

# Introduction to AUXLAB

## version 1.49

AUXLAB is an integrated environment for sound generation, processing, visualization and audio playback, based on a programming language AUX (AUDIO syntaX). AUXLAB allows audio manipulations, plotting of audio and non-audio data, playback of audio data, implementation of algorithms, creation of user interfaces and interfacing with programs written in other programming languages such as C/C++ and MATLAB.

Although AUXLAB is primarily intended for audio processing, it offers versatile functionalities allowing the users to compute and handle non-audio data.

While the syntax of AUXLAB resembles that of MATLAB, there are many unique syntax features of AUXLAB and differences in syntax conventions.

This is a work in progress, more so for documentation. The following is the list of features of the software that are not yet included in this documentation.

- User defined functions, general (how-to info) info and the Debugger feature
- Custom user interface module--you can design your own window components (dialog box, buttons, etc), so you can create your own program to be used for an experimental procedure with functionalities of AUX playback and graphics
- Complete descriptions on properties of graphics objects---Similar to MATLAB, but with numerous differences to note
- "# Reserves"--special keywords assigned to frequently performed tasks or operations

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Last updated 10/2/2018

# License & Credit

AUXLAB is released under Academic Free License 3.0.

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Languages used: C++11 with yacc/lex. All codes written with Win32 API

Internal libraries:

- sigproc: syntax tokenizing, parsing; signal generation and processing
- graffy: visualization and screen processing
- xcom: console handing, variable display, history window, managing debugger and coordinating with sigproc
- wavplay: audio playback
- auxp: private user-defined functions
- auxcon: module for the custom user interface environment

Source codes will be available in GitHub soon. If you can't wait, let me know.

External libraries (open source) utilized:

- FFTW 3.3.4
- libsndfile 1.0.26, libsamplerate 0.1.8; Erik de Castro Lopo
- ELLF (2014-10-03 release); Stephen L Moshier
- Win32++ 7.3; David Nash
- Bison 2.4.1
- Flex 2.5.4a

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<http://auxlab.org>  
Last updated 9/25/2018

# **System Requirements**

- Windows 7, 8 and 10
- Minimum RAM: 128 MB
- Microsoft Visual C++ Redistributable for Visual Studio 2017

# Data types in AUX

The following data types are used in AUX/AUXLAB:

NUL	null/empty data
SCAL	scalar
TEXT	text string
VCT	vector; array
AUD	audio
CEL	cell array
CLAS	class; structure
TSEQ	<a href="#">Time Sequence</a>
HAUD	Handle to audio playback
HGO	Handle to graphic object

- Values can be either real or complex.
- Matrix is treated as a "grouped" VCT according to the row.
- The difference between VCT and AUD is that the latter has the information of 1) the sample rate, and 2) the time marker. In addition, one AUD object can have many chunks of audio in different times.
- Another difference: for some functions that do not allow negative values but could be useful in sound processing or computations are treated as an even-function for AUD (for example, the sqrt function), whereas for VCT, it is either an error or produce an imaginary value (i.e.,  $\sqrt{-1}$ )

# TSEQ: Time sequence

## Introduction

A data type TSEQ is an array container consisting of a time point and the data corresponding to the time point.

## Defining a TSEQ object

A TSEQ object has the following form:

```
[x] [y]
```

where x is the time point vector in milliseconds and y is the data vector. Here x and y must have the same length. In this form, at each time point the data is a scalar. In general, the data do not need to be a scalar; but can be in any form. To define such time sequence,

```
[t1 | y1; t2 | y2; ...]
```

where tn and yn are time marker and the corresponding value array in any length. (note: this is not implemented yet as of AUXLAB 1.47).

## Relative TSEQ

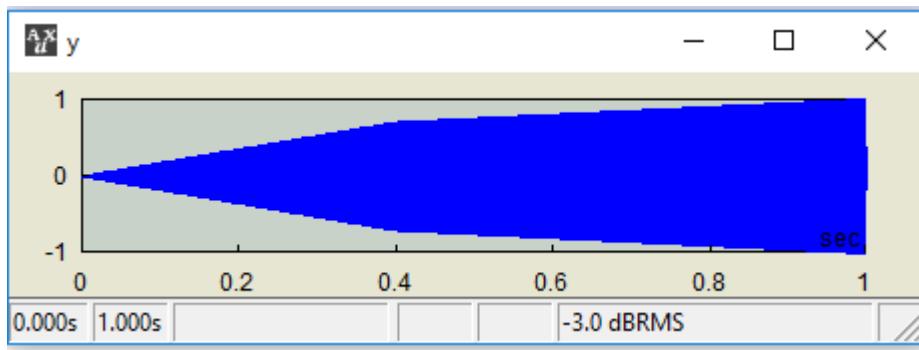
Sometimes it is very useful to have time values relative to another audio signal. In such cases, define a relative time sequence as follows:

```
[x;] [y]
```

## Example 1

The amplitude of a tone is scaled with a TSEQ, 0 at t=0, .7 at t=250ms, .3 at t=500ms, and 1 at t=1000ms. The multiplication operation with a TSEQ involves linear interpolation between specified time points.

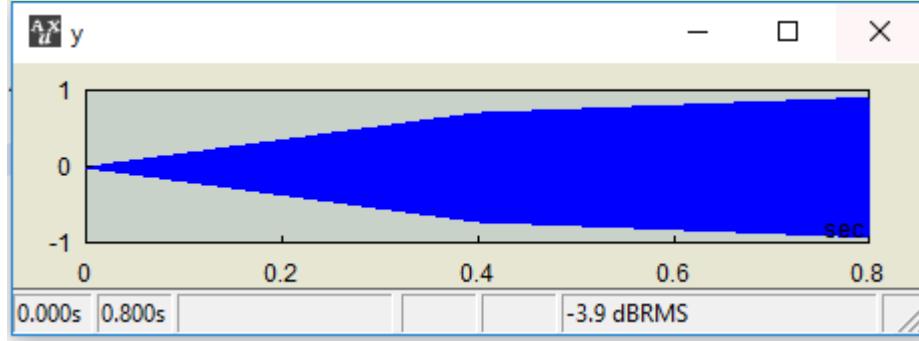
```
AUX> x = tone(500,1000);
AUX> ts = [0 250 500 1000] [0 .7 .3 1];
AUX> y = ts * x;
```



## Example 2

The same TSEQ as above but the audio signal with a different duration. Is this what you want?

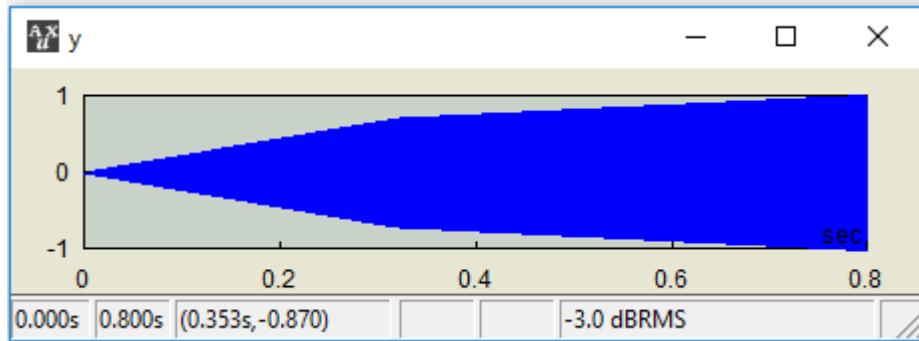
```
AUX> x = tone(500,800);
AUX> ts = [0 250 500 1000][0 .7 .3 1];
AUX> y = ts * x;
```



## Example 3

If you wanted to scale the audio with the same relative time course as Example 1, then go with a relative TSEQ.

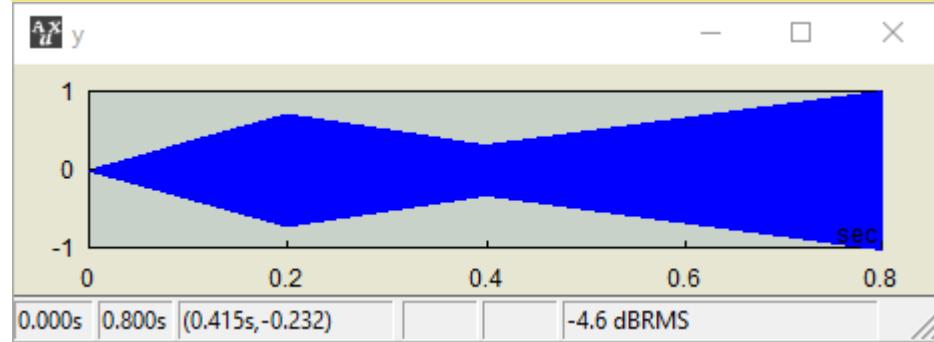
```
AUX> x = tone(500,800);
AUX> ts = [0 .25 .5 1;][0 .7 .3 1];
AUX> y = ts * x;
```



## Example 4

To adjust the amplitude of the audio signal with a desired time course in terms of dB, use the @ operator:

```
AUX> x = tone(500,800);  
AUX> ts = [0 .25 .5 1;] [-100 -3 -10 0];  
AUX> y = x @ ts;
```



# Operators in AUX

In addition to arithmetic operators (+ - \* / ^ %) and logical operators (&& || !), that you can find in most programming or scripting languages, AUX offers unique operators designed for audio signals, such as @ (amplitude scaling), >> (time-shift), ~ (time-compression/spectrum-expansion), -> (spectrum-shift), <> (change of duration), and # (change of pitch).

**+** -

```
z = x + y
z = x - y
```

Arithmetic Plus or Minus

### Commutative

Yes for + No for -

### Data Types

#### Allowed

SCAL, VCT, AUD, TEXT, TSEQ, NUL

#### Not allowed

CLASS, CELL

x      y      z

NUL    anythin anything

g

SCAL   anythin anything

g

VCT    VCT    VCT (see below)

VCT    AUD    AUD (see below)

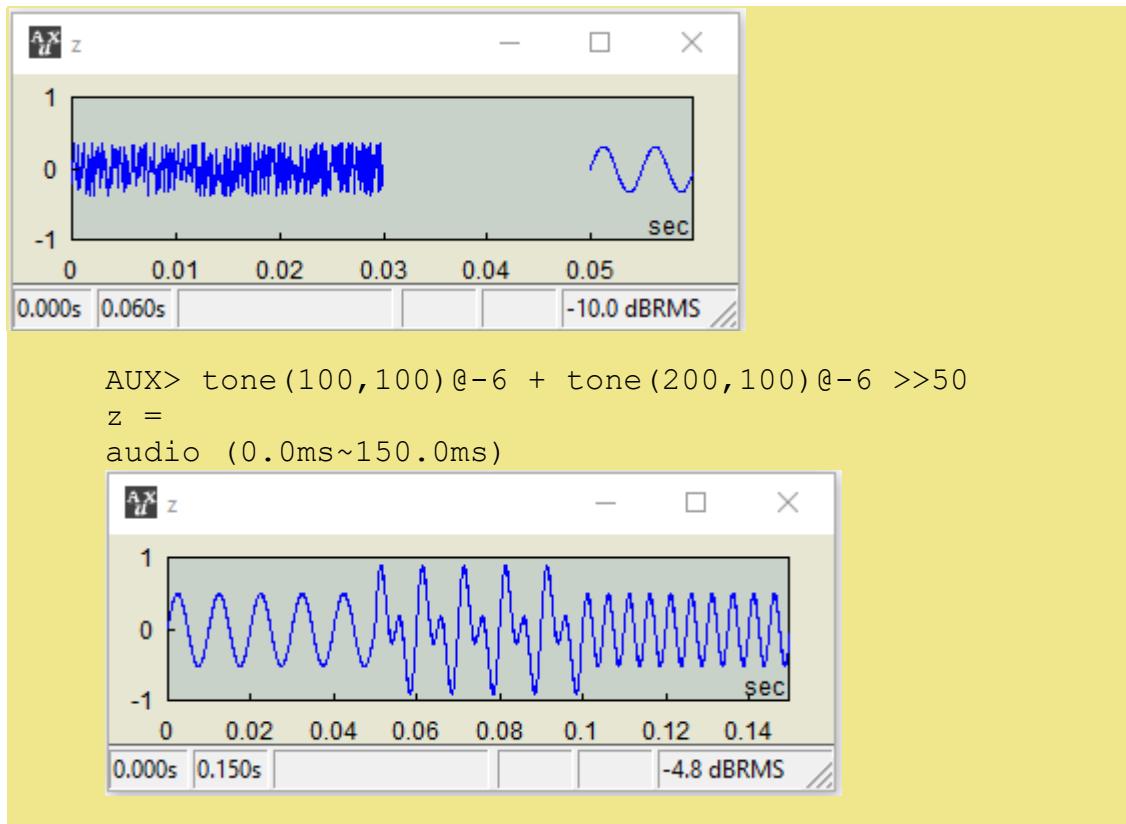
TSEQ   array   not allowed

### Notes

- In the case of SCAL + VCT or SCAL + AUD, the scalar is applied (added) to the entire array.
- In the case of VCT + VCT, if the lengths of operands are different, the operation takes place until the two values are available.
- In the case of VCT + AUD, if the lengths of operands are different, the operation takes place until the two values are available.
- In the case of AUD + AUD, the operation is time-based: i.e., if the signal is available at the particular time, the operation takes place.
- In the case of SCAL + TSEQ, the operation applies to the value of the TSEQ: i.e., the output TSEQ has the value added to SCAL at each time point.
- In the case of TSEQ + TSEQ, both must have the same number of time points (individual time points don't need to be the same).
- When both operands are grouped (i.e., matrix), both must have the equal number of groups(i.e., rows)

### Examples

```
AUX> noise(30)@-10 + tone(200,10)@-10 >>50
z =
audio (0.0ms~30.0ms) (50.0ms~60.0ms)
```



\* /

```
z = x * y
z = x / y
```

Arithmetic Multiplication or Division

### Commutative

Yes for \* No for /

### Data Types

#### Allowed

SCAL, VCT, AUD, TEXT, TSEQ, NUL

#### Not allowed

CLASS, CELL

x      y      z

NUL     anythin anything

g

SCAL    anythin anything

g

VCT    VCT    VCT (see below)

VCT    AUD    AUD (see below)

TSEQ    AUD    AUD (special meaning; see below)

TSEQ    VCT    TSEQ

### Notes

- In essence, this is "dot-multiplication," in the MATLAB language.
- In the case of SCAL \* VCT or SCAL \* AUD, the scalar is applied (multiplied) to the entire array.
- In the case of VCT \* VCT, if the lengths of operands are different, the operation takes place until the two values are available.
- In the case of VCT \* AUD, if the lengths of operands are different, the operation takes place until the two values are available.
- In the case of AUD \* AUD, the operation is time-based: i.e., if the signal is available at the particular time, the operation takes place.
- In the case of SCAL \* TSEQ, the operation applies to the value of the TSEQ: i.e., the output TSEQ has the value multiplied by SCAL at each time point.
- In the case of TSEQ \* TSEQ, both must have the same number of time points (individual time points don't need to be the same).
- TSEQ \* AUD multiplies each value of the audio signal with a linear interpolated version of TSEQ (see example below).
- When both operands are grouped (i.e., matrix), both must have the equal number of groups(i.e., rows)
- As of 9/20/2018, AUXLAB does not support matrix multiplication (in the mathematical sense). I will add the feature in future releases (it is not difficult to do it, anyway; we need to choose the operator symbol, though), if there are enough requests from users.



