

X	Data frame with results containing columns for sound file name (sound.files), selection number (selec), and start and end time of signal (start and end). The ouptut of <a href="#">manualoc</a> or <a href="#">autodetec</a> can be used as the input data frame.
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
flim	A numeric vector of length 2 for the frequency limit of the spectrogram (in kHz), as in <a href="#">spectro</a> . Default is c(0, 22).
wn	Character vector of length 1 specifying window name. Default is "hanning". See function <a href="#">ftwindow</a> for more options.
pal	A color palette function to be used to assign colors in the plot, as in <a href="#">spectro</a> . Default is reverse.gray.colors.2.
ovlp	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 70.
inner.mar	Numeric vector with 4 elements, default is c(5,4,4,2). Specifies number of lines in inner plot margins where axis labels fall, with form c(bottom, left, top, right). See <a href="#">par</a> .
outer.mar	Numeric vector with 4 elements, default is c(0,0,0,0). Specifies number of lines in outer plot margins beyond axis labels, with form c(bottom, left, top, right). See <a href="#">par</a> .
picsize	Numeric argument of length 1. Controls relative size of spectrogram. Default is 1.
res	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.

<code>cexlab</code>	Numeric vector of length 1 specifying the relative size of axis labels. See <a href="#">spectro</a> .
<code>title</code>	Logical argument to add a title to individual spectrograms. Default is TRUE.
<code>propwidth</code>	Logical argument to scale the width of spectrogram proportionally to duration of the selected call. Default is FALSE.
<code>x1</code>	Numeric vector of length 1. A constant by which to scale spectrogram width. Default is 1.
<code>osci</code>	Logical argument to add an oscillogram underneath spectrogram, as in <a href="#">spectro</a> . Default is FALSE.
<code>gr</code>	Logical argument to add grid to spectrogram. Default is FALSE.
<code>sc</code>	Logical argument to add amplitude scale to spectrogram, default is FALSE.
<code>bp</code>	A numeric vector of length 2 for the lower and upper limits of a frequency bandpass filter (in kHz). Default is <code>c(0, 22)</code> .
<code>cex</code>	Numeric vector of length 2, specifies relative size of points plotted for frequency measurements and legend font/points, respectively. See <a href="#">spectro</a> .
<code>threshold</code>	amplitude threshold (%) for fundamental frequency and dominant frequency detection. Default is 15.
<code>contour</code>	Character vector, one of "df", "ff" or "both", specifying whether the dominant or fundamental frequencies or both should be plotted. Default is "both".
<code>col</code>	Vector of length 1 or 2 specifying colors of points plotted to mark fundamental and dominant frequency measurements respectively (if both are plotted). Default is <code>c("skyblue", "red2")</code> .
<code>pch</code>	Numeric vector of length 1 or 2 specifying plotting characters for the frequency measurements. Default is <code>c(21, 24)</code> .
<code>mar</code>	Numeric vector of length 1. Specifies the margins adjacent to the selections to set spectrogram limits. Default is 0.05.
<code>lpos</code>	Character vector of length 1 or numeric vector of length 2, specifying position of legend. If the former, any keyword accepted by <code>xy.coords</code> can be used (see below). If the latter, the first value will be the x coordinate and the second value the y coordinate for the legend's position. Default is "topright".
<code>it</code>	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
<code>parallel</code>	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing).
<code>path</code>	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.
<code>img.suffix</code>	A character vector of length 1 with a suffix (label) to add at the end of the names of image files. Default is NULL.
<code>custom.contour</code>	A data frame with frequency contours for exactly the same sound files and selection as in X. The frequency values are assumed to be equally spaced in between the start and end of the signal. The first 2 columns of the data frame should contain the 'sound.files' and 'selec' columns and should be identical to the corresponding columns in X.
<code>pb</code>	Logical argument to control progress bar. Default is TRUE. Note that progress bar is only used when <code>parallel = 1</code> .

## Details

This function provides visualization of frequency measurements as the ones made by [specan](#). Frequency measures can be made by the function or input by the user (see 'custom.contour' argument). Arguments that are accepted by xy.coords and can be used for 'lpos' are: "bottomright", "bottom", "bottomleft", "left", "topleft", "top", "topright", "right" and "center". Setting inner.mar to c(4,4.5,2,1) and outer.mar to c(4,2,2,1) works well when picsize = 2 or 3. Title font size, inner.mar and outer.mar (from mar and oma) don't work well when osci or sc = TRUE, this may take some optimization by the user. Note that if no amplitude was detected for a particular time bin, then the image will show a dark dot at the bottom of the time bin.

