

REFERENCE SECTION

VOICE PLAY MODE

VOICE PLAY MODE

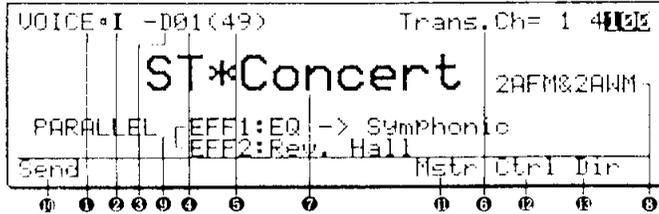
You will normally play the SY99 in Voice Play mode. In Voice Play mode you can do the following things.

- Select voices from preset, internal, or card memory.
- View a directory of the 16 voices in the currently selected bank of internal, card, or preset memory.
- Copy the currently selected voice to any internal or card memory.
- View the controller assignments for the currently selected voice.
- Send a program change to an external device.

Voice select

JUMP #100

Press VOICE to enter voice play mode. The following LCD will appear:



- ❶ VOICE indicates that you are in Voice Play mode.
- ❷ Voice memory (I, C, P1, P2): This indicates the voice memory: Internal, Card, Preset 1, or Preset 2.
- ❸ Bank (A-D): This indicates the voice memory bank.
- ❹ Voice number in individual bank (1-16): This indicates the number of the voice in the bank.
- ❺ Voice number in banks A-D (1-64): This indicates the voice as a number between 1 (voice 1 of bank A) to 64 (voice 16 of bank D).
- ❻ Transmit channel (1-16): This indicates the transmit channel you selected in *MIDI Utility I. Setting* (JUMP #807). The SY99 keyboard will transmit data from MIDI OUT on this channel, except when a master control setup is selected. You can also set the transmit channel at any time by holding SHIFT and pressing a voice select key 1-16.
- ❼ Voice name: The voice name is displayed in large characters.
- ❽ Voice mode: This indicates the type and number of elements used by this voice. For details refer

to *Voice Edit mode, Voice Mode Select* (page 95).

- ❾ Effect settings: This area displays the effect mode (OFF, SERIAL, or PARALLEL), and the effect type for each of the two effect units. For details refer to *Voice edit, Common data 10.Effect set*, page 104.
- ❿ Press F1 to send bank select and program change messages via MIDI. Refer to the following section *Send bank select and program change* (page 88).
- ⓫ Press F6 to jump to the master control select display. Refer to the following section *Master control select* (page 89).
- ⓬ Press F7 to view the controller assignments for the currently selected voice. Refer to the following section *Controller view* (page 87).
- ⓭ Press F8 to view the voice directory. Refer to *Voice directory*, below.

To select a voice use the following procedure. The voice will not actually change until you specify the voice number 1-16. If you want to play a different voice in the same bank, simply specify a different number 1-16.

1. Select the voice memory: INTERNAL, CARD (only if a card is inserted into the DATA slot), PRESET 1, or PRESET 2. The selected LED will blink.
2. Select a bank A, B, C, or D. The selected LED will blink.
3. Select a voice 1-16. The selected LED will light, and the LCD display will show the newly selected voice name.

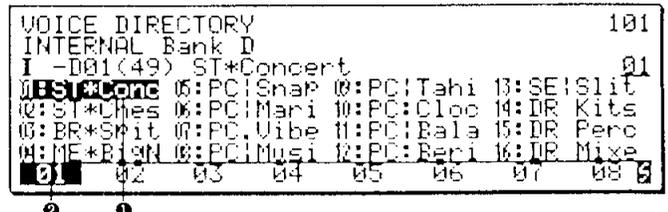
Voice directory

JUMP #101

Summary: This function allows you select voices while viewing a directory of the sixteen voices in the currently selected voice bank.

Procedure:

- From: voice play mode (JUMP #100)
- Select: F8 (Dir) (JUMP #101)
- Specify: one of the displayed voices
- To quit: and return to the voice play display press EXIT.



- ❶ The first seven characters of each ten-character voice name will be displayed. When you select a

different voice memory (internal, card, preset 1, or preset 2) and voice bank (A–D) the sixteen voices in the newly selected bank will be displayed. In addition to the usual methods of selecting a voice, you can also use the arrow keys to select a voice. When the voice directory is displayed, pressing a memory select button or bank select button will immediately select a voice.

