



looping the unloopable for over a decade

Infinity

DSP SAMPLE LOOPING TOOLS

Owner's Manual

©2002 Antares Audio Technologies. All rights reserved.

All trademarks are the property of their respective owners.

Antares Audio Technologies
231 Technology Circle
Scotts Valley, California 95066 USA

voice: (831) 461 7800
fax: (831) 461 7801
email: info@antarestech.com
web: www.antarestech.com

Rev 1.0rgb v3.18 11/2002

THE OBLIGATORY LEGAL MUMBO-JUMBO

The Antares Infinity software and this User's Manual are protected by copyright law. Making copies, adaptations, or derivative works without the prior written authorization of Antares Audio Technologies, is prohibited by law and constitutes a punishable violation of the law.

Antares Audio Technologies retains all ownership rights to the Infinity software and its documentation. Use of Infinity is limited by the following license agreement.

Please carefully read all the terms and conditions of this license agreement. At the time of installation of the Infinity software you will be presented with a copy of the agreement and asked whether or not you agree to it. Continuing with the installation process beyond that point constitutes such agreement.

INFINITY LICENSE AGREEMENT

Antares Audio Technologies grants you a nontransferable, nonexclusive license to use Infinity under the terms and conditions stated in this agreement. Use of Infinity indicates your agreement to the following terms and conditions.

You may:

1. Use Infinity on only one computer.

You may not:

1. Make copies of Infinity or of the user manual in whole or in part except as expressly provided for in this agreement. Your right to copy Infinity and the user manual is limited by copyright law. Making copies, verbal or media translations, adaptations, derivative works, or telecommunication data transmission of Infinity without prior written authorization of Antares, is prohibited by law and constitutes a punishable violation of the law.
2. Make alteration or modifications to Infinity (or any copy) or disassemble or de-compile Infinity (or any copy), or attempt to discover the source code of Infinity.
3. Sublicense, lease, lend, rent, or grant other rights in all or any portion of Infinity (or any copy) to others.

TERM OF THE AGREEMENT

This agreement is effective until terminated by you or Antares. You may terminate the agreement at any time by notifying Antares and destroying all copies of the manual, and erasing Infinity from all machine-readable media, whether on-line or on archival copies.

In the event of breach of any of the terms of this agreement, you shall pay the attorney's fees of Antares that are reasonably necessary to enforce the agreement, plus resulting damages.

LIMITED WARRANTY AND DISCLAIMER

INFINITY AND ACCOMPANYING MATERIALS ARE PROVIDED "AS IS" WITHOUT WARRANTY OF ANY KIND, EITHER EXPRESS OR IMPLIED, INCLUDING, BUT NOT LIMITED TO, THE IMPLIED WARRANTIES OF MERCHANTABILITY AND FITNESS FOR A PARTICULAR PURPOSE.

Antares Audio Technologies does not warrant that the functions contained in the program will meet your requirements. The entire risk as to the use, quality, and performance of Infinity is with you.

SOME JURISDICTIONS DO NOT ALLOW LIMITATIONS ON HOW LONG AN IMPLIED WARRANTY LASTS, SO THE ABOVE LIMITATION MAY NOT APPLY TO YOU. THIS WARRANTY GIVES YOU SPECIFIC LEGAL RIGHTS. YOU MAY ALSO HAVE OTHER RIGHTS WHICH VARY FROM JURISDICTION TO JURISDICTION.

LIMITATION OF LIABILITY

IN NO EVENT WILL ANTARES BE LIABLE FOR ANY DAMAGES, INCLUDING LOSS OF DATA, LOST PROFITS OR OTHER SPECIAL, INCIDENTAL, CONSEQUENTIAL OR INDIRECT DAMAGES ARISING FROM THE USE OF Infinity OR ACCOMPANYING MATERIALS. THIS LIMITATION WILL APPLY EVEN IF ANTARES OR ITS AUTHORIZED AGENT HAS BEEN ADVISED OF THE POSSIBILITY OF SUCH DAMAGE. YOU ACKNOWLEDGE THAT THE LICENSE FEE REFLECTS THIS ALLOCATION OF RISK. SOME JURISDICTIONS DO NOT ALLOW LIMITATION OR EXCLUSION OF LIABILITY FOR INCIDENTAL OR CONSEQUENTIAL DAMAGES, SO THE ABOVE LIMITATION MAY NOT APPLY TO YOU.

Whew! Now that that's over, let's get on to the good stuff.

Contents

	Welcome	
	Infinity Technical Support	vii
	Installation and Authorization	viii
Chapter 1	Getting Started	
	Infinity Features	1
Chapter 2	Introducing Infinity	
	The Problem with Looping	4
	Crossfade Loops	5
	The Soundfile Window	7
	The Palette: Mode Icons	10
	Status Display + Action Icons	11
	Get Info	12
	Envelope Files	13
Chapter 3	The Basics of Infinity	
	Basic Commands	15
	Viewing the Soundfile	16
	Changing the Display Resolution	17
	Listening to the Soundfile	20
	Editing the Soundfile	21
	Editing Functions	24
	Saving the Soundfile	34
	Customizing Your Environment	36

Chapter 4	Looping Tools	
	Managing Loop Points	39
	The Crossfade Looper	41
	The Freeze Looper	44
	The Rotated Sums Looper	48
	The SPR Looper	51
	The Synthesis Looper	55
	How To Loop a Sound	65
	Appendix A: Keyboard Shortcuts	67
	Index	70

Welcome!



At Antares Audio Technologies offer both our thanks and congratulations on your decision to purchase Infinity, the world's best sample-looping toolkit.

Before proceeding, please register and authorize your copy of Infinity. (You can skip ahead a few pages to the Authorization and Installation instructions. We'll wait.) Also, if you plan to discard that lovely Infinity jewel case, write down the serial number that appears on the back for future reference.

As a registered Infinity owner, you'll receive notification of any software upgrades, technical support, and advance announcements of upcoming products. But we can't send you stuff unless we know who and where you are. So please, register. Note that we do not share this information with people who will pester you.

At Antares, we are committed to excellence in quality, customer service, and technological innovation. With your purchase of Infinity, you have created a relationship with Antares which we hope will be long and gratifying. Let us know what you think. You can count on us to listen.

Again, thanks.

The Whole Antares Crew

Technical Support



In the unlikely event that you experience a problem using Infinity, try the following:

1. Make another quick scan through this manual. Who knows? You may have stumbled onto some feature that you didn't notice the first time through.
2. Check our web page for tips, techniques, or any late-breaking information: <http://www.antarestech.com>
3. Call your local Antares dealer.
4. Email our tech support department by pointing your web browser to: <http://www.antarestech.com/support/etech.shtml> and filling in the form there.
5. Call us at (831) 461-7814 Monday through Friday between 10am and 5pm USA Pacific Standard Time (or leave a message any time and we'll get back to you).

For options 3, 4, or 5, please be prepared to provide the Registration Code of your copy of Infinity.

Installation and Authorization

INSTALLING INFINITY

To install Infinity, simply copy the Infinity installer to your hard drive, double-click the installer and follow the resulting directions. If you have purchased Infinity as a download, decompress the downloaded file and double-click the installer. In either case, be sure to read the Read Me file for any late breaking news.

AUTHORIZING INFINITY

To run Infinity, the computer on which it runs needs to be authorized using a “challenge/response” authorization technique. The first time you launch this software, you will be presented with a string of words (the Challenge) and will be asked to enter another string of words (the Response).

***Note:** When initially installed, this software will run for ten days without authorization.*

So even if you can’t authorize it right away, you can still use your software. (During this period, click the “Try It” button whenever you are presented with the Trial Period screen at launch.) But don’t procrastinate too long. After those ten days are up, you can’t launch Infinity until you enter the correct Response.

CHALLENGE/RESPONSE AUTHORIZATION

You can obtain a Response via the Antares web site, email, or fax. Web authorization, an automated process, is available 24 hours a day. Email and fax authorization involve the participation of actual live people and, depending on when you send your challenge, may take from one to three days to provide a response.

For web authorization, follow these directions:

- Launch Infinity. To access Challenge/Response authorization, press “Next” at the Trial Period screen to access the Challenge screen.

The Challenge screen displays the Challenge string. To receive the appropriate Response you will need to provide us with both the Challenge string and the registration code that was included on the yellow card in your software package (if you bought a retail package) or supplied with your download.

- To obtain your Response, have your registration code and Challenge string at hand. (If you will be accessing the web from the same computer that will be running your software, you can press “Copy Challenge” at the Challenge screen and then simply paste the Challenge into the appropriate field on the web page below.)
- Point your web browser to:
<https://transactions.antarestech.com/>
and simply follow the directions. (Be extremely careful to enter the registration code and Challenge string accurately.)

If, for some reason, you do not have access to the web but do have email, copy and paste your Challenge string and registration code into an email along with your name and the product name (e.g., Infinity for MAS), and send it to:

register@antarestech.com

You will receive your Response string by return email.

If you do not have access to the web or email, fax your Challenge string along with your registration code to us at (831) 461-7801. Be sure to include your fax number. We will fax the Response string back to you at that number.

Note: *You may optimize (defragment) the hard disk containing the Challenge/Response authorization, or even reinstall your system, without losing the authorization. However, if you reformat the hard drive or if it fails, contact Antares for the re-authorization procedure (you are a registered user, right?).*

Chapter 1: Getting Started



Imagine being able to take a raw sample, and then, with just a few mouse clicks, create a loop so perfect that you can't tell when it started or where it repeats. Imagine doing that with sounds that are impossible to loop smoothly using your current sample editor, like string orchestra, choirs, fat analog synths, chorused guitar, and so on. Infinity lets you do that, quickly and easily.

Infinity is a powerful DSP which uses unique DSP technologies to automate and improve the looping process (the part of sample editing that takes the most time and skill).

Some of Infinity's special features are:

REAL TIME LOOP ADJUST

Infinity makes fine tuning the loop easy by letting you move the loop points anywhere in the sound file during real time playback.

SYNTHESIS LOOPER

The Synthesis Looper tool calculates the energy at each fundamental and harmonic frequency and allows you to specify the width of the sidelobes the resulting looped sound will have. In this way, you could take the lumpy sound of a highly out-of-tune ensemble and change it to the smoother sound of a higher quality ensemble, making the variations in the loop repetition less noticeable.

SPR LOOPER

This tool automatically randomizes the spectral components of the sound inside the loop points to create a virtually seamless result in sounds that have irregular waveforms like string orchestra, chorus, analog synths, cymbal rolls, and so on. Stereo samples retain their ambience and dimensionality. The SPR Looper is protected by US patent law.

ROTATED SUMS LOOPER

This looping tool uses a convolutional approach to randomizing irregularities inside the loop points. Like the SPR Looper, it is excellent for use with all sounds that have irregular (non-periodic) waveforms.

FREEZE LOOPER

The Freeze Looper creates perfect loops in sounds like winds, brass, bells, and other sounds that have a pure harmonic series. It's also good for "freezing" the timbre of decaying strings like piano or guitar.

CROSSFADE LOOPER WITH SMART AUTO-SCAN

This is an enhanced crossfade tool with an intelligent loop point scanner. It uses a sophisticated DSP algorithm to find the right loop point automatically.

FULLY ADJUSTABLE EDIT BLENDING

This editing feature allows the taper and length of the blend crossfades used during edit operations to be adjusted to suit the sound you are working on.

COMPLETE SAMPLE EDITING ENVIRONMENT

The Infinity environment provides all the basic tools necessary to create professional quality samples easily and quickly.

Infinity is robust. It lets beginners get results which are superior to professionals using other programs. For professionals, Infinity provides a level of quality, speed, and ease of use inapproachable by any other sample editor.

HOW TO USE THIS MANUAL

Infinity has a transparent user interface, has excellent Balloon Help and is extraordinarily easy to use. However, the operation of the DSP Looping Tools and some functions of the user interface will not be immediately obvious because they do things which have never been done before. We recommend that you read the following sections of the manual to take full advantage of the quality and speed that Infinity makes possible.

- The Infinity Environment on page 7
- Viewing the Soundfile on page 16
- Editing the Soundfile on page 21
- Saving the Soundfile on page 34
- Customizing Your Environment on page 36
- Chapter 4: Infinity Looping Tools on page 39

Chapter 1: Getting Started

This chapter explains everything about starting Infinity and communicating with Antares Audio Technologies.

Chapter 2: Introducing Infinity

This chapter explains the scope of Infinity's functions and how it solves current problems in the electronic music studio. Basic concepts in sample editing and file management are also covered. The user interface is explained with emphasis on the anatomy of the Palette, and the Soundfile Window.

Chapter 3: The Basics of Infinity

Infinity Basics explains how to perform all the basic functions in manipulating and editing a soundfile. The topics covered are: basic commands and editing functions, changing the display resolution, viewing, listening to, editing and saving the soundfile, and customizing the Infinity environment.

Chapter 4: Infinity Looping Tools

This chapter contains detailed instructions on how to use the looping tools. We recommend this chapter as must-reading for everyone.

Appendix A: Keyboard Equivalent Commands

This appendix contains a list of all the keyboard equivalent commands and shortcuts that can be used with Infinity.

Chapter 2: Introducing Infinity



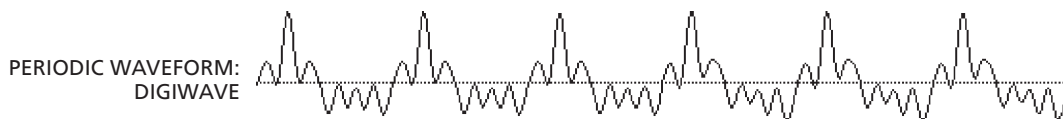
amplers and sample players have become the most popular electronic musical instruments because of the realism and fidelity made possible by digital recording techniques. This realism comes at the price of ever increasing memory requirements needed to produce the sustained tone of natural instruments. Since samplers have finite memories, rudimentary looping techniques have been used to create the illusion of sustained tones, but often with less than satisfactory results.

THE PROBLEM WITH LOOPING

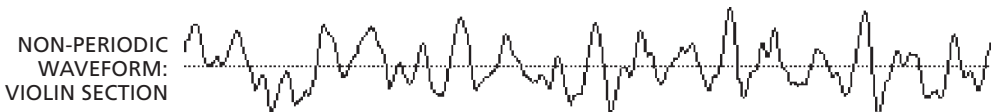
An ideal loop sounds like a uniform sustained tone. An imperfect loop is detectable because one hears repetitive changes in the tone. Using standard looping techniques, it is possible to create an ideal loop only under special conditions. These conditions are:

1. The waveform being played must be perfectly uniform during the duration of the loop (which may be thousands of cycles long), and
2. The loop must start and end at the same point in the period of the waveform.

This means that a perfect loop can only be created by using a sustained periodic waveform. But this type of sound is only a small subset of the sounds used in music making. Sustained sounds come in two varieties:



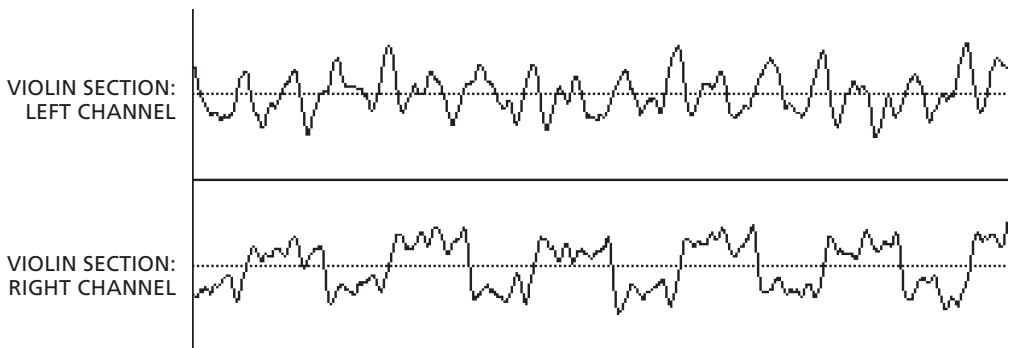
PERIODIC SOUNDS Sounds recorded monophonically which have stable tone, loudness, and pitch. Examples are solo wind and brass instruments, organ, and so on.



NON-PERIODIC SOUNDS Some examples of non-periodic waveforms are instrumental ensembles like string sections, brass sections, choirs, chorused guitar and keyboard sounds and many waveforms from the older analog synthesizers. Sounds that change tone as they decay like piano, harp, timpani, bells, cymbal rolls, etc. are also non-periodic.

Clearly, most of the sustained sounds that are used for music making are non-periodic. It is impossible to make an ideal loop of a non-periodic sound. The simple fact that the sound changes with time will cause the repetitive nature of the loop to show up.

The problem becomes even more difficult when considering the stereo case. If there are even slight changes to the shape of the two waveforms then it becomes impossible to find loop points where the waveforms precisely match. (A stereo loop cannot have a separate loop point for the left and right channel because doing so would not preserve the stereo relationship between the channels in the reproduced output.) The illustration below shows a segment of a stereo recording of a violin section playing a single pitch. Note that the two waveforms don't even resemble each other.