**++**

**x ++ y**

Append (Concatenate) y at the end of x

**Commutative**

No

**Data Types**

x

AUD

y

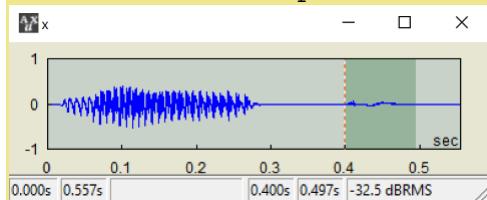
AUD

**Notes**

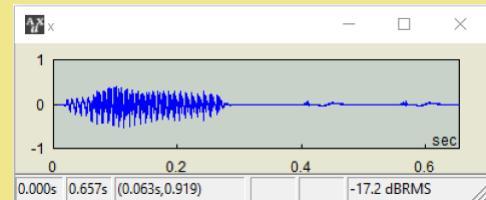
**Examples**

Open a .wav file "mop" and repeat the /p/ portion at the end.

AUX> x=wave ("mop");



AUX> x=x++x (400~500);



•

```
a = x : y
```

Make an array  $a$  beginning from  $x$  to  $y$  with step of 1 or -1

```
a = x : y : z
```

Make an array  $a$  beginning from  $x$  to  $z$  with step of  $y$

### Commutative

No

### Data Types

$x \quad y \quad z$   
SCAL SCAL SCAL

### Notes

- In  $x:y$ , the step size is either 1, if  $x < y$ , or -1, if  $x > y$ . Note that this is different from the MATLAB convention.

### Examples

```
AUX> 1:5  
ans = 1 2 3 4 5
```

```
AUX> 5:-1:  
ans = 5 4 3 2 1
```

```
AUX> 1:.5:3  
ans = 1 1.5 2 2.5 3
```

# **:** (indexing)

```
a = x(:)
```

"Serialize" the matrix  $x$  and turn to a vector.

## Commutative

N/A

## Data Types

$x$	$z$
VCT	VCT

## Notes

- If  $x:$  is a vector, it won't have any effect.

## Examples

```
AUX> x=(1:6).matrix(2)
ans =
1 2 3
4 5 6
AUX> x(:)
ans = 1 2 3 4 5
```

# ~ (time-indexing)

**x(t1 ~ t2)**

Portion of audio x from t1 ms to t2 ms

## Commutative

N/A

## Data Types

x	t1	t2
AUD	SCAL	SCAL

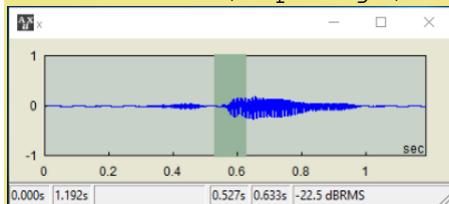
## Notes

- If  $t_1 > t_2$ , the extracted signal is time-reversed.

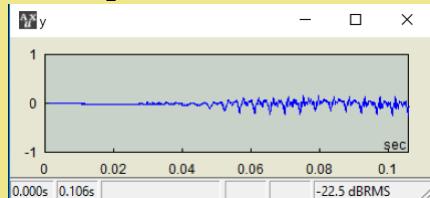
## Examples

Open a .wav file "spring" and extract from 527 ms to 633 ms.

AUX> `x=wave("spring");`

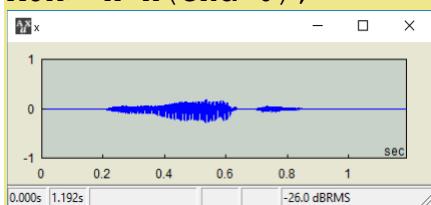


AUX> `y=x(527~633);`



To time-reverse the sound,

AUX> `x=x(end~0);`



• •

```
x = any expression with ...
x(id) = any expression with ...
x(id1:id2) = any expression with ...
x(id1:id2,jd1:jd2) = any expression with ...
x(t1~t2) = any expression with ...
```

Replicator; Replicate what's in the left hand side to the right hand side

### Commutative

N/A

### Data Types

This applies to multiple cases of expressions, not necessarily tied to one data type.

### Notes

- Whatever is on the left hand side is being replicated to the right hand side with ... Therefore,  
`very_long_variable_name = any_function(..)`  
is equivalent to:  
`very_long_variable_name =  
any_function(very_long_variable_name)`
- The dot(.) notation also applies; therefore, the following is also an equivalent statement:  
`very_long_variable_name = ...any_function`  
as long as `any_function` supports the object function style.
- Conceptually, you might understand this as a sort of an advanced compound assignment operator:  
`x = x + 50` can be written as either `x += 50` or `x = .. + 50`  
On the other hand, `x = sqrt(x)` can't be written with a regular compound assignment operator, but .. works perfectly: `x = sqrt(..)`
- This feature will save you from tedious typings or copy-pastings across left and right hand sides.

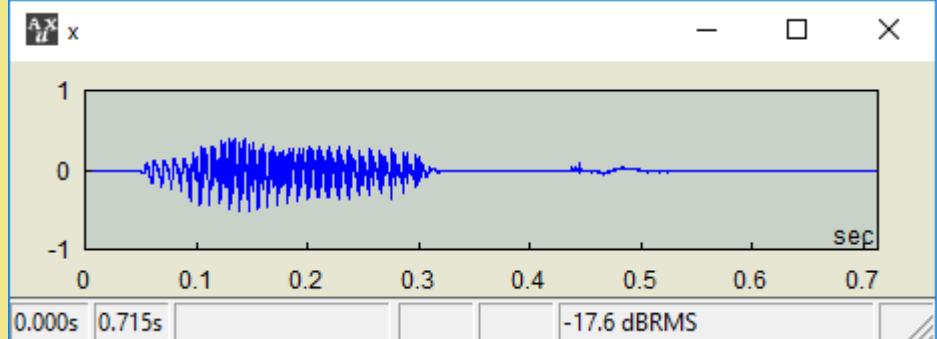
### Examples

- You have a vector with a long variable name, if you append a value to it. You may write it in the vector form or use ++  
`long_var_name = [long_var_name more_values]`  
`long_var_name = long_var_name ++ more_values` or  
`long_var_name += more_values`

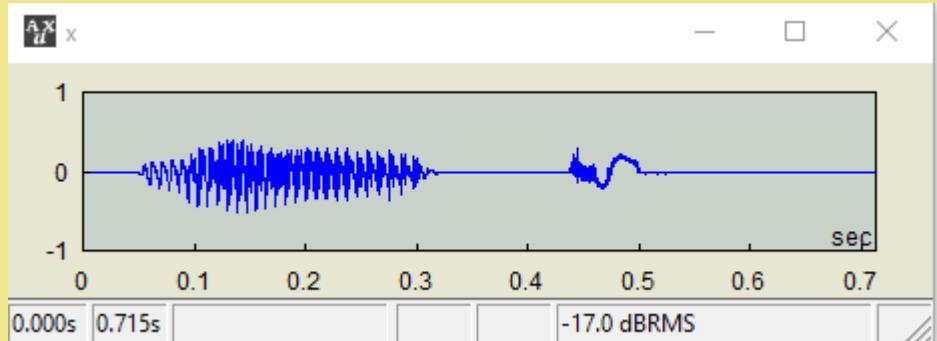
But if values are inserted before it, you can't use `++=`. Then, instead of typing `long_var_name` again on the RHS, just

```
long_var_name = [values ..] or  
long_var_name = values ++ .. will do it.
```

- You want to modify a part of a signal: for example,  
AUX> `x=wave ("mop") ;`



To distort the signal from 400 to 500 ms by applying sqrt, do this:  
AUX> `x(400~500) = ...sqrt;`



- This is a more efficient way of coding. Internally, the expression `x(400~500) = x(400~500).sqrt` evaluates `x(400~500)` twice for the LHS and RHS, but the use of `..` bypasses one.

!

```
a = x'
```

Transpose; Swap the row and column of a matrix  $x$ .

**Commutative**

N/A

**Data Types**

$x$	$z$
VCT	VCT

**Notes**

**Examples**

```
AUX> x=(1:6).matrix(2)
ans =
1 2 3
4 5 6
AUX> x'
ans =
1 4
2 5
3 6
```

$\% \circ$

**z = x % y**

Remainder after division (modulo operation)

**Commutative**

No

**Data Types**

x        y        z  
AUD,     SCAL    AUD, VCT, SCAL  
VCT,  
SCAL

**Notes**

- This is equivalent to  $z=\text{mod}(x, y)$  or  $z=x.\text{mod}(y)$
- See [mod](#) for examples.

**^**

**z = x ^ y**

z is x raised to the power y

**Commutative**

No

**Data Types**

x        y        z  
AUD,     SCAL    AUD, VCT, SCAL  
VCT,  
SCAL

**Notes**

- 
- See \_\_\_ for examples.

@

**z = x @ y**

"At" operator; Amplitude scaling in terms of dB

### Commutative

No

### Data Types

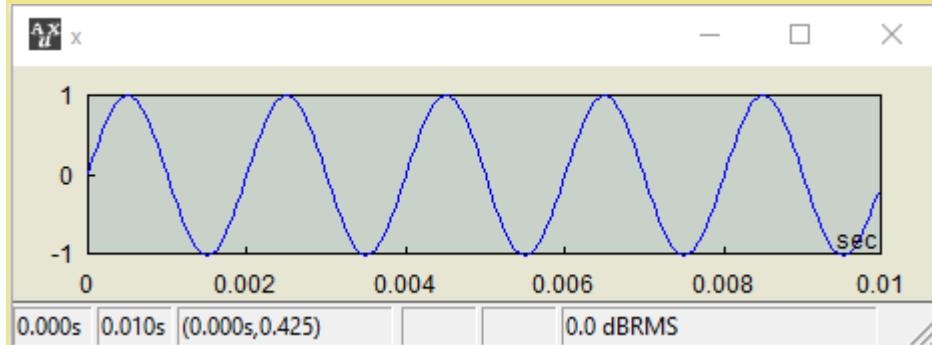
x	y	z
AUD	SCAL	AUD
AUD	TSEQ	AUD

### Notes

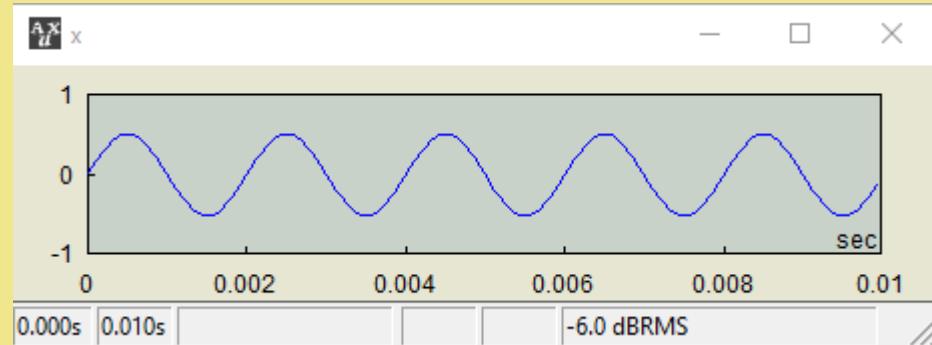
- The rms level of  $z$  is to be  $y$  dB, if  $y$  is a scalar.
- 0 dB is defined as the RMS level of a full-scale sinusoid. This means that the RMS of a full-scale square wave is to be 3 dB.