- ② Pressing F1–F8 (01)–(08) will select a voice 1–8 from the displayed voice bank. Holding SHIFT and pressing F1–F8 (09)–(16) will select a voice 9–16 from the displayed voice bank.

To return to the voice play display with the voice name displayed in large characters press EXIT.

Copy voice

Summary: You can copy the currently selected voice to another voice memory anytime you are in voice play mode.

Procedure:

From: voice play mode (JUMP #100)

Press: COPY

Specify: the destination to which the voice will be copied.

To execute: the copy operation press F8 (Go).

To quit: without executing press EXIT.

The names of the sixteen voices in the currently selected bank of Internal or Card memory are displayed as explained in *Voice Directory*. Press INTERNAL or CARD, press a bank button A–D, and press a memory select button 1–16 to specify the copy destination.

After specifying the copy destination press F8 (Go). You will be asked “Are you sure?” If you are sure you want to copy the voice, press YES and the data will be copied. To quit without copying press NO.

```

COPY VOICE
I -D01(49) ST*Concert
INTERNAL Bank A 01
Rocks 06:EP:76St 08:PL:Rock 13:KY:Smok
09:HP:Crsk 06:EP:Clas 10:EP:DXis 14:KY:Crsc
08:AP:Conc 07:EP:Nite 11:EP:GrnD 15:KY:Clav
04:AP:Stal 08:EP:Belr 12:EP:UoxL 16:KY:Reso
Go
    
```

Controller view

JUMP #102

Summary: This function allows you to view the controller assignments for the voice as a reminder of how the voice can be controlled.

Procedure:

From: voice play mode (JUMP #100)

Select: F7 (Ctrl) (JUMP #102)

To quit: and return to voice play mode press EXIT.

```

CONTROLLER VIEW 102
Pitch Bend Wheel:Range= 2
After Touch:Md=all Pnt=C 3 PB=+ 0
F Mod :Mk1 Pan LFO :off
A Mod :off Pan Bias:off
F Mod :off EG Bias :off
Cutoff :013 Mk2 Volume :off
Efl Par1:off Efl Par1:off
    
```

- ① The left side of each column displays the parameter which is being controlled. The actual effect that a controller will have on the parameter to which it is assigned will depend on the depth that is specified for each controller assignment as explained in *Voice Common job 12. Controller* (page 127).
- ② The right side of each column displays the controller which has been assigned to control the parameter ①. Controllers which have standard definitions will be displayed as an abbreviation (see below). Other controllers will be indicated by their MIDI control change number. In cases where the controller has been assigned but will have no effect (such as when the control range is 0), an inverse “X” will be displayed.

VOICE PLAY MODE

- ③ Pitch Bend Wheel Range: This displays the range over which the PITCH wheel can raise or lower the pitch.
- ④ After Touch: This area displays the zoned after-touch mode, the split point, and the aftertouch pitch bend setting.
- ⑤ The bottom line displays the two effect parameters which have been selected for realtime control, and the controller which is affecting each effect parameter.

Controller abbreviations: The following abbreviations will be displayed to indicated commonly-used control sources

MW1	Modulation Wheel
MW2	Assignable Modulation Wheel

BC	Breath Controller
VOL	Volume
FC	Foot Controller
DE	Data Entry Slider
SUS	Sustain Switch
FS	Assignable Foot Switch
AFT	Aftertouch
VEL	Velocity
SCL	Scaling
LFO	Effect LFO

This function only allows you to view the controller assignments. To edit them, refer to *Voice Common job 12. Controller* (page 127), and *Voice Common job 10.5 Effect control* (page 123).

Send bank select and program change

Summary: While in voice play mode you can transmit bank select and program change messages from MIDI OUT without affecting the SY99's own tone generator. This allows you to switch a tone generator module connected to the SY99 MIDI OUT to another memory without changing the SY99's own memory selection. (An identical function is available in multi play mode.)

Procedure:

From: voice play mode (JUMP #100)

Select: F1 (Send)

Specify: a bank select number (1-16,384) and a program change number (1-128).

To transmit: the bank select and program change messages press ENTER.

To quit: without sending a bank select or program change message, press EXIT.

1. Use the numeric key pad to enter a bank number between 1 and 16,384, then press ENTER to move the cursor to the program change item. (To send a program change message but no bank select message, simply press ENTER.)
2. Use the numeric keypad to enter a program change number between 1 and 128.