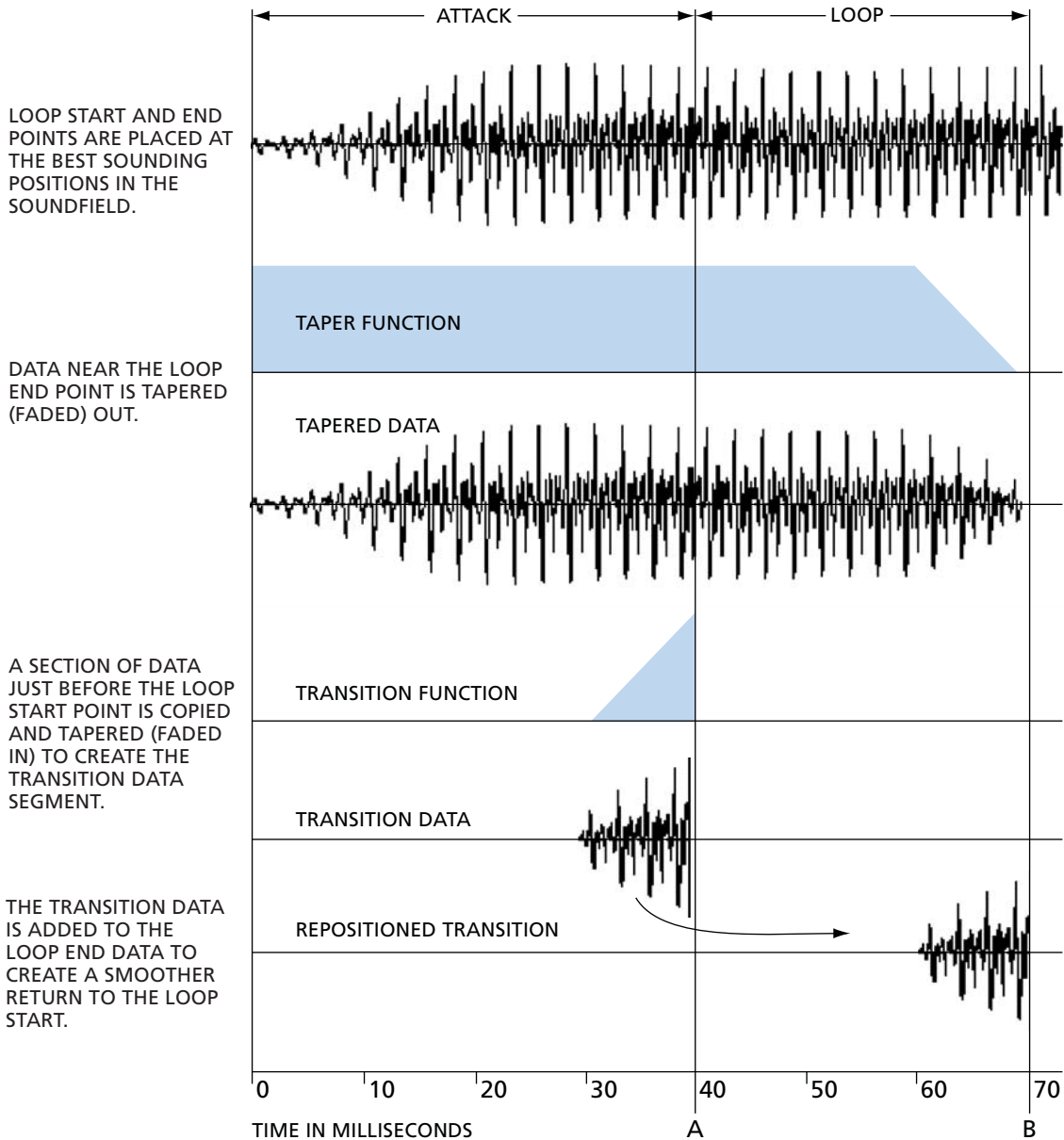
CROSSFADE LOOPS

One common compromise for looping non-periodic sounds is to blend the sound leading up to the loop end point with sound around the loop start point. This is called a crossfade. This tends to make the transition from loop end to loop start less abrupt.

No technique will ever loop a non-periodic sound without leaving irregularities somewhere in the loop. The problem with a crossfade loop is all of the different frequencies in the sound make a transition from one phase to another during the crossfade. This causes different combinations of constructive and destructive interference. Because this happens for all frequencies at the same point in the loop, a noticeable bump is produced by the crossfade loop.

The illustration below shows the process of creating a crossfade loop. The waveform segment labeled "attack" is the part of the sound that is heard before the loop begins. The line labeled "A" is the loop start point. The line labeled "B" is the loop end point.

Used with skill and patience, crossfade looping techniques can result in marginally usable loops in some non-periodic sounds. The level of quality that results is currently considered acceptable, but is, in fact, mediocre compared to what can be achieved by Infinity's unique DSP tools.



THE COSTS OF CURRENT METHODS

The time and effort required to make a good set of looped samples is more than most musicians are willing to invest. With complex sounds like pianos, string sections, chorused synths, and so on, it is extraordinarily difficult to make good crossfade loops. Lumpy loops have necessarily become acceptable.

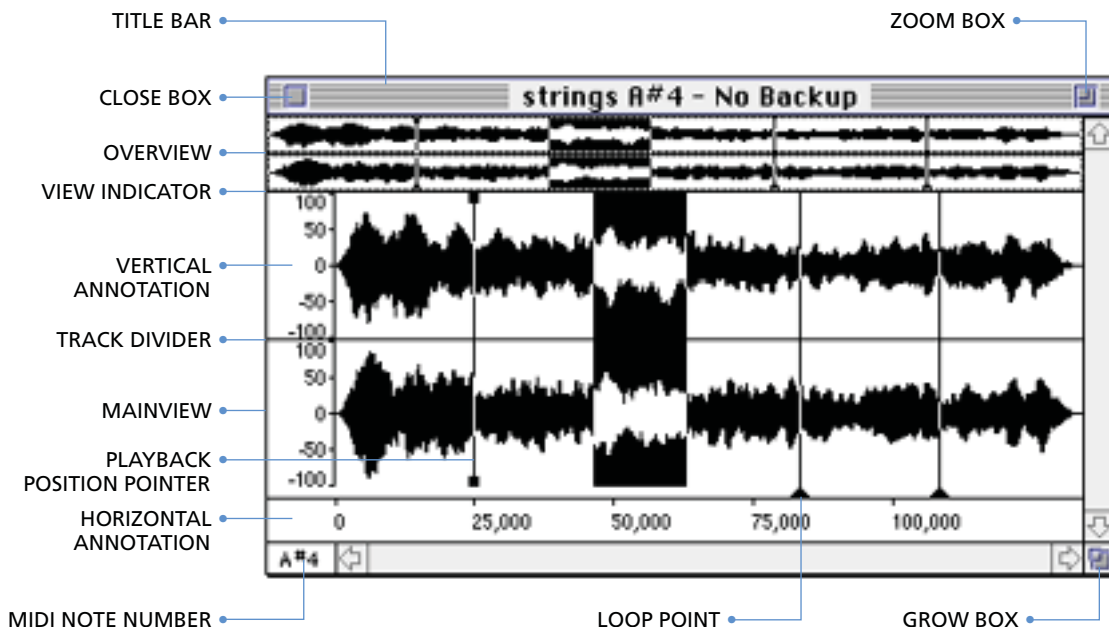
The third party sound design business is booming. License to use pre-looped sounds which are often of less than satisfactory quality costs hundreds of dollars. While this saves time and effort, even the best commercial samples have audible loops.

THE INFINITY ENVIRONMENT

Even experienced sound designers are advised to read this section carefully. It contains basic information on the display and manipulation of digital sound that is essential to using Infinity.

THE SOUNDFILE WINDOW

The soundfile window is a visual representation of the soundfile on the desktop. Much of the work you do on the soundfile will be done inside the soundfile window. The soundfile window has several components whose functions are listed below.



TITLE BAR

The Title Bar contains the name of the soundfile, the close box, the zoom box, and either “No Backup” or “Backup Created” written next to the title of the soundfile.

Infinity will create a backup file when the soundfile is first opened provided the corresponding Preferences dialog item is enabled. When “Backup Created” is displayed in the title bar, the Revert to Backup command can be used in the File Menu. See “Create Backup Files” on page 36.

CLOSE BOX

Clicking on the close box will close the soundfile window.

ZOOM BOX

Clicking on the zoom box will increase the size of the window to the maximum allowed by the size of your video monitor. Clicking it again will return the window to its original size.

OVERVIEW

The Overview displays the whole soundfile, the insertion point, loop points, and the Playback Position Pointer. The Overview allows you to navigate around the soundfile easily. In a stereo soundfile, the left track is displayed above the right track.

Data can be chosen for display in the Mainview by using the view indicator described below. Data can also be selected in the Overview by placing it in the Selection Mode using the Palette. See “Using Selection Mode in the Overview” on page 23 for details of how this is done.

VIEW INDICATOR

The view indicator is a dotted rectangular box which shows the borders of the region of the soundfile which is currently displayed in the Mainview. Any change in the Mainview is reflected by a change in the view indicator. The region displayed in the Mainview can be changed by redrawing the view indicator. The details are discussed in “Using Zoom Mode in the Overview” on page 17.

PLAYBACK POSITION POINTER

The Playback Position Pointer appears in both the Overview and the Mainview. It is a non-blinking vertical line with terminals at each end that moves across the screen as the soundfile plays. It shows where playback is occurring so that you can correlate the sound and the graphic display of the soundfile.

VERTICAL ANNOTATION

The Vertical Annotation allows you to measure the amplitude of the waveform in the Mainview. The units displayed are either percent of full scale or decimal sample values. The units for the Vertical Annotation are chosen in the Preferences dialog. The Vertical Annotation and the soundfile data can be moved up or down by using the vertical scroll bar when the scale is expanded.

MAINVIEW

The Mainview always displays the part of the soundfile shown in the view indicator in the Overview. When a file is opened, the Mainview initially displays the whole soundfile. The amount of data displayed in the Mainview can be changed by redrawing the view indicator in the Overview, or by using the Display Scale Icon on the Palette. The Mainview can also be placed in the Zoom Mode. See "Using Zoom Mode in the Mainview" on page 18. The data in the Mainview can be scrolled using several methods. See "Scrolling the Data File" on page 16.

TRACK DIVIDER AND ZERO CROSSING

The tracks in both the Overview and the Mainview can be separated by a track divider line. There can also be a dotted zero crossing line in each channel. These choices are made in the Preferences Dialog.

THE HORIZONTAL ANNOTATION

The Horizontal Annotation is located below the Mainview and runs the full width of the window. It allows you to measure time in the soundfile. The units for the Horizontal Annotation are selected in the Preferences dialog. There are three display formats: Samples, Time, and SMPTE formats.

LOOP POINTS

The Loop Points show the region of data in the soundfile that will be looped when the soundfile is played. The Loop Points can be moved by dragging them horizontally. They can also be dragged during sound playback.

Loop Points can be created by selecting the Create Loop Points menu item in the Edit menu or by clicking on the Loop Points Icon on the Palette. The Loop Points can be deleted by clicking on the Loop Points Icon or by selecting the Delete Loop Points item in the Edit Menu.

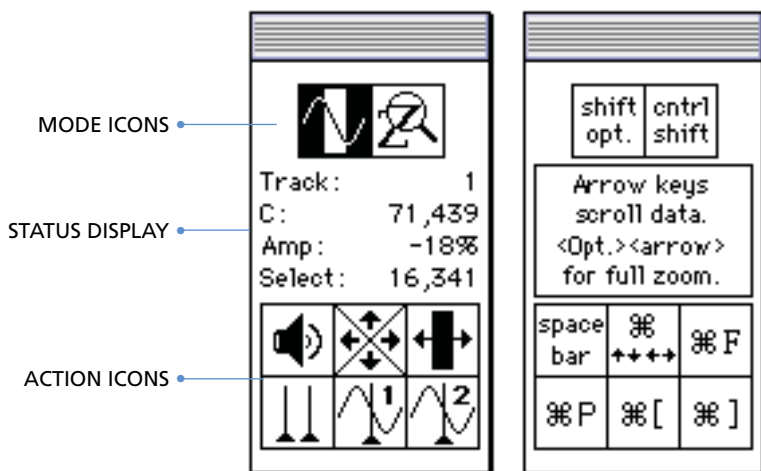
GROW BOX

Dragging the grow box allows you to resize the window.

THE PALETTE

The Palette is always present on the desktop. It contains information and tools for editing the soundfile data. It is divided into three sections;

- Mode Icons
- Status Display
- Action Icons



The area in the middle of the Palette is the Status Display. It is used for verbal messages and as a cursor tracking display. The icons above the Status Display are the Mode Icons. The icons below the Status Display are the Action Icons. Icons which cannot be selected are grayed out.

The Palette can be dragged anywhere on the desktop by pressing on the grey bar at the top of the Palette and dragging it to the desired location. Pressing on the Palette in an area where there are no icons will reveal the keyboard equivalent commands for the icons.

MODE ICONS

When Infinity starts, the Mainview is in Selection Mode and the Overview is in Zoom Mode. The mode of a display determines what work can be done on the data in that display. See "Using Zoom Mode in the Mainview" on page 18 and "Using Selection Mode in the Overview" on page 23 for instructions on changing the mode of a display.

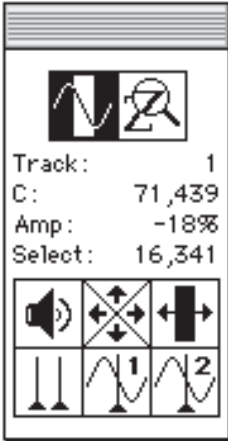
The mode icons are:



Selection Mode Icon: Selection Mode allows data to be selected in the display. See "Specifying Where to Change Sound Data" on page 21.



Zoom Mode Icon: Zoom Mode allows specification of a region to be displayed in the Mainview. See "Specifying Where to Change Sound Data" on page 17.



STATUS DISPLAY

The Status Display on the Palette gives information about the location of the cursor. Four fields of information are displayed which show various fields, depending on the current usage of the software:

Track: This field displays the track where the cursor is located.

C: This field displays the absolute time at the cursor location.

Amp: This field displays the vertical location of the cursor's active pixel.

C-I: This field displays the horizontal position of the cursor relative to the insertion point when the insertion point is present.

During a selection, this field reads "Select:" and shows how much data has been selected.

When the loop points are being dragged and the cursor is in the Mainview, this field reads "Loop:" and displays the amount of data inside the loop.

When the cursor is in the Soundview of the Crossfade Looper, this field reads either "Side: Loop Start" or "Side: Loop End" depending on which side the cursor is located.

THE ACTION ICONS



Speaker: The Speaker icon is used to play the soundfile. If a region is selected, only that region will play.



Display Scale: The Display Scale icon is used to change the display resolution of the Mainview.



Fit Selection: The Fit Selection icon fills the Mainview completely with the currently selected region.



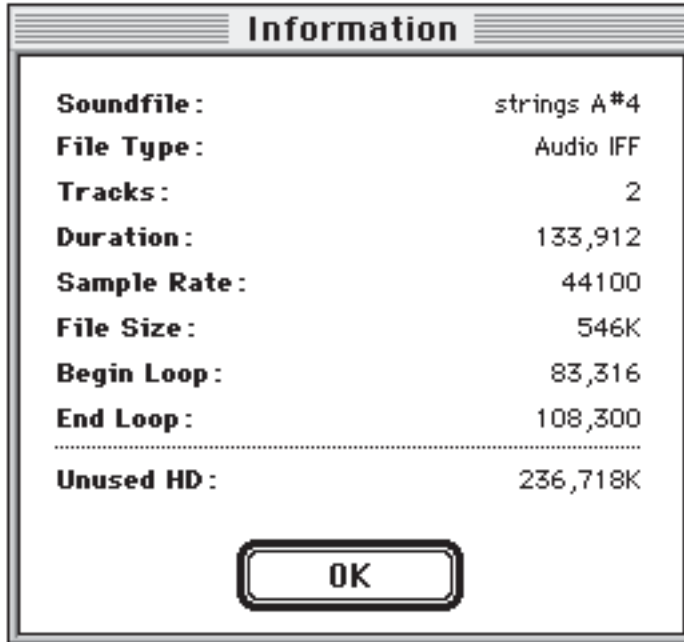
Loop Points: The Loop Points icon creates or deletes loop points in a soundfile.



Loop Start: The Loop Start icon scrolls the Mainview so that the Loop Start point is in the center of the screen.



Loop End: The Loop End icon scrolls the Mainview so that the Loop End point is in the center of the screen.



GET INFO...

Keyboard Equivalent Command: ⌘-I. Selecting Get Info will cause the Information dialog shown above to appear. It contains information on the currently active soundfile. Data cannot be edited in the Information dialog.

Sound File refers to the title of the active soundfile.

File Type refers to the file format in which the soundfile is stored.

Tracks refers to the number of data channels that exist in the file: one track for mono files, two tracks for stereo files.

Duration of the file is given in the current time format.

Sample Rate refers to the sample rate of the soundfile.

File Size of the soundfile is shown in kilobytes.

Unused HD states the amount of empty space that is currently available on the hard disk of the currently active soundfile.

DISK BASED FILES

Unlike most Macintosh applications which store their open files in RAM, Infinity uses disk based files. Changes made to the soundfile are automatically written to the disk when the change is made. Soundfiles, however, are played and processed from RAM. If a file is too large to fit in RAM, you can select and play a smaller portion.

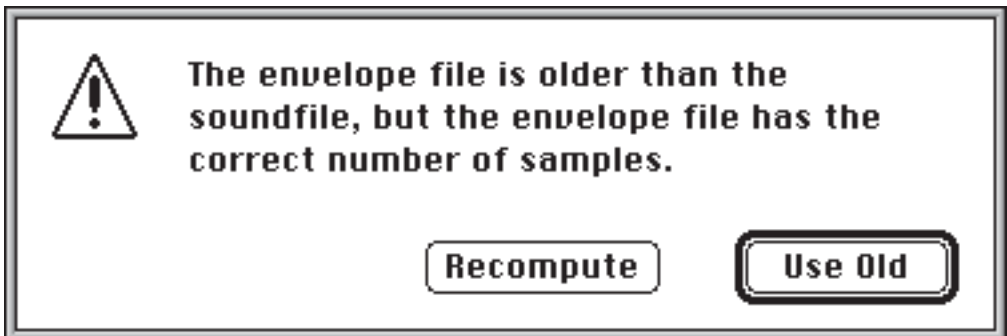
ENVELOPE FILES

In order to display a soundfile, Infinity creates an envelope file. These files are stored in a folder called "Envelope Files" in the same folder as the sound file. Envelope files are used to store information which allows Infinity to quickly draw the soundfile on the screen. They occupy about one tenth the disk space that the soundfile uses and are stored in a special Envelope Folder which Infinity creates. Envelope files are saved in the Envelope Folder after the soundfile is closed.

When data in the soundfile is shifted as a result of an edit, the envelope is also shifted, hence avoiding more lengthy computations involving the soundfile. Unfortunately, the envelope can only be shifted to an accuracy of ± 10 samples. Hence, the envelope may be slightly inaccurate. In this circumstance a "stale" flag is set in the envelope file.

Only the first envelope is shifted. The remaining envelopes are recomputed from the first envelope. This means the remaining envelopes will have fewer errors. Specifically, each envelope frame (sample) will contain errors of up to ten sound samples, however the second, third, fourth, etc. envelopes represent 400, 8000, 160,000, etc. sound frames each. Thus, the worst error will be 10 out of 400, 10 out of 8,000, 10 out of 160,000, etc.

The next time the soundfile is opened, the "stale" flag is checked, and if the file is sufficiently large, the following message appears:



Choosing either command button will reset the "stale" flag and the message will not reappear until the next edit.

When a soundfile is opened, if the envelope file is not found or if the sample counts don't agree with the envelope frame count, then the envelope file is recomputed automatically. If the soundfile has been modified by some other sample editing program, the envelope is considered to be "out of date". If this "out of date" condition exists, and the file is sufficiently large, the dialog shown above is posted. If "Recompute" is chosen, the envelope is recomputed. If "Use Old" is chosen, then the envelope file is touched so the condition does not reoccur, and no prompt will occur for the "stale" condition, even though the envelope file may be stale.

