

# **Reference Manual for the Auditory Research Soundcard Application Programming Interface**

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## Table of Contents

ARSC API Overview .....	4
ARSC Function List by Category .....	5
ARSC Examples .....	6
Streaming Output .....	7
Synchronous Input/Ouput .....	9
ARSC Functions .....	10
ar_adjust_rate .....	11
ar_close_all .....	12
ar_dev_name .....	13
ar_err_msg .....	14
ar_find_dev .....	15
ar_find_dev_name .....	16
ar_get_fmt .....	17
ar_get_cardinfo .....	18
ar_get_gdsr .....	19
ar_get_rate .....	20
ar_get_sfs .....	21
ar_get_vfs .....	22
ar_io_close .....	23
ar_io_cur_seg .....	24
ar_io_open .....	25
ar_io_prep .....	26
ar_io_prepare .....	27
ar_io_start .....	28
ar_io_stop .....	29
ar_num_devs .....	30
ar_out_open .....	31
ar_out_prepare .....	32
ar_out_seg_fill .....	33
ar_set_fmt .....	34
ar_set_latency .....	35
ar_set_sfs .....	36
ar_set_vfs .....	37
ar_set_xfer .....	38
ar_wind .....	39
ar_version .....	40
ar_xruns .....	41
ARSC DLL .....	42
ar_wait_seg .....	42
ar_io_prepare_vb .....	42
ar_out_prepare_vb .....	43
ar_set_xfer_stdcall .....	44
ar_err_msg_matlab .....	44
ar_dev_name_matlab .....	44
ar_version_DLL .....	44

## ARSC API

ar_fill_tone.....	44
SIO API.....	46
sio_open.....	46
sio_close.....	46
sio_get_nioch.....	46
sio_get_device .....	46
sio_set_device.....	46
sio_get_info .....	47
sio_get_vfs.....	47
sio_set_size .....	47
sio_set_output.....	47
sio_set_input.....	48
sio_set_average.....	48
sio_set_rate .....	48
sio_set_att_in .....	48
sio_set_att_out .....	48
sio_set_vfs .....	49
sio_set_escape.....	49
sio_io.....	49
sio_io_chk.....	49
sio_set_latency.....	50
sio_get_cardinfo.....	50
sio_set_cardinfo .....	50
ARSC Pre-defined Constants.....	51
ARSC Error Codes.....	52

## ARSC API Overview

This reference manual describes an application programming interface (API) for auditory research soundcard (ARSC) applications. The set of functions provided by this API simplify the programming of soundcard input and output for typical auditory research requirements, including (1) output that is synchronized to visual cues and (2) output that is synchronized to input. These functions can be accessed directly from a C program by linking to the arsc.lib static library or accessed indirectly through the arsc.dll dynamic library.

Not all soundcards will synchronize input and output. A few (such as DAL CardDeluxe, Echo Indigo, and Lynx L22) are able to perform synchronous input/out (with consistent latency) using standard Windows drivers. Soundcards for which ASIO drivers are available will also perform synchronous i/o. The ARSC API can be used with any Windows soundcard, even when i/o is not synchronized properly. The ARSC API works with either ASIO or WDM drivers.

The ARSC API also works with ALSA drivers under Linux. Soundcards from Echo (such as Gina24 and Indigo) have ALSA drivers and have been tested to perform synchronous i/o with consistent latency. The ALSA website <http://www.alsa-project.org> provides a list of supported soundcards and, if needed, soundcard firmware. Under Linux, users should belong to the “audio” group to obtain permission to use the soundcard with ARSC applications.

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## ARSC Function List by Category

### Input / Output functions

ar_io_open	Open device for i/o
ar_io_close	Close device
ar_io_prepare	Prepare device for i/o
ar_io_start	Start i/o
ar_io_stop	Stop i/o
ar_set_fmt	Specify app's data format
ar_set_xfer	Install data-transfer functions
ar_io_cur_seg	Get current segment number
ar_close_all	Close all devices

### Output-only functions

ar_out_open	Open device for output
ar_out_prepare	Prepare device for output
ar_out_seg_fill	Refill output segments

### Information functions

ar_find_dev	Find device given preferences
ar_find_dev_name	Find device given name
ar_dev_name	Get device name
ar_err_msg	Get error message
ar_get_cardinfo	Get soundcard information
ar_get_fmt	Get device's data format
ar_get_gdsr	Get device's good sampling rates
ar_get_rate	Get sampling rate
ar_adjust_rate	Return closest sampling rate
ar_get_vfs	Set ADC/DAC volts-full-scale
ar_num_devs	Get number of devices
ar_set_vfs	set ADC/DAC volts-full-scale
ar_version	Get ARSC version
ar_wind	Specify window to receive messages
ar_xruns	Get number of under & over runs

## ARSC Examples

Code fragments of two short examples are included in this section to demonstrate the proper calling sequence of ARSC functions for simple applications. Testing of error codes returned by the ARSC functions is recommended, but such details have been omitted in these examples for clarity.

tstfm.c – Demonstrates continuous streaming output of a frequency-modulated tone with coordinated visual cues.

tstlat.c – Demonstrates synchronous averaging of the recorded response to a repeated stimulus output.