3. Press ENTER and the specified bank select and program change messages will be transmitted on the keyboard transmit channel (Kyb Trans Ch) specified in *MIDI Utility job 1.Setting* (JUMP #807).

If, for the program change, you enter a number below 1 it will be transmitted as 1. If you enter a number above 128, it will be transmitted as 128. In addition to the program change message transmitted by this function, a program change message will be transmitted every time you select a SY99 voice or multi unless Program Change has been turned off using *MIDI utility job 1.Setting*, page 258.

Remark: Refer to the *MIDI Data Format* booklet for details regarding the use of MIDI bank select numbers.

Note: If a master control setup is activated, the transmit filter for that setup may prevent the sending of bank select and program change messages using this function. Refer to the explanation of the Master control utility on page 284 for further details.

Master control select

Summary: While the SY99 is in voice play mode, you can jump to the master control select display (JUMP #832) by pressing a single function key. This feature makes it easy to use the MIDI master control function while playing the SY99.

Procedure:

From: voice play mode (JUMP #100)
Select: F6 (Mstr)

The controller select display will appear, just as if you had pressed JUMP, entered 832 using the numeric keypad, and then pressed ENTER. This display can be used to send a variety of control information to MIDI instruments connected to the SY99.

If you perform live using the SY99, you will probably want to use the master control function often while playing the SY99's keyboard. You will probably find the master control function especially convenient because it is available with the press of a single button. See page 284 for details.

VOICE PLAY MODE

VOICE EDIT MODE

This section explains the details of all voice edit parameters.

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VOICE EDIT MODE

The organization of Voice Edit mode will depend on whether the voice is a normal voice or a drum voice.

Normal voice (voice modes 1–10)

F1 (Mode)	F2 (Com)	F3 (E1)	F4 (E2)	F5 (E3)	F6 (E4)
Specify the Voice Mode	Common data edit job directory	AFM element edit job directory	OR	AWM element edit job directory	
1. 1AFM mono 2. 2AFM mono 3. 4AFM mono 4. 1AFM poly 5. 2AFM poly 6. 1AWM poly 7. 2AWM poly 8. 4AWM poly 9. 1AFM&1AWM poly 10. 2AFM&2AWM poly 11. Drum set	1. Element level 2. Element detune 3. Element note shift 4. Element note limit 5. Element velocity limit 6. Element dynamic pan 7. Output group select 8. Random pitch 9. Portamento 10. Effect set 11. Micro tuning set 12. Controller set 13. Voice name ... 15. Initialize voice 16. Recall voice	1. AFM algorithm 2. AFM oscillator 3. AFM EG 4. AFM operator output 5. AFM sensitivity 6. AFM LFO 7. AFM pitch EG 8. AFM filter ... 15. Initialize AFM element 16. Recall AFM element		1. _____ 2. AWM waveform set 3. AWM EG 4. AWM output 5. AWM sensitivity 6. AWM LFO 7. AWM pitch EG 8. AWM filter ... 15. Initialize AWM element 16. Recall AWM element	

Drum voice (voice mode 11)

F1 (Mode)	F2 (Com)
Specify the Voice Mode	Drum Set edit job directory
1. 1AFM mono 2. 2AFM mono 3. 4AFM mono 4. 1AFM poly 5. 2AFM poly 6. 1AWM poly 7. 2AWM poly 8. 4AWM poly 9. 1AFM&1AWM poly 10. 2AFM&2AWM poly 11. Drum set	1. Voice volume 2. Wave data set 3. Effect set 4. Controller set 5. Name ... 7. Initialize 8. Recall

Compare

When you are in edit mode but have not yet modified the data, a small square is displayed at the left of the voice number to indicate that the voice has not yet been edited. If the data is edited in any way, this will change to an inverse "E".

If you want to see and hear the original data press EDIT (COMPARE) and the inverse "E" will change to a "C" indicating that you are in compare mode.

To return to edit mode press EDIT (COMPARE) once again and the "C" will change back to an "E".

Note:

- The Compare function is not available in the job directory displays, nor while editing Dynamic Pan, Micro Tuning, or waveform data.
- If the Voice Mode has been changed, the Compare function will not be available.
- While comparing, it is not possible to modify parameter values. (However there are some exceptions.)
- If you compare while editing a card voice, a card error will cancel compare after displaying an error message.
- While comparing, EXIT, mode select, page, cursor, JUMP, COPY, and some of F1–F8 will not function.