### Examples

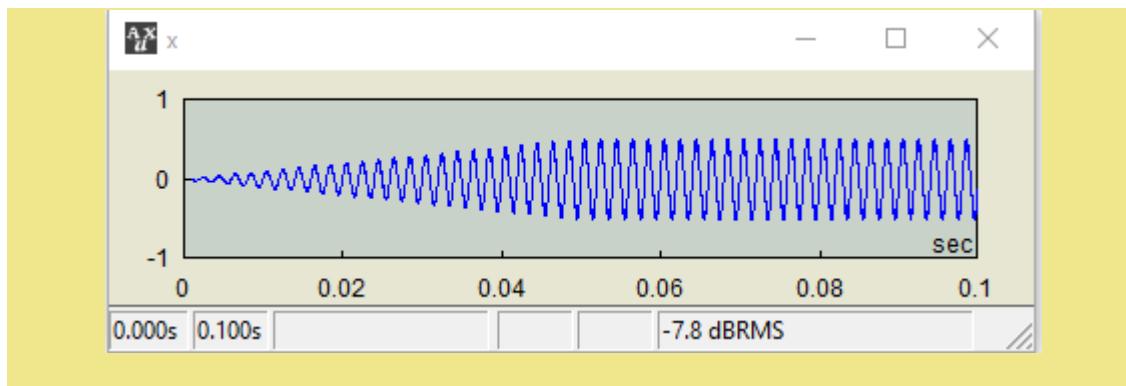
AUX> `x=tone(500,10)`



AUX> `x=tone(500,10) @-6`



AUX> `x=tone(500,100) @ [0 .5 1;] [-100 -6 -6]`



>>

```
z = x >> y
```

Time-shift: Shift the audio signal  $x$  by  $y$  milliseconds

### Commutative

No

### Data Types

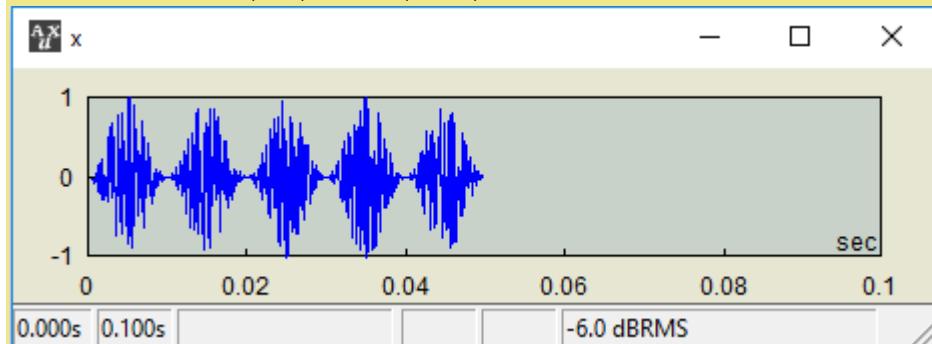
$x$	$y$	$z$
AUD	SCAL	AUD

### Notes

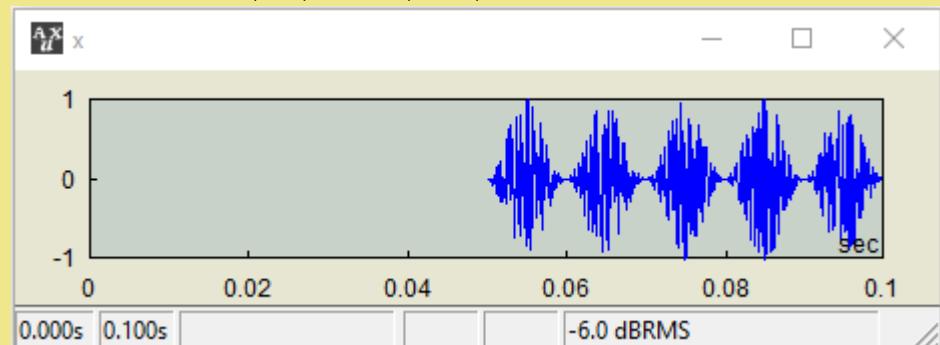
- If  $x$  is present in the interval  $(t_1, t_2)$ ,  $z$  will be present in the interval  $(t_1+y, t_2+y)$

### Examples

```
AUX> x=noise(50).sam(100)@-6
```



```
AUX> x=noise(50).sam(100)@-6 >> 50
```



[ ; ]

**z = [x ; y]**

Make a stereo signal  $z$  from  $x$  and  $y$

### Commutative

No

### Data Types

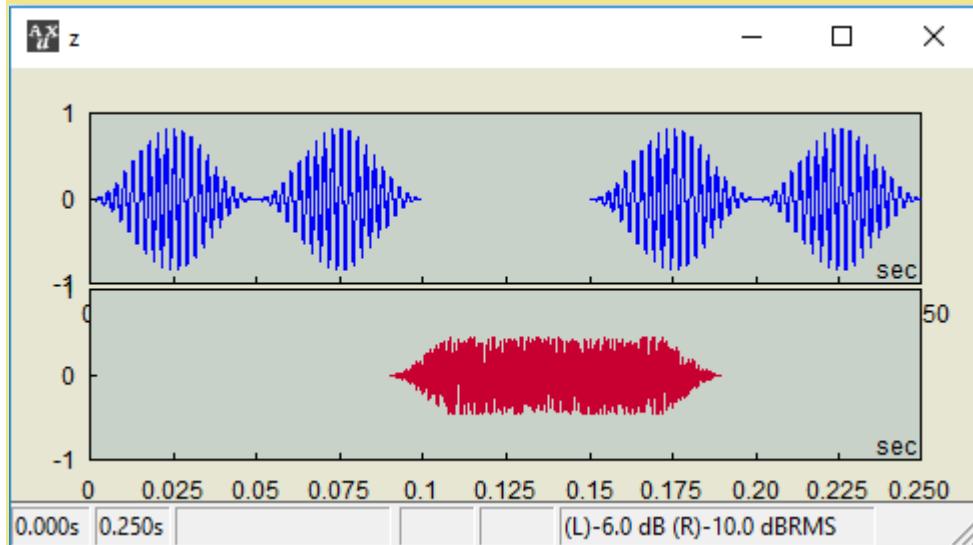
$x \quad y \quad z$   
AUD AUD AUD

### Notes

- $x$  and  $y$  do not need to be the same length or even in the same time period.

### Examples

```
AUX> x=tone(400,100).sam(20)@-6;
AUX> x = x+x>>150;
AUX> y=noise(100).ramp(50)@-10>>90;
AUX> z=[x;y]
z =
audio(L) (0.0ms~100ms) (150.0ms~250.0ms)
audio(R) (90.0ms~190ms)
```



~

**z = x ~ y**

Time-Frequency Scaling Operation; i.e., Time-Compress (or Frequency-Expand at the same time) an audio signal  $x$  by a factor of  $y$

**Commutative**

No

**Data Types**

x	y	z
AUD	SCAL	AUD

**Notes**

- This simulates a playback of audio samples with a different rate of its original sample rate.

**Examples**

->

**$z = x \rightarrow y$**

Frequency-Shift Operator; Shift the spectrum of an audio signal  $x$  by  $y$  Hz

#### Commutative

No

#### Data Types

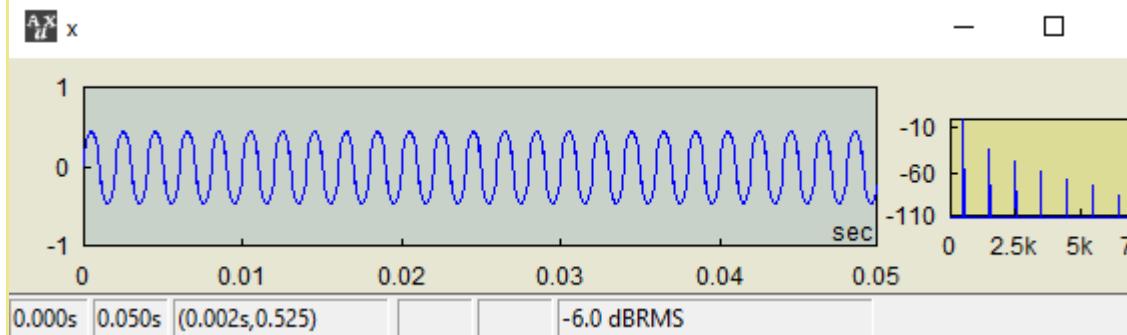
$x$	$y$	$z$
AUD	SCAL	AUD

#### Notes

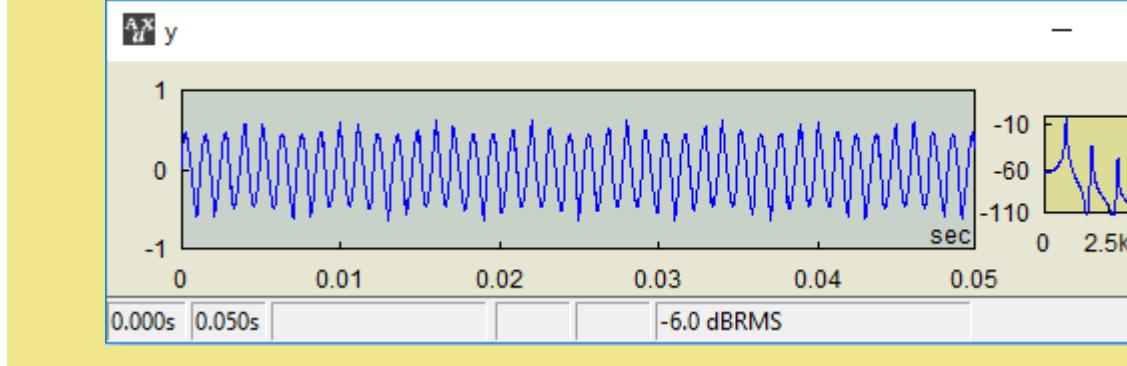
- This is equivalent to  $z=\text{movespec}(x, y)$  or  $z=x.\text{movespec}(y)$
- This does not change the pitch of a harmonic tone  $x$  except for a sinesoid. Frequencies of the harmonics are shifted equally by  $y$  Hz, so it won't sound like a natural harmonic tone.

#### Examples

AUX>  $x=\text{tone}(500, 50).\text{sqrt}@-6$



AUX>  $y=x\rightarrow 330$



# Compound Assignment Operators in AUX

AUX supports compound assignment operators, also known as augmented operators. The following examples are self-explanatory, as their usage is the same in other programming languages such as C and Java:

`+ = - = * = / =`

The following examples are unique in AUX:

`++ = @ = @@ = >> = ~ = -> = <> = # =`

**$+=$   $-=$   $*=$   $/=$**

```
x += y  
x -= y  
x *= y  
x /= y
```

Compound operators for arithmetic operations, + - \* and /

#### *Equivalent expressions*

$x = x + y$   
 $x = x - y$   
 $x = x * y$   
 $x = x / y$

#### **Commutative**

No

#### **Data Types**

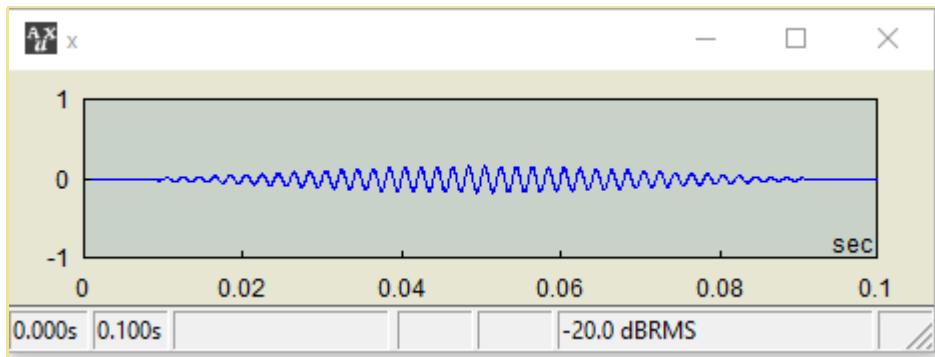
<b>Allowed</b>		<b>Not allowed</b>
SCAL, VCT, AUD, TEXT, TSEQ, NUL		CLASS, CELL
$x$	$y$	$x$ becomes
NUL	$T^*$	$T$
SCAL	$T$	$T$
VCT	VCT	VCT
AUD	VCT	AUD
VCT	AUD	AUD
TSEQ	VCT	TSEQ
TSEQ	AUD	not allowed
AUD	TSEQ	AUD

\*  $T$  means an allowed type.