Other programming examples are included in the ARSC source code distribution, which can be downloaded from <http://audres.org/downloads/arscsc.zip>.

tstout.c – Demonstrates the use of multiple waveform segments to control the timing of a two-interval stimulus output with coordinated visual cues.

tstsio.c – Demonstrates synchronous averaging using an older API called SIO, which is supported in the ARSC library for backward compatibility.

tst3ch.c – Demonstrates *three-channel* continuous streaming output. Similar to tstfm.c

## Streaming Output

This code fragment provides an example of continuous streaming output. For a complete version, see the file `tstfm.c`.

```
#include "arsclib.h"

double rate = 44100;    // sampling rate
int dvid = 0;           // device identifier
int stop = 0;           // terminates output
int nsmp = 2500;        // samples per segment
int nseg = 2;           // number of segments
long *stm[2];           // list of stimulus buffers

void
tone_seg(int seg)       // output callback function
{
    // fill output buffer here
}

void
test_tone(double f, double t) // tone frequency and duration
{
    long siz[2], fmt[2];
    void *out[4];
    int nchn = 2;
    int nswp = 0;        // 0 is for streaming

    // set up i/o

    dvid = ar_find_dev(ARSC_PREF_NONE);
    ar_out_open(dvid, rate, nchn);
    fmt[0] = ARSC_DATA_I4;    // 32-bit integer data
    fmt[1] = 0;              // samples non-interleaved
    ar_set_fmt(dvid, fmt);    // specify data format

    // set up stimulus

    rate = ar_get_rate(dvid);
    stm[0] = calloc(nsmp, sizeof(long));
    stm[1] = calloc(nsmp, sizeof(long));
    out[0] = out[1] = stm[0]; // put stm on left
    out[2] = out[3] = stm[1]; // put stm on right
    siz[0] = siz[1] = nsmp;

    // perform i/o

    ar_set_xfer(dvid, NULL, tone_seg); // specify callback
    ar_out_prepare(dvid, out, siz, nseg, nswp); // prepare
    ar_io_start(dvid); // start

    while (!stop) {        // loop until user terminates
        chk_msg();          // check for user input
    }
}
```

## ARSC API

```
        cue(); // provide visual cue
        ar_io_cur_seg(dvid);
    };
    ar_io_close(dvid);

    // clean up stimulus

    free(stm[0]);
    free(stm[1]);
}
```



## ***Synchronous Input/Output***

This code fragment provides an example of simultaneous input and output. For a complete version see the file `tstlat.c`.

```
#include "arsclib.h"

int
test_io()
{
    double sr = 48000; // sampling rate
    int d, m;
    long *sl, *rl, fmt[2];
    void *od[1], *id[1];
    int io_dev = 0;
    long nseg = 1; // number of segments
    long nsmp = 4800; // samples per segment
    long nswp = 1; // number of sweeps

    // set up i/o

    d = io_dev;
    ar_io_open(d, sr, 1, 1); // one-channel i/o
    fmt[0] = ARSC_DATA_I4; // int32
    fmt[1] = 0; // non-interleaved
    ar_set_fmt(d, fmt);
    sr = ar_get_rate(d);
    id[0] = rl = calloc(nsmp, sizeof(long)); // input data
    od[0] = sl = calloc(nsmp, sizeof(long)); // output data
    sl[0] = (long) pow(2, 31) - 1; // click

    // perform i/o

    ar_io_prepare(d, id, od, &nsmp, nseg, nswp);
    ar_io_start(d)
    while (ar_io_cur_seg(d) < nseg) {
        continue;
    }

    // clean up i/o

    ar_io_stop(d);
    ar_io_close(d);
    free(rl);
    free(sl);
}
```

## **ARSC Functions**

The following pages provide a description of each ARSC function in alphabetical order.

## ***ar\_adjust\_rate***

Adjust desired sampling rate to nearest usable rate

```
double ar_adjust_rate(  
    long dev,  
    double rate  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

The nearest sampling rate possible on the specified soundcard device.

### **Remarks**

If the soundcard device can't provide the desired rate, then it will be set to the closest rate possible.

### **See Also**

ar\_io\_get\_rate, ar\_io\_set\_rate

### ***ar\_close\_all***

Close all devices

```
void ar_close_all(  
    );
```

#### **Parameters**

None.

#### **Return Value**

None

#### **Remarks**

Use this function to close all open ASRC devices before program termination.

## ***ar\_dev\_name***

Get device name

```
long ar_dev_name(  
    long dev,  
    char *name,  
    long len  
);
```

### **Parameters**

dev	Device identifier.
name	Array to receive name of device.
len	Length of name array. Recommended length is ARSC_NAMLEN.

### **Return Value**

Error code.

### **Remarks**

Returns the name of the device associated with the specified identifier. Device may be opened or closed at the time this function is called.

## ***ar\_err\_msg***

Get error message

```
void ar_err_msg(  
    long err,  
    char *msg,  
    long len  
);
```

### **Parameters**

<code>err</code>	Error code.
<code>msg</code>	Array to receive name of device.
<code>len</code>	Length of <code>msg</code> array. Recommended length is <code>ARSC_MSGLEN</code> .

### **Return Value**

None.

### **Remarks**

Retrieves an error message for a specified error code.

## ***ar\_find\_dev***

Find device given preferences

```
long ar_find_dev(  
    long flags  
);
```

### **Parameters**

dev            Device identifier.

flags          Hints about desired device, which may be set to one of the following values.

- ARSC\_PREF\_NONE
- ARSC\_PREF\_SYNC
- ARSC\_PREF\_ASIO
- ARSC\_PREF\_ALSA
- ARSC\_PREF\_WIND

### **Return Value**

Device identifier

### **Remarks**

Retrieves a device identifier for a soundcard device that matches, if possible, with the indicated preference. Defaults to 0, which indicates the system default soundcard device.

### ***ar\_find\_dev\_name***

Find device given name