Store voice

When you press EXIT or use the JUMP button to exit Voice Edit mode after editing the data, the top line of the display will blink "AUTO-STORE VOICE"

```

AUTO-STORE VOICE  Push Return/Quit/Stor
BI -D01(49) ST*Concert
INTERNAL Bank D
Use Bank D
0:ST*Ches 0:PC|Mari 10:PC|Cloc 14:DR Kits
1:ME*BigN 0:PC|Musj 12:PC|Beri 16:DR Mixe
Ret Quit Stor

```

Note 1:

Four-element voices, i.e., voices using voice mode 3 (4AFM mono), 8 (4AWM poly), or 10 (2AFM&2AWM) occupy extra memory, and can be stored only in bank D. The AUTO-STORE display for such voices will automatically show bank D, and blink "Use bank D" in the bottom line as a reminder.

Voices which use other voice modes can be stored in bank D as well.

Drum voices may be stored in any of the four banks A through D. However, since these voices contain a large volume of data, only the data for notes C1 to C6 will be saved if a drum voice is stored in banks A, B, or C. Store drum voices in bank D if you wish to save the data for the full range of notes from E0 to G6.

Note 2:

When storing a voice which uses an AWM waveform card, make sure that the correct card is inserted when you store, since the waveform card ID number is stored as part of the voice.

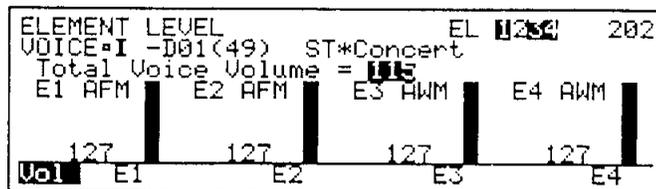
The LCD will show the first seven characters of the voice names in the currently selected bank of voices. The voice name displayed in inverse indicates the voice memory into which the edited data will be stored.

1. Use INTERNAL or CARD to specify the voice memory, select a bank A–D, and select the voice memory 1–16 in which you want to store your newly edited voice.
2. Press F8 (Stor), and the bottom line will ask "Are you sure ? (Yes or No)".
3. If you are sure you want to store the edited voice, press +1/YES and the bottom line of the LCD will show "Store completed". If you decide not to store, press -1/NO and the bottom line of the LCD will show "Store cancelled".
4. You will then return to voice play mode or the jump destination.

Element on/off

When editing a voice which uses two or more elements, it is often useful to hear only the element being edited. At any time while editing a normal voice, pressing the ELEMENT ON/OFF buttons located at the upper right of the front panel will turn individual elements off/on. When editing voice common data, the number of elements used by the voice will be shown in the upper right corner of the display. Elements that are on will be displayed in inverse. In addition the LEDs above the ELEMENT ON/OFF buttons will light if the element is on and be darkened when the element is turned off. In the following LCD, element 2 has been turned off, and will not be heard.

Element 2 has been turned off.



When editing Drum Set data the ELEMENT ON/OFF buttons will have no effect.

When you move to the Voice Mode Select display, the element on/off settings will automatically be cancelled.

Element select

At any time while editing the element data of a normal voice, you can use the ELEMENT SELECT buttons located at the upper right of the front panel to select an element to edit. This is often faster than returning to the top level of voice edit mode and pressing F3-F6 to select the job directory of a different element.

When editing Voice Common data, Drum Set data, or waveforms, the ELEMENT SELECT buttons will have no effect.

Voice mode select

Summary: The voice mode setting determines whether a voice will consist of one, two, or four AWM or AFM elements (modes 1–10), or 76 AWM waves (mode 11).

Procedure:

From: the top level of voice edit mode
(JUMP #200, #201, #230, #256)

Press: F1 (Mode) to get the following display.
(JUMP #200)

Specify: the desired voice mode.