#### **Notes**

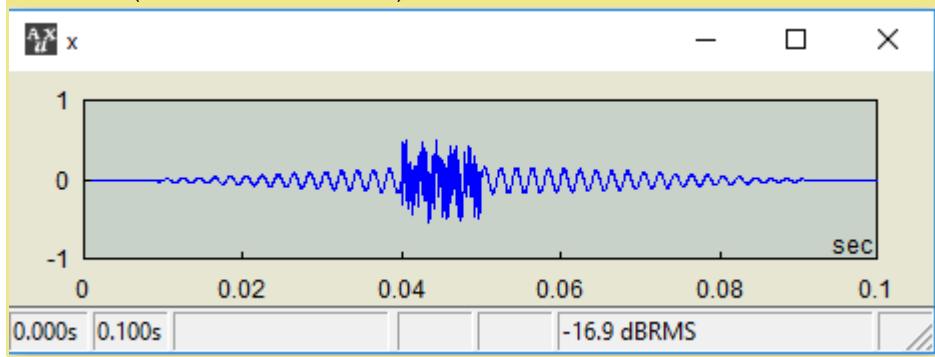
#### **Examples**

```
AUX> x=rand(5)  
x =  
0.5853 0.2218 0.6290 0.1231 0.2736  
AUX> x*=10  
x =  
5.8531 2.2182 6.2896 1.2313 2.7363  
AUX> x=tone(500,100).hann@-20  
x =  
audio (0.0ms~100.0ms)
```



```
AUX> y=noise(10)@-10>>40
```

```
y =  
audio (40.0ms~50.0ms)  
AUX> x += y  
x =  
audio (0.0ms~100.0ms)
```



# **++=**

**x += y**

Compound operator for the append operation ++

*Equivalent expression*

$x = x ++ y$

**Commutative**

No

**Data Types**

<b>x</b>	<b>y</b>
AUD	AUD

**Notes**

**Examples**

## $\text{@}=$

**x @= y**

Compound operator for the level operator @

*Equivalent expression*

$x = x @ y$

**Commutative**

No

**Data Types**

<b>x</b>	<b>y</b>
AUD	SCAL

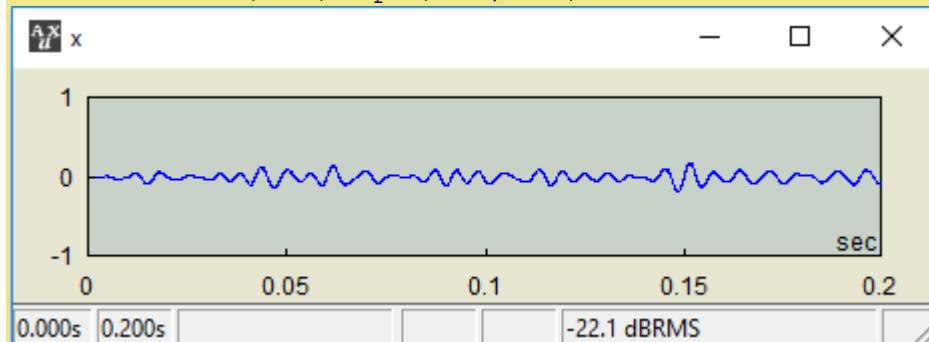
**Notes**

This is used to set the level at the target level y (with an absolute level)

**Examples**

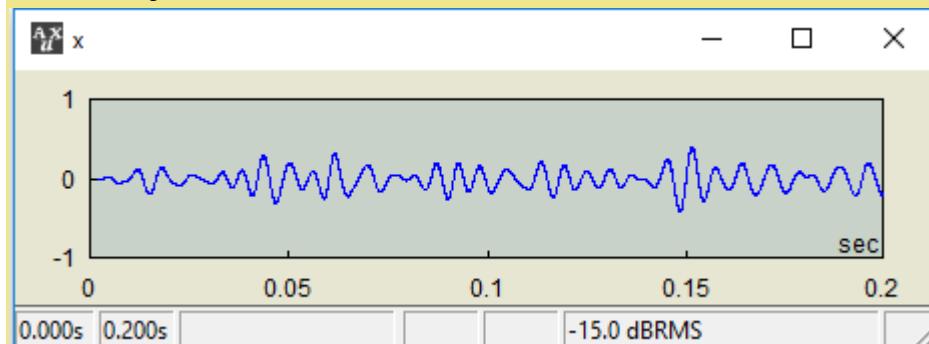
Bandpass filter the white noise x

```
AUX> x=noise(100).bpf(100,200);
```



The RMS level of x, -22.1 dB, as seen in the plot, is simply a byproduct of filtering. To set it specifically at -15 dB, do the following:

```
AUX> x @= -15
```



## $\text{@} @=$

**x @= y**

Compound operator for the level operator @@

Also known as relative level operator

**Equivalent expression**

$x = x @ x @ y$

**Commutative**

No

**Data Types**

x

AUD

y

SCAL

**Notes**

$x @ x @ y$  literally means the signal x with the level at y dB re. x

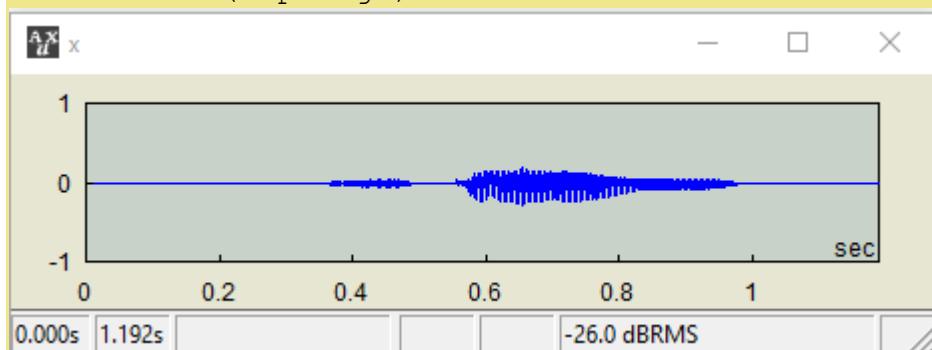
In other words, the signal x with the level adjusted by y dB from its own rms.

Therefore, this is also known as an "incremental" level operator.

**Examples**

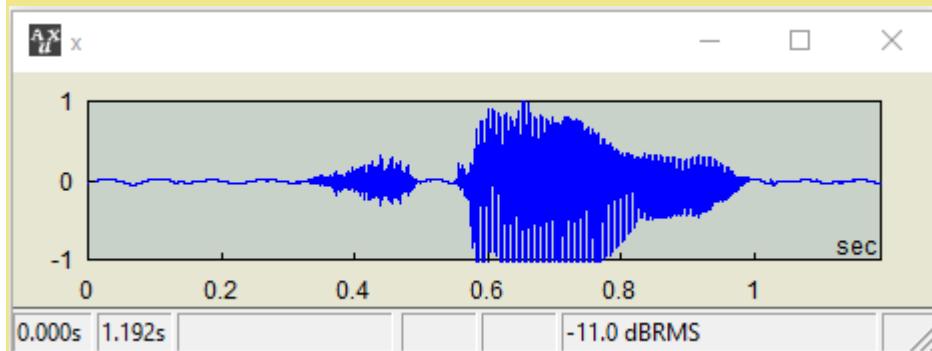
Open a .wav file. The RMS level of this audio happens to be -26 dB.

AUX> x=wave("spring")



I want to increase its RMS level by 15 dB.

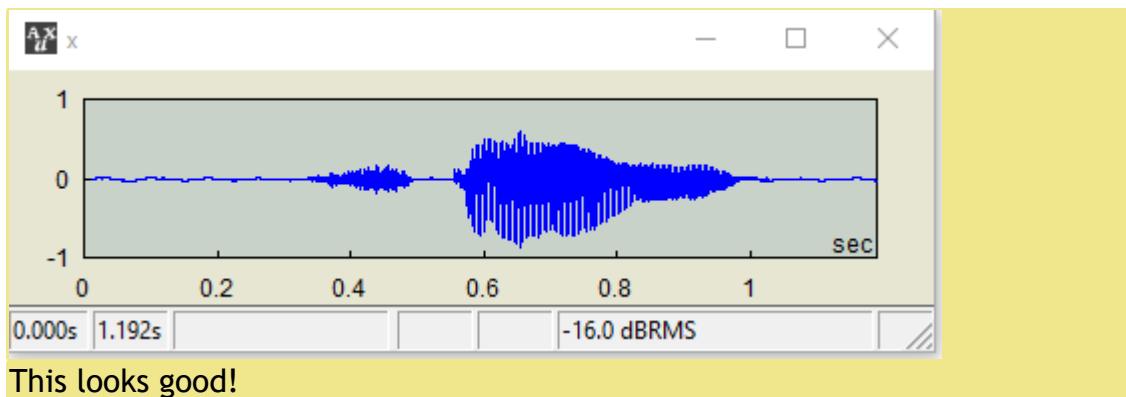
AUX> x @@= 15



Oops, we have a peak-clipping situation. A 15-dB increase was too much.

Let's go down by 5 dB.

AUX> x @@= -5



>>=

**x >= y**

Compound operator for the time-shift operator >>

## *Equivalent expression*

**x = x >> v**

### Commutative

No

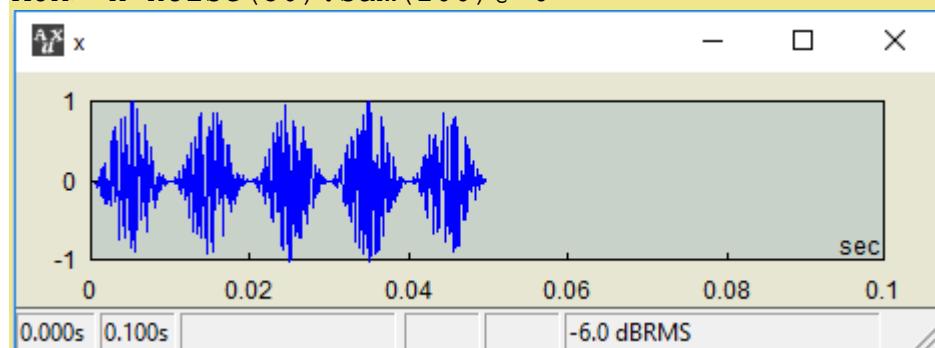
## Data Types

x AUD y SCAL

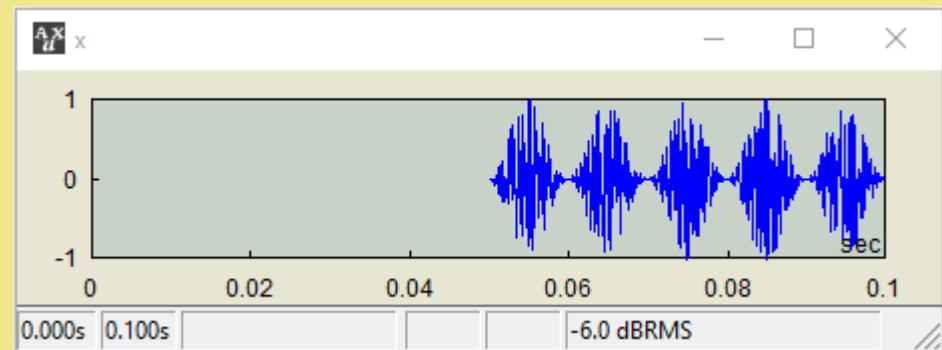
Notes

## Examples

AUX> x=noise(50),sam(100)@-6



AUX> x >>= 50



## $\sim=$

**x ~= y**

Compound operator for the time-frequency scale operator  $\sim$

*Equivalent expression*

$x = x \sim y$

**Commutative**

No

**Data Types**

$x$	$y$
AUD	SCAL

**Notes**

**Examples**

## **->=**

**x ->= y**

Compound operator for the frequency-shift operator ->

*Equivalent expression*

$x = x -> y$

**Commutative**

No

**Data Types**

X	y
AUD	SCAL

**Notes**

**Examples**

# Functions that generate a signal or vector

Here are the functions that generate a signal or vector with a given list of parameters. Notable example include [tone](#) (to create a pure tone), [noise](#) (to create white noise), [wave](#) (to read .wav file), and [rand](#) (to create an array of random numbers).

# cell

```
y = cell ( n )
y = n.cell ( )
```

Create a cell array with specified blank elements

arg	type	Description	Unit or Value scale
n	SCL	Number of blank elements	

## Outputs

- **y** is a cell array with blank elements.

## Notes

## Examples

## See Also

# dc

```
y = dc ( duration )
```

Generate a dc signal (constant amplitude at the full scale)

arg	type	Description	Unit or Value scale
duration	SCAL	duration of the signal to generate	
<b>Outputs</b>			
• <b>y</b> is an audio signal (not really) of all values of one with the duration <b>duration</b> milliseconds.			

## Notes

- Although this is not technically "sound", this is treated as an audio signal.

## Examples

## See Also

[ones](#) | [silence](#) | [zeros](#)

# fm

```
y = fm ( freq1,freq2,mod_rate,duration,phase[=0] )
```

Generate a frequency-modulated tone signal

arg	type	Description	Unit or Value scale
<b>freq1</b>	SCAL	FM swing from	
<b>freq2</b>	SCAL	FM swing to	
<b>mod_rate</b>	SCAL	how many times the frequency will swing	
<b>duration</b>	SCAL	duration of the signal to generate	
<b>phase</b>	SCAL	The initial modulation phase between 0 and 1	

## Outputs

- **y** is a frequency-modulated tone.

## Notes

## Examples

## See Also

# gnoise

```
y = gnoise ( duration )
```

Generate a white noise signal with a Gaussian distribution

arg	type	Description	Unit or Value scale
-----	------	-------------	---------------------

**duration** SCAL duration of the signal to generate

## Outputs

- **y** is Gaussian white noise (random numbers from a Gaussian distribution).

## Notes

- Unlike noise, the amplitude of gnoise is not bounded between -1 and 1. → Scale down the output appropriately to avoid clipping at  $\pm 1$ .

## Examples

## See Also

[noise](#)

# input

```
y = input( str )
y = str.input( )
```

Prompt a user response.

arg	type	Description	Unit or Value scale
str	STR	message to display	

## Outputs

- `y` is the string input from the user.

## Notes

- This does not return until the user presses the Return/Enter key.

## Examples

## See Also

[inputdlg](#)

# irand

```
y = irand ( n )
y = n.irand ( )
```

Create an integer random number, uniformly distributed in the interval [1, n]

arg	type	Description	Unit or Value scale
n	SCAL	the upper limit of the range	

## Outputs

- y is an integer random number between 1 and n

## Notes

## Examples

## See Also

[rand](#) | [randperm](#)

# noise

```
y = noise ( duration )
```

Generate a white noise signal

arg	type	Description	Unit or Value scale
duration	SCAL	duration of the signal to generate	

## Outputs

- `y` is uniform white noise (random numbers from a uniform distribution) with amplitude bound between -1 and 1.

## Notes

- The random signal is generated from a uniform distribution.