```
long ar_find_dev_name(  
    char *name  
);
```

#### **Parameters**

name            Device name.

#### **Return Value**

Device identifier

#### **Remarks**

Retrieves a device identifier for a soundcard device which the specified name. Defaults to 0, which indicates the system default soundcard device.



## ***ar\_get\_fmt***

Get device's data format

```
long ar_get_fmt(  
    long dev,  
    long *fmt  
);
```

### **Parameters**

dev            Device identifier.

fmt            Array of parameters to describe the format of the soundcard device's sample data. The length of this array is 2.

fmt[0]	<i>data type</i> — one of the following values: ARSC_DATA_I2 – 16-bit integer ARSC_DATA_I4 – 32-bit integer ARSC_DATA_F4 – 32-bit floating-point ARSC_DATA_F8 – 64-bit floating-point
fmt[1]	<i>channel interleave</i> — It's value will be 0 to indicate <i>non-interleaved</i> sample data or 1 to indicate <i>interleaved</i> sample data.

### **Return Value**

Error code.

### **Remarks**

Retrieves the format of the soundcard device's sample data, which can be *interleaved* or *non-interleaved* and one of four data types. The ARSC input/output functions will supply any conversions necessary to make the application's sample data format compatible that the format expected by the soundcard device.

### **See Also**

ar\_set\_fmt

***ar\_get\_cardinfo***

Get soundcard information

```
long ar_get_cardinfo(
    long dev,
    CARDINFO *ci
);
```

**Parameters**

**dev**            Device identifier.

**ci**            Pointer to CARDINFO structure:

```
typedef struct {
    char    name[MAX_CT_NAME]; /* card name          */
    int     bits;              /* precision          */
    int     left;              /* MSB shift          */
    int     nbps;              /* bytes per sample   */
    int     ncad;              /* # A/D channels     */
    int     ncda;              /* # D/A channels     */
    int     gdsr;              /* good sampling rates */
    double  ad_vfs[MAXNCH];    /* A/D volts full scale */
    double  da_vfs[MAXNCH];    /* D/A volts full scale */
} CARDINFO;
```

**Return Value**

The sequential number of the card type with the soundcard features that are returned in the CARDINFO structure.

**Remarks**

This soundcard information is based on hardcoded defaults for a few soundcards, which may be overridden or extended by the registry (on Windows) or the configuration file /usr/local/etc/arscr (on Linux).

## ***ar\_get\_gdsr***

Get device's good sampling rate

```
long ar_get_gdsr(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

Bitwise list of good sampling rates or zero if device identifier is out of range or device has not yet been opened. Bits 0 – 26 in the returned value correspond to these sampling rates: 4000, 5512, 6000, 8000, 8269, 10000, 11025, 12000, 16000, 16538, 20000, 22050, 24000, 25000, 32000, 33075, 44100, 48000, 50000, 64000, 88200, 96000, 100000, 128000, 176400, 192000, 200000.

### **Remarks**

Device should be opened prior to calling this function, but need not be still open when this function is called.

### **See Also**

ar\_get\_rate

## ***ar\_get\_rate***

Get sampling rate

```
double ar_get_rate(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

The sampling rate currently being used by the specified soundcard device.

### **Remarks**

The soundcard device's sampling rate will differ from the rate requested by the application when the device was opened. If the soundcard device can't provide the requested rate, then it will be set to the closest rate possible.

### **See Also**

ar\_io\_adjust\_rate, ar\_io\_set\_rate

## ***ar\_get\_sfs***

Set sample-full-scale value for floating-point transfer buffers

```
void ar_get_sfs(  
    long dev,  
    double *i_sfs,  
    double *o_sfs  
);
```

### **Parameters**

dev	Device identifier.
i_vfs	input sample-full-scale.
o_vfs	output sample-full-scale.

### **Return Value**

None.

### **Remarks**

When ARSC inputs samples to a floating-point buffer or outputs samples from a floating-point buffer the floating-point value that corresponds to full-scale voltage at the convertor is i\_vfs or o\_vfs, respectively. The same sample-full-scale parameter is used for both single-precision and double-precision floating-point buffers. The default value of sample-full-scale value is 1 for both i\_vfs and o\_vfs.

### **See Also**

ar\_set\_sfs, ar\_set\_fmt, ar\_set\_xfer

## ***ar\_get\_vfs***

Get ADC and DAC volts-full-scale

```
void ar_get_vfs(  
    long dev,  
    double *da_vfs,  
    double *ad_vfs  
);
```

### **Parameters**

dev            Device identifier.  
da\_vfs        DAC volts-full-scale array (size = 8).  
ad\_vfs        ADC volts-full-scale array (size = 8).

### **Return Value**

None.

### **Remarks**

If the name of the soundcard is one of a few recognized by the ARSC library (i.e., CardDeluxe, Gina3G, Indigo, or Layla3G), then default values for the full-scale voltage of the A/D and D/A converters will be set automatically. Otherwise, these values can be set by the `ar_set_vfs` function. The VFS values do not affect the operation of any ARSC input/output function. They are provided only for informational purposes.

### **See Also**

`ar_set_vfs`

## ***ar\_io\_close***

Close device

```
long ar_io_close(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

Error code.