## Value

Spectrograms of the signals listed in the input data frame showing the location of the dominant and fundamental frequencies.

## Author(s)

Grace Smith Vidaurre and Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## See Also

[speccreator](#) for creating spectrograms from selections, [snrspecs](#) for creating spectrograms to optimize noise margins used in [sig2noise](#)

Other spectrogram.creators: [dfDTW](#), [dfts](#), [ffDTW](#), [ffts](#), [snrspecs](#), [speccreator](#)

## Examples

```
## Not run:
#Set temporal folder as working directory
setwd(tempdir())

#load data
data("Cryp.soui")
writeWave(Cryp.soui, "Cryp.soui.wav") #save sound files

#autodetec location of signals
ad <- autodetec(threshold = 6, bp = c(1, 3), mindur = 1.2,
maxdur = 3, img = FALSE, ssmooth = 600, wl = 300, flist = "Cryp.soui.wav")

#track dominant frequency graphs
trackfreqs(X = ad[!is.na(ad$start),], flim = c(0, 5), ovlp = 90, it = "tiff",
bp = c(1, 3), contour = "df", wl = 300)

# Check this folder
getwd()

#track both frequencies
trackfreqs(X = ad[!is.na(ad$start),], flim = c(0, 5), ovlp = 90, it = "tiff",
bp = c(1, 3), contour = "both", wl = 300)
```

```
## End(Not run)
```

---

warbleR

*warbleR: A package to streamline bioacoustic analysis*


---

## Description

warbleR is a package designed to streamline analysis of animal acoustic signals in R. This package allows users to collect open-access avian vocalizations data or input their own data into a workflow that facilitates spectrographic visualization and measurement of acoustic parameters. warbleR makes fundamental sound analysis tools from the R package seewave, as well as new tools not yet offered in the R environment, readily available for batch process analysis. The functions facilitate searching and downloading avian vocalizations from Xeno-Canto <http://www.xeno-canto.org/>, creating maps of Xeno-Canto recordings, converting .mp3 files to .wav files, checking .wav files, automatically detecting acoustic signals, selecting them manually, printing spectrograms of whole recordings or individual signals, measuring signal to noise ratio, cross-correlation and performing acoustic measurements.

The warbleR package offers three overarching categories of functions:

- Obtaining avian vocalization data
- Sound file management
- Streamlined (bio)acoustic analysis in R

## Details

License: GPL ( $\geq 2$ )

## Obtaining avian vocalization data

[querxc](#): Download recordings and metadata from Xeno-Canto

[xcmeps](#): Create maps to visualize the geographic spread of Xeno-Canto recordings

[imp.syrinx](#): Importing Syrinx selections

[imp.raven](#): Importing Raven selections

## Managing sound files

[mp32wav](#): Convert several .mp3 files in working directory to .wav format

[checkwavs](#): Check whether .wav files can be read by subsequent functions and the minimum windows length ("wl" argument) that can be used

[wavedur](#): Determine the duration of sound files

**Streamlining analysis of animal acoustic signal structure in R**

`autodetec`: Automatically detect start and end of acoustic signals

`manualoc`: Interactive spectrographic view to measure start and end of acoustic signals

`autodetec`: Automatic detection of acoustic signals based on amplitude

`lspec`: Produce spectrograms of whole recordings split into multiple rows

`lspec2pdf`: Combine lspec images to single pdf files

`speccreator`: Create spectrograms of manualoc selections

`snrspecs`: Create spectrograms to visualize margins over which noise will be measured by sig2noise

`sig2noise`: Measure signal-to-noise ratio across multiple files

`trackfreqs`: Create spectrograms to visualize frequency measurements

`filtersels`: Filter selection data frames based on filtered image files

`specan`: Measure acoustic parameters on selected acoustic signals

`xcorr`: Pairwise cross-correlation of multiple signals

`xcorr.graph`: Pairwise cross-correlation of multiple signals

`dfts`: Extract the dominant frequency values as a time series

`ffts`: Extract the fundamental frequency values as a time series

`dfDTW`: Calculate acoustic dissimilarity using dynamic time warping on dominant frequency contours