## Examples

## See Also

[gnoise](#)

# ones

```
y = ones ( n )
y = n.ones ( )
```

Create a non-audio array of all ones

arg	type	Description	Unit or Value scale
n	SCAL	Size of array to create	

## Outputs

- y is an array of value one with size n

## Notes

## Examples

## See Also

[zeros](#)

# rand

```
y = rand ( sz )
y = sz.rand ( )
```

Create an array of random numbers, uniformly distributed between 0 and 1, with the size of **sz**

arg	type	Description	Unit or Value
			scale

**sz** SCAL the size of the array to create

## Outputs

- **y** is an array of random numbers

## Notes

## Examples

## See Also

[irand](#) | [randperm](#)

# randperm

```
y = randperm ( n )
y = n.randperm ( )
```

Random permutation of the integers from 1 to **n**

arg	type	Description	Unit or Value scale
<b>n</b>	SCAL	the upper limit of the range, or the size of the array to create	

## Outputs

- **y** is an array of integers between 1 and **n** in a random order

## Notes

## Examples

## See Also

[irand](#) | [rand](#)

# silence

```
y = silence ( duration )
```

Generate a silence signal

arg	type	Description	Unit or Value scale
duration	SCAL	duration of the signal to generate	

## Outputs

- `y` is AUDIO with zero values.

## Notes

- zeros generates a non-audio vector; silence generates an audio signal

## Examples

## See Also

[dc](#) | [zeros](#)

# sprintf

```
y = sprintf ( format,... )
```

Generate a formatted text

arg	type	Description	Unit or Value scale
<b>format</b>	TXT	String that contains the text to be written to the file, following the same convention as in C language.	
...	...	n/a	

## Outputs

- **y** is TXT containing the formatted text.

## Notes

## Examples

## See Also

[fprintf](#)

# str2num

```
y = str2num ( str )
y = str.str2num ( )
```

Read a string and convert it to a numerical array.

arg	type	Description	Unit or Value scale
<b>str</b>	STR	String (it must not have a non-numerical character)	

## Outputs

- **y** is an array from the string **str**.

## Notes

- If there's a non-numerical character, an empty array will be returned.

## Examples

## See Also

[eval](#) | [sprintf](#)

# tone

```
y = tone ( freq,duration,phase[=0] )
```

Generate a pure tone

arg	type	Description	Unit or Value scale
<b>freq</b>	SCAL	frequency or VCT	
<b>duration</b>	SCAL	duration	
<b>phase</b>	SCAL	initial phase (0 to 1)	

## Outputs

- **y** is a pure tone.

## Notes

- If **freq** is a two-element vector, a tone glide is generated with beginning and ending frequencies, as specified.

## Examples

- `tone(100, 50, .25) // 100-Hz, 50-ms tone with a starting phase of 90°`

## See Also

# wave

```
y = wave ( filename )
```

Open a .wav file

arg	type	Description	Unit or Value scale
-----	------	-------------	---------------------

**filename** STR the name of the file to open.

## Outputs

- **y** is an audio signal retrieved from the file **filename**.

## Notes

- If the extension is not specified, .wav is used.
- If path is not included, the current folder (the same directory as AUXLAB)

## Examples

- `wave ("c:\soundData\specialnoise")`

## See Also

[wavwrite](#)

# **zeros**

```
y = zeros ( n )
y = n.zeros ( )
```

Create a non-audio array of all zeros

<b>arg</b>	<b>type</b>	<b>Description</b>	<b>Unit or Value scale</b>
<b>n</b>	SCAL	Size of array to create	

## **Outputs**

- **y** is a non-audio array with **n** zeros

## **Notes**

- zeros generates a non-audio vector; silence generates an audio signal

## **Examples**

## **See Also**

[ones](#)

# **Functions that produce an signal or vector based on an existing signal or vector**

These are the functions that produce a modified version of an existing signal or vector with a given list of parameters. They include functions for filtering ([filt](#), [filtfilt](#), [lpf](#), [hpf](#), [bpf](#), [bsf](#)), windowing ([ramp](#), [blackman](#), [hann](#), [hamming](#)), or altering features of audio signals, such as time-scaling [tscale](#), frequency-scaling [fscale](#) or spectrum-shifting [movespec](#).

# audio

```
y = audio( x )
y = x.audio( )
```

Convert a non-audio vector to an audio signal

arg	type	Description	Unit or Value scale
x	VCT	non-audio vector	

## Outputs

- y is a converted audio signal of x.

## Notes

- This is used when hacking aux functions requiring audio signal arguments.
- In AUX, all audio signals should be bound between -1 and 1. If the maximum of x exceeds the boundary, clipping will occur.

## Examples

## See Also

[vector](#)

# blackman

```
y = blackman ( x,alpha )
y = x.blackman ( alpha )
```

Apply a blackman window

arg	type	Description	Unit or Value scale
x	AUDIO	Signal to apply the windowing to or VCT	
alpha	SCAL	$\alpha$	

## Outputs

- $y$  is  $x$  with a blackman window applied.

## Notes

## Examples

- `blackman(ones(128),.4)` or `ones(128).blackman(.16)`: 128-pt blackman window with  $\alpha=.16$
- `blackman(dc(100))` or `dc(100).blackman(.16)`: 100-ms blackman window  $\alpha=.16$
- `blackman(tone(500,100))` or `tone(500,100).blackman(.16)`: blackman window  $\alpha=.16$  applied to a 500-Hz, 100-ms tone

## See Also

[blackman](#) | [hamming](#)

# bpf

```
y = bpf (   
x,fcut1,fcut2,order[=8],type[=1],dBpass[=.5],dBstop[=-40]   
)  
y = x.bpf (   
fcut1,fcut2,order[=8],type[=1],dBpass[=.5],dBstop[=-40] )
```

IIR, band-pass filtering; Apply a band-pass filter to the audio signal

arg	type	Description	Unit or Value scale
<b>x</b>	AUDIO	audio signal	
<b>fcut1</b>	SCAL	cut-off frequency1	
<b>fcut2</b>	SCAL	cut-off frequency2	
<b>order</b>	SCAL	order of the IIR filter	
<b>type</b>	SCAL	IIR filter type (1: Butterworth, 2: Chebyshev, 3: Elliptic)	
<b>dBpass</b>	SCAL	Passband ripple allowed	
<b>dBstop</b>	SCAL	Stopband attenuation	

## Outputs

- **y** is a bandpass-filtered version of **x**.

## Notes

- The output is not normalized; i.e., the rms of **x** is adjusted according to the filter gain.
- **fcut1** and **fcut2** should be less than the Nyquist frequency.
- IIR filter coefficients are designed by the specification **requested** in the argument list, **but not guaranteed**. The user is responsible for making sure the output follows the spec.

## Examples

- `bpf(x, 2000, 4000)`

## See Also

[hpf](#) | [bpf](#) | [bsf](#)

## Algorithm

[ELLF Digital Filter Calculator](#)

# bsf

```
y = bsf (   
x,fcut1,fcut2,order[=8],type[=1],dBpass[=.5],dBstop[=-40]  
)  
y = x.bsf (   
fcut1,fcut2,order[=8],type[=1],dBpass[=.5],dBstop[=-40] )
```

IIR, band-stop filtering; Apply a band-stop filter to the audio signal

arg	type	Description	Unit or Value scale
<b>x</b>	AUDIO	audio signal	
<b>fcut1</b>	SCAL	cut-off frequency1	
<b>fcut2</b>	SCAL	cut-off frequency2	
<b>order</b>	SCAL	order of the IIR filter	
<b>type</b>	SCAL	IIR filter type (1: Butterworth, 2: Chebyshev, 3: Elliptic)	
<b>dBpass</b>	SCAL	Passband ripple allowed	
<b>dBstop</b>	SCAL	Stopband attenuation	

## Outputs

- **y** is a bandstop-filtered version of **x**.

## Notes

- The output is not normalized; i.e., the rms of **x** is adjusted according to the filter gain.
- **fcut1** and **fcut2** should be less than the Nyquist frequency.
- IIR filter coefficients are designed by the specification **requested** in the argument list, **but not guaranteed**. The user is responsible for making sure the output follows the spec.

## Examples

- `bsf(x, 2000, 4000)`

## See Also

[lpf](#) | [hpf](#) | [bpf](#)

## Algorithm

[ELLF Digital Filter Calculator](#)

# filt

```
y = filt ( x,num,den )
y = x.filt ( num,den )
```

1-D digital filter

arg	type	Description	Unit or Value scale
x	AUD or Input data VECT		
num	VECT	Numerator coefficients of rational transfer function	
den	VECT	Denominator coefficients of rational transfer function	

## Outputs

- y is filtered data.

## Notes

- For a grouped or "2-D" array, filtering takes place for each group.

## Examples

## See Also

[filtfilt](#)

# **filtfilt**

```
y = filtfilt ( x,num,den )
y = x.filtfilt ( num,den )
```

Zero-phase digital filtering

<b>arg</b>	<b>type</b>	<b>Description</b>	<b>Unit or Value scale</b>
<b>x</b>	AUD or Input data VECT		
<b>num</b>	VECT	Numerator coefficients of rational transfer function	
<b>den</b>	VECT	Denominator coefficients of rational transfer function	

## Outputs

- **y** is filtered data.

## Notes

- For a grouped or "2-D" array, filtering takes place for each group.

## Examples

## See Also

[filt](#)

# fscale

```
y = fscale ( x,freq )
y = x.fscale ( freq )
y = x # freq
```

Increase/decrease the pitch of the signal without changing the duration

arg	type	Description	Unit or Value scale
x	AUD	audio signal	
freq	VCT	The desired pitch change in semitones	

## Outputs

- y is audio with the same duration as x

## Notes

## Examples

## See Also

[tscale](#) | [movespec](#)

## Algorithm

First the signal time is rescaled with the phase vocoder(Flanagan & Golden, "Phase Vocoder" 1966); then the signal is resampled to equalize the duration.

# hamming

```
y = hamming( x )
y = x.hamming( )
```

Apply a hamming window

arg	type	Description	Unit or Value scale
x	AUDIO	Signal to apply the windowing to or VCT	

## Outputs

- **y** is **x** with a hamming window applied.

## Notes

## Examples

- `hamming(ones(128))` or `ones(128).hann`: 128-pt hamming window
- `hamming(dc(100))` or `dc(100).hamming`: 100-ms hamming window
- `hamming(tone(500,100))` or `tone(500,100).hamming`: hamming window applied to a 500-Hz, 100-ms tone

## See Also

[hann](#) | [blackman](#)

# **hann**

```
y = hann ( x )
y = x.hann ( )
```

Apply a hann (hanning) window

<b>arg</b>	<b>type</b>	<b>Description</b>	<b>Unit or Value scale</b>
<b>x</b>	AUDIO	Signal to apply the windowing to or VCT	

## Outputs

- **y** is **x** with a hanning window applied.

## Notes

## Examples

- `hann(ones(128))` or `ones(128).hann`: 128-pt hann window
- `hann(dc(100))` or `dc(100).hann`: 100-ms hann window
- `hann(tone(500,100))` or `tone(500,100).hann`: hann window applied to a 500-Hz, 100-ms tone

## See Also

[hamming](#) | [blackman](#)

# hpfilter

```
y = hpfilter ( x,fcut,order[=8],type[=1],dBpass[=.5],dBstop[=-40] )
y = x.hpfilter ( fcut,order[=8],type[=1],dBpass[=.5],dBstop[=-40] )
```

IIR, high-pass filtering; Apply a high-pass filter to the audio signal

arg	type	Description	Unit or Value scale
x	AUDIO	audio signal	
fcut	SCAL	cut-off frequency	
order	SCAL	order of the IIR filter	
type	SCAL	IIR filter type (1: Butterworth, 2: Chebyshev, 3: Elliptic)	
dBpass	SCAL	Passband ripple allowed	
dBstop	SCAL	Stopband attenuation	

## Outputs

- **y** is a highpass-filtered version of **x**.

## Notes

- The output is not normalized; i.e., the rms of **x** is adjusted according to the filter gain.
- **fcut** should be less than the Nyquist frequency.
- IIR filter coefficients are designed by the specification **requested** in the argument list, **but not guaranteed**. The user is responsible for making sure the output follows the spec.

## Examples

- `hpfilter(x, 2000)`

## See Also

[lpfilter](#) | [bpfilter](#) | [bsfilter](#)

## Algorithm

[ELLF Digital Filter Calculator](#)

# interp

```
y = interp ( x,factor )
y = x.interp ( factor )
y = x ~ factor
```

Interpolate the array. This is equivalent to spectrum compression / time expansion (or spectrum expansion / time compression) for an audio signal.

arg	type	Description	Unit or Value scale
x	AUD or Audio signal VCT		
factor	SCL	The factor to compress or expand the signal in time and frequency	

## Outputs

- y is AUD or VCT

## Notes

- This changes the number of samples in an audio signal. Internally the signal is treated as if it was resampled. Note that this is not actual resampling unless the sample rate in AUXLAB was adjusted.

## Examples

## See Also

## Algorithm

The library from [Secret Rabbit Code \(aka libsamplerate\)](#) is used.

# lpf

```
y = lpf ( x,fcut,order[=8],type[=1],dBpass[=.5],dBstop[=-40] )
y = x.lpf ( fcut,order[=8],type[=1],dBpass[=.5],dBstop[=-40] )
```

IIR, low-pass filtering; Apply a low-pass filter to the audio signal

arg	type	Description	Unit or Value scale
<b>x</b>	AUDIO	audio signal	
<b>fcut</b>	SCAL	cut-off frequency	
<b>order</b>	SCAL	order of the IIR filter	
<b>type</b>	SCAL	IIR filter type (1: Butterworth, 2: Chebyshev, 3: Elliptic)	
<b>dBpass</b>	SCAL	Passband ripple allowed	
<b>dBstop</b>	SCAL	Stopband attenuation	

## Outputs

- **y** is a lowpass-filtered version of **x**.

## Notes

- The output is not normalized; i.e., the rms of **x** is adjusted according to the filter gain.
- **fcut** should be less than the Nyquist frequency.
- IIR filter coefficients are designed by the specification **requested** in the argument list, **but not guaranteed**. The user is responsible for making sure the output follows the spec.