### **Remarks**

Closes the specified soundcard device.

### **See Also**

`ar_io_open`, `ar_close_all`

## ***ar\_io\_cur\_seg***

Get current unwrapped segment

```
long ar_io_cur_seg(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

Returns the number of the input/output segment which has most recently been completed. This unwrapped segment number is not reset to 0 at the beginning of each new sweep. Instead it continues to increment with each successive sweep. The number of the segment within the current sweep can be recovered by taking the unwrapped segment number modulo the number of sweeps per segment.

### **Remarks**

This function can be used to synchronize visual cues to the user with input/output segment boundaries. It is important that the application poll the `ar_io_cur_seg` function often because it also performs important input/output processing functions. This function needs to be called at least once during each input/output segment to insure the successive segment are queued for the soundcard device.

Under Windows, one way to implement calling of the `ar_io_cur_seg` function with sufficient frequency is to respond to WM\_ARSC messages.

### **See Also**

`ar_set_wind`



***ar\_io\_open***

Open device for i/o

```
long ar_io_open(
    long dev,
    double rate,
    long in_chan,
    long out_chan
);
```

**Parameters**

dev	Device identifier. This can be an identifier provided by the <code>ar_find_dev</code> or <code>ar_find_dev</code> name functions. It's value can also be zero to specify use of the system default soundcard.
rate	Desired sampling rate in samples/second. If the soundcard can't provide the requested rate, the closes possible sampling rate will be used. The actual sampling rate can be obtained by calling <code>ar_get_rate</code> .
in_chan	Desired number of input channels. This indicates the number of input channels that will be expected in the application's input data buffers that are specified to <code>ar_io_prepare</code> . If the requested number of channels is more than the soundcard can provide, any excess channel data will be filled with zero. The value of <code>in_chan</code> can be set to zero if no input is desired.
out_chan	Desired number of output channels. This indicates the number of output channels for which data will be provided in the application's output data buffers that are specified to <code>ar_io_prepare</code> . If the requested number of channels is more than the soundcard can provide, any excess channel data will be ignored. The value of <code>out_chan</code> can be set to zero if no input is desired.

**Return Value**

Error code.

**Remarks**

If calling `ar_io_open` is successful, then call `ar_io_prepare` to specify the input/output buffers and `ar_io_start` to initiate the input/output process. When both input and output is specified, these processes will be synchronized, if possible.

**See Also**

`ar_io_close`, `ar_io_prepare`, `ar_out_open`

***ar\_io\_prep***

Prepare device for i/o

```
long ar_io_prepare(
    long dev,
    void *in_data[],
    void *out_data[],
    long size[],
    long nseg,
    long tseg
);
```

**Parameters**

<code>dev</code>	Device identifier.
<code>in_data</code>	List of pointers to the input data buffers for each segment. The value <code>in_data</code> may be NULL to indicate no input. Or, the value of any pointer in the list may be NULL to indicate no input for a particular segment.
<code>out_data</code>	List of pointers to the output data buffers for each segment. The value <code>out_data</code> may be NULL to indicate no output. Or, the value of any pointer in the list may be NULL to indicate no output for a particular segment.
<code>size</code>	List of sizes indicating the length of each segment.
<code>nseg</code>	The number of input/output segments in each sweep.
<code>tseg</code>	The total number of segments, which equals the total number of sweeps times <code>nseg</code> . Set <code>tseg</code> to zero for continuous output.

**Return Value**

Error code.

**Remarks**

The data type of the input/output buffers can be *long* or *short* and the channels can be *interleaved* or *non-interleaved* depending on the format specified through the `ar_set_fmt` function. If the data buffers are *non-interleaved*, then the `in_data` and `out_data` lists must provide pointers to each channel for each segment. Whenever the number of channels is greater than one, either the data or the pointers are interleaved. The input/output process is not initiated until `ar_io_start` is called. Output buffers are filled by this function, so call `ar_set_xfer` before this function.

**See Also**

`ar_io_start`, `ar_io_prepare`, `ar_out_prepare`, `ar_set_xfer`

***ar\_io\_prepare***

Prepare device for i/o

```
long ar_io_prepare(
    long dev,
    void *in_data[],
    void *out_data[],
    long size[],
    long nseg,
    long nswp
);
```

**Parameters**

dev	Device identifier.
in_data	List of pointers to the input data buffers for each segment. The value in_data may be NULL to indicate no input. Or, the value of any pointer in the list may be NULL to indicate no input for a particular segment.
out_data	List of pointers to the output data buffers for each segment. The value out_data may be NULL to indicate no output. Or, the value of any pointer in the list may be NULL to indicate no output for a particular segment.
size	List of sizes indicating the length of each segment.
nseg	The number of input/output segments in each sweep.
nswp	The total number of input/output sweeps. Set nswp to zero for continuous output.

**Return Value**

Error code.

**Remarks**

The data type of the input/output buffers can be *long* or *short* and the channels can be *interleaved* or *non-interleaved* depending on the format specified through the `ar_set_fmt` function. If the data buffers are *non-interleaved*, then the `in_data` and `out_data` lists must provide pointers to each channel for each segment. Whenever the number of channels is greater than one, either the data or the pointers are interleaved. The input/output process is not initiated until `ar_io_start` is called. Output buffers are filled by this function, so call `ar_set_xfer` before this function.

**See Also**

`ar_io_start`, `ar_io_prepare`, `ar_out_prepare`, `ar_set_xfer`

## ***ar\_io\_start***

Start i/o

```
long ar_io_start(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

Error code.