`ffDTW`: Calculate acoustic dissimilarity using dynamic time warping on fundamental frequency contours

`compare.methods`: Produce graphs to visually assess performance of acoustic distance measurements

`coord.graph`: Create graphs of coordinated singing

`coord.test`: Assess statistical significance of singing coordination

`seltailor`: Interactive view of spectrograms to tailor start and end of selections

**Author(s)**

Marcelo Araya-Salas & Grace Smith Vidaurre

Maintainer: Marcelo Araya-Salas (<araya-salas@cornell.edu>)

wavdur

*Measure the duration of sound files*

---

**Description**

wavdur measures the duration of sound files in '.wav' format

**Usage**

```
wavdur(files = NULL, path = NULL)
```

**Arguments**

files	Character vector with the names of the sound files to be measured. The sound files should be in the working directory or in the directory provided in 'path'.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

**Details**

This function returns the duration (in seconds) of sound files.

**Value**

A data frame with the duration (in seconds) of the sound files.

**Author(s)**

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

**Examples**

```
## Not run:
# First create empty folder
setwd(tempdir())

data(list = c("Phae.long1", "Phae.long2", "Phae.long3"))
writeWave(Phae.long1, "Phae.long1.wav")
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")

wavdur()

## End(Not run)
```



xcmaps

*Maps of Xeno-Canto recordings by species***Description**

xcmaps creates maps to visualize the geographic spread of Xeno-Canto recordings.

**Usage**

```
xcmaps(X, img = TRUE, it = "jpeg", res = 100, labels = F)
```

**Arguments**

X	Data frame output from <a href="#">querxc</a> .
img	A logical argument specifying whether an image file of each species map should be returned, default is TRUE.
it	A character vector of length 1 giving the image type to be used. Currently only "tiff" and "jpeg" are admitted. Default is "jpeg".
res	Numeric argument of length 1. Controls image resolution. Default is 100 (faster) although 300 - 400 is recommended for publication/ presentation quality.
labels	A logical argument defining whether dots depicting recording locations are labeled. If TRUE then the Recording_ID is used as label.

**Details**

This function creates maps for visualizing the geographic spread of recordings from the open-access online repository Xeno-Canto (<http://www.xeno-canto.org/>). The function takes the output of [querxc](#) as input. Maps can be displayed in the graphic device or saved as images in the working directory.

**Value**

A map of Xeno-Canto recordings per species (image file), or a faceted plot of species map(s) in the active graphic device.

**Author(s)**

Marcelo Araya-Salas (<araya-salas@cornell.edu>) and Grace Smith Vidaurre

**Examples**

```
## Not run:
X <- querxc("Phaethornis anthophilus", download = FALSE)
View(X)
xcmaps(X)
xcmaps(X, img = FALSE, it = "jpeg")

## End(Not run)
```

xcorr

*Spectrogram cross-correlation***Description**

xcorr estimates the similarity of two spectrograms by means of cross-correlation

**Usage**

```
xcorr(X, wl = 512, frange = NULL, ovlp = 90, dens = 0.9, bp = NULL, wn = 'hanning',
      cor.method = "pearson", parallel = 1, path = NULL)
```

**Arguments**

X	Data frame containing columns for sound files (sound.files), selection number (selec), and start and end time of signal (start and end).
wl	A numeric vector of length 1 specifying the window length of the spectrogram, default is 512.
frange	A numeric vector of length 2 setting the upper and lower frequency limits (in kHz) in which to compare the signals. If not provided (NULL) the <a href="#">dfts</a> function is used internally to define the higher and lower dominant frequency in the signals to be analyzed. This method is more adequate for pure tone signals. Default is NULL.
ovlp	Numeric vector of length 1 specifying % of overlap between two consecutive windows, as in <a href="#">spectro</a> . Default is 90. High values of ovlp slow down the function but produce more accurate results.
dens	Numeric vector of length 1 specifying the approximate density of points in which to sample amplitude. See <a href="#">makeTemplate</a> . Default is 0.9.
bp	A numeric vector of length 2 for the lower and upper limits of a frequency bandpass filter (in kHz) in which to detect dominant frequency. Only applied when frange is NULL. Default is NULL.
wn	A character vector of length 1 specifying the window name as in <a href="#">ftwindow</a> .
cor.method	A character vector of length 1 specifying the correlation method as in <a href="#">cor</a> .
parallel	Numeric. Controls whether parallel computing is applied. It specifies the number of cores to be used. Default is 1 (i.e. no parallel computing). Not available in Windows OS.
path	Character string containing the directory path where the sound files are located. If NULL (default) then the current working directory is used.