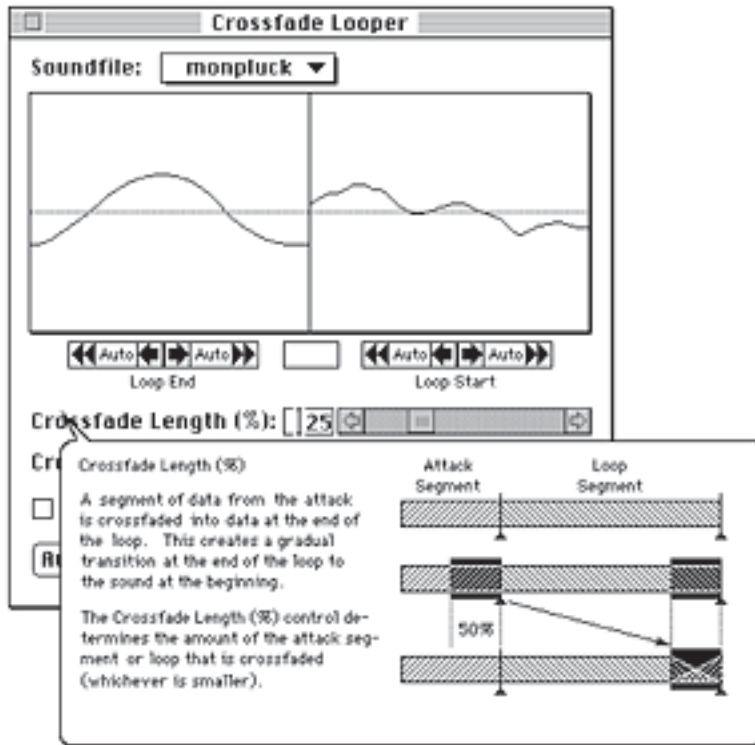
BACKUP FILES

If the Create Backup Files function in the Preferences dialog is enabled, a backup file is created when a soundfile is opened. Upon closing a soundfile, the backup file is deleted. Note that other temporary files are created as a result of edit operations involving the Clipboard and undo-able edits. These are automatically managed by Infinity.

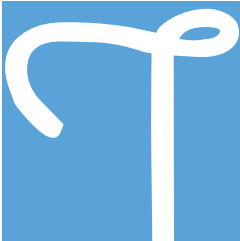
BALLOON HELP

Infinity incorporates on-line help using the Balloon Help feature. Choose Show Balloons in the menu to turn on Balloon Help.

BALLOON HELP IS AVAILABLE FOR EVERY ITEM IN INFINITY'S USER INTERFACE. SOME ITEMS HAVE GRAPHICS TO EXPLAIN COMPLEX PROCESSES.



Chapter 3: The Basics of Infinity



This chapter introduces you to the basic functions of Infinity. We recommend that you read it through once and then refer to it as needed while becoming familiar with Infinity.

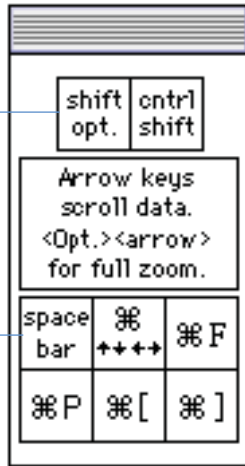
USING COMMANDS

USING KEYBOARD COMMAND EQUIVALENTS

The keyboard can be used to choose many commands. If a command has a keyboard equivalent, it usually appears on the menu or on the Palette.

PRESSING SHIFT+OPTION
WILL TEMPORARILY
PUT INFINITY INTO
SELECTION MODE

PRESSING THE
SPACE BAR
ACTUATES THE
SPEAKER ICON



Edit

Undo	⌘ Z
Undo Loop Move	

Cut	⌘ H
Copy	⌘ C
Paste	⌘ U
Clear	⌘ B
Clear Unselected	⌘ T
Clear Beyond Loop ...	

Mix	⌘ M
Invert	
Reverse	
Silence	

The shift, option, command (⌘) and control keys are called modifier keys. They don't do anything by themselves, but they can modify the next keystroke or mouse click. In Infinity, they are also used in combination to momentarily put the program in either Selection Mode or Zoom Mode. For details of how this is done, see "Using Momentary Selection Mode" on page 23 and "Using Momentary Zoom Mode" on page 18.

The following are examples of keyboard command equivalents and instructions on how to execute them. For a full listing of all the keyboard command equivalents, see page 67.

TO EXECUTE:	DO THIS:
⌘-X	Hold down the Command key and press the x key.
Shift-Option	Press the Shift and Option keys at the same time.
⌘-Shift-[Hold down the Shift and Command keys and press the [key.

UNDOING A COMMAND

Commands which change the soundfile can be undone by choosing Undo from the Edit menu. The keyboard command equivalent is ⌘-Z.

This function is possible only if Enable Undo is checked in the Preferences... dialog. See "Enable Undo" on page 36.

VIEWING THE SOUNDFILE

After you open or create a soundfile, Infinity allows the soundfile to be viewed in a variety of ways designed to increase the speed and ease of editing the data. This section explains how to scroll the data, change the magnification of the display, and manage multiple windows.

SCROLLING THE SOUNDFILE DATA

Infinity provides several convenient options for scrolling the soundfile data in the Mainview. The soundfile data can be scrolled both horizontally and vertically.

TO:	DO THIS:
Scroll by small increments	Click on an arrow in a scroll bar, or press the appropriate arrow key.
Scroll by one window increments	Click on a grey area in a scroll bar, or press Shift and the appropriate arrow key.
Scroll full left and center vertically.	Press the Home key.

CHANGING THE DISPLAY RESOLUTION



USING THE DISPLAY SCALE ICON

The display resolution of the Mainview can be changed by using the Display Scale icon on the Palette or the keyboard command equivalents.

- To double the vertical display resolution, click on the up arrow or press ⌘-up arrow
- To halve the vertical display resolution, click on the down arrow or press ⌘-down arrow
- To double the horizontal display resolution, click on the right arrow or press ⌘-right arrow
- Halve the horizontal display resolution, click on the left arrow or press ⌘-left arrow
- Zoom to the extreme setting, press Option and click on the appropriate arrow or press the appropriate arrow key.



USING THE FIT SELECTION ICON

Clicking on the Fit Selection icon will cause a selected region to fill the Mainview Display. The keyboard command equivalent is ⌘-F.



USING ZOOM MODE IN THE OVERVIEW

When a soundfile is opened, the Overview is in Zoom Mode. In Zoom Mode, a view indicator can be dragged around the area of the soundfile you want to see in the Mainview. This area then fills the Mainview display.

To chose an area of the soundfile for viewing in the Mainview:

1. Move the cursor into the Overview display.

The cursor turns into a cross. The Zoom icon on the Palette video-reverses to show that this display is in Zoom mode.

2. Place the cursor at the corner of the area you want to display and then click-drag the mouse over the area.

A view indicator box appears around the area you have chosen.

3. Release the mouse.

The view indicator remains around the area you have chosen. The Mainview now displays the selected area.

USING ZOOM MODE IN THE MAINVIEW

The view indicator can be used directly in the Mainview itself, by placing the Mainview in Zoom Mode.

To put the Mainview in Zoom Mode:

1. Click on the Zoom Mode icon in the Palette.

The icon begins to blink and the message "Choose a Window for Zoom Mode" appears in the Status Display area of the Palette.

2. Move the cursor into the active window and click in the Mainview.

The cursor becomes the cross cursor. The Zoom Mode icon on the Palette video-reverses to show that the Mainview is in Zoom Mode.

You may now drag a view indicator around any region in the Mainview. The display will automatically scroll if the view indicator is dragged past the edge of the display. The view indicator can also be dragged from the first click inside the Mainview.

To put the Mainview back into Selection Mode,

1. Click on the Selection Mode icon in the Palette.

The icon begins to blink and the message "Choose a Window for Selection Mode" appears in the Status Display area of the Palette.

2. Move the cursor into the active window and click in the Mainview.

The cursor becomes the I-beam cursor. The Selection Mode icon on the Palette video-reverses to show that the Mainview is in Selection Mode.

You may now select data in the Mainview as before.

USING MOMENTARY ZOOM MODE

Infinity can be put into Zoom Mode temporarily by pressing <control-shift>. This is useful for quickly zooming a region in the Mainview. The cursor will become the cross cursor and all active displays will remain in Zoom Mode as long as this key combination is pressed. When the key combination is released, all displays go back to their previous modes.

USING FULL ZOOM IN AND FULL ZOOM OUT

The horizontal display resolution of the Mainview can be set to its extreme in one step.

TO:

DO THIS:

To zoom to minimum resolution

Select Full Zoom Out from the Display menu, or double-click on the Mainview in Zoom Mode.

To zoom to maximum resolution

Select Full Zoom In from the Display menu.

CHANGING SEPARATORS, UNITS AND SCALES

SHOW ZERO CROSSING

This item places dotted lines at the zero crossing of tracks in the Overview and Mainview.

SHOW TRACK DIVIDERS

When checked, this item places solid lines between the tracks in the Overview and Mainview.

VERTICAL SCALE UNITS

This item allows either percentage of full scale or sample value to be displayed on the Vertical Annotation in the Mainview and the Status Display Area of the Palette.

HORIZONTAL SCALE UNITS

This item allows the units shown on the Horizontal Annotation of the Mainview and all other time displays to be chosen. The pop-up menu choices are Samples, Hours:Minutes:Seconds:Milliseconds, SMPTE 24 FPS, SMPTE 25 FPS, SMPTE 30 FPS, and SMPTE 30 FPS Drop.

MANAGING MULTIPLE WINDOWS

Windows can be arranged by using the Tile Windows and Stack Windows commands in the Windows menu.

- To show all open windows without overlapping, select Tile Windows.
- To stagger all open windows at maximum size, select Stack Windows.

HIDING THE OVERVIEW

More space in the active soundfile window can be created by using the Hide Overview command in the Display Menu. Selecting this command causes the Overview of the active window to disappear. When the active window has its Overview hidden, the menu item reads "Show Overview". Selecting Show Overview will put the Overview back in the active window.

SETTING-UP SOUND PLAYBACK

The Sound Playback menu contains a list of the devices which are currently installed. Choosing an item will direct the soundfile playback through the selected device. Choosing Mac 8, or Mac 16 connects Infinity the Apple Sound Manager. 'Mac 8 bit' presents 8 bit data to the Sound Manager and 'Mac 16 bit' presents 16 bit data to the Sound Manager. (The Sound Manager 3.0 or later must be present. If not, 'Mac 16 bit' will present 8 bit data to the Sound Manager.) Interpolation of data occurs when the sample rate of the Mac is different than the sample rate of the data. The Sound Manager distorts the sound when there are not enough machine cycles to interpolate the data. If this occurs, there are two things to try. The first is to use 'Mac 8 bit', since the Mac is faster at interpolating 8 bit data. The second is to set the sample rate of the Mac to the Sample rate of the data. See the Sound control panel to adjust Sound Manager sample rates.



USING THE SPEAKER ICON

The soundfile can be played by clicking on the Speaker icon in the Palette. Click the icon again to stop playback. Pressing on the <space bar> is the keyboard command equivalent for the Speaker icon. Playback in the active soundfile window is determined by the following conditions:

IF:	THEN:
No data is selected	The track(s) containing the insertion point play from the beginning of the soundfile.
The soundfile has loop points	The track(s) containing the insertion point are played from the beginning and continue to loop until the Speaker icon is clicked again.
Data is selected	Just the selected region is played. The selection contains both loop points The selection is played and loops until the Speaker icon is clicked again.
Only the loop data is selected	Just the loop will play until the Speaker icon is clicked again.
A looping tool is the active window	Both tracks of the soundfile are played from the beginning regardless of any insertion point or selection.

Note: *In a stereo soundfile, data is played back through the audio channel that corresponds to the track number. Data on Track 1 plays through the Left channel; data on Track 2 plays through the Right channel.*

EDITING THE SOUNDFILE

In addition to automated looping tools, Infinity has a sophisticated waveform editing environment. The following section describes how to select data and use the editing tools to change it.

SPECIFYING WHERE TO CHANGE SOUND DATA

The first step in editing the soundfile data is to identify which data are to be changed. In Infinity, as in many other Macintosh applications, data is selected by using the insertion point and the I-beam cursor. Data can be selected in both the Overview and the Mainview.



USING SELECTION MODE IN THE MAINVIEW

When a soundfile opens, the Mainview is in Selection Mode. Selection Mode is used to change the current selection or insertion point. When the cursor is moved into the active Mainview display, it becomes the I-beam cursor. The Selection icon on the Palette video-reverses to indicate that the Mainview is in Selection Mode. The I-beam cursor can be used to place the insertion point or make a selection anywhere in the soundfile.

PLACING THE INSERTION POINT

The insertion point is the blinking vertical line which appears in both the Overview and the Mainview. It can be moved only in Selection Mode by using the I-beam cursor. It marks the place where data will be inserted or a boundary of a region to be selected.

TO:	DO THIS:
Place the insertion point	Click once in the Mainview at the desired location. In a stereo soundfile, the insertion point appears in both tracks.
Place it in only one track	Hold down the Option key and click at the desired location.
Extend it to the other track	Hold down the Control key and click in the other track. Click again to remove it.

SELECTING DATA

There is a variety of ways to select data in Infinity. Selected data become video reversed to show that they are selected. The Mainview will automatically scroll during the selection if the cursor is dragged beyond the data displayed in the Mainview.

TO:	DO THIS:
Select a region in one track	Drag through the region, or Optionclick at the start , then Shift-click at the end.
Select a region in both tracks	Drag across the two tracks, or click at the start of the region, then Shift-click in the other track at the end.
Lengthen the selection	Point at the new start or end and Shift-click. You can then drag to a different endpoint.
Shorten the selection	Point near the current end, Shift-click, then drag to a different endpoint.
Extend selection to the other track	Hold down the Control key and click in the other track. Click again to remove it.
Select a whole track	Double-click in the track; or place the insertion point in the track and choose Select All (⌘-A) from the Edit menu.
Select all the data in the soundfile	Double-click in one track and drag to the other; or Shift double-click in the other track; or place the insertion point in both tracks and choose Select All from the Edit menu.
Select data inside the loop points	Choose Select Loop from the Edit menu.
Deselect a region	Place an insertion point anywhere.

USING SELECTION MODE IN THE OVERVIEW

Data can also be selected in the Overview by placing the Overview in Selection Mode. The insertion point and data selection behaviors are exactly the same as in the Mainview.

To put the Overview in Selection mode:

1. Click on the Selection icon in the Palette.

The icon begins to blink and the message "Choose a Window for Selection Mode" appears in the Status Display area of the Palette.

2. Move the cursor into the active window and click in the Overview.

The cursor becomes the I-beam cursor. The Selection icon on the Palette video-reverses. Use the I-beam cursor to select any region in that Overview.

To put the Overview back into Zoom Mode,

1. Click on the Zoom Mode icon in the Palette.

The icon begins to blink and the message "Choose a Window for Zoom Mode" appears in the Status Display area of the Palette.

2. Move the cursor into the active window and click in the Overview.

The cursor becomes the cross cursor. The Zoom Mode icon on the Palette video-reverses to show that the Overview is in Zoom Mode.

USING MOMENTARY SELECTION MODE

Infinity can be put into Selection Mode temporarily by pressing <shift-option>. This is useful for quickly selecting a region in the Overview. The cursor will become the I-beam cursor and all active displays will remain in Selection Mode as long as this key combination is pressed. When the key combination is released, all displays go back to their previous modes.

USING THE EDITING FUNCTIONS

This section describes the edit functions shown in the Edit menu as they operate without the use of the Blending feature. See “Using Edit Blending” on page 31 for the details of how this feature effects the edit functions.

All of the functions described below can be reversed using the Undo command in the Edit menu, provided that Undo is enabled in the Preferences... dialog. See “Customizing Your Environment” on page 36 for the details.

When an edit operation is done on a soundfile, the contents of the Clipboard is stored as the soundfile’s previous Clipboard. For a Cut or Copy, the following steps take place:

- The application Clipboard is updated.
- The soundfile stores the new Clipboard as its new Clipboard.
- For a Clear, Clear Unselected, or Paste, the soundfile stores undo information to undo the edit.
- You can go back to any document, any time after it has been edited, and do an undo.
- The soundfile uses the new Clipboard or the undo information, whichever is available, to undo the edit.
- If the soundfile’s new Clipboard information was used, and if that information is the same as the application Clipboard, then the application Clipboard is replaced with the soundfile’s previous Clipboard.
- This has the effect of undoing the Clipboard as well as the soundfile edit.
- You can go back to any document, any time after it has had an undo and do a redo.
- If the undo had the effect of undoing the Clipboard, then redo will have the effect of restoring it.

CUT

Cut removes precisely the number of samples that are selected. Any data to the right of the cut is shifted left by exactly that amount. Blending is performed by combining data just inside the selection with data that was not cut on the opposite end of the selection. Hence, if you use cut to remove a noise spike, you may find, depending on the size of the cut and the amount of blending used, that part of the spike still remains.

If an Undo Cut occurs later, the cut data is replaced in the soundfile and the previous Clipboard data is returned to the Clipboard.

- To cut data, select the data to be removed. Then choose Cut from the Edit menu. The keyboard command equivalent is ⌘-X.

COPY

The Copy function copies data into the Clipboard without removing it from the soundfile. Since the Clipboard can hold only one item at a time, any data previously in the Clipboard is lost.

- To copy data, select the region to be copied. Then select Copy from the Edit menu. The keyboard command equivalent is ⌘-C.

PASTE

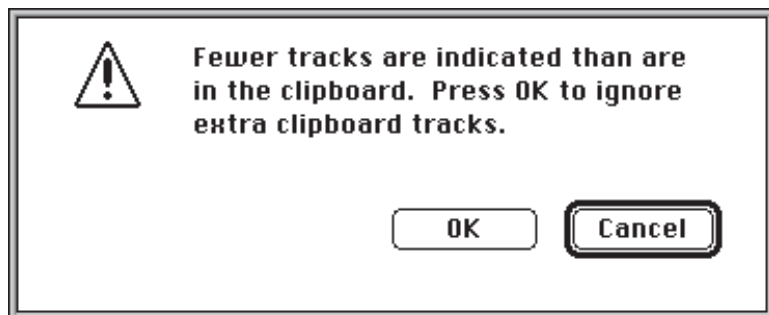
The Paste function inserts data from the Clipboard into the soundfile. Infinity will paste data into the soundfile at the insertion point or replace a selected region with the Clipboard data. Since the original data remains on the Clipboard, it can be pasted any number of times. The keyboard command equivalent is ⌘-V.

When Pasting into a soundfile that has no samples, the soundfile will acquire the sample rate and MIDI Note Number of the first data pasted into it.