## Examples

- `lpf(x, 2000)`
- `noise(500).lpf(500)`

## See Also

[hpfilter](#) | [bpfilter](#) | [bsfilter](#)

## Algorithm

[ELLF Digital Filter Calculator](#)

# matrix

```
y = matrix ( x,m )
y = x.matrix ( m )
```

Turns an array into a matrix

arg	type	Description	Unit or Value scale
x	VCT or input array AUD		
m	SCAL	Number of rows (also known as the number of "groups")	

## Outputs

- **y** is a matrix with **m** rows.

## Notes

- The length of **x** must be a multiple of **m**. Number of columns is (length of **x**) / **m**

## Examples

- (1:12).matrix(3) →
- 1 2 3 4  
5 6 7 8  
9 10 11 12

## See Also

# movespec

```
y = movespec ( x,freq )
y = x.movespec ( freq )
y = x -> freq
```

Shift the spectrum in the frequency domain

arg	type	Description	Unit or Value scale
x	AUD	audio signal	
freq	VCT	how much the signal is shifted by frequency	

## Outputs

- **y** is audio with the same duration as **x**

## Notes

- This function shifts the frequencies all harmonic components; i.e., this is not the same as shifting the pitch.

## Examples

## See Also

[fscale](#)

# ramp

```
y = ramp ( x,dur_ramp )
y = x.ramp ( dur_ramp )
```

Smooth out beginning and ending portions of the audio signal; Apply the cosine square envelope

arg	type	Description	Unit or Value scale
x	AUDIO	audio signal	
dur_ramp	SCAL	ramping duration (i.e., duration to smooth out)	

## Outputs

- **y** is an audio signal with fade-in and fade-out.

## Notes

- Ramping is applied only to the portions specified by **dur\_ramp**. Windowing functions, such as hann hamming or blackman, apply the window for the whole duration.

## Examples

- `tone(440,100).ramp(40)`

## See Also

[hann](#) | [hamming](#) | [blackman](#)

# sam

```
y = sam ( x,rate,depth[=1],phase[=0] )
y = x.sam ( rate,depth[=1],phase[=0] )
```

Apply Sinusoidal-Amplitude-Modulation to an audio signal.

arg	type	Description	Unit or Value scale
<b>x</b>	AUD	audio signal	
<b>rate</b>	SCAL	modulation frequency	
<b>depth</b>	SCAL	degree of modulation, 0 (no modulation) to 1 (full modulation)	
<b>phase</b>	SCAL	initial AM phase (0 to 1)	

## Outputs

- **y** is AUDIO with sinusoidally amplitude modulation of **x**

## Notes

## Examples

## See Also

# sort

```
y = sort ( x,order[=1] )
y = x.sort ( order[=1] )
```

Sort the input array

arg	type	Description	Unit or Value scale
x	VCT	array to sort	
order	SCL	Direction: positive value for ascending (default); negative value for descending	

## Outputs

- **y** is the sorted array of **x**. If **x** is a matrix, each row is sorted.

## Notes

- Row-wise operation

## Examples

- Given  $y = [3 \ 1 \ 8 \ -2]$ ,  
 $y.sort \rightarrow [-2 \ 1 \ 3 \ 8]$ .

Given  $y =$   
[3 1 8 -2  
3 5 2 2],  
 $y.sort \rightarrow$   
[-2 1 3 8  
2 2 3 5]  
 $y.sort(-1) \rightarrow$   
[8 3 1 -2  
5 3 2 2].

## See Also

# tscale

```
y = tscale ( x,freq )
y = x.tscale ( freq )
y = x <> freq
```

Increase/decrease the duration of the signal without affecting the spectrum

arg	type	Description	Unit or Value scale
x	AUD	audio signal	
freq	VCT	The desired change of duration in ratio (> 1 to make longer or < 1 for shorter)	

## Outputs

- y is audio with the same frequency as x

## Notes

## Examples

## See Also

[fscale](#) | [movespec](#)

## Algorithm

The algorithm is based on Dan Ellis's matlab code (<http://www.ee.columbia.edu/ln/rosa/matlab/pvoc/>), which was originally based on (Flanagan & Golden, "Phase Vocoder" 1966). This function in AUMLAB has minor modifications from Dan Ellis's code 1) adjustment of short-term RMS to reduce unwanted fluctuations after processing and to adjust the duration properly.

# vector

```
y = vector( x )
y = x.vector( )
```

Convert an audio signal to a non-audio vector

arg	type	Description	Unit or Value scale
x	AUDIO		

## Outputs

- y is

## Notes

- This is used when hacking some aux functions requiring a non-audio vector argument

## Examples

## See Also

[audio](#)

# **Functions that perform computation with the given signal or vector or display properties of the signal.**

These are the functions that perform computation with an existing signal or vector with a given list of parameters, or display the properties associated with the signal. Some are relevant to only audio signals, while others are applicable both audio and non-audio. They include functions for the rms-level ([rms](#)), duration ([dur](#)), the timing information ([begint](#), [endt](#)), statistics ([mean](#), [std](#)), [length](#), [size](#)).

# begint

```
y = begint ( x )
y = x.begint ( )
```

Get the begin time of an audio (the same as tmark)

arg	type	Description	Unit or Value scale
x	AUD	audio input	

## Outputs

- **y** is a T-sequence showing the begin time of an audio signal **x**. **y** is a constant if there's a single signal chunk and tmark is 0.

## Notes

- Operation executed by chunk

## Examples

- tone(40,100).begint → 0 (constant)
- (tone(40,100)>>1000).begint → 0: 0 (time sequence)

## See Also

[dur](#) | [endt](#)

# cumsum

```
y = cumsum ( x )
y = x.cumsum ( )
```

Cumulative sum

arg	type	Description	Unit or Value scale
x	SCAL	the array	

## Outputs

- **y** is the cumulative sum of **x**.

## Notes

- If **x** is a matrix, **y** is a matrix of row-wise operations

## Examples

- [1 2 3 4 5].cumsum → [1 3 6 10 15]
- (1:12).matrix(3).cumsum →
- ans =  
1 3 6 10  
5 11 18 26  
9 19 30 42

## See Also

[diff](#) | [reshape](#)

# diff

```
y = diff ( x,n[=1] )
y = x.diff ( n[=1] )
```

Calculates the n-th difference between adjacent elements of **x**

arg	type	Description	Unit or Value scale
<b>x</b>	VCT or the array or signal AUDIO		
<b>n</b>	SCAL	skip count	

## Outputs

- **y** is the array of n-th order difference.

## Notes

- If **x** is a matrix, **y** is a matrix of row-wise operations

## Examples

- `a=1:5; a.diff` → [1 1 1 1]
- `a.diff(2)` → [2 2 2]
- `s=(1:12).matrix(3); s(2,:)*= 10; ss.diff` →  
`ans =`  
1 1 1  
10 10 10  
1 1 1

## See Also

[cumsum](#)

# dur

```
y = dur ( x )
y = x.dur ( )
```

Get the duration of an audio

arg	type	Description	Unit or Value scale
x	AUD	audio input	

## Outputs

- **y** is a T-sequence showing the duration of audio **x**. **y** is a constant if there's a single signal chunk and tmark is 0.

## Notes

- Operation executed by chunk

## Examples

- tone(40,100).dur → 100 (constant)
- (tone(40,100)>>1000).dur → 1000: 100 (time sequence)

## See Also

[begin](#) | [endt](#)

# endt

```
y = endt ( x )
y = x.endt ( )
```

Get the end time of an audio

arg	type	Description	Unit or Value scale
x	AUD	audio input	

## Outputs

- **y** is a T-sequence showing the end time of an audio signal **x**. **y** is a constant if there's a single signal chunk and tmark is 0.

## Notes

- Operation executed by chunk

## Examples

- tone(40,100).endt → 100 (constant)
- (tone(40,100)>>1000).endt → 1000: 1100 (time sequence)

## See Also

[begin](#) | [dur](#)

# envelope

```
y = envelope ( x )
y = x.envelope ( )
```

The Hilbert envelope (the magnitude of the analytic signal)

arg	type	Description	Unit or Value scale
x	AUDIO	the audio signal	

## Outputs

- y is the Hilbert envelope of x

## Notes

## Examples

## See Also

[hilbert](#)

# fft

```
y = fft ( x,n[=(size_of_x)] )  
y = x.fft ( n[=(size_of_x)] )
```

Computes the FFT of the array

arg	type	Description	Unit or Value scale
x	VCT or the array AUDIO		
n	SCAL	FFT size	

## Outputs

- y is complex array of size n

## Notes

## Examples

## See Also

[ifft](#)

## Algorithm

[FFTW](#)

# hilbert

```
y = hilbert ( x )
y = x.hilbert ( )
```

Computes the hilbert transform of the array--90 ° shift version; the imaginary part of the analytic signal

arg	type	Description	Unit or Value scale
-----	------	-------------	------------------------

**x**      AUDIO the audio signal

## Outputs

- **y** is the 90 ° -shift version of **x**.

## Notes

## Examples

## See Also

[envelope](#)

# ifft

```
y = ifft ( cx )
y = cx.ifft ( )
```

Computes the inverse FFT

arg	type	Description	Unit or Value scale
<b>cx</b>	comple	complex array x VCT	

## Outputs

- **y** is the inverse FFT of **cx**

## Notes

## Examples

## See Also

[fft](#)

## Algorithm

FFTW

# left

```
y = left ( x )
y = x.left ( )
```

Extract the left channel from a stereo audio

arg	type	Description	Unit or Value scale
x	AUDIO	audio signal	

## Outputs

- y is the left channel of x.

## Notes

- If x is not stereo, the function returns x

## Examples

- [.2\*tone(400,100); .1\*noise(300)].left // the 400-Hz tone channel is extracted

## See Also

# length

```
y = length ( x )
y = x.length ( )
```

Get the length of the array; for a 2-D array, it returns the entire length, i.e., the length of the "serialized" version of the array.

arg	type	Description	Unit or Value scale
x	STR AUD VCT SCL CEL	input array	

## Outputs

- For a non-audio variable or a contiguous audio signal, **y** is a constant scalar showing the length of the array. If **x** has null portions, **y** is a time sequence showing the length of each segment.

## Notes

- If **x** is an audio signal, it must not be a stereo signal.

## Examples

## See Also

# max

```
[y, id] = max ( x1,x2,... )  
[y, id] = x1.max ( x2,... )
```

Maximum element of an array if there is a single array argument; the maximum value from all values in the arguments

arg	type	Description	Unit or Value scale
x1	SCAL or VCT		
x2	SCAL or VCT		
...	...	...	

## Outputs

- **y** is the maximum element.
- **id** is the index of the maximum element found first.

## Notes

- Row-wise operation

## Examples

- a=[8 2 4 6]; max(a) or a.max → 8
- b=[3 9 -1]; max(a,b) or a.max(b) → 9
- Multiple output [a,b]=(1:10:120).matrix(3).max() returns
- a=  
31  
71  
111
  
- b=  
4  
4  
4

## See Also

[min](#)

# min

```
[y, id] = min ( x1,x2,... )  
[y, id] = x1.min ( x2,... )
```

Minimum element of an array if there is a single array argument; the minimum value from all values in the arguments

arg	type	Description	Unit or Value scale
x1	SCAL or VCT		
x2	SCAL or VCT		
...	...	...	

## Outputs

- **y** is the minimum element.
- **id** is the index of the minimum element found first.

## Notes

## Examples

- a=[8 2 4 6]; min(a) or a.min → 2
- b=[3 0 -1]; min(a,b) or a.min(b) → -1
- Multiple output [a,b]=(1:10:120).matrix(3).max() returns
- a=  
1  
41  
81
  
- b=  
1  
1  
1

## See Also

[max](#)

# right

```
y = right ( x )
y = x.right ( )
```

Extract the right channel from a stereo audio

arg	type	Description	Unit or Value scale
x	AUDIO	audio signal	

## Outputs

- y is the right channel of x.

## Notes

- If x is not stereo, the function returns x

## Examples

- [.2\*tone(400,100); .1\*noise(300)].right // the noise signal is extracted

## See Also

# rms

```
y = rms ( x )
y = x.rms ( )
```

Calculates the rms energy in dB

arg	type	Description	Unit or Value scale
x	AUDIO	the signal	

## Outputs

- y is SCAL

## Notes

- In AUX, by definition, the rms of a sinusoid with the full magnitude is 0 dB.

## Examples

- The rms of a sinusoid with the magnitude half of the full scale is -6 dB.

## See Also

# size

```
y = size ( x )
y = x.size ( )
```

Get the dimension of the matrix.

arg	type	Description	Unit or Value scale
x	STR AUD VCT SCL CEL	input array	

## Outputs

- y is a two-element vector showing the count of row and column, respectively.

## Notes

- For an array, first element of the output is one.

## Examples

## See Also

[length](#)

# std

```
y = std ( x,w[=0] )  
y = x.std ( w[=0] )
```

Calculates the standard deviation

arg	type	Description	Unit or Value scale
x	VCT or the array AUDIO		
w	SCAL	weight, 0: normalized by the (size-1) of #p1, 1: normalized by the size of #p1	

## Outputs

- y is SCAL or grouped SCAL

## Notes

## Examples

## See Also

[sum](#)

# sum

```
y = sum ( x )
y = x.sum ( )
```

Calculates the sum of the array elements

arg	type	Description	Unit or Value scale
x	VCT or the array AUDIO		

## Outputs

- y is

## Notes

## Examples

## See Also

# **Mathematical Functions**

These are the collections of mathematical functions.