### **Remarks**

The application must call `ar_io_prepare` or `ar_io_open` before calling `ar_io_start`.

### **See Also**

`ar_io_prepare`, `ar_out_prepare`

## ***ar\_io\_stop***

Stop i/o

```
long ar_io_stop(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Value**

Error code.

### **Remarks**

Use this function to close all stop the input/output on a specified device immediately. Because the soundcard input/output also stops when the specified number of sweeps has completed or if the device is closed while still performing input/output, it is never necessary to call `ar_io_stop`. On the other hand, it is not an error to call `ar_io_stop` when input/output has already been stopped, provided that device is open. Calling `ar_io_stop` also *unprepares* the input/output process, so it can not be restarted without again calling `ar_io_prepare` or `ar_out_prepare`.

### **See Also**

`ar_io_prepare`, `ar_io_start`, `ar_out_prepare`

## ***ar\_num\_devs***

Get number of devices.

```
long ar_num_devs(  
    );
```

### **Parameters**

None.

### **Return Value**

Device count

### **Remarks**

Returns the total number of soundcard devices currently present.

### **See Also**

`ar_find_dev`

## ***ar\_out\_open***

Open device for output

```
long ar_out_open(  
    long dev,  
    double rate,  
    long out_chan  
);
```

### **Parameters**

dev	Device identifier. This can be an identifier provided by the <code>ar_find_dev</code> or <code>ar_find_dev</code> name functions. It's value can also be zero to specify use of the system default soundcard.
rate	Desired sampling rate in samples/second. If the soundcard can't provide the requested rate, the closes possible sampling rate will be used. The actual sampling rate can be obtained by calling <code>ar_get_rate</code> .
out_chan	Desired number of output channels. This indicates the number of output channels for which data will be provided in the application's output data buffers that are specified to <code>ar_io_prepare</code> . If the requested number of channels is more than the soundcard can provide, any excess channel data will be ignored.

### **Return Value**

Error code.

### **Remarks**

Calling `ar_out_open` is the same as calling `ar_io_open` with the number of input channels set to zero. It simplifies programming when no soundcard input is needed.

### **See Also**

`ar_io_open`

## ***ar\_out\_prepare***

Prepare device for output

```
long ar_out_prepare(  
    long dev,  
    void *out_data[],  
    long size[],  
    long nseg,  
    long nswp  
);
```

### **Parameters**

dev	Device identifier.
out_data	List of pointers to the output data buffers for each segment. The value of any pointer in the list may be NULL to indicate no output for a particular segment.
size	List of sizes indicating the length of each segment.
nseg	The number of input/output segments in each sweep.
nswp	The total number of input/output sweeps. The value of nswps can be zero to indicate continuous output.

### **Return Value**

Error code.

### **Remarks**

Calling `ar_out_prepare` is the same as calling `ar_io_prepare` with a NULL value for the list of input data buffers. It simplifies programming when no soundcard input is needed.

### **See Also**

`ar_io_prepare`



## ***ar\_out\_seg\_fill***

Refill output segments

```
long ar_out_seg_fill(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Values**

Error code.

### **Remarks**

This function can be used to change the contents of output buffers that have already been queued for output. It is only useful when the number of segments is greater than 2 and an output callback function has been specified through `ar_set_xfer`. Calling `ar_out_seg_fill` causes the applications output callback function to be called again for every segment except the segment currently being processed by the soundcard device and the next segment, in case a segment transition occurs during before the callback function has finished filling the output buffer. The `ar_out_seg_fill` function may be useful in streaming applications with many small output segments in order to reduce the delay in transitioning to a new type of output.

### **See Also**

`ar_set_xfer`

***ar\_set\_fmt***

Specify app's data format

```
long ar_set_fmt(
    long dev,
    long *fmt
);
```

**Parameters**

dev	Device identifier.
fmt	Array of parameters to describe the format of the application's sample data. The length of this array is 2.
fmt[0]	<i>data type</i> — one of the following values: ARSC_DATA_I2 – 16-bit integer ARSC_DATA_I4 – 32-bit integer ARSC_DATA_F4 – 32-bit floating-point ARSC_DATA_F8 – 64-bit floating-point
fmt[1]	<i>channel interleave</i> — It's value can be 0 to indicate <i>non-interleaved</i> sample data or 1 to indicate <i>interleaved</i> sample data.

**Return Value**

Error code.

**Remarks**

Sets the format of the soundcard device's sample data, which can be *interleaved* or *non-interleaved* and one of four data types. The ARSC input/output functions will supply any conversions necessary to make the application's sample data format compatible that the format expected by the soundcard device.

**See Also**

ar\_get\_fmt

## ***ar\_set\_latency***

Set ASIO device latency offset and return current latency setting

```
long ar_set_latency(  
    long dev,  
    long nsmp  
);
```

### **Parameters**

dev	Device identifier.
nsmp	Number of samples added to ASIO latency. Number may be negative to decrease latency. Special value nsmp=ASIO_GET_LATENCY or nsmp=9999 bypasses setting just returns current value.

### **Return Value**

Number of samples of current latency offset

### **Remarks**

Some ASIO device drivers center the loopback impulse response at zero latency, which is less than the WDM loopback latency. The ar\_set\_latency function allows the ASIO interface to have the same latency as the WDM interface.