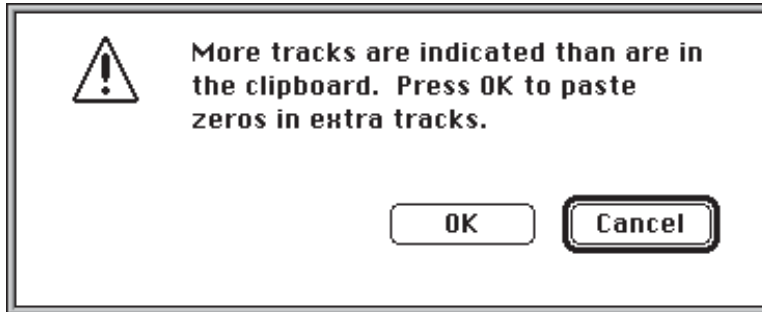
Normally, when Pasting or Mixing with an insertion point, the new data is inserted before the first sample of the group of samples represented by the pixel under the insertion point. The exception to this rule is when the insertion point is on the last pixel in the soundfile window. In that case, the new data is added after the last sample in the soundfile. Blending occurs between the end of the original sound and the pasted data.

- To paste at the insertion point, place the insertion point where the region of pasted data is to begin. Then select Paste from the Edit menu. The data to the right of the insertion point is moved over and the Clipboard data is inserted.
- To replace a selected region with the Clipboard, select the region to be replaced. Then select Paste from the Edit menu. The selected data is deleted and the Clipboard data is inserted in its place. Data to the right of the selection is moved to fit the newly pasted region.

Attempting to paste a stereo Clipboard into one track, evokes this dialog.



Attempting to paste a mono Clipboard into two tracks, evokes this dialog.



CLEAR

The Clear function is the same as Cut except the Clipboard remains unchanged. The data to the right of the cleared region are moved to fill up the gap.

- To clear data, select the region to be cleared. Then select Clear from the Edit menu. The keyboard command equivalent is ⌘-B or Delete.

CLEAR UNSELECTED

The Clear Unselected function clears all the data around a selected region of the soundfile. After the unselected data are cleared, the selected region is moved to the front of the soundfile.

- To clear unselected data, select the region to be left unchanged. Then choose Clear Unselected from the Edit menu. The keyboard command equivalent is ⌘-T.

CLEAR BEYOND LOOP END

This option clears data after the loop.

MIX

The Mix function combines the contents of the Clipboard with the soundfile data in a 1:1 ratio. Infinity will mix data into the soundfile starting at the insertion point or mix into a selected region. The keyboard command equivalent is ⌘-M. There must be data in the Clipboard for the Mix function to be enabled. You can mix into a selection or at an insertion point.

- To mix data starting at the insertion point, place the insertion point where the mixture is to begin. Then select Mix from the Edit menu. The data in the Clipboard will be mixed with the soundfile in a 1:1 ratio, starting at the insertion point and going to the end of the Clipboard. When mixing to an insertion point, the Clipboard selection is placed at the insertion point. Initial blending (if enabled) occurs before the insertion point. After the insertion point, all of the Clipboard is mixed, even if this means extending the length of the soundfile.

- To mix data into a selected region, choose Mix from the Edit menu. The Clipboard is mixed with the selected data from the beginning of the selection through the duration of the Clipboard or the selected region, whichever is shorter. When mixing into a selection (with no blend), the mixed data is limited to the minimum of the soundfile selection or the Clipboard selection. This keeps changes within the selection.

When there is a blend, the blend is done outside the selection and different blends are used for beginning and end, whatever can be worked out between the soundfile and the Clipboard.

In blending during the mix, the soundfile is never tapered. The Clipboard is tapered up in the beginning and down at the end (depending on the amount of data available to construct the blend).

When mixing into a soundfile that has no samples, the soundfile will acquire the sample rate and MIDI Note Number of the first data pasted or mixed into it.

Normally, when Pasting or Mixing with an insertion point, the new data is inserted before the first sample of the group of samples represented by the pixel under the insertion point. The exception to this rule is when the insertion point is on the last pixel in the soundfile window. In that case, the new data is added after the last sample in the soundfile.

INVERT

The Invert function changes the sign of the soundfile data. In other words, it inverts the phase of the waveform by 180 degrees.

- To invert the soundfile data, select the region to be inverted. Choose Invert from the Edit menu. The selected data are then changed.

REVERSE

The Reverse function reverses the order of the sample data in time, end for end. (e.g. a backwards cymbal roll)

- To reverse the soundfile data, select the region to be reversed. Choose Reverse from the Edit menu. The data are then switched in order, the last sample coming first.

SILENCE

The Silence function zeroes selected data in the soundfile.

- To silence soundfile data, select the region to be silenced. Choose Silence from the Edit menu. The sample values of the selected data are then zeroed.

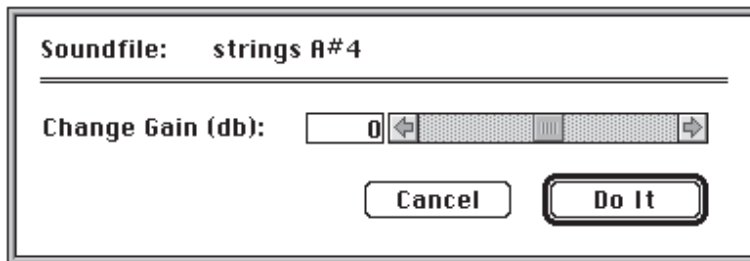
NORMALIZE

The Normalize function increases the gain of a selected region until the largest sample in the region is at 100% amplitude. Normalize insures the maximum signal to noise ratio from your sample playback instrument.

- To normalize soundfile data, select the region to be normalized. Choose Normalize from the Edit menu. The selected region will be increased in amplitude until the largest sample in the region is at full scale.

CHANGE GAIN...

The Change Gain dialog allows the level of the soundfile to be adjusted to avoid clipping during processing. Selecting Change Gain... from the Edit menu will cause the following dialog to appear:



The gain of the selected data can be changed by using the slider or by typing in the text box. The range of the control is from -12 dB to +12db. Clicking on the arrows will increment the setting by 1 dB. Clicking in the grey area will increment the setting by 3 dB.

A typical application for the Change Gain function is adjusting the level of a normalized soundfile downwards before mixing it with another soundfile. After processing, the soundfile can be returned to its original level by changing the gain an equal amount in the opposite direction.

- To change the amplitude of soundfile data, select the data to be scaled. Choose Change Gain... from the Edit menu. Set the desired gain change using the controls. Click OK to process the data. Changing Gain can be undone.



CREATE/DELETE LOOP POINTS

The Loop Points icon adds loop points to a file or deletes them from a file which has them.

- To create loop points in a soundfile, click on the Loop Points icon in the Palette, or choose Create Loop Points from the Edit menu. The keyboard command equivalent is ⌘-P. The loop points appear in the soundfile and the menu item changes to read Delete Loop Points.

- To delete loop points from a soundfile, click on the icon again. The loop points will disappear and the menu item will change back to read Create Loop Points. The deleted loop points can be returned to their original positions by clicking on the icon a third time.

LOOP THE SELECTION

This command place the loop points at the ends of the current selection.



LOCATING LOOP POINTS

The Mainview can be scrolled to the position of either the loop start or loop end points. The Loop Start icon is labeled 1. The Loop End icon is labeled 2.

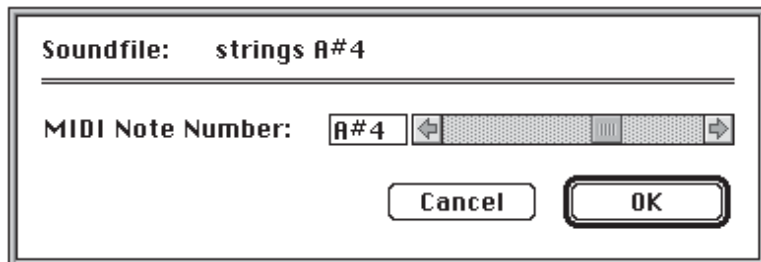
To center the Mainview on the loop start point, click on the Loop Start icon. The keyboard command equivalent is ⌘-Shift-].

- To center the Mainview on the loop end point, click on the Loop End icon. The keyboard command equivalent is ⌘-Shift-].

Hold the Option key while clicking these buttons does a full zoom out of the mainview display.

SET MIDI NOTE...

Choosing this item from the Edit menu or clicking on the MIDI Note Number in the soundfile window displays the following dialog.

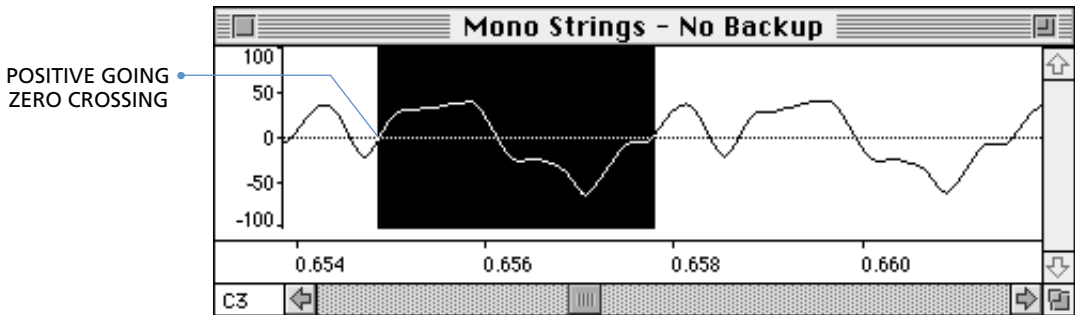


The MIDI Note Number tells Infinity the pitch of the soundfile. It is used in Infinity's looping algorithms and is critical to the operation of the Freeze Looper. Empty soundfiles will acquire the sample rate and MIDI Note Number of the first data pasted or mixed into them.

- To change the MIDI Note Number, choose Set MIDI Note... from the Edit menu. Data can be entered by typing in the text box or by using the scroll bar. Either the MIDI Note Number or the musical pitch (e.g. C#3) can be entered. Clicking in the grey area increments by one octave.

AUTO-ZERO

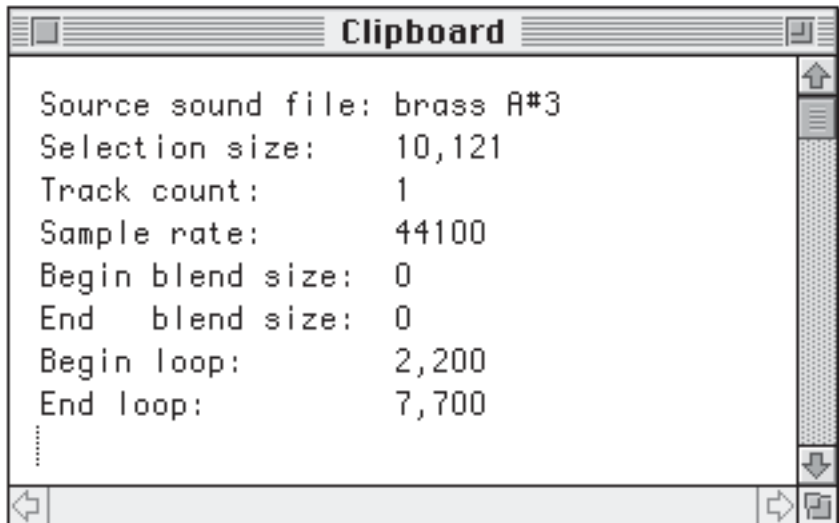
When this item is checked in the Edit menu, Infinity constrains the boundaries of selected regions to positive going zero crossings in the sample data. This helps you to avoid creating pops and clicks when you cut and paste. Infinity will always expand the selection to the positive going zero crossings closest to the initial selection boundaries. When two tracks are selected, Track 1 is used as the guide track.



To constrain the selection boundaries to positive going zero crossings in the soundfile, check Auto-Zero.

SHOW CLIPBOARD...

Choosing Show Clipboard... in the Edit menu calls the following dialog.



The dialog gives information about what data is currently in the Clipboard.

Source Sound File: The name of the soundfile from where the current Clipboard contents originated.

Selection Size: The size of the selected region in terms of the currently selected units used in the Horizontal Annotation. (in this case, samples)

Track Count: The number of tracks in the Clipboard.

Sample Rate: Expressed in Hertz

Begin Blend Size: The number of samples (or duration) before the selected region which will be used to blend the region when pasting or mixing.

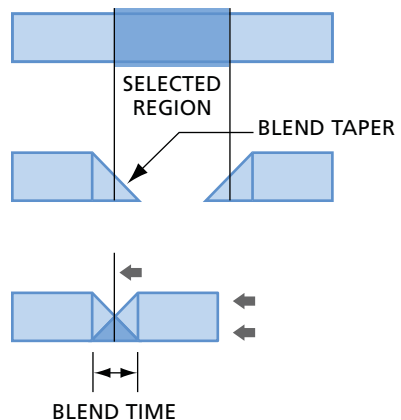
End Blend Size: The number of samples (or duration) after the selected region which will be used to blend the region when pasting or mixing.

Sound data which has been copied into the Clipboard in Infinity cannot be shared with Sound Designer or Alchemy, as these programs have incompatible Clipboard formats.

USING EDIT BLENDING

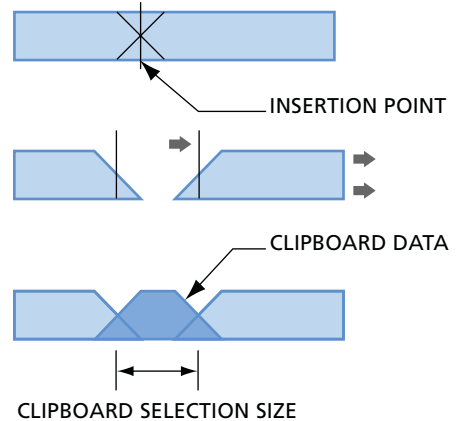
Enabling the Blending function causes Infinity to blend the edges of data segments that are moved during Cut, Paste, Clear, Mix, Invert, Reverse, Silence, Normalize, and Change Gain operations. This smooths the transition so that abrupt changes in the sound are avoided. The following figures show the effect of blending as used with the Cut and Clear functions.

1. Select a region of data.
2. Choose either Cut or Clear from the Edit menu.
3. Data on both sides of the selection boundaries are tapered. The remaining selected data are removed from the soundfile.
4. The data on the right are moved to close the gap and are crossfaded with the data on the left.



The illustration below describes the effect of Blending on the Paste function.

1. Place the insertion point at the location where data is to be pasted.
2. Select Paste from the Edit menu.
3. Data on either side of the insertion point is tapered. The data on the right are moved to create a gap to fit the Clipboard data.
4. Clipboard data are tapered and crossfaded with the soundfile data.



Cut or Copy will put the selected data in the Clipboard. If blending is enabled and the blend time is greater than zero, the leading and trailing samples will also be put on the Clipboard. If the selected region is at the end of a file, fewer or no trailing samples will be captured. If the selected region is at the beginning of a file, fewer or no leading samples will be captured. You can see how many leading and trailing samples are captured with "Show Clipboard". These are labeled "Begin Blend Size" and "End Blend Size", respectively.

Paste will similarly identify independent leading and trailing blend regions around the selection or insertion point. Paste will then compare its leading blend region with the leading blend region of the Clipboard and choose the smallest. Similarly, for the trailing blend time. The Paste is then performed with these new blend times. Note that blending occurs outside of the original selection and outside of the Paste's selection.

Cut removes precisely the number of samples selected in the file. In order to blend the result, samples on both side of the beginning of the selection are merged with samples on both sides of the ending selection, provided that blending is enabled and there is a non-zero blend time. The size of the merge is given by the blend time. It must be noted, therefore, that cutting some samples does not necessarily completely remove them. To completely remove the center-most samples, the selection size must be larger than the blend time.

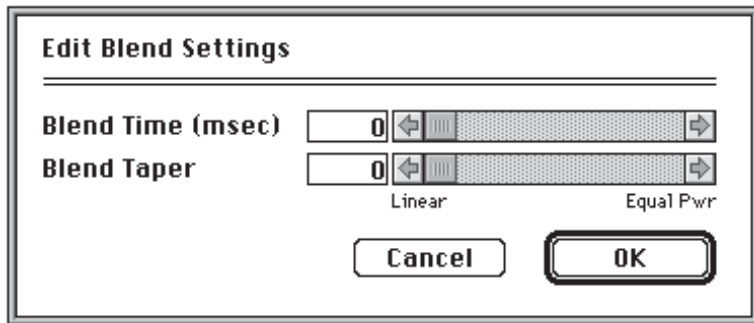
Clear Unselected does no blending.

Clear is the same as Cut, except no samples go to the Clipboard. Blending in Clear works the same as blending in Cut.

Normally, when Pasting or Mixing with an insertion point, the new data is inserted before the first sample of the group of samples represented by the pixel under the insertion point. The exception to this rule is when the insertion point is on the last pixel in the soundfile window. In that case, the new data is added after the last sample in the soundfile. Blending occurs between the end of the original sound and the pasted data.

ADJUSTING THE BLEND PARAMETERS

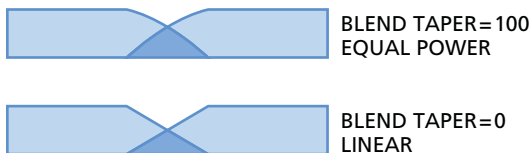
Infinity allows the length and taper of the crossfade used in edits to be adjusted. Clicking on the Blend... command button in the Preferences... dialog will evoke the Edit Blend Settings dialog shown below.



Blend Time: To change the length of the crossfade, use the Blend Time scroll bar or type directly into the text box to adjust the duration. The initial setting is 10 milliseconds. The range is from 0 milliseconds to 2,000 milliseconds. The most useful range for normal editing functions is from 10 to 50 milliseconds. (The scroll bar has exponential scaling so there is plenty of resolution around the most useful settings.) The illustration below shows the effect of changing the Blend Time control.



Blend Taper: The slope of the crossfade can be changed by using the Blend Taper scroll bar or text box. The range of settings is from 0 (linear) to 100 (equal power). The initial setting is 0 (full linear). The illustration below shows the effect of changing the Blend Taper control.



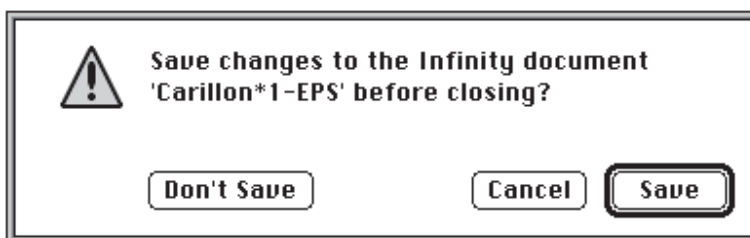
Linear taper should be used for editing simple periodic waveforms like solo instruments. Equal power taper should be used when editing complex non-periodic waveforms like string sections or noise.

SAVING THE SOUNDFILE

SAVE

Because Infinity uses disk based documents, changes to the soundfile are saved as they occur. The Save command in the Edit menu updates the backup file (if one exists) so that a Revert to Backup command will revert the soundfile to the most recently saved backup. The Save command in the Edit menu is selectable after a change has been made in the soundfile.