VOICE EDIT		E1: AFM	E3: AWM	200
		E2: AFM	E4: AWM	
• I -D01(49)	ST*Concert			10
01: 1AFM mono	05: 2AFM Poly	09: 1AFM&1AWM		
02: 2AFM mono	06: 1AWM Poly	10: 2AFM&2AWM		
03: 4AFM mono	07: 2AWM Poly	11: Drum Set		
04: 1AFM poly	08: 4AWM Poly			
0000 Com	E1	E2	E3	E4

- ❶ This area shows the number (1, 2, or 4) and type (AWM or AFM) of elements in the selected voice mode.
- ❷ Move the cursor to the desired voice mode 1–11. The selected voice mode will become effective immediately.
 - 01: 1AFM mono: The voice consists of one AFM element.
 - 02: 2AFM mono: The voice consists of two AFM elements.
 - 03: 4AFM mono: The voice consists of four AFM elements. (See note)
 - 04: 1AFM poly: The voice consists of one AFM element.
 - 05: 2AFM poly: The voice consists of two AFM elements.
 - 06: 1AWM poly: The voice consists of one AWM element.
 - 07: 2AWM poly: The voice consists of two AWM elements.
 - 08: 4AWM poly: The voice consists of four AWM elements. (See note)
 - 09: 1AFM&1AWM: The voice consists of one AFM and one AWM element.
 - 10: 2AFM&2AWM: The voice consists of two AFM and two AWM elements. (See note)

11: Drum Set: The voice consists of seventy-six AWM samples.

Mono modes (1–3): Voices which use modes 1–3 are monophonic. Only one note can be produced at a time. If a note is played while the previous note is still sounding, the previous note will be cut off. Mono mode is useful when simulating instruments that naturally produce only one note at a time. Mono mode also allows you to use a special type of portamento; *fingered portamento*. For details refer to *Voice Common 9. Portamento* (page 104).

Polyphonic modes (4–10): Voices which use modes 4–10 are polyphonic, and will allow you to play chords of as many notes as can be produced by the SY99's tone generator. The AWM and AFM tone generators can each produce up to 16 simultaneous notes. For some voice modes more than one element may be sounded by a single key, and this will correspondingly reduce the number of simultaneous notes you can play.

Drum Set mode (11): Drum set voices use only the AWM tone generator, and up to 16 AWM samples can be sounded simultaneously.

Note: Four-element voices (modes 3, 8, and 10) can be stored only in bank D.

Drum voices may be stored in any of the four banks A through D. However, since these voices contain a large volume of data, only the data for notes C1 to C6 will be saved if a drum voice is stored in banks A, B, or C. Store drum voices in bank D if you wish to save the data for the full range of notes from E0 to G6.

A slight delay may occur in the sounding of some notes when a large number of notes are played simultaneously (either manually or in response to MIDI note on messages) using four-element voices. To avoid this delay in such cases, you may wish to use a voice using fewer elements, or reduce the number of notes being played simultaneously.

Common data

COMMON DATA

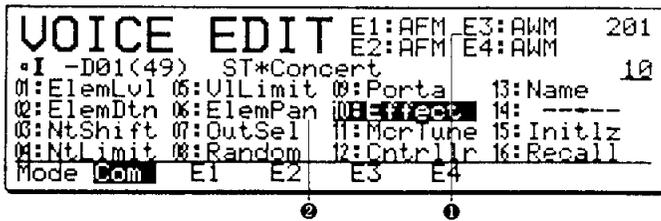
Common data job directory

JUMP #201

Summary: This job directory shows the jobs containing data that affects all elements in the voice.

Procedure:

- From: the top level of voice edit mode
- When: editing a normal voice
- Press: F2 (Com) (JUMP #201)
- Select: the desired job



- ❶ This area shows the number (1, 2, or 4) and type (AWM or AFM) of elements in the selected voice mode.
- ❷ Move the cursor in this area to select a job:
 - 01: ElemLvl (Element level): Total voice volume, and element level
 - 02: ElemDtn (Element detune): Fine tuning for each element
 - 03: NtShft (Element note shift): Transpose each element

- 04: NtLimit (Element note limit): Range of notes that play each element
- 05: VILimit (Element velocity limit): Range of key-on velocities that play each element
- 06: ElemPan (Element dynamic pan): Dynamic panning table for each element
- 07: OutSel (Output group select): Output group for each element
- 08: Random (Random pitch): Random pitch variation for entire voice
- 09: Porta (Portamento): Portamento mode and time
- 10: Effect (Effect set): Effect set job directory
- 11: Mcrtune (Micro tuning): Micro tuning select for entire voice, and micro tuning on/off for each element.
- 12: Cntrlr (Controller set): Controller assignments and depth for pitch bend, modulation, pan, etc.
- 13: Name (Voice name): Ten-character voice name
- 15: Initlz (Initialize voice): Initialize the voice common data being edited
- 16: Recall (Recall voice): Recall all data (common and element) of the previously edited voice

COMMON DATA

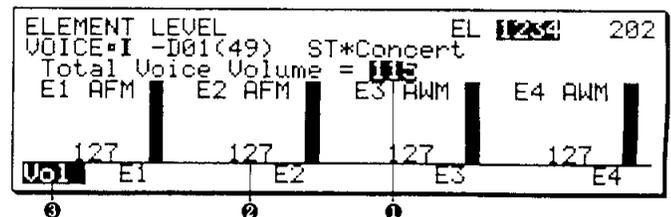
1. Element level

JUMP #202

Summary: Adjust the overall volume of the entire voice, and the volume of individual elements 1-4.

Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 01:ElemLvl (JUMP #202)
- Specify: the total voice volume and the levels of each element



- ❶ Total voice volume (0...127): This determines the overall volume of the entire voice.

- ② Element level (0...127) E1-E4: These determine the volume level of each element. Press F2, F4, F6, F8 to move the cursor to elements 1-4. The level of each element is displayed as a vertical bar graph.
- ③ Pressing F1 will move the cursor to Total Voice Volume. Pressing F2, F4, F6, F8 will move the cursor to elements 1-4.

Remarks: Since the total voice volume setting is part of the voice data, it can be used to even out the volume differences between voices. This is important when editing a set of voices for live performance, and allows you to avoid any sudden jumps in volume when a voice is selected.

COMMON DATA

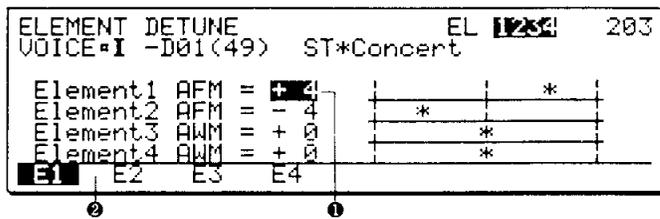
2. Element detune

JUMP #203

Summary: Adjust the fine tuning of individual elements 1-4.

Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 02:ElemDtn (JUMP #203)
- Specify: the tuning of each element



- ① Detune (-7...+7) E1-E4: When this is set to 0, the element will play the correct pitch for the key that was pressed. Negative settings will lower

the pitch, and positive settings will raise the pitch. The tuning of each element is indicated graphically by the position of the asterisk on a horizontal scale.

- ② Pressing F1-F4 will move the cursor to elements 1-4.

Remarks: If you are creating a voice that plays two or more elements for a single note, slightly detuning the elements will create a natural chorus effect, giving a richer quality to the sound.

Element detune is intended to change the *relative* pitch of two or more elements. Setting all elements to the same detune value will not be useful, nor will this setting be useful if the voice contains only one element.

COMMON DATA

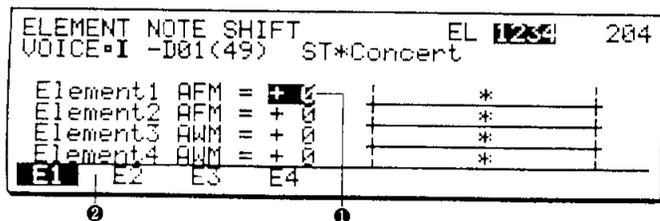
3. Element note shift

JUMP #204

Summary: Transpose the pitch of individual elements 1-4.

Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 03:NtShft (JUMP #204)
- Specify: the transposition of each element



- ① Note Shift (-64...+63) E1-E4: When this is set to 0, the element will play the correct pitch for the key that was pressed. This setting adjusts the pitch in semitone increments. For example if set to -12 the pitch will be one octave lower than normal, and if set to +24 the pitch will be two octaves higher than normal.

- ② Pressing F1-F4 will move the cursor to elements 1-4.

Remarks: The note shift setting can be useful when you need play notes that are beyond the range of the SY99's 76-note keyboard. For voices which play two or more elements for each key, note shift can be used to create automatic parallel harmony.

COMMON DATA 4. Element note limit JUMP #205

Summary: Specify the range of notes that will play each element.

Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 04:NtLimit (JUMP #205)
- Specify: the note range for each element

ELEMENT NOTE LIMIT				EL	205
VOICE=I -D01(49) ST*Concert				Low	High
Element1	AFM	C-2	G8		
Element2	AFM	C-2	G8		
Element3	AWM	C-2	G8		
Element4	AWM	C-2	G8		
E1	E2	E3	E4		Kbd