**Details**

This function calculates the pairwise similarity of multiple signals by means of spectrogram cross-correlation. This method "slides" one spectrogram over the other calculating a correlation of the amplitude values at each step. The function runs pairwise cross-correlations on several signals

and returns a list including the correlation statistic for each "sliding" step as well as the maximum (peak) correlation for each pairwise comparison. To accomplish this the margins of the signals are expanded by half the duration of the signal both before and after the provided time coordinates. The correlation matrix could have NA's if some of the pairwise correlation did not work (common when sound files have been modified by band-pass filters). This function is a modified version of the [corMatch](#) and [makeTemplate](#) from the awesome R package 'monitoR'.

## Value

A list that includes 1) a data frame with the correlation statistic for each "sliding" step, 2) a matrix with the maximum (peak) correlation for each pairwise comparison, and 3) the frequency range.

## Author(s)

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## Source

H. Khanna, S.L.L. Gaunt & D.A. McCallum (1997). Digital spectrographic cross-correlation: tests of sensitivity. *Bioacoustics* 7(3): 209-234

## See Also

[xcorr.graph](#)

## Examples

```
## Not run:
#First set temporal working directory
setwd(tempdir())

#load data
data(list = c("Phae.long1", "Phae.long2", "Phae.long3", "Phae.long4", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav") #save sound files
writeWave(Phae.long2, "Phae.long2.wav")
writeWave(Phae.long3, "Phae.long3.wav")
writeWave(Phae.long4, "Phae.long4.wav")

xcor <- xcorr(X = manualoc.df, wl = 300, frange = c(2, 9), ovlp = 90,
dens = 1, wn = 'hanning', cor.method = "pearson")

## End(Not run)
```

---

`xcorr.graph`*Pairwise plots of spectrogram cross-correlation scores*

---

## Description

`xcorr.graph` generates pairwise plots showing the spectrogram cross-correlation scores against the time sliding.

## Usage

```
xcorr.graph(X, cex.cor = 1, cex.lab = 1, cex.axis.lab = 1, rel.cex = FALSE, labs = NULL)
```

## Arguments

<code>X</code>	Output from <code>xcorr</code> function.
<code>cex.cor</code>	A numeric vector of length 1 giving the amount by which correlation scores (in the upper triangle of the multipannel plot) should be magnified. Default is 1.
<code>cex.lab</code>	A numeric vector of length 1 giving the amount by which signal selection labels (in diagonal of the multipannel plot) should be magnified. Default is 1.
<code>cex.axis.lab</code>	A numeric vector of length 1 giving the amount by which the axis labels should be magnified. Default is 1.
<code>rel.cex</code>	Logical. Controls whether the size of the correlation scores (in the upper triangle of the multipannel plot) should be relative to the correlation score.
<code>labs</code>	Alternative selection labels. If not provided the combined name of sound files and selection numbers are used as labels. Default is FALSE.

## Details

This function generates pairwise plots of the spectrogram cross-correlation scores by sliding step. The function takes the output of `xcorr` as input. The colors of the lines in the lower triangle of the plot matrix represent the strenght of the similarity between the two signals. The x axis shows the time difference between the two signals for each sliding step (0 means perfectly centered signals). Note that large number of signals may not display well in the default graphic device. In such cases saving the plot as and image file is advised.

## Author(s)

Marcelo Araya-Salas (<araya-salas@cornell.edu>)

## See Also

`xcorr`

**Examples**

```
## Not run:
#load data
#First set temporal working directory]
setwd(tempdir())

#load data
data(list = c("Phae.long1", "Phae.long2", "manualoc.df"))
writeWave(Phae.long1, "Phae.long1.wav") #save sound files
writeWave(Phae.long2, "Phae.long2.wav")

#run cross correlation first
xcor<-xcorr(X = manualoc.df[1:5,], wl =300, frange= c(2, 9), ovlp=90, dens=0.8, wn='hanning',
cor.method = "pearson")

#plot pairwise scores
xcorr.graph(X = xcor, cex.cor = 2, cex.lab = 1, rel.cex = FALSE)

## End(Not run)
```

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