If the soundfile has been changed and a backup file exists, closing the soundfile will cause the following dialog to appear.



SAVE A COPY...

Use the Save a Copy... command to save the soundfile in a different folder, with a different file name, or in a different file format. This command is useful to make a series of samples for later comparison.

To use the Save a Copy... command:

1. Choose Save a Copy... from the File menu. The Save a Copy dialog box appears.



2. Type a new file name in the Save Soundfile As text box.
3. Choose a file format from the four options listed.

USE THIS FORMAT:

Audio IFF

Audio IFC

Sound Designer I

Sound Designer II

TO SAVE AS A:

Apple Audio Interchange File Format document

Apple Audio Interchange File Format-C document

Digidesign Sound Designer I mono soundfile

Digidesign Sound Designer II mono or stereo soundfile

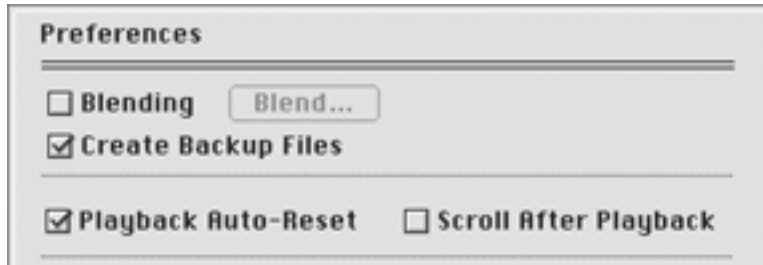
4. Click on Save. A copy of the active soundfile will be stored on disk with the name and file type specified.

REVERTING TO THE LAST SAVED VERSION

Choosing Revert to Backup from the File menu causes the soundfile to revert to the version last saved with the Save command. This command is enabled only if a backup file has been created for the active soundfile.

CUSTOMIZING YOUR ENVIRONMENT

When Infinity is first started, the look of the soundfile windows is predetermined. The appearance of the soundfile windows can be customized and some of Infinity's behaviors can be modified by using the Preferences... dialog in the Display menu.



BLENDING

Checking this item enables the Blending function.

CREATE BACKUP FILES

When this item is checked, opening or saving a soundfile will create a temporary backup copy of the file. This allows the Revert to Backup command in the File menu to be used.

Files that are opened with Create Backup Files checked will display the phrase "Backup Created" in their title bar. This indicates that there is a copy of the last saved version of the soundfile. Each time Save is used, this backup copy is deleted and the current sound is used to create a new backup file.

Creating backup files can take an inconveniently long time when working with large soundfiles. Also, there may be insufficient space on your hard disk to store the backups. If either of these conditions apply, uncheck the Create Backup Files item. The title bar for a file opened in this condition will display the phrase "No Backup".

NOTE: Be forewarned that when "No Backup" is displayed in the title bar, any changes you make to the soundfile will be permanent unless Undo is enabled.

PLAYBACK AUTO-RESET

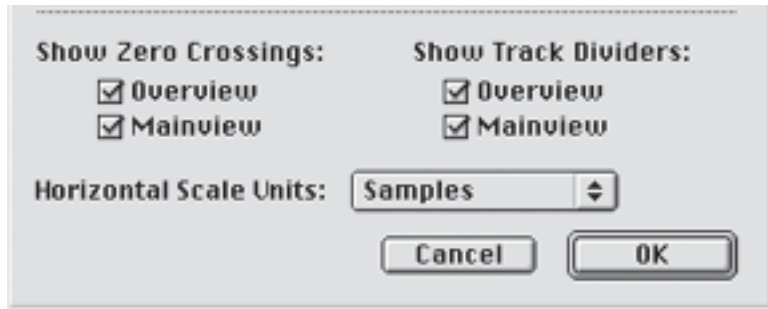
If this item is checked, playback will occur from the beginning of the soundfile unless a region is selected. If this item is not checked, playback will begin from the point where it stopped the last time. This item should remain checked for most applications.

SCROLL AFTER PLAYBACK

This item is initially not checked when you start Infinity. Checking this item causes the Mainview to automatically scroll to the position where playback stopped. This feature is most useful when working with the Mainview at high display resolutions.

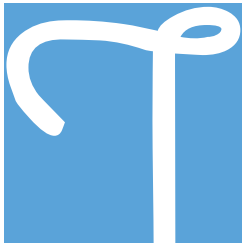
CUSTOMIZING THE SOUNDFILE WINDOW

The appearance of the soundfile window can be changed by using the soundfile window options in the Preferences... dialog.



- To display the dotted zero crossing lines, check the box next to Overview or Mainview.
- To display track divider lines, check the box next to Overview or Mainview.
- To change the units on the Horizontal Annotation, use the pop-up menu. See the reference section for more information about these options.

Chapter 4: Looping Tools



The looping tools in Infinity are easy to use and provide extraordinary results. To use them most effectively requires some understanding of the types of sound you are working on and the way the tools operate on the sound data. The following chapter explains, in detail, the operation of Infinity's looping tools.

MANAGING LOOP POINTS

REAL TIME LOOP ADJUST

Infinity allows the location of the loop points to be changed during playback. To move a loop point during playback, drag it to its new location by pressing somewhere near the base of the loop point in the Horizontal Annotation. The loop points can be dragged over each other. The loop point which is on the left then becomes the Loop Start.

- To move both loop points simultaneously, while preserving the loop length, press the Option key while dragging either loop point.

The real time loop adjust feature can be used to rough set the loop points when working with sounds that change over time. If you are working on a stable loop, set the loop points around a region which is relatively stable in pitch, timbre, and loudness. If you are making a loop which includes a vibrato or tremolo, set the loop start and end to include several vibrato cycles.

A more precise way to set the loop points is to use the Smart Auto-Scan feature described below.

UNDO LOOP MOVE

Undo Loop Move is on the edit menu. Undo Loop Move is enabled only after loop points have been dragged in the sound window or moved using the crossfade dialog window. Pressing Undo Loop Move undoes the previous loop move. Pressing a second time redoes the previous loop move. Edits: cut clear, paste, clear unselected, and reverse, when they change the loop positions, will disable Undo Loop Move. If the loop points are not changed by edits, Undo Loop Move remains enabled. Please remember that normal edit Undo recovers any loop points changed by the edit. Using any of the looping tools (except the Crossfade Looper) will disable Undo Loop Move.

LOOP POINTS AND CLIPBOARD

The loop points are copied onto the clipboard. This occurs with a cut or copy if both loop points are included in selection. Valid loops points in clipboard show up as at least one non-zero value for the begin or end loop time (see the Show Clipboard option in the Edit menu). The values displayed in Show Clipboard are relative to the beginning of the selection when the cut or copy was performed. When Paste is used and loop points are in clipboard, they are always pasted into the document.

LOOP POINTS DURING EDITS

The loop points are moved only if all tracks are selected in Cut, Clear, Paste, and Reverse operations. Loop points are always moved for Clear Unselected. The loop points are never moved for Silence and Invert operations.

If both loop points are in an area that is cut from the data, they are deleted. Undo restores the previous loop points.

During Cut and Clear operations, a loop point that is within the selection is moved to the new insertion point. (If this happens to both loop points, the loop is deleted.) Loop points after the selection end point are moved forward, "stuck" to their respective sample.

During Paste operations, where there is a selection, a loop point in the selected region stays at the same number of samples after the selection start point unless the pasted region is smaller than that offset. In that case, it is moved to the first sample after the new pasted region. Loop points after the selection end point are "stuck" to their respective sample. During Paste operations where there is an insertion point, loop points after the insertion point are moved backwards, "stuck" to their respective sample.

During the Clear Unselected operation, loop points in the discarded regions are moved to the first or last sample of the soundfile. Loop points within the selected region are "stuck" to their respective sample.

During the Reverse operation, loop points in the selected region are reversed as if "stuck" to their respective sample.

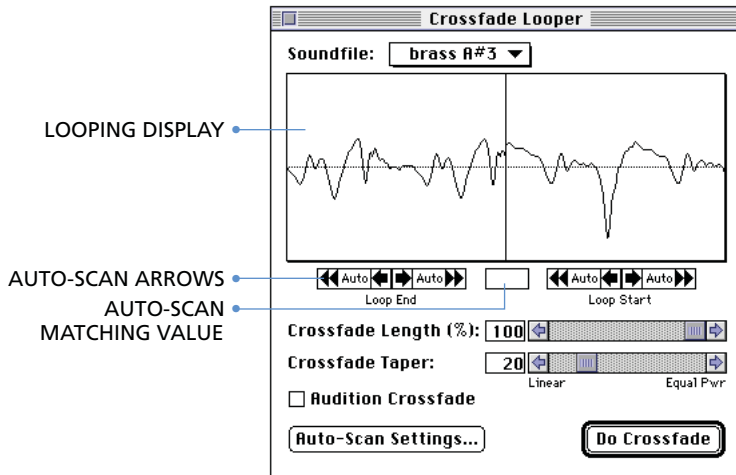
LOOP POINTS IN OTHER FILE FORMATS

Two loop points exist in the AIFF and the Sound Designer I formats, the sustain loop and the release loop. Currently, Infinity only deals with the sustain loop.

In Sound Designer II, the sustain loop is called Loop 1 and the release loop is called Loop 2. Infinity will read only Loop 1 from a Sound Designer file as the sustain loop. When Infinity writes to a Sound Designer II file, all other loops are lost.

THE CROSSFADE LOOPER

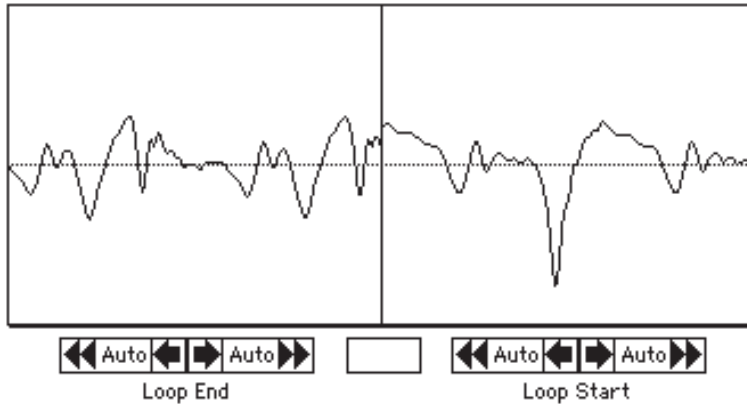
The standard crossfade loop is created by blending part of the attack segment into the end of the loop segment of the soundfile to create a smooth transition back to the loop start. Infinity's Crossfade Looper allows for precise control of the process and contains many enhancements including a scalable looping display, a non-destructive audition feature, fully adjustable crossfade parameters, and an intelligent loop point scanning technology called Smart Auto-Scan. This section explains these features and their controls.



The soundfile can be auditioned while the Crossfade Looper is the active window. Playback works by pressing the space bar or clicking on the Speaker icon. If the cursor is in a text box, playback can be started only with the Speaker Icon. Terminate playback by clicking on the Speaker icon or pressing the space bar. During playback, unusable controls are grayed.

ABOUT SMART AUTO-SCAN

This algorithm searches the soundfile for places where the sound at one loop point matches the sound at the other loop point. The closeness of the match and the number of places in the soundfile that are searched are controllable using the Auto-Scan Settings dialog shown below.



Conventional “auto” functions move the loop points to the nearest zero crossing or the nearest zero crossing with the same slope. Smart Auto-Scan compares the amplitude and the phase of the major harmonics at the loop points and searches until a good match is found. The result is a loop point which always creates good crossfade loops, i.e. without unwanted phase cancellations of the major harmonics. The illustration below shows the result of a Smart Auto-Scan; the alignment of the waveforms is exact. The slopes at the loop interface will blend after the crossfade is complete.

UNDERSTANDING THE DIALOG CONTROLS

Soundfile Menu: The Soundfile Menu shows the name of the last active soundfile. All open soundfiles are listed in the menu. The Crossfade Looper operates on the soundfile which is displayed in the menu.

Looping Display: The Looping Display shows the interface between the loop end and the loop start. The vertical line in the middle of the display shows the interface point. The left half shows the Loop End region, the right shows the Loop Start region. Certain combinations of display resolution and loop point positions will show the loop point in the opposite display. In this case, the loop point marker is shown as a vertical line.

The display resolution can be changed using the Display Scale Arrows on the Palette. The position of the cursor is also shown in the Palette cursor tracking display.

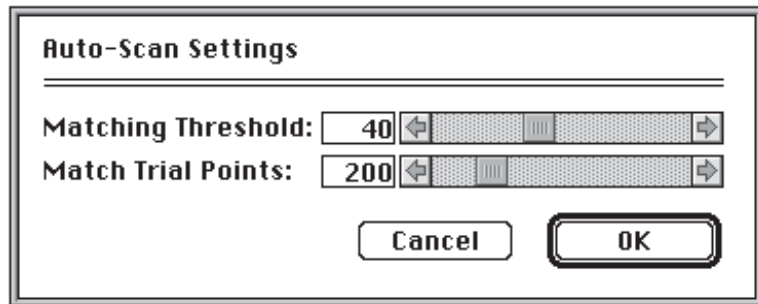
Auto-Scan Arrows: Clicking on a double arrow will cause Infinity to begin searching for a matching loop point in the direction of the arrow. When one is found, the loop point marker is moved to the new location.

Infinity will continue to search until it finds a matching value which is greater than or equal to the value of the Matching Threshold set in the Auto-Scan Settings dialog. If none is found, an audio signal will sound, indicating the end of the search. The loop points will remain in their original locations. Pressing <esc> or <command-period> will terminate the search.

Clicking on a single arrow will move the loop point one pixel in the direction of the arrow. Pressing on a single arrow will cause the loop point to move and slowly accelerate. The longer the arrow is pressed, the faster the loop point will move.

Auto-Scan Matching Value: This displays the matching values found during Auto-Scan searches. Watching this value during the search gives you an idea of how closely the sound at the loop points is matching. Use this information to change the Matching Threshold in the Auto-Scan Settings dialog, if necessary.

Auto-Scan Settings: Clicking on the Auto-Scan Settings command button will bring up the dialog shown below.

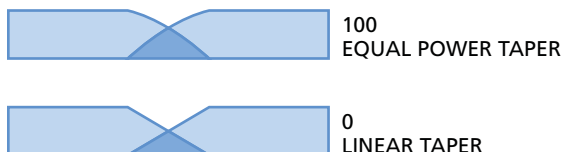


Matching Threshold: The Matching Threshold determines how closely the loop end must match the loop start. The range of the control is from zero to 100. Higher numbers require a closer match. A value of 100 equals a perfect match. Non-periodic sounds typically have lower matching values than periodic sounds.

Match Trial Points: This control changes the number of locations in the soundfile which are searched during an Auto-Scan. The soundfile is first divided by the number of Match Trial Points, then the search begins in the direction of the arrow which was clicked. The range is from 10 to 1000.

Crossfade Length (%): This control adjusts the amount of attack data which is crossfaded with the loop end to create a smooth transition back to the loop start. The range is from 0% to 100% of the attack segment. If the loop segment is shorter than the attack segment, then the crossfade length percentage refers to the amount of the loop data which is effected by the crossfade.

Crossfade Taper: This control adjusts the taper shape of the crossfade that is used to blend the attack and loop segments. It is adjustable from 0 (linear taper) to 100 (equal power taper). It should be adjusted to provide the smoothest transition between the attack and loop segments of the sample. Use equal power taper for string sounds, choruses, and other non-periodic sounds. Linear taper is best used with periodic waveforms like solo instruments.



Linear taper should be used for editing simple periodic waveforms like solo instruments. Equal power taper should be used when editing complex non-periodic waveforms like string sections or noise.

Audition Crossfade: When the Audition Crossfade checkbox is checked, clicking on the Speaker icon (or pressing the space bar) will compute the crossfade loop and play it until the Speaker icon is clicked again. This feature allows you to hear the crossfade without permanently changing the soundfile.

Do Crossfade: Clicking on the Do Crossfade command button executes the crossfade loop process. Pressing <esc> or <command-period> terminates the process.

THE FREEZE LOOPER

The Freeze Looper is an automatic looping algorithm that creates a perfectly seamless loop with no sonic variations. It is excellent for making short loops in relatively periodic sounds like solo instruments, and for “freezing” the timbre of decaying instruments that have clearly defined harmonics like harp, piano, guitar notes, etc.

If the sound to be looped has nearly uniform waveshapes, the Freeze Looper will change the aural qualities of the sound very little and result in a continuous uniform loop. If the Freeze Looper is used on a more complex sound, the resulting loop will sound very different from the original.

The crossfade used in the Freeze Looper is different from a normal crossfade loop in that a portion of the loop segment is blended into the attack rather than a portion of the attack being blended into the loop. This modification of the attack segment occurs every time a Freeze Loop is created.

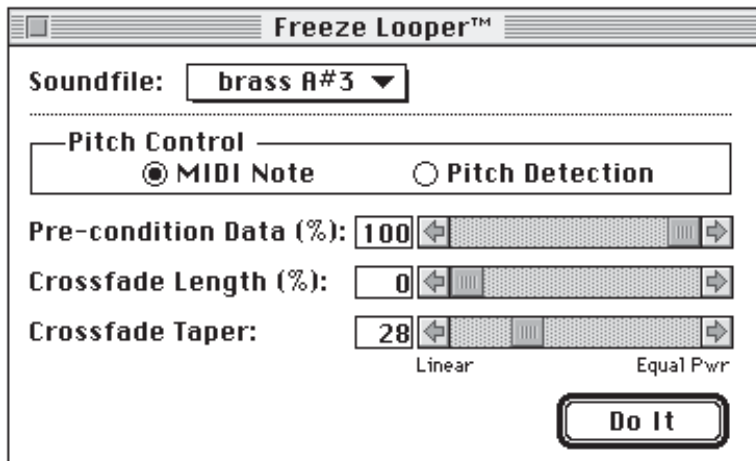
HOW IT WORKS

The Freeze Looper quantizes all the sound energy in the loop to a precise harmonic series, giving a waveform that is perfectly repeated in every cycle. The Freeze Looper outputs a constant pitch equal to the average pitch of the original loop segment.

The Freeze Looper uses the MIDI Note Number to set internal processing parameters. Because these settings effect the results, the MIDI Note Number should be set to the approximate pitch of the sound being looped.

The size of the loop segment is reduced by processing with the Freeze Looper. The loop end point is moved to shorten the loop. The loop start is left where it was before processing. This shortening is necessary because the Freeze Looper uses an FFT algorithm that requires the number of samples in the loop be equal to a power of two. The finished loop is also shortened slightly so that it contains an integral number of waveform periods. So, the output loop end point will be earlier than that which was input.