## ***ar\_set\_sfs***

Set sample-full-scale value for floating-point transfer buffers

```
void ar_set_sfs(  
    long dev,  
    double *i_sfs,  
    double *o_sfs  
);
```

### **Parameters**

dev	Device identifier.
dev	Device identifier.
i_sfs	input sample-full-scale.
o_sfs	output sample-full-scale.

### **Return Value**

None.

### **Remarks**

When ARSC inputs samples to a floating-point buffer or outputs samples from a floating-point buffer the floating-point value that corresponds to full-scale voltage at the convertor is i\_vfs or o\_vfs, respectively. The same sample-full-scale parameter is used for both single-precision and double-precision floating-point buffers. The default value of sample-full-scale value is 1 for both i\_vfs and o\_vfs.

### **See Also**

ar\_get\_sfs, ar\_set\_fmt, ar\_set\_xfer

## ***ar\_set\_vfs***

Set ADC and DAC volts-full-scale

```
void ar_set_vfs(  
    long dev,  
    double *da_vfs,  
    double *ad_vfs  
);
```

### **Parameters**

dev            Device identifier.  
da\_vfs        DAC volts-full-scale array (size = 8).  
ad\_vfs        ADC volts-full-scale array (size = 8).

### **Return Value**

None.

### **Remarks**

If the name of the soundcard is one of a few recognized by the ARSC library (i.e., CardDeluxe, Gina24, Indigo, or WaveTerminal), then default values for the full-scale voltage of the A/D and D/A converters will be set automatically. Otherwise, these values can be set by the application. The VFS values do not affect the operation of any ARSC input/output function. They are provided only for informational purposes.

### **See Also**

ar\_get\_vfs

***ar\_set\_xfer***

Specify app data transfer functions

```
long ar_set_xfer(  
    long dev,  
    void (*in_xfer)(long),  
    void (*out_xfer)(long)  
);
```

**Parameters**

dev	Device identifier.
in_xfer	Input callback function which will be called for each input segment to allow the application to retrieve data from the input buffer. The unwrapped segment number is provided as an argument to this function. The value of in_xfer can be NULL if no input callback function is desired.
out_xfer	Output callback function which will be called for each output segment to allow the application to fill the output buffer with data. The unwrapped segment number is provided as an argument to this function. The value of out_xfer can be NULL if no output callback function is desired.

**Return Value**

Error code.

**Remarks**

The function allows streaming input and/or output to be implemented. The input/output process can be made continuous by setting the number of sweeps to zero. The segment number within the current sweep is the unwrapped segment number provided to the callback function modulo the number of segments per sweep. To reduce the delay in switching to new output when many output segments have been queued, the ar\_out\_seg\_fill function can be called to refill output data buffers with new data. Call this function prior to calling ar\_io\_prep or ar\_io\_prepare to allow output buffers to be filled.

**See Also**

ar\_io\_prep, ar\_io\_prepare, ar\_out\_seg\_fill

## ***ar\_wind***

Specify a window receives messages

```
void ar_wind(  
    long hwind  
);
```

### **Parameters**

hwind            Handle to window that is to receive WM\_ARSC messages.

### **Return Values**

None.

### **Remarks**

The specified window will receive WM\_ARSC messages whenever an input or output segment has completed. The wParam associated with this message will be 1 for an input segment and 2 for an output segment. The lParam will have the value of the device identifier. The ar\_io\_cur\_seg function can be called in response to this message to insure that it is called at least once per segment. Visual cue might also be provided to the user at this time that indicate a segment transition.

### ***ar\_version***

Get ARSC version.

```
char *ar_version(  
    );
```

#### **Parameters**

None.

#### **Return Values**

ARSC version string.

#### **Remarks**

The returned value is a string containing the ARSC version number and date, similar to the line below.

```
ARSC version 0.01, 24-Apr-05
```



## ***ar\_xruns***

Get combined number of underruns and overruns.

```
void ar_xruns(  
    long dev  
);
```

### **Parameters**

dev            Device identifier.

### **Return Values**

- 1     device identifier out of range
- 2     device not open
- >0    number of xruns since device was opened

### **Remarks**

Because the ARSC functions handling input and output simultaneously, underruns and overruns are also occur simultaneously. These events are counted together as xruns.

## ARSC DLL

The dynamic library for Windows, arsc.dll, contains the same static library functions described in the previous section; however, the Windows DLL calling convention is used in arsc.dll instead of the standard C calling convention. For example, the function declared in the static library as “long ar\_find\_dev( ),” is declared in the DLL as “WIN32DLL\_API long STDCALL ar\_find\_dev( ).”

The DLL also contains additional functions to help support its use with languages other than C.

### ***ar\_wait\_seg***

Wait until current segment has finished.

```
WIN32DLL_API long STDCALL
ar_io_wait_seg(
    long dev
);
```

Ordinarily the calling program polls to see when a segment has finished. This function provides a different methodology where the looping is done internal to the DLL, then control is released to the calling program only after the segment is finished.

Because this function resides within a windows DLL, an escape mechanism has been provided. Specifically, hit the ESC key any time the sound engine is running to stop.