# abs

```
y = abs ( x )
y = x.abs ( )
```

Absolute value or complex magnitude

arg	type	Description	Unit or Value scale
x	SCAL	n/a or VCT or AUDIO	

## Outputs

- **y** is the absolute value for a real number, or the magnitude for a complex number

## Notes

## Examples

## See Also

# acos

```
y = acos ( x )
y = x.acos ( )
```

Inverse cosine in radians

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is the Inverse Cosine,  $\cos^{-1}$ , of x

## Notes

## Examples

## See Also

[sin](#) | [cos](#) | [tan](#) | [asin](#) | [atan](#)

# angle

```
y = angle ( z )
y = z.angle ( )
```

Returns the phase angles

arg	type	Description	Unit or Value scale
z	SCAL or VCT or AUDIO	complex value(s)	

## Outputs

- **y** is the phase angles of **z** in radian.

## Notes

## Examples

## See Also

[abs](#) | [atan](#)

# asin

```
y = asin ( x )
y = x.asin ( )
```

Inverse sine in radians

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is the Inverse Sine,  $\sin^{-1}$ , of x

## Notes

## Examples

## See Also

[sin](#) | [cos](#) | [tan](#) | [acos](#) | [atan](#)

# atan

```
y = atan ( x )
y = x.atan ( )
```

Inverse tangent in radians

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- **y** is the Inverse Tangent,  $\tan^{-1}$ , of **x**

## Notes

## Examples

## See Also

[sin](#) | [cos](#) | [tan](#) | [asin](#) | [acos](#)

# ceil

```
y = ceil ( x )
y = x.ceil ( )
```

Round toward positive infinity

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is an "always-round-up" version of x.

## Notes

## Examples

- ceil(-2.3) returns -2.
- ceil(2.3) return 3.

## See Also

[fix](#) | [floor](#) | [round](#)

# conj

```
y = conj ( cx )
y = cx.conj ( )
```

Complex conjugate

arg	type	Description	Unit or Value scale
cx	SCAL or VCT or AUDIO	complex value(s)	

## Outputs

- `y` is complex conjugate of `cx`.

## Notes

## Examples

- `conj(3+2*sqrt(-1))` → `3-2*i`
- `(3+2*sqrt(-1)).conj` → `3-2*i`

## See Also

[imag](#) | [real](#)

# COS

```
y = cos ( x )
y = x.cos ( )
```

Cosine of argument in radians

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is Cosine of x.

## Notes

## Examples

## See Also

[sin](#) | [tan](#) | [acos](#) | [asin](#) | [atan](#)

# exp

```
y = exp ( x )
y = x.exp ( )
```

Exponential

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is the exponential  $e^x$

## Notes

## Examples

- $\text{exp}(1) \rightarrow 2.71828 \leftarrow e$

## See Also

[log](#)

# fix

```
y = fix ( x )
y = x.fix ( )
```

Round toward zero

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- If you take out the decimal portion of **x**, you get **y**.

## Notes

## Examples

- `fix(3.99) → 3`
- `fix(-3.99) → -3`

## See Also

[ceil](#) | [floor](#) | [round](#)

# floor

```
y = floor( x )
y = x.floor( )
```

Round toward negative infinity

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is a "round-down" version of x.

## Notes

## Examples

- floor(3.99) → 3
- floor(-3.99) → -4

## See Also

[ceil](#) | [fix](#) | [round](#)

# imag

```
y = imag ( z )
y = z.imag ( )
```

Imaginary part of complex number

arg	type	Description	Unit or Value scale
z	SCAL or VCT or AUDIO	complex value(s)	

## Outputs

- $y$  is the imaginary part of  $z$

## Notes

## Examples

- $(3).imag \rightarrow 0$   
 $(2+sqrt(-1)).imag \rightarrow 1$

## See Also

[conj](#) | [real](#) | [abs](#) | [angle](#)

# log

```
y = log ( x )
y = x.log ( )
```

Natural logarithm

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is natural log of x

## Notes

## Examples

## See Also

[exp](#) | [log10](#)

# log10

```
y = log10 ( x )
y = x.log10 ( )
```

Common logarithm

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is the common log of x

## Notes

## Examples

## See Also

[log](#)

# mod

```
y = mod ( x,div )
y = x.mod ( div )
y = x % div
```

Remainder after division (modulo operation)

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		
div	SCAL n/a		

## Outputs

- **y** is the remainder after division of **x** by **div**

## Notes

## Examples

- mod(17,5) rarr; 2
- mod(-1,3) rarr; -1
- (12).mod(5) rarr; 2

## See Also

# pow

```
y = pow ( x )
y = x.pow ( )
```

Raise to power

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

## Notes

- base raised to the power exponent;  $\text{base}^{\text{power}}$

## Examples

## See Also

[log](#)

# real

```
y = real ( z )
y = z.real ( )
```

Real part of complex number

arg	type	Description	Unit or Value scale
z	SCAL or VCT or AUDIO	complex value(s)	

## Outputs

- $y$  is the real part of  $z$

## Notes

## Examples

- $(3).imag \rightarrow 3$   
 $(2+sqrt(-1)).imag \rightarrow 2$

## See Also

[imag](#) | [conj](#) | [abs](#) | [angle](#)

# round

```
y = round ( x )
y = x.round ( )
```

Round to nearest decimal or integer

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- **y** is a "half-round-up" version of **x**.

## Notes

## Examples

- `round(2.6) → 3`
- `round(2.4) → 2`
- `round(-2.6) → -3`
- `round(-2.4) → -2`

## See Also

[ceil](#) | [fix](#) | [floor](#)

# sign

```
y = sign ( x )
y = x.sign ( )
```

Sign function (signum function)

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is

## Notes

## Examples

## See Also

# sin

```
y = sin ( x )
y = x.sin ( )
```

Sine of argument in radians

arg	type	Description	Unit or Value scale
x	SCAL, n/a VCT or AUDIO		

## Outputs

- y is Sine of x.

## Notes

## Examples

## See Also

[cos](#) | [tan](#) | [acos](#) | [asin](#) | [atan](#)

# **sqrt**

```
y = sqrt ( x )
y = x.sqrt ( )
```

Square root

<b>arg</b>	<b>type</b>	<b>Description</b>	<b>Unit or Value scale</b>
<b>x</b>	SCAL	n/a or VCT or AUDIO	

## Outputs

- **y** is square root of **x**.

## Notes

- If **x** is non-audio, square root of negative values produce imaginary values. If **x** is audio, this produces sign(sqrt(**x**)).

## Examples

## See Also

# **tan**

```
y = tan ( x )
y = x.tan ( )
```

Tangent of argument in radians

<b>arg</b>	<b>type</b>	<b>Description</b>	<b>Unit or Value scale</b>
<b>x</b>		SCAL, n/a VCT or AUDIO	

## **Outputs**

- **y** is tangent of **x**.

## **Notes**

## **Examples**

## **See Also**

[sin](#) | [cos](#) | [acos](#) | [asin](#) | [atan](#)

# Audio Playback Functions

In AUXLAB, you can [play](#) an audio signal, [pause](#), [resume](#) or [stop](#) it prematurely. Multiple audio playback events can exist concurrently and can be controlled or managed. The handle to audio playback, the output of each play call, is used to specify the particular audio event to handling it. Note: [play](#) is non-blocking and will return immediately.

For example,

```
AUX> noi = noise(1000).bpf(500,2000);  
To play just one and only one, no need to have the output  
AUX> noi.play
```

But when playing a signal with a long duration, it is necessary to have the output,

```
AUX> noi = noise(10000).bpf(500,2000);x =  
tone(500,10000)@-20;  
AUX> h=noi.play  
AUX> h2=x.play
```

Then,

```
AUX> h.pause or h2.pause  
or  
AUX> h.stop
```

in the middle of the playback will do the trick.

The audio handle, h or h2 in this example, contains the information about the audio play back including the progress of the playback and is constantly updated in the background. The user can view the status of the audio event in the variable view window or by retrieving it in the AUXLAB command window.

# pause

```
y = pause( h )
y = h.pause( )
```

Pause an on-going audio playback

arg	type	Description	Unit or Value scale
h	AUDIOHANDL	Audio playback event handle to	
E		pause	

## Outputs

- **y** is the audio playback handle

## Notes

## Examples

## See Also

[resume](#) | [play](#) | [stop](#) | [qstop](#) | [status](#)

# play

```
y = play ( x,repeat[=1],devID[""] )  
y = play ( handle,x,repeat[=1],devID[""] )  
y = handle.play ( x,repeat[=1],devID[""] )
```

Audio playback

arg	type	Description	Unit or Value scale
<b>handle</b>	HAUD	handle to audio playback	
<b>x</b>	AUD	audio signal	
<b>repeat</b>	SCAL	number of repeats	
<b>devID</b>	TXT	string identifier of the playback device (NOT YET IMPLEMENTED--as of AUXLAB 1.44)	

## Outputs

- **y** is either an audio handle, either newly created or existing. **y** is -1 if the playback fails or the specified handle is invalid.

## Notes

- If devID is not specified, the default device or the last device selected for playback will be used.
- If play is called for the audio handle, **handle** is queued in the back of the playback list and played when the existing list is exhausted.  
If play is called for an audio signal, a new audio handle is generated.

## Examples

- To play x (and don't care about asynchronous playing),  
`play(x)`
- To play x and, then, y in sequence  
`h=play(x)`  
`h.play(y)` while x is played.  
If `h.play(y)` is given after playing x is done, it doesn't play and returns -1

## See Also

[pause](#) | [resume](#) | [status](#) | [stop](#) | [qstop](#)

# qstop

```
y = qstop( h )
y = h.qstop( )
```

Stop an on-going audio playback

arg	type	Description	Unit or Value scale
h	AUDIOHANDL	Audio playback event handle to stop	E

## Outputs

- **y** is the audio playback handle

## Notes

- Unlike stop, this function terminates the on-going audio event immediately.

## Examples

## See Also

[play](#) | [stop](#) | [pause](#) | [resume](#) | [status](#)

# resume

```
y = resume( h )
y = h.resume( )
```

Resume a paused audio playback

arg	type	Description	Unit or Value scale
h	AUDIOHANDL	Audio playback event handle to E resume	

## Outputs

- **y** is the audio playback handle

## Notes

## Examples

## See Also

[pause](#) | [play](#) | [stop](#) | [qstop](#) | [status](#)

# status

```
y = status ( h )
y = h.status ( )
```

Displays the status of the audio playback handle

arg	type	Description	Unit or Value scale
h	AUDIOHANDL	Audio playback event handle	E

## Outputs

- **y** is an object variable showing the status of the audio handle.

## Notes

## Examples

## See Also

# stop

```
y = stop ( h )
y = h.stop ( )
```

Stop an on-going audio playback

arg	type	Description	Unit or Value scale
h	AUDIOHANDL	Audio playback event handle to E stop	

## Outputs

- **y** is the audio playback handle

## Notes

- The audio playback fades out (i.e., "ramps" out) with a cosine amplitude function for 350 milliseconds.

## Examples

## See Also

[play](#) | [qstop](#) | [pause](#) | [resume](#) | [status](#)

# Graphics Functions

In addition to displaying the signal as a graph by pressing the enter key in the variable show window, graphic windows can be created and controlled with AUX functions. As of 9/20/2018, while there are five functions--[plot](#), [figure](#), [axes](#), [delete](#), [text](#), each graphic object is equipped with numerous properties and the appearance and display can be controlled by modifying the graphic properties.

## Example

```
AUX> x=(1:10).sqrt;
AUX> hLine=plot(x); // plot returns the handle to the line object
AUX> hAx = hLine.parent; // hAx is the handle to the axes object
AUX> hFig = hAx.parent; // hFig is the handle to the figure object
AUX> hAx.pos(4) /=2; // The axes size adjusted--the height becomes half
AUX> hAx2 = hFig.axes(hAx.pos+[0 .45 0 0]); // A new axis is
added to the figure window
AUX> hAx2.plot((1:10).log); // The second plot
AUX> hTxt = hFig.text(.5,.55,"Two plots") // A text added
AUX> hTxt.fontsize(15) // Font size adjusted
```

# axes

```
y = axes( h_or_pos )
y = h_or_pos.axes( )
```

Set the current axes or create a new axes

arg	type	Description	Unit or Value scale
<b>h_or_pos</b>	HGO or handle to the axes or 4-element VECT	vector with the position to create a new axes in	

## Outputs

- **y** is the handle to the axes object

## Notes

- The position is a 4-element vector in proportion to the window size showing the following: top left width height, with the reference of the bottom left corner of the screen.
- This function creates an axes in the current figure window. If no current figure window is available, a new one is created.

## Examples

- `h = axes([.08 .18 .86 .5])` // create a new axes in the current figure window
- `axes(hPrev)` // set hPrev as the current axes

## See Also

[figure](#) | [plot](#) | [text](#) | [delete](#)

# delete

```
y = delete ( object )
y = object.delete ( )
```

Delete the graphic object

arg	type	Description	Unit or Value scale
object	HGO	The graphic handle to the object to delete	

## Outputs

- `y` is empty.

## Notes

- Upon success, the variable `object` will become empty.

## Examples

## See Also

[figure](#) | [plot](#) | [axes](#) | [text](#)

# figure

```
y = figure ( h_or_pos )
```

Get the handle of an existing figure window or create a blank figure window

arg	type	Description	Unit or Value scale
h_or_pos	HGO or handle to the figure window or 4-VECT	HGO or handle to the figure window or 4-element vector with the position to create a new figure at	

## Outputs

- **y** is the handle to the figure window

## Notes

- The position is a 4-element vector in pixel count showing the following: top left width height, with the reference of the top left corner of the screen.