If the computations cause the loop data to exceed 90% of maximum, the entire file will be scaled so that the loop data are 90% of maximum and thereby do not clip. Thus, data are scaled so no overflow occurs. When scaling occurs, the user is notified with an alert which indicates how much the data was scaled. The scaling information will allow you to maintain volume control of your samples. If you are getting clipping in the fade from the attack segment, check out that the crossfade taper may be set more towards linear taper.



UNDERSTANDING THE DIALOG CONTROLS

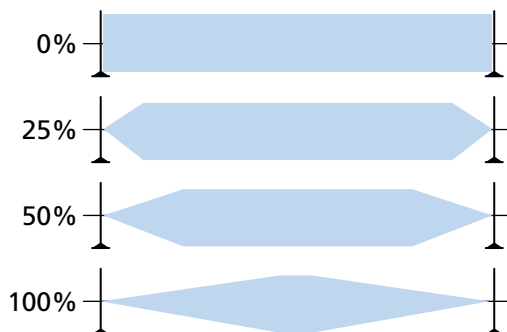
Soundfile Menu: The Soundfile Menu shows the name of the last active soundfile. All open soundfiles are listed in the menu. The Freeze Looper operates on the soundfile which is displayed in the menu.

Pitch Control Radio: Previously, Freeze Looper had Pitch Detection built-in. This means the harmonic series energy distribution and output tuning was determined by an algorithm that detected the exact pitch of the original loop. The MIDI note was used as an initial guide such that if the MIDI note was way off, an incorrect result would occur. This algorithm is used when you select the Pitch Detection radio.

Unfortunately, if you process a short loop (less than 20 cycles), the Pitch Detection algorithm will not be precise and the result may be slightly out of tune. Also, it may be the original sample was out of tune and you wish to tune it up. In these cases, you should select the MIDI Note radio. This will force the harmonic series energy distribution and output tuning to be determined using the MIDI Note number of the sample and the standard that A3 is 440Hz.

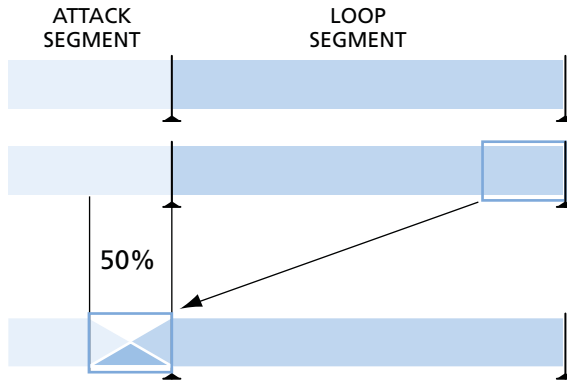
Pre-condition Data (%): The purpose of the Pre-condition Data feature is to taper the data at the loop start and end so that all the energy components in the loop begin and end on zero. The control has a range from 0% to 100%. Any setting larger than a few percent leaves little or no energy at the loop points to cause a click which might inject unwanted noise into the process.

Increasing the Pre-conditioning percentage favors the timbre of the sound near the middle of the loop. Using different values gives different aural qualities to the resulting sound. The diagram below shows the effect of changing the amount of the loop tapered by the Pre-conditioning process. (The taper does not show up in the finished loop.)

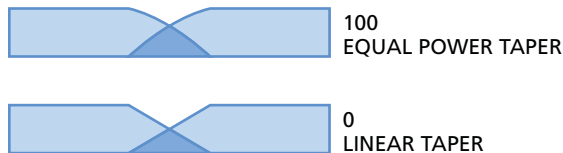


LOOP DATA AMPLITUDE AS A FUNCTION
OF PRE-CONDITION DATA PERCENTAGE

Crossfade Length: This control sets the amount of the loop that is blended with the attack segment. It is variable from 0% to 100% of the attack duration. A large crossfade length will cause the attack to slowly blend into the loop sound. A smaller crossfade length will create a quick but blended transition between the two segments of the sample. A setting of zero creates no crossfade.



Crossfade Taper: This control adjusts the taper shape of the crossfade that is used in the attack/loop merge. It is adjustable from 0 (linear taper) to 100 (equal power taper). It should be adjusted to provide the smoothest transition between the attack and loop segments of the sample.

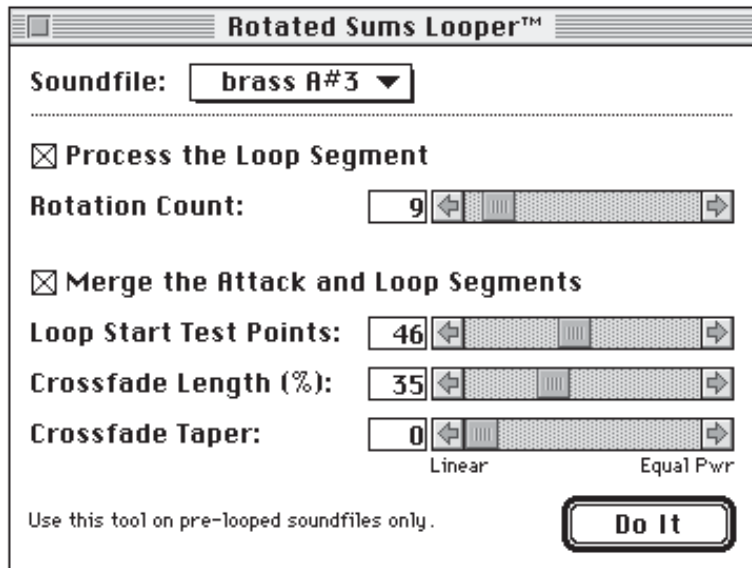


Linear taper should be used for editing simple periodic waveforms like solo instruments. Equal power taper should be used when editing complex non-periodic waveforms like string sections or noise.

Do It: Clicking on the Do It command button executes the Freeze looping process. Pressing <esc> or <command-period> terminates the process.

THE ROTATED SUMS LOOPER

The Rotated Sums Looper creates seamless loops by randomly distributing sonic irregularities throughout the loop. It is very effective with sounds that have non-periodic waveforms like chorus, string orchestra, complex synth patches, sounds that include noise, and so on.



NOTE: Sounds must be pre-looped before using the Rotated Sums Looper. Any clicks, pops or other irregularities will be distributed throughout the loop and show up as noise. Use the Crossfade Looper to smooth out the loop data before processing with the Rotated Sums Looper.

Use the Smart Auto-Scan feature of the Crossfade Looper to place the loop points before processing with the Rotated Sums Looper. This will insure that the resulting loop will more closely match the tone color of the original data.

If the computations cause the loop data to exceed 90% of maximum, the entire file will be scaled so that the loop data are 90% of maximum and thereby do not clip. Thus, data are scaled so no overflow occurs. When scaling occurs, the user is notified with an alert which indicates how much the data was scaled. The scaling information will allow you to maintain volume control of your samples. If you are getting clipping in the fade from the attack segment, check out that the crossfade taper may be set more towards linear taper.

HOW IT WORKS

In a crossfade loop, all of the different frequencies in the sound make a transition from one phase to another during the crossfade. This causes different combinations of constructive and destructive interference at the location of the crossfade. Because this happens for all frequencies at the same point in the loop, a noticeable “bump” is produced by the crossfade looping process.

The Rotated Sums Looper creates a seamless loop by adding together many random rotations of the input loop, thereby distributing the irregularity of the crossfade bump throughout the loop. This is a time domain solution to the crossfade bump problem in complex sounds. The more rotations that are chosen, the better the results, up to a limit that exists because of the natural variations in the sound. The Rotated Sums Looper does not change the length of the loop.

Because the Rotated Sums Looper chooses random amounts by which to perform loop rotations, different trials using the same input sound will yield different results.

UNDERSTANDING THE DIALOG CONTROLS

The Rotated Sums Looper divides the looping process into two distinct phases; processing the loop data, and merging the attack and loop segments. These can be controlled separately and executed at different times.

With both phases of the process enabled, the loop data and the attack data are altered every time the Rotated Sums process is run. It is sometimes desirable to disable the attack/loop merge phase of the process, and do several takes of processing the loop data without merging the loop with the attack. This saves time and allows for selecting the best sounding loop before the attack segment is altered by merging. After the best sounding loop is found, the merge can be done separately by disabling the loop processing phase.

Soundfile Menu: This menu has the same function as in the other loopers.

Process The Loop Segment: Check this item if you wish to apply the Rotated Sums process to the data inside loop. When this item is not checked, the slider labeled Rotation Count is inactive and no change will be made to the loop data.

Rotation Count: The Rotation Count is the number of times that the loop data will be rotated and added in the accumulator during the loop processing phase. The range of the control is from 1 to 100. More rotations give better results, up to a point, but take more time.

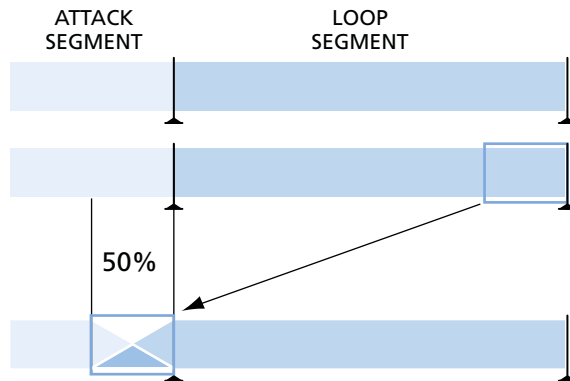
Merge The Attack And Loop Segments: Checking this item enables the second phase of the looping process; merging the loop segment with the attack segment of the sample. Enable this part of the process only when you want to crossfade the loop data with the attack segment of the sound. When this item is not checked, the sliders labeled Loop Start Test Points, Crossfade Length, and Crossfade Taper are inactive.

When Merge the Attack and Loop Segments is checked in the dialog, Infinity will automatically search a number of points in the loop segment to find the point which most closely correlates to the end of the attack segment. The number of test points that are tried during the search can be controlled using the Loop Start Test Points control. Processing time increases as more points are tried.

When the best merge point is found, Infinity rotates the data in the loop until the merge point is located at the loop start. The loop is then merged with the attack segment. The type of crossfade that is used only modifies the attack segment. The length and taper of the crossfade that is used to merge the two segments can be controlled.

Loop Start Test Points: This controls the number of points that are tested during the search phase of the merging process. It is variable from 0 to 100 points. A large number of test points results in a better attack/loop merge but will also increase the processing time.

Crossfade Length: This control sets the amount of the loop that is blended with the attack segment. It is variable from 0% to 100% of the attack duration. A large crossfade length will cause the attack to slowly blend into the loop sound. A smaller crossfade length will create a quick but blended transition between the two segments of the sample. A setting of zero creates no crossfade.



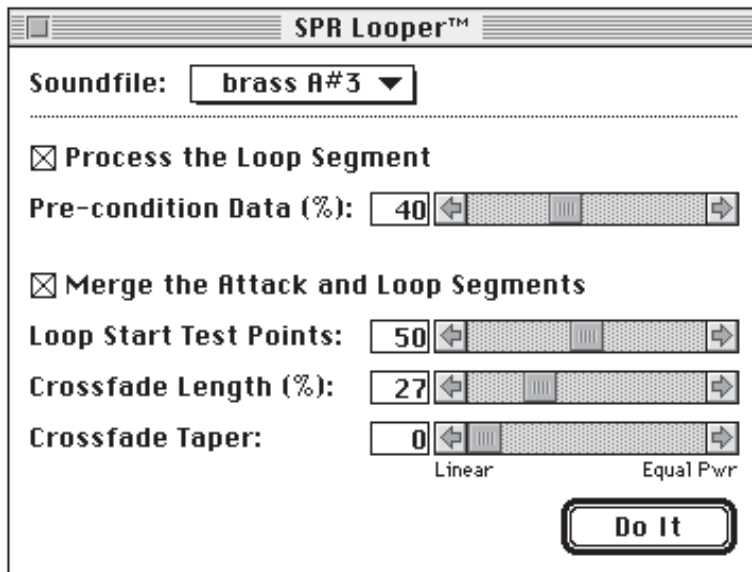
Crossfade Taper: This control adjusts the taper shape of the crossfade that is used in the attack/loop merge. It is adjustable from 0 (linear taper) to 100 (equal power taper). It should be adjusted to provide the smoothest transition between the attack and loop segments of the sample.



Linear taper should be used for editing simple periodic waveforms like solo instruments. Equal power taper should be used when editing complex non-periodic waveforms like string sections or noise.

THE SPR LOOPER

The SPR Looper uses a patented process which randomizes the phase of each spectral component in the loop. SPR means Spectral Phase Randomization. It is excellent for looping sounds which have very non-periodic waveforms. It is especially good at looping complex sounds which require exact preservation of the original tone color like string sections, choirs, and so on.



HOW IT WORKS

The SPR Looper is similar to the Rotated Sums Looper in that it rotates loop data a random amount through the loop. The difference with the SPR Looper is that it treats every constituent frequency of the loop separately. This approach is more thorough than the Rotated Sums Looper. Because the SPR Looper rotates each spectral component randomly, different trials using the same input sound will yield different results.

If the computations cause the loop data to exceed 90% of maximum, the entire file will be scaled so that the loop data are 90% of maximum and thereby do not clip. Thus, data are scaled so no overflow occurs. When scaling occurs, the user is notified with an alert which indicates how much the data was scaled. The scaling information will allow you to maintain volume control of your samples. If you are getting clipping in the fade from the attack segment, check out that the crossfade taper may be set more towards linear taper.

***Performance Note:** Sometimes the Rotated Sums Looper and the SPR Looper have an undesirable effect on the output loop. The energy of any transient sound in the loop will be distributed evenly over the length of the loop. Clicks and pops will be output as relatively steady state noise.*

For example, when looping a close miked 'cello, the bow rasp, which normally occurs in phase with the rest of the waveform, is distributed randomly across the loop. The result sounds like the 'cello mixed with tape hiss. We recommend that these two tools be used only with instrumental groups or other random non-periodic waveforms.

UNDERSTANDING SPR CONTROLS

The SPR Looper divides the looping process into two distinct phases; processing the loop data, and merging the attack and loop segments. These can be controlled separately and executed at different times.

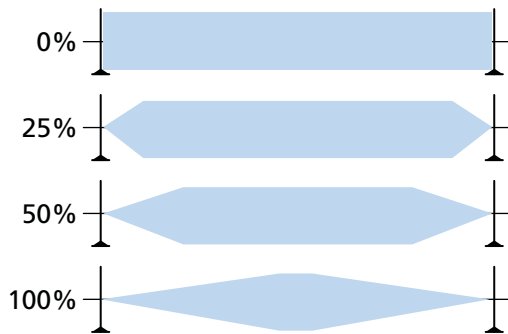
With both phases of the process enabled, the loop data and the attack data are altered every time the SPR process is run. It is sometimes desirable to disable the attack/loop merge phase of the process, and do several takes of processing the loop data without merging the loop with the attack. This saves time and allows for selecting the best sounding loop before the attack segment is altered by merging. After the best sounding loop is found, the merge can be done separately by disabling the loop processing phase.

Soundfile Menu: This menu has the same function as in the other loopers.

Process The Loop Segment: Check this item if you wish to apply the SPR process to the data inside loop. When this item is not checked, the slider labeled Pre-condition Data (%) is inactive and no change will be made to the loop data.

Pre-condition Data (%): The Freeze Looper and the SPR Looper both use the Pre-condition Data feature before processing the loop data. The purpose of this feature is to taper the data at the loop start and loop end so that all the energy components in the loop begin and end on zero. The control has a range from 0% to 100%. Any setting larger than a few percent leaves little or no energy at the loop points to cause a click which might inject unwanted noise into the process.

Increasing the Pre-conditioning percentage favors the timbre of the sound near the middle of the loop. Using different values gives different aural qualities to the resulting sound. The diagram below shows the effect of changing the amount of the loop tapered by the Pre-conditioning process. (The taper does not show in the finished loop.)



LOOP DATA AMPLITUDE AS A FUNCTION OF PRE-CONDITION DATA PERCENTAGE

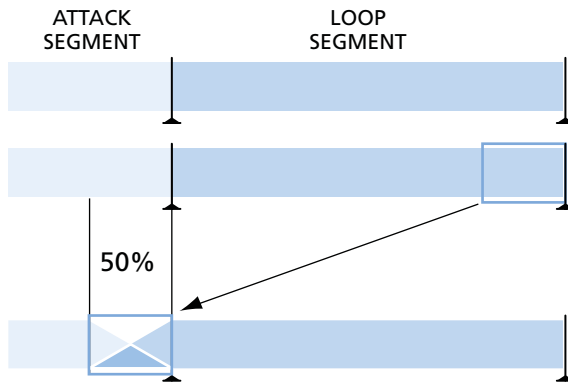
Merge The Attack And Loop Segments: Checking this item enables the second phase of the looping process; merging the loop segment with the attack segment of the sample. Enable this part of the process only when you want to crossfade the loop data with the attack segment of the sound. When this item is not checked, the sliders labeled Loop Start Test Points, Crossfade Length, and Crossfade Taper are inactive.

When Merge the Attack and Loop Segments is checked in the dialog, Infinity will automatically search a number of points in the loop segment to find the point which most closely correlates to the end of the attack segment. The number of test points that are tried during the search can be controlled using the Loop Start Test Points control. Processing time increases as more points are tried.

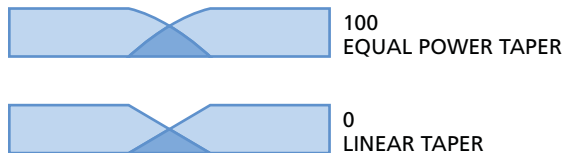
When the best merge point is found, Infinity rotates the data in the loop until the merge point is located at the loop start. The loop is then merged with the attack segment. The type of crossfade that is used only modifies the attack segment. The length and taper of the crossfade that is used to merge the two segments can be controlled.

Loop Start Test Points: This controls the number of points that are tested during the search phase of the merging process. It works as in the Rotated Sums Looper.

Crossfade Length: This control sets the amount of the loop that is blended with the attack segment. It is variable from 0% to 100% of the attack duration. A large crossfade length will cause the attack to slowly blend into the loop sound. A smaller crossfade length will create a quick but blended transition between the two segments of the sample. A setting of zero creates no crossfade.



Crossfade Taper: This control adjusts the taper shape of the crossfade that is used in the attack/loop merge. It is adjustable from 0 (linear taper) to 100 (equal power taper). It should be adjusted to provide the smoothest transition between the attack and loop segments of the sample.



Linear taper should be used for editing simple periodic waveforms like solo instruments. Equal power taper should be used when editing complex non-periodic waveforms like string sections or noise.

THE SYNTHESIS LOOPER

The Synthesis Looper tool can be used to selectively smooth out lumpy sounding frequencies in chorused and ensemble sounds. The new sound is more stable and loop repetitions are less noticeable. This control is achieved by allowing you to specify the sidelobe widths of the harmonic series as a function of frequency. You can control the sidelobe widths to continuously range from the wide sidelobes of the original ensemble sound to the zero width sidelobes of the Freeze Looper.