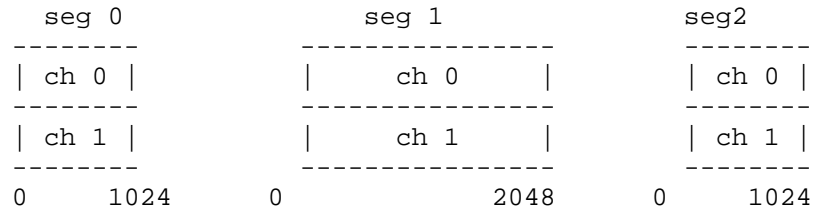
### ***ar\_io\_prepare\_vb***

Wrapper to help Visual Basic perform I/O with ARSC.

```
WIN32DLL_API long STDCALL
ar_io_prepare_vb (
    long dev,                // device identifier
    void *in_data,           // pointer to input data
    void *out_data,          // pointer to output data
    long size[],              // array of segment sizes
    long nseg,               // number of segments
    long nswp                // number of sweeps
);                           // returns error code
```

This wrapper assumes a pointer to a single block of memory then re-defines it to the ar\_io\_prepare() arguments. This has been tested and works well with VB.NET, MatLab, and even C#.

A visual interpretation may help to clarify how this works. Consider a scenario of 3 segments (0,1,2) and 2 channels (0,1) and that we are considering output only. The first segment is 1024; the second is 2048; the third is 1024 samples.

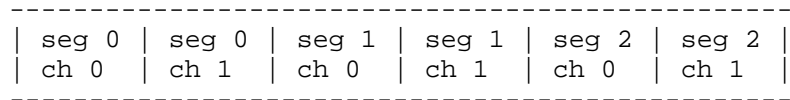


**\*\* FIGURE 1 \*\***

Figure 1 shows the segments as separate entities. The underlying C library only needs pointers, so this layout shows the segments in a logical train of left to right.

Channels are interleaved in the underlying code. It is useful to think of channels as being layers of the segment.

This next figure shows how this function takes the data as one contiguous block.



**\*\* FIGURE 2 \*\***

The sizes array is populated with segment lengths. This is independent of the number of channels. That is, if you have 1 or 1000 channels, the segment length must be the same.

For this example, sizes[3] = { 1024, 2048, 1024 }

## ***ar\_out\_prepare\_vb***

Wrapper to ar\_io\_prepare\_vb() to skip input channels.

```
WIN32DLL_API long STDCALL
ar_out_prepare_vb (
    long dev,                // device identifier
    void *out_data,          // pointer to output data
    long size[],              // array of segment sizes
    long nseg,               // number of segments
    long nswp                // number of sweeps
);                          // returns error code
```

### ***ar\_set\_xfer\_stdcall***

Special function for stdcall transfer (xfer) functions.

```
WIN32DLL_API long STDCALL
ar_set_xfer_stdcall (
    long dev,
    void (STDCALL *in_xfer)(long),
    void (STDCALL *out_xfer)(long)
);
```

### ***ar\_err\_msg\_matlab***

Help MATLAB access error message.

```
WIN32DLL_API char * STDCALL
ar_err_msg_matlab (
    long code           // error code
);
```

### ***ar\_dev\_name\_matlab***

Help MATLAB access device name.

```
WIN32DLL_API char * STDCALL
ar_dev_name_matlab (
    long dev           // device identifier
);
```

### ***ar\_version\_DLL***

Returns the DLL version and the ARSC version.

```
WIN32DLL_API long STDCALL
ar_version_dll (
    char *astrVersionInfo, // pointer to string
    long len               // Length of string
);
```

### ***ar\_fill\_tone***

Fill a memory block with a tone of frequency.

```
WIN32DLL_API long STDCALL
```

## ARSC API

```
ar_fill_tone (  
    long *ptrBlock,                // Output  
    long aintBlockSize,           // Memory block size (bytes)  
    double adblDesiredFrequency,   // Frequency (Hz)  
    double adblSampleRate,         // e.g. 44100.  
    double adblAmplitude           // sine wave amplitude  
);                                // returns error code
```

## SIO API

The SIO functions provide an alternative synchronous-I/O API that uses floating-point buffers. The SIO functions are implemented on top of the ARSC functions. The dynamic library file `sio.dll` functions contains the entire set of ARSC functions, in addition to the SIO functions listed below. However, the ARSC\_DLL helper functions are not included.

### ***sio\_open***

Initializes soundcard and internal SIO variables.

```
int sio_open(  
    );           // returns non-zero if successful
```

### ***sio\_close***

Terminate I/O and free any allocated resources.

```
void sio_close(  
    );
```

### ***sio\_get\_nioch***

Returns the number of I/O channels available on the soundcard.

```
int sio_get_nioch(  
    int *ni,      // Number of input channels returned here  
    int *no       // Number of output channels returned here  
    );           // Device open?
```

### ***sio\_get\_device***

Obtain short description of I/O device.

```
int sio_get_device(  
    char *s       // Device description returned here  
    );           // Device open?
```

### ***sio\_set\_device***

Select I/O device.

```
int sio_set_device(  
    int n        // card number  
);               // returns card number
```

### ***sio\_get\_info***

Obtain one-line of information about I/O device.

```
int sio_get_info(  
    char *s      // Device information returned here  
);               // Device open?
```

### ***sio\_get\_vfs***

Obtain full-scale voltage of ADCs and DACs.

```
int sio_get_vfs(  
    double *ad_vfs, // ADC volts-full-scale returned  
    double *da_vfs  // DAC volts-full-scale returned  
);                 // Device open?
```