## Examples

## See Also

[plot](#) | [axes](#) | [text](#) | [delete](#)

# plot

```
y = plot ( x,y[],opt[] )  
y = plot ( handle,x,y[],opt[] )  
y = handle.plot ( x,y[],opt[] )
```

Plot the data

arg	type	Description	Unit or Value scale
<b>handle</b>	HGO	handle to the figure window or the axes	
<b>x</b>	VECT	x-data or AUD	
<b>y</b>	VECT	y-data or AUD	
<b>opt</b>	TXT	plot option string--specifying color marker and line style (see notes)	

## Outputs

- **y** is the handle to the line object

## Notes

- If **handle** is omitted, this function creates a new figure window and plots
- If **y** is specified, **y** is plotted as a function of **x**.
- If **y** is omitted, **x** is plotted with the index (if non-audio) or the time (if audio) on the x-axis.
- **opt** is a text not exceeding 4 characters specifying the color, the marker type and the line style.  
Color: r(ed) g(reen) b(lue) y(ellow) c(yan) m(agenta) h(white) (blac)k  
Marker: o circle s square . point \* asterisk x cross + plussign d diamond  
^ upward-pointingtriangle v downward-pointingtriangle > right-pointingtriangle < left-pointingtriangle  
Linestyle: - solid : dotted -- dashed -. dash-dot
- The default marker type is . (point)
- The default line style is solid, but if only the marker type is specified, the line style will become none (i.e., in order to make the line tyle none, specify ".")
- The default color is blue
- All the other graphic properties can be directly manipulated with the relevant member variables.

## Examples

- `plot(x)` // plot x with the default parameters

- `plot(x,y) // plot y as a function of x with the default parameters`
- `plot(x,"r") // plot x with the red line`
- `plot(x,"o") // plot x with the marker "o" and no line`
- `plot(x,"*":) // plot x with the marker "*" and dotted line`
- `plot(x,y,"o") // plot y as a function of x with the marker "o" and no line`
- `plot(x,y,"*:") // plot y as a function of x with the marker "*" and dotted line`
- `plot(h,x) // plot x in h (either a figure window handle or axes handle)`
- `plot(h,x,y)`
- `plot(h,x,y,"r")`

## See Also

[figure](#) | [axes](#) | [text](#) | [delete](#)

# text

```
y = text ( h,pos_x,pos_y,text )
y = _h.text ( pos_x,pos_y,text )
```

Display a text in the figure window or axes.

arg	type	Description	Unit or Value scale
<code>_h</code>	HGO	Graphic handle, either figure or axes handle	
<code>pos_x</code>	SCAL	x position proportion of the window or axes	
<code>pos_y</code>	SCAL	y position proportion of the window or axes	
<code>text</code>	TXT	Text to display	

## Outputs

- `y` is the handle to the text displayed.

## Notes

- If `_h` is omitted, the text is displayed in the current figure window. If the current figure window doesn't exist, it creates one.

## Examples

- `text(.5, .5, "hello world!") // hello world!` is displayed at the center point in the current window (left-aligned)

## See Also

[figure](#) | [plot](#) | [axes](#) | [delete](#)

# Logical Functions

# and

```
y = and ( b1,b2 )  
y = b1.and ( b2 )
```

Point-wise boolean operation `&&` for two arguments; Check if all logical elements are true.

arg	type	Description	Unit or Value scale
<b>b1</b>	LOGICA L VCT or SCL	Logical array or logical scalar	
<b>b2</b>	LOGICA L VCT or SCL	Logical array or logical scalar	

## Outputs

- `y` is a logical array for two arguments with the length of a shorter argument or a logical scalar for one argument.

## Notes

- If two arguments have different length, the boolean operation is applied only to the common length.

## Examples

## See Also

[or](#)

# isaudio

```
y = isaudio ( x )
y = x.isaudio ( )
```

Checks if the input array is an audio array.

arg	type	Description	Unit or Value scale
x	anythin g		

## Outputs

- y is true if x is audio.

## Notes

## Examples

## See Also

[isvector](#)

# isbool

```
y = isbool ( x )
y = x.isbool ( )
```

Checks if the input is logical.

arg	type	Description	Unit or Value scale
x	anything		

## Outputs

- y is true if x is logical.

## Notes

## Examples

## See Also

# iscell

```
y = iscell ( x )
y = x.iscell ( )
```

Checks if the input is a cell array.

arg	type	Description	Unit or Value scale
x	anythin g		

## Outputs

- y is true if x is a cell array.

## Notes

## Examples

## See Also

# isclass

```
y = isclass ( x )
y = x.isclass ( )
```

Checks if the variable is a class object.

arg	type	Description	Unit or Value scale
x	anythin g		

## Outputs

- y is boolean

## Notes

## Examples

- isaudio isvector isstring isbool isempty

## See Also

# isempty

```
y = isempty ( x )
y = x.isempty ( )
```

Checks whether the input array is empty.

arg	type	Description	Unit or Value scale
x	anything		

## Outputs

- `y` is true if the input is empty.

## Notes

- For a time-seq variable, the array can have zero length but shouldn't be considered empty because it carries the tmark information.

## Examples

## See Also

# isstereo

```
y = isstereo ( x )
y = x.isstereo ( )
```

Checks if the audio signal is stereo.

arg	type	Description	Unit or Value scale
x	AUD	n/a	

## Outputs

- y is logical indicating whether the signal is stereo.

## Notes

## Examples

## See Also

# isstring

```
y = isstring ( x )
y = x.isstring ( )
```

Checks if the input is string.

arg	type	Description	Unit or Value scale
x	anythin g		

## Outputs

- y is true if x is string.

## Notes

## Examples

- x="bj kwon";
- x.isstring → true

## See Also

# isvector

```
y = isvector ( x )
y = x.isvector ( )
```

Checks if the input array is a vector (non-audio array).

arg	type	Description	Unit or Value scale
x	anythin g		

## Outputs

- y is true if x is a vector (non-audio array).

## Notes

## Examples

## See Also

[isaudio](#)

## or

```
y = or ( b1,b2 )
y = b1.or ( b2 )
```

Point-wise boolean operation || for two arguments; Check if any one element is true.

arg	type	Description	Unit or Value scale
b1	LOGICA L VCT or SCL	Logical array or logical scalar	
b2	LOGICA L VCT or SCL	Logical array or logical scalar	

### Outputs

- **y** is a logical array for two arguments with the length of a shorter argument or a logical scalar for one argument.

### Notes

- If two arguments have different length, the boolean operation is applied only to the common length.

### Examples

### See Also

[and](#)

# File/Directory Functions

# dir

```
y = dir ( s )
```

Retrieve the files in the specified directory

arg	type	Description	Unit or Value scale
s	STR	File name with or without a wild card, or directory	

## Outputs

- `y` is a cell array showing the directory contents. Each cell is a directory class with the following members: `name`, `ext`, `path`, `isdir`, `date`, and `bytes`.

## Notes

## Examples

- `y = dir ("c:\Temp\auxlab\1.42")` or
- `y = dir ("c:\Temp\auxlab\1.42\*.ini")`
- `y{1} =`  
  `.bytes = 1341`  
  `.date = "06/18/2018, 04:18:28"`  
  `.ext = ".ini"`  
  `.isdir = (logical) 0`  
  `.name = "auxcon32.AUDITORY"`  
  `.path = "C:\Temp\auxlab\1.42"`
- `y{2} =`  
  `.bytes = 160`  
  `.date = "06/18/2018, 04:22:06"`  
  `.ext = ".ini"`  
  `.isdir = (logical) 0`  
  `.name = "auxlab32.AUDITORY"`  
  `.path = "C:\Temp\auxlab\1.42"`

## See Also

# fclose

```
y = fclose ( file_id )
```

Closes a file stream specified by the file identifier

arg	type	Description	Unit or Value scale
file_id	SCAL	File identifier from the fopen call	

## Outputs

- y is SCAL. 0 for success, -1 for failure.

## Notes

## Examples

## See Also

[fopen](#) | [fprintf](#)

# fdelete

```
y = fdelete ( filename )
```

Delete a file

arg	type	Description	Unit or Value scale
-----	------	-------------	---------------------

**filename** STR The file name to delete

## Outputs

- **y** is 1 for success, 0 for failure.

## Notes

- The file name may include the path.

## Examples

- `fdelete("c:\delete.me")`

## See Also

[fopen](#) | [fprintf](#) | [fclose](#)

# file

```
y = file ( filename )
```

Read from a file (TO BE DONE)

arg	type	Description	Unit or Value scale
-----	------	-------------	---------------------

**filename** TEXT .....

## Outputs

- **y** is anything

## Notes

## Examples

## See Also

# fopen

```
y = fopen ( filename,mode )
```

Opens a file with the given mode

arg	type	Description	Unit or Value scale
<b>filename</b>	TXT	Name of the file to open	
<b>mode</b>	TXT	File open mode, such as "r" "w" "a"	

## Outputs

- **y** is SCAL indicating the file identifier. **y** is -1 if there's an error.

## Notes

- Internally, the C fopen function is called with the given arguments. This means that this follows all the conventions in C language.

## Examples

## See Also

[fclose](#) | [fprintf](#) | [fread](#) | [fwrite](#)

# fprintf

```
y = fprintf ( file,format,... )
```

Write formatted data to a file

arg	type	Description	Unit or Value scale
<b>file</b>	TXT or SCAL	file name string or file identifier	
<b>format</b>	TXT	String that contains the text to be written to the file, following the same convention as in C language.	
...	...	n/a	

## Outputs

- **y** is the number of characters written. If an error occurs, **y** is negative and one of the following:
  - 1: fopen error
  - 2: invalid file identifier
  - 3: fwrite error
  - 4: fclose error
  - 999: Unknown error

## Notes

- **file** can be either a text of the file name (if an extension is not specified, .txt is added by default) or a file identifier, which is the output of fopen
- When the file name is specified for **file**, fprintf opens the file with the mode "at," meaning that the content will be appended to the existing content in the file, writes the content as specified, and closes the file.
- If a file identifier is used for **file**, fprintf writes the content to the file stream opened by a prior call to fopen.

## Examples

### See Also

[fopen](#) | [fclose](#) | [sprintf](#)

# include

```
include ( filename )
```

Run a script

arg	type	Description	Unit or Value scale
-----	------	-------------	---------------------

**filename** TXT file name

## Outputs

- No output

## Notes

- This is simply to run a batch AUX script, which is different running a UDF (user-defined function).

## Examples

## See Also

# wavwrite

```
y = wavwrite( x,filename,opt )
y = x.wavwrite( filename,opt )
```

Generate a .wav file from the audio signal

arg	type	Description	Unit or Value scale
<b>x</b>	AUDIO		
<b>filename</b>	STR	.wav audio file name	
<b>opt</b>	STR	file encoding format	

## Outputs

- The output **y**, if specified, is the audio signal **x**.

## Notes

- If **filename** doesn't indicate the extension, .wav is used.
- **opt** is one of the following
  - "8" 8-bit PCM
  - "16" 16-bit PCM (default)
  - "24" 24-bit PCM
  - "32" 32-bit PCM
  - "alaw" a-law encoding
  - "ulaw" μ-law encoding
  - "adpcm1" ADPCM encoding 1
  - "adpcm2" ADPCM encoding 2

## Examples

## See Also

[wave](#)

# Miscellaneous Functions

# clear

```
clear ( x )
x.clear ( )
```

Clear the variable (or the member variable) from the workspace

arg	type	Description	Unit or Value scale
x		any variable or a member variable of a class variable	

## Outputs

- No output

## Notes

## Examples

- a.clear or clear(a) → clear the variable a  
a.obj.clear or clear(a.obj) → clear the member obj from the class variable a

## See Also

# eval

```
n/a = eval ( expression )
```

Execute AUXLAB expression in text string

arg	type	Description	Unit or Value scale
<b>expression</b>	TXT	String that contains a valid AUXLAB expression.	

## Outputs

- n/a

## Notes

- Similar to the eval function in MATLAB

## Examples

## See Also

# getfs

```
y = getfs( )
```

Retrieve the sample rate in the AUXLAB workspace.

arg	type	Description	Unit or Value scale
-----	------	-------------	---------------------

## Outputs

- `y` is the sample rate in the AUXLAB workspace.

## Notes

- no argument

## Examples

## See Also

[setfs](#)

# inputdlg

```
y = inputdlg ( title,content )
y = title.inputdlg ( content )
```

Create a simple message dialog box with OK and cancel buttons and an edit box for user input.

arg	type	Description	Unit or Value scale
<b>title</b>	TEXT	The title of the dialog box	
<b>content</b>	TEXT	The content to display in the dialog box. May use the printf format	

## Outputs

- **y** is the string typed by the user.

## Notes

- This does not return until the user responds with OK or cancel.

## Examples

## See Also

[input](#)

# msgbox

```
msgbox ( format,arg )
```

Display a messagebox

arg	type	Description	Unit or Value scale
<b>format</b>	TEXT	printf-style format	
<b>arg</b>	anythin	variables	
	g ...		

## Outputs

- No output

## Notes

## Examples

- for k=1:100, value = 2^k; if value > 1000 msgbox("2 raised by the power of %d is greater than 1000.", k); break; end; end

## See Also

[input](#) | [inputdlg](#)

# setfs

```
y = setfs ( new_Fs )
```

Adjusts the sample rate to the specified value

arg	type	Description	Unit or Value scale
<b>new_Fs</b>	SCL	The new sample rate	

## Outputs

- **y** is undefined. Don't try to use it, as in `setfs(16000)+1`

## Notes

- The usage of this function is limited to the user-defined functions or the auxcon module. In general, it is better to use a hook command (as in `#setfs 16000`), because it involves UI and expressions. This will not update existing variables according to the new sample rate, as done by the hook command, so this functionality is pretty limited.

## Examples

## See Also