This tool is appropriately used on chorused and ensemble sounds where the instruments are playing the same notes or notes in octaves. This tool can also be used as a sophisticated chorusing effect, to generate an ensemble sound from a solo instrument sound.

This tool is designed for use on monaural sounds. Stereo sounds can be processed, however, the phase relationships that determine stereo positioning will be lost. The resulting stereo sound will have a random phase relationship, yielding a distributed spatial quality.

COMMENTARY

There are two classifications of sound. The first classification is solo instruments (sans vibrato). These generate uniformly repeated waveforms. A generic crossfade loop can be used to loop these sounds, however the best results will be had using the Freeze Looper. The second classification is all other sounds. These are sounds that change over time. Crossfade loop quality becomes worse as the sounds become more complex. The SPR Looper and Rotated Sums are most often used to loop these sounds. However, a basic problem remains: No matter how perfectly the end of the loop is connected to the beginning of the loop, the sound in between is still of a quality that changes over time. Hence, you will always “hear the loop” because the varying sound repeats itself, over and over again.

This quality of hearing the loop because the varying sound repeats itself, is sometimes judged as a poor quality loop. In this case, the loop-maker is really wanting for a tool to change the sound to one which does not have as much temporal variation.

It is for this reason we have created the Synthesis Looper. This tool gives you control over that aspect of the sound which creates temporal variation: the harmonic sidelobe widths.

SOME THEORY

Temporal variation occurs when two sinusoidal frequencies are not exactly in tune. For example, if one sine wave generator is tuned to 1000 Hz (Hertz or cycles per second) and another is tuned to 1001 Hz, a 1 Hz “beat note” will be produced. A temporal variation will also occur when there is any integer relation between the frequencies involved: if one sine wave generator is tuned to 500 Hz and another is tuned to 1001 Hz, a 1 Hz beat note will also be produced.

Many solo instruments (sans vibrato) generate uniformly repeated waveforms. These sounds contain energy at the fundamental pitch, and overtone pitches, which have frequencies that are EXACT integer multiples of the fundamental frequency. These frequencies are called the harmonic series. For example, an oboe play A-440 generates energy at frequencies 440 Hz, 880 Hz, 1320 Hz, 1760 Hz, etc. Zero energy is generated at any other frequencies (except for perhaps white noise generated at all frequencies by air rushing through the instruments).

The second chair oboe, who does not have as much precision in pitch, might play an A-440 out of tune, yielding energy at frequencies 441 Hz, 882 Hz, 1323 Hz, 1764 Hz, etc. Notice how the differences in pitch increase proportionately with the harmonic index (1, 2, 3, etc.). These two players, playing together will generate a sound with a 1 Hz beat note (a 1 second repeated variation in sound).

Now suppose there were 100 oboes playing together (with vibratos). What would the energy as a function of frequency look like? Remember the bell-shaped curve? Draw a line in the dirt, stand back and throw 100 pennies at the line. Most of them will land close to the line and some of them will land far from the line. The number of pennies per inch will be highest at the line, and lower farther from the line. If you draw a graph of the number of pennies per inch verses distance, you will plot a bell-shaped curve. This is technically described as a Normal Probability Density Function or Normal PDF. If there were 100 oboes players, each playing their version of A- 440 together (with or without vibratos) , the energy distribution of the fundamental pitch as a function of frequency would form a Normal PDF around 440 Hz. The width of the distribution would depend on how good the players were. If they were very good players playing in tune and no vibratos, the width would narrow. If they were unskilled and/or added vibratos, the width would be wider. Since this Normal PDF is energy as a function of frequency, concentrated at a single frequency, it is described as a frequency with Sidelobe energy.

These 100 players would also generate energy at the harmonic frequencies. The energy distribution of the harmonics as a function of frequency would also form a Normal PDF. As with two players in the example above, these differences in pitch increase proportionately with the harmonic index. Add to this some elementary arithmetic about playing in octaves and we get the following rules:

- **Rule 01:** An ensemble composed of instruments playing the same note or notes in octaves will form an energy distribution as a function of frequency having energy at the fundamental frequency and integer multiples of the fundamental frequency.
- **Rule 02:** The energy distribution at the fundamental frequency will form a Normal PDF whose sidelobe width depends on the intonation quality of the players: in-tune players will generate a narrower sidelobes than players who are more out of tune.
- **Rule 03:** Each harmonic will have an energy distribution at its respective harmonic frequency that forms a Normal PDF. The sidelobe width of each harmonic will increase in proportion to frequency with the first harmonic being two times wider than the fundamental, etc.

There is a final rule for which the authors do not take credit: Energy cannot be created or destroyed. Nor can energy change its frequency. Acoustic energy can change its form to heat. None the less, if you are listening to two flutes playing slightly out of tune, then at the moment at which there is silence at your ears, there is also at least one loud sound somewhere else in the room. Be advised that by moving your head, you will change what you hear.

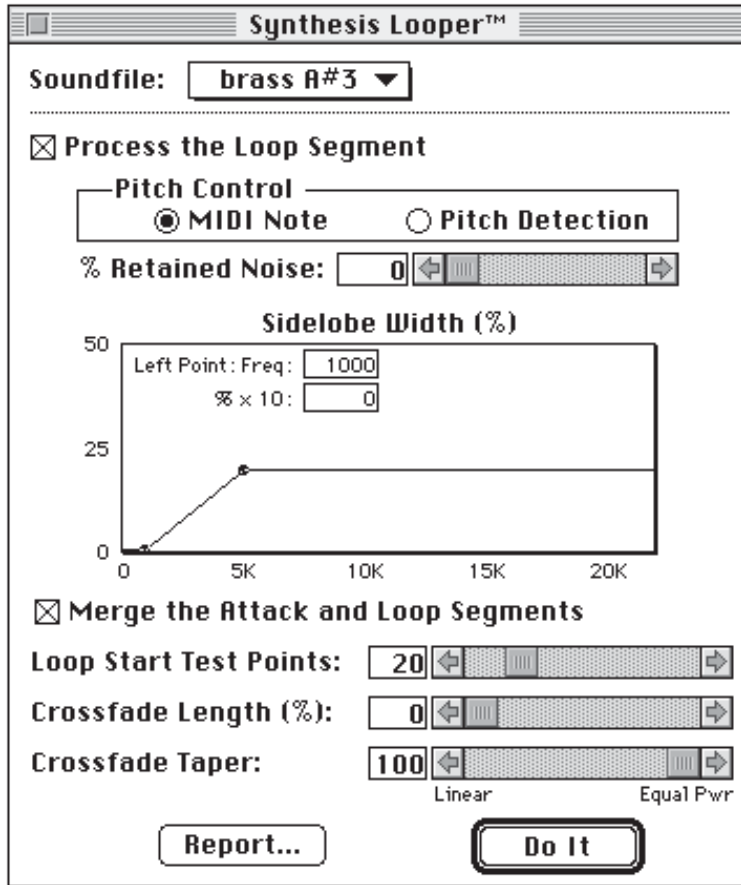
HOW IT WORKS

The Synthesis Looper tool calculates the energy at each fundamental and harmonic frequency and allows you to specify the width of the sidelobes the resulting looped sound will have. In this way, you could take the lumpy sound of a highly out-of-tune ensemble and change it to the smoother sound of a higher quality ensemble, making the variations in the loop repetition less noticeable.

There are other possibilities, too. For example, you can change the first few harmonics to focus the energy in the sidelobes to zero. This will stabilize the sound and still provide the warmth of the ensemble at the higher frequencies. You could also, in the extremes, make a soloist into an ensemble or an ensemble into a soloist.

You can view the Synthesis Looper window by selecting it from the Looping Tools menu item: First, you can use the pop-up menu at the top of the dialog to select any of the open soundfiles for processing or playback. You can also press the space bar or the speaker icon in the palette window to play back the selected sound.

This tool performs one, or both, of two functions chosen using the checkboxes Process the Loop Segment and Merge the Attack and Loop Segments. The latter function is the same as that in the SPR Looper and Rotated Sums Looper. It allows you to combine a looped sound into the attack portion of the sound. You can use both tools simultaneously or you can first create a satisfactory loop with Process the Loop Segment and later, merge it using Merge the Attack and Loop Segments.



If the computations cause the loop data to exceed 90% of maximum, the entire file will be scaled so that the loop data are 90% of maximum and thereby do not clip. Thus, data are scaled so no overflow occurs. When scaling occurs, the user is notified with an alert which indicates how much the data was scaled. The scaling information will allow you to maintain volume control of your samples. If you are getting clipping in the fade from the attack segment, check out that the crossfade taper may be set more towards linear taper.

UNDERSTANDING SYNTHESIS LOOPER CONTROLS

The Synthesis Looper divides the looping process into two distinct phases; processing the loop data, and merging the attack and loop segments. These can be controlled separately and executed at different times.

With both phases of the process enabled, the loop data and the attack data are altered every time the Synthesis Looper process is run. It is sometimes desirable to disable the attack/loop merge phase of the process, and do several takes of processing the loop data without merging the loop with the attack. This saves time and allows for selecting the best sounding loop before the attack segment is altered by merging. After the best sounding loop is found, the merge can be done separately by disabling the loop processing phase.

Process the Loop Segment: In the Process the Loop Segment section, there are three controls. The Pitch Control allows you to choose MIDI Note or Pitch Detection. Selecting the MIDI Note radio button will cause the harmonic series frequencies to be determined using the MIDI note. If the loop data is not in tune, incorrect harmonic amplitudes will be computed. Selecting the Pitch Detection radio button will cause the harmonic series frequencies to be determined by the loop data using the MIDI note as a guide. This is accurate unless the data is too short.

% Retain Noise: The second control is the % Retain Noise slider and number box. Some sounds have a characteristic background noise, that when removed, makes the sound less natural. An example of this would be a cello section, where the action of the bow on the string adds noise at all frequencies. Other sounds may have background noise that is undesirable. If this control set to 100, the noise floor on the output will be the same as on the input. If it is set to 0, the noise floor will be completely removed. Intermediate values set the noise floor (energy) proportionately.

Sidelobe Width (%): The third control is the Sidelobe Width (%) graph. This specifies the output sidelobe width as a percentage of the fundamental frequency. This percentage is multiplied by the fundamental frequency to get a frequency variation that is used as the standard deviation of the Normal PDF energy distribution. You use this graph by dragging either of the points to a desired location to specify the desired sidelobe widths. Note that whenever this window is selected and your cursor is over this graph, you can view the coordinates of the cursor location on the Status Display Area palette window. The sound of the result is very sensitive to the position of the left point. For this reason, the two text boxes are provided so that you can type in numbers, giving you more accuracy than dragging the point on the graph. Note that the sidelobe width is specified as 10 times the percentage, that is, typing a 7 means .7 %.

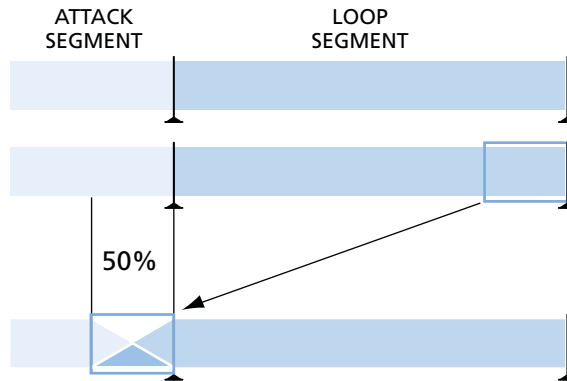
Merge The Attack And Loop Segments: Checking this item enables the second phase of the looping process; merging the loop segment with the attack segment of the sample. Enable this part of the process only when you want to crossfade the loop data with the attack segment of the sound. When this item is not checked, the sliders labeled Loop Start Test Points, Crossfade Length, and Crossfade Taper are inactive.

When Merge the Attack and Loop Segments is checked in the dialog, Infinity will automatically search a number of points in the loop segment to find the point which most closely correlates to the end of the attack segment. The number of test points that are tried during the search can be controlled using the Loop Start Test Points control. Processing time increases as more points are tried.

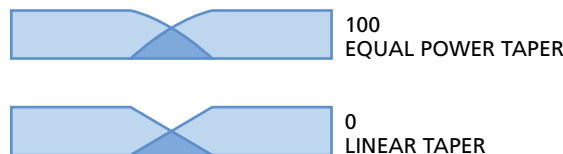
When the best merge point is found, Infinity rotates the data in the loop until the merge point is located at the loop start. The loop is then merged with the attack segment. The type of crossfade that is used only modifies the attack segment. The length and taper of the crossfade that is used to merge the two segments can be controlled.

Loop Start Test Points: This controls the number of points that are tested during the search phase of the merging process. It works as in the Rotated Sums Looper.

Crossfade Length: This control sets the amount of the loop that is blended with the attack segment. It is variable from 0% to 100% of the attack duration. A large crossfade length will cause the attack to slowly blend into the loop sound. A smaller crossfade length will create a quick but blended transition between the two segments of the sample. A setting of zero creates no crossfade.



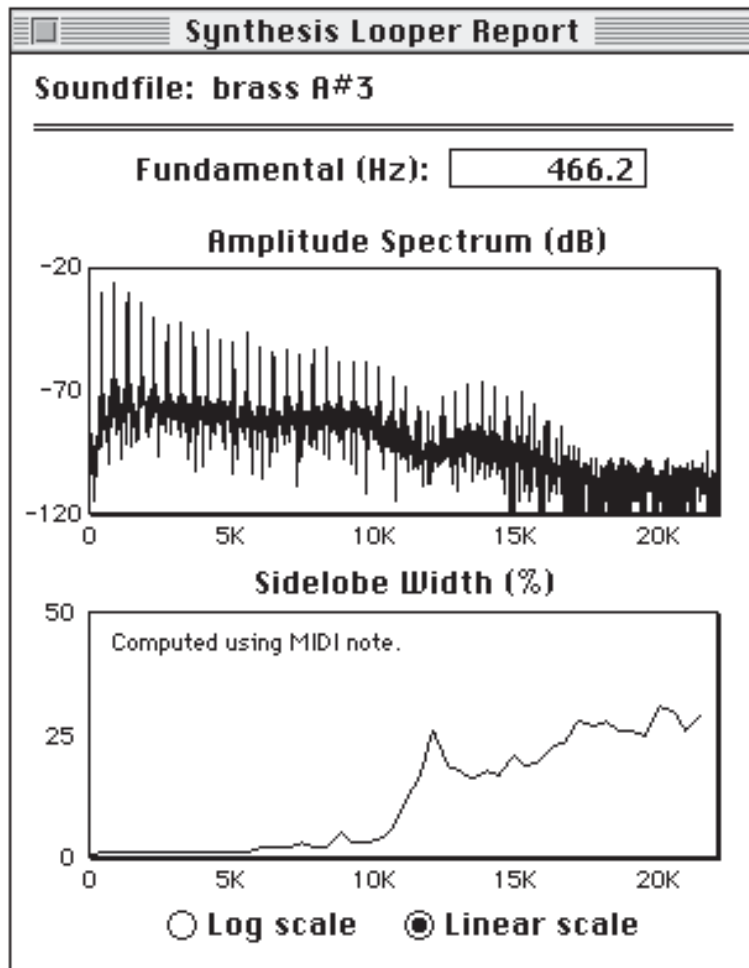
Crossfade Taper: This control adjusts the taper shape of the crossfade that is used in the attack/loop merge. It is adjustable from 0 (linear taper) to 100 (equal power taper). It should be adjusted to provide the smoothest transition between the attack and loop segments of the sample.



Linear taper should be used for editing simple periodic waveforms like solo instruments. Equal power taper should be used when editing complex non-periodic waveforms like string sections or noise.

THE SYNTHESIS LOOPER REPORT

But how should you set the Sidelobe Width (%) control? How should you choose MIDI Note or Pitch Detection? The Synthesis Looper Report will assist you in deciding these issues. Select the Report... button at the bottom of the Synthesis Looper window. This causes the Synthesis Looper Report window to be generated:



This window “belongs” to the soundfile. The data in the window is computed from the soundfile’s loop data at the moment you pressed the Report... button. You may generate as many reports for a soundfile as you desire. You may close a report by clicking the “go-away” box in the top left. If you close a soundfile, all the reports for that soundfile are closed automatically. You can also press the space bar or the speaker icon in the palette window to play back the soundfile to which this report belongs. Remember though, if you modified the soundfile after generating the report, you will hear the new soundfile.

Fundamental (Hz): The Fundamental (Hz) number is the fundamental frequency. It is computed using the MIDI Note number or the Pitch detection method selected when you pressed Report....

Amplitude Spectrum (dB): The Amplitude Spectrum (dB) is the amplitude (square root of energy) of the loop expressed in Decibels. (Decibels are $20 * \log(\text{value})$. Every 6 dB is a approximately factor of 2. in amplitude.)

Sidelobe Width (%): The Sidelobe Width (%) graph shows the sidelobe width computed from the loop as a percentage of the fundamental frequency. The scale presented is the same as the Sidelobe Width (%) input graph in the tool. This sidelobe width is computed for the fundamental and each harmonic as the standard deviation of the energy distribution around the fundamental or harmonic. It is expressed as a percentage of the fundamental frequency.

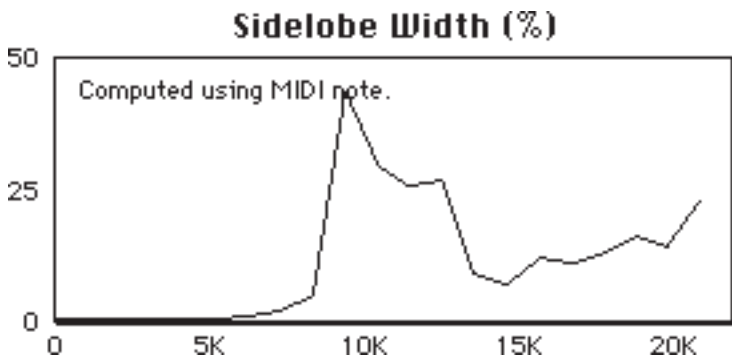
Note that whenever this window is selected and your cursor is over the Amplitude Spectrum (dB) or Sidelobe Width (%) graphs, you can view the coordinates of the cursor location on the Status Display Area palette window.

Log Scale or Linear Scale: At the bottom of the report are radio buttons labeled Log Scale and Linear Scale. The Log scale allows more resolution at the low frequencies where the audio content is the most important. The Linear scale should show an approximate linear increase in sidelobe width as a function of frequency for ensembles where the instruments are playing the same notes (with octaves allowed).