### ***sio\_set\_size***

Specify size of buffers and gap between buffers.

```
void sio_set_size(  
    int ns,      // size of I/O buffers (samples)  
    int ng,      // size of gap between I/O buffers (samples)  
    double pg    // size of gap before I/O (seconds)  
);
```

### ***sio\_set\_output***

Specify output buffers.

```
void sio_set_output(  
    int nch,      // number of output channels to be used  
    int bpc,      // number of buffers per output channel  
    int *inc,     // increments for each output buffer  
    float **bpt   // pointers to each output buffer  
);
```

### ***sio\_set\_input***

Specify input buffers.

```
void sio_set_input(  
    int nch,          // number of input channels to be used  
    int bpc,          // number of buffers per input channel  
    int *inc,         // increments for each input buffer  
    float **bpt       // pointers to each input buffer  
);
```

### ***sio\_set\_average***

Specify averaging.

```
void sio_set_average(  
    float **apt,      // pointers to accumulate buffers  
    int *acp          // number of sweeps in each buffer  
);
```

### ***sio\_set\_rate***

Specify sampling rate.

```
double sio_set_rate(  
    double r          // desired rate (samples/second)  
);                  // returns actual rate (samples/second)
```

### ***sio\_set\_att\_in***

Specify attenuation on input all channels.

```
double sio_set_att_in(  
    double a          // desired input attenuation (dB)  
);                  // returns actual input attenuation (dB)
```

### ***sio\_set\_att\_out***

Specify attenuation on output all channels.

```
double sio_set_att_out(  
    double a          // desired output attenuation (dB)  
);                  // returns actual output attenuation (dB)
```



***sio\_set\_vfs***

Specify full-scale voltage on ADCs and DACs

```
void sio_set_vfs(
    double *ad_vfs, // volts-full-scale for each ADC
    double *da_vfs  // volts-full-scale for each DAC
);
```

***sio\_set\_escape***

Specify callback function for early I/O termination. The callback function `esc` requests an escape by returning a nonzero value.

```
void sio_set_escape(
    int (*esc)() // escape callback function
);
```

***sio\_io***

Perform input/output/averaging.

```
void sio_io(
    int nskip, // samples to skip before averaging
    int nswps, // maximum total sweeps
    int navgs, // maximum sweeps averaged
    int nrejs  // maximum rejected
);
```

***sio\_io\_chk***

Perform input/output/averaging with callback. The function `resp_check` is called at the completion of each sweep with a pointer to an “escape” flag `esc`. The `resp_check` function assesses the status of the average and requests termination by setting `esc` to a nonzero value.

```
void sio_io_chk(
    void (*resp_check)(int *esc) // sweep callback function
);
```

***sio\_set\_latency***

Set ASIO internal latency.

```
int sio_set_latency(
    int nsmp
);
```

***sio\_get\_cardinfo***

Get card info.

```
CARDINFO sio_get_cardinfo(
    int ct          // card type
);
```

***sio\_set\_cardinfo***

Set card info.

```
void sio_set_cardinfo(
    CARDINFO ci, // card info
    int ct       // card type
);
```

The CARDINFO structure:

```
typedef struct {
    char    name[MAX_CT_NAME]; /* card name          */
    int     bits;              /* precision          */
    int     left;              /* MSB shift          */
    int     nbps;              /* bytes per sample   */
    int     ncad;              /* # A/D channels     */
    int     ncda;              /* # D/A channels     */
    int     gdsr;              /* good sampling rates */
    double  ad_vfs[MAXNCH];    /* A/D volts full scale */
    double  da_vfs[MAXNCH];    /* D/A volts full scale */
} CARDINFO;
```

## ARSC Pre-defined Constants

ARSC_MSGLEN	80
ARSC_NAMLEN	40
ARSC_PREF_NONE	0
ARSC_PREF_SYNC	1
ARSC_PREF_ASIO	2
ARSC_PREF_ALSA	3
ARSC_PREF_WIND	4
ARSC_DATA_UNKNOWN	0
ARSC_DATA_U1	1
ARSC_DATA_I2	2
ARSC_DATA_P3	3
ARSC_DATA_I4	4
ARSC_DATA_X3	5
ARSC_DATA_F4	6
ARSC_DATA_M1	7
ARSC_DATA_F8	8
WM_ARSC	( WM_USER+555 )

## ARSC Error Codes

These error codes may be returned by one or more of the ARSC functions. The corresponding error message can also be obtained by using the `ar_error_message` function.

```
0   no errors
1   device identifier out of range
2   device not open
101 io_open - MMSYSERR = unspecified error
102 io_open - MMSYSERR = device ID out of range
103 io_open - MMSYSERR = driver enable failed
104 io_open - MMSYSERR = device already allocated
105 io_open - MMSYSERR = unknown error
106 io_open - MMSYSERR = no device driver present
107 io_open - MMSYSERR = memory allocation error
108 io_open - WAVERR = unsupported wave format
109 io_open - WIND unknown device identifier
110 io_open - WIND open error
120 io_open - ASIO open error
130 io_open - ALSA open error
201 io_prepare - device not opened
202 io_prepare - unsupported data format conversion
203 io_prepare - size matters
204 io_prepare - too many segments per sweep
205 io_prepare - too many sweeps
206 io_prepare - failed low level prepare
301 io_start - device not prepared
401 io_set_fmt - NULL format
402 io_set_fmt - unsupported data format
403 io_set_fmt - unsupported interleave
404 io_set_fmt - unsupported format conversion
```