The report above is shown using the log scale. The report below is the same report shown in linear scale. The data is that of a string section. Notice how the sidelobe width increases approximately linearly with frequency. Use the linear scale on a report of unprocessed loop data to help you establish what settings to use in the Synthesis Looper Sidelobe Width (%)graph.

In this graph, you can easily see that the harmonics have frequencies which are integer multiples of the fundamental frequency. Also, the Normal PDF distributions in energy show up as narrow concave downward hyperbola (in Decibels verses linear frequency). Notice how the higher frequency harmonics are wider. These sidelobe widths are proportional to frequency.

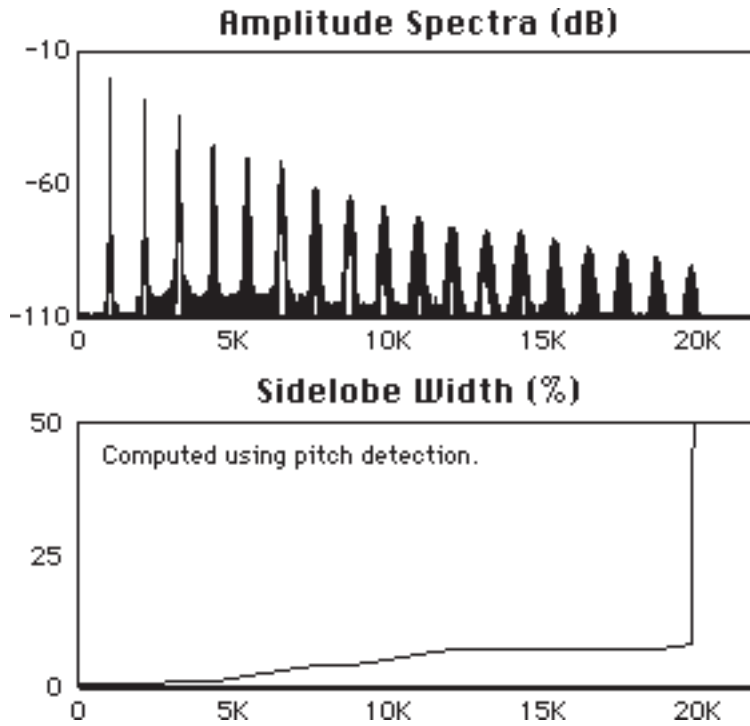
Note that the above Sidelobe Width (%) graphs were generated using pitch detection (see the note enclosed in the graph). Pitch detection was used here because this note was slightly out of tune. You can determine this because if the MIDI Note were used instead, the graph would appear:



Normally, the sidelobe widths do not exceed 30% unless the expected harmonic frequency is not aligned to the actual harmonic frequency. This will happen when the MIDI Note is used for pitch control and the note is out of tune.

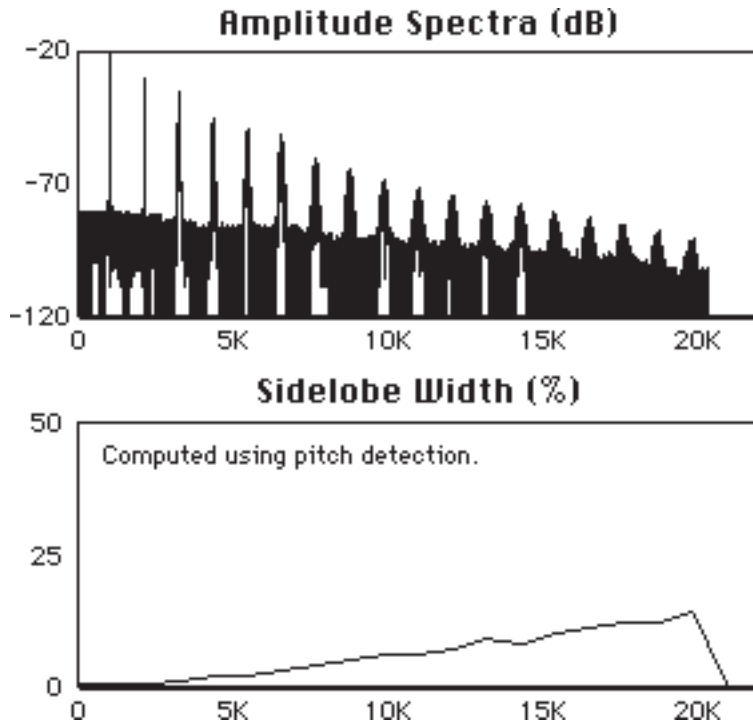
The following graphs show the result of processing the data with the Retain Noise checkbox clear and checked, respectively.

Retain Noise unchecked:



Notice how the Sidelobe Width (%) becomes flat above 12k Hz, as was specified in the Sidelobe Width (%) of the Synthesis Looper tool.

With Retain Noise checked:



Compare the above graph to the report of the unprocessed data and notice how the noise floor has been computed to vary with frequency similar to the initial data. Also notice that the presence of noise causes the Sidelobe Width (%) computation to be slightly excessive in frequencies where the noise is nearly the same amplitude as the harmonic.

HOW TO LOOP A SOUND:

First, set the sample's MIDI Note to as near the correct pitch as possible using A3 = 440Hz. Second, set the loop points. Note that the Synthesis Looper will make the loop smaller to the next power of two (approximately). Open the Synthesis Looper tool. Set the Pitch Control radio to MIDI Note if you believe your sample should be in tune according to the A3 = 440Hz equal tempered scale. If not, set it to Pitch Detection.

Next, press the Report... button. On the report, select Log scale and place your cursor precisely at the fundamental peak. Check that the Fundamental (Hz): number is within a pixel's accuracy of the Hertz number on the floating palette. If it is not, then you are going to have pitch problems. If you had checked MIDI Note you either specified the wrong note or the data is not tuned to A3 = 440Hz equal tempered scale. Either re-specify the note or use Pitch Detection. If you had checked Pitch Detection, then your MIDI note may be off. The MIDI note was used as an initial guide by the pitch detection algorithm. If the MIDI note is way off, an incorrect result would occur. Unfortunately, if you process a short loop (less than 20 cycles), the Pitch Detection algorithm will not be precise and the result may be slightly out of tune.

Next, review the Sidelobe width (%) graph. Normally, the sidelobe widths do not exceed 30% unless the expected harmonic frequency is not aligned to the actual harmonic frequency. This will happen when the MIDI Note is used for pitch control and the note is out of tune. In this case, use Pitch Detection.

Next, go back to the Synthesis Looper tool and check Retain Noise.

Finally, specify a Sidelobe Width (%) curve that resembles the one on your report and press the Do It button. This is a quality control step and should result in a loop very close to your original sound.

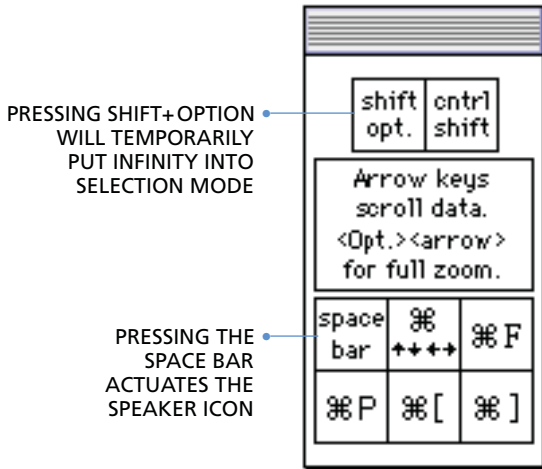
Now the fun begins: Select Retain Noise as desired. Experiment by carefully adjusting the Sidelobe Width (%) curve to have less width than the original and press Do It. (Remember to Undo between trails.)

A FINAL NOTE

Remember this: Ensemble sounds vary. When you make a loop you will hear the variations repeat, over, and over, and over, and over... You can make a static sound that doesn't vary (use zero Sidelobe Widths or use Freeze Looper) but that won't sound like an ensemble. You can't have it both ways.

However, this tool lets you determine the trade-off, giving you unprecedented control of the of the result.

Appendix A: Keyboard Shortcuts



The keyboard command equivalents listed below are grouped by function. For instance, commands that scroll the data in the Mainview are grouped together. Keys that have names, like Shift, will be shown as <shift>. Two named keys that are used together look like <shift-option>.

Pressing on the Palette and using Balloon Help will show many keyboard shortcuts.

SCROLLING DATA

TO:

Scroll Mainview to the right
 Scroll Mainview to the left
 Scroll Mainview up
 Scroll Mainview down
 Scroll right by one windowful
 Scroll left by one windowful
 Scroll up by one windowful
 Scroll down by one windowful
 Scroll full left and center
 Scroll Mainview to the loop start
 Scroll Mainview to the loop end
 Full zoom to loop start function
 Full zoom to loop end function

KEY COMMAND:

<right arrow>
 <left arrow>
 <up arrow>
 <down arrow>
 <shift-right arrow>
 <shift-left arrow>
 <shift-up arrow>
 <shift-down arrow>
 <home>
 ⌘ [
 ⌘]
 <option> [
 <option>]

CHANGING DISPLAY SCALING

TO:	KEY COMMAND:
Double vertical display resolution	⌘ <up arrow>
Halve vertical display resolution	⌘ <down arrow>
Double horizontal display resolution	⌘ <right arrow>
Halve the horizontal display resolution	⌘ <left arrow>
Go to minimum vertical display resolution	<option-down arrow>
Go to maximum vertical display resolution	<option-up arrow >
Go to minimum horizontal display resolution	<option-left arrow>
Go to maximum horizontal display resolution	<option-right arrow>
Fit the current selection in the Mainview	⌘-F
Full zoom to the loop start function	<option> [
Full zoom to the loop end function	<option>]

FILE COMMANDS

TO:	KEY COMMAND:
Create a new soundfile	⌘-N
Open an existing soundfile	⌘-O
Close a window	⌘-W
Save a soundfile (update the backup file)	⌘-S
Get information about a soundfile	⌘-I
Quit Infinity	⌘-Q

EDIT COMMANDS

TO:	KEY COMMAND:
Cut data from the soundfile into the Clipboard	⌘-X
Copy data into the Clipboard	⌘-C
Paste data from the Clipboard into the soundfile	⌘-V
Clear data from the soundfile (delete)	⌘-B or <delete>
Clear Unselected data from the soundfile	⌘-T
Mix	⌘-M
Create or delete loop points	⌘-P
Select the loop data	⌘-L
Select all the data in the soundfile	⌘-A
To drag both loop points simultaneously	<option> drag

LOOPING TOOLS

TO:	KEY COMMAND:
Call the SPR Looper	⌘-1
Call the Rotated Sums Looper	⌘-2
Call the Freeze Looper	⌘-3
Call the Crossfade Looper	⌘-4

PALETTE

TO:	KEY COMMAND:
Momentarily put the active window in Zoom Mode	<control-shift>
Momentarily put the active window in Selection Mode	<shift-option>
Play/Stop the active soundfile	<space bar>
Create or delete loop points	⌘-P
Scroll the Mainview to the loop start	⌘- [
Scroll the Mainview to the loop end	⌘-]
Fit the current selection in the Mainview	⌘-F

OTHER

To cancel playback and many processing operations <esc> or ⌘-. (period)

Index

Symbols

% Retain Noise 59

A

Action Icons 11
 Display Scale 11
 Fit Selection 11
 Loop End 11
 Loop Points: 11
 Loop Start 11
 Speaker 11
Adjusting Blend Parameters 33
Amplitude Spectrum (dB) 62
Antares Audio Technologies ii
Audio IFF 37
Authorization ix
Auto-Scan Arrows 42
Auto-Scan Matching Value 43
Auto-Zero 30, 37

B

Backup Files 14
Blend Parameters 37
Blend Taper 37
Blend Time 37
Blending 36

C

Change Gain... 28
Changing Display Scaling 68
Changing The Display Resolution 17, 26
Clear 26
Clear Beyond Loop End 26
Clear Unselected 26
Clipboard 37
Copy 25, 26
Create Backup Files 36, 37
Create/Delete Loop Points 28
Crossfade 33

Crossfade Length 47, 50, 54, 60
Crossfade Length (%) 43
Crossfade Looper 2, 41
Crossfade Loops 5
Crossfade Taper 44, 47, 51, 54, 60
Customizing Your Environment 36
 Blending 36
 Create Backup Files 36
 Playback Auto-Reset 36
 Scroll After Playback 37
 Soundfile Window 37
Cut 24, 26

D

Disk Based Files 12, 13
Display Scale 11
Display Scale Icon 17, 26
Do It 47

E

Edit Blend Settings dialog 37
Edit Blending 31, 37
Edit Commands 68
Editing Functions 24, 26
 Adjusting Blend Parameters 33
 Auto-Zero 30
 Change Gain... 28
 Clear 26
 Clear Beyond Loop End 26
 Clear Unselected 26
 Copy 25
 Create/Delete Loop Points 28
 Cut 24
 Edit Blending 31
 Invert 27
 Locating Loop Points 29
 Loop the Selection 29
 Mix 26
 Normalize 28
 Paste 25
 Reverse 27
 Set MIDI Note... 29
 Show Clipboard... 30
 Silence 27

- Editing the Soundfile 26
- Enable Clipboard 37
- Enable Undo 37
- Envelope Files 13

F

- File Commands 68
- Fit Selection 11
- Fit Selection Icon 17
- Fit Selection icon 26
- Freeze Looper 2, 29, 44
 - Crossfade Length 47
 - Crossfade Taper: 47
 - Do It 47
 - How It Works 45
 - Pitch Control Radio 46
 - Pre-condition Data (%) 46
 - Soundfile Menu 46
- Full Zoom In 26
- Full Zoom Out 26
- Fundamental (Hz) 62

G

- Get Info... 12

H

- Hide Overview 26
- Horizontal Annotation 37
- Horizontal Scale Units 19
- How To Loop A Sound 65

I

- Installation ix
- Invert 27, 37

K

- Keyboard Command Equivalents 26
- Keyboard Shortcuts 67
 - Changing Display Scaling 68
 - Edit Commands 68
 - File Commands 68
 - Looping Tools 69
 - Other 69
 - Palette 69
 - Scrolling Data 67

L

- License Agreement iii
- Listening to The Soundfile 26
- Locating Loop Points 29
- Log Scale or Linear Scale 62
- Loop End 11
- Loop Points 11
- Loop Points In Other File Formats 41
- Loop Points, locating 37
- Loop Start 11
- Loop Start Test Points 50, 54, 60
- Loop the Selection 29
- Looping 4
 - Looping Display 42
 - Looping Tools 39, 69
 - Crossfade Looper 41
 - Freeze Looper 44
 - Managing Loop Points 39
 - Rotated Sums Looper 48
 - SPR Looper 51
 - Synthesis Looper 55

M

- Mainview 9, 18, 21
 - Placing the Insertion Point 21
 - Selection Mode 18, 21
 - Zoom Mode 18
- Managing Loop Points 39
 - Loop Points and Clipboard 40
 - Loop Points During Edits 40
 - Loop Points In Other File Formats 41
 - Real Time Loop Adjust 39
 - Undo Loop Move 40
- Match Trial Points: 43
- Matching Threshold 43
- Merge The Attack And Loop Segments 49, 53, 59
- MIDI Note 37
- Mix 26, 37
- Mode Icons 10
 - Selection Mode Icon 10
 - Zoom Mode Icon 10

N

- Normalize 28

O

Overview 8, 19, 23, 26
Hiding 19
Using Selection Mode 23

P

Palette 10, 15, 69
Paste 25, 26
Pitch Control Radio 46
Placing the Insertion Point 21
Playback Auto-Reset 26, 36
Pre-condition Data (%) 46, 53
Process The Loop Segment 52
Process the Loop Segment 59
Process The Loop Segment: 49

R

Real Time Loop Adjust 1, 39
Retain Noise checked 65
Reverse 27, 37
Revert to Backup 35, 37
Rotated Sums Looper 1, 48
Crossfade Length 50
Crossfade Taper 51
How It Works 49
Loop Start Test Points 50
Merge The Attack And Loop Segments 49
Process The Loop Segment: 49
Rotation Count 49
Soundfile Menu 49
Rotation Count 49

S

Save 37
Save A Copy... 34, 37
Scales 19
Scroll After Playback 26, 37
Scrolling The Soundfile Data 26
scrolling the soundfile data 26
Selecting Data 22, 26
Selection Mode 18, 26
Momentary 23
Selection Mode Icon 10
Selection Mode In The Mainview 26

Selection Mode In The Overview 26
Selection Mode, momentary 26
Separators 19
Set MIDI Note... 29
Show Clipboard... 30, 37
Show Overview 26
Show Track Dividers 19, 26
Show Zero Crossing 19, 26
Sidelobe Width (%) 59, 61, 62
Silence 27, 37
Smart Auto-Scan 2, 39, 42
Arrows 42
Crossfade Length (%) 43
Crossfade Taper 44
Looping Display 42
Match Trial Points 43
Match Trial Points: 43
Matching Threshold 43
Matching Value 43
Settings 43
Soundfile Menu 42
Sound Designer I 37
Sound Designer II 37
Sound Playback 26
Soundfile 7, 13, 16, 20, 21, 34
Editing 21
Listening to 20
Revert to Backup 35
Save A Copy... 34
Saving 34
Specifying where to change sound data 21
Viewing 16
Soundfile Menu 42, 46, 49, 52
Soundfile window 7
Speaker 11
Speaker Icon 20, 26
Spectral Phase Randomization 51
SPR Looper 1, 51
Crossfade Length 54
Crossfade Taper 54
How It Works 52
Loop Start Test Points 54
Merge The Attack And Loop Segments 53
Performance Note 52
Pre-condition Data (%) 53
Process The Loop Segment 52

- Soundfile Menu 52
- Stack Windows 26
- Status Display 11
- Synthesis Looper 1, 55, 61, 65
 - % Retain Noise 59
 - Commentary 55
 - Crossfade Length 60
 - Crossfade Taper 60
 - How It Works 57
 - Loop Start Test Points 60
 - Merge The Attack And Loop Segments 59
 - Process the Loop Segment 59
 - Report 61
 - Sidelobe Width (%) 59
 - Some Rules 57
 - Some Theory 56
- Synthesis Looper Report
 - Amplitude Spectrum (dB) 62
 - Fundamental (Hz): 62
 - Log Scale or Linear Scale 62
 - Retain Noise checked 65
 - Sidelobe Width (%) 62

T

- Technical Support viii
- Tile Windows 26
- Track divider lines 37

U

- Undoing A Command 26
- Units 19
- Using Selection Mode 21, 23

V

- Vertical Annotation 37
- Vertical Scale Units 19, 26
- Viewing The Soundfile 26

W

- Windows 19
 - Get Info... 12
 - Mainview 9
 - Managing Multiple 19
 - Overview 8
 - Soundfile 7
 - Title Bar 8

Z

- Zero crossing lines 37
- Zoom Mode 17, 18, 26
 - Momentary 18
- Zoom Mode Icon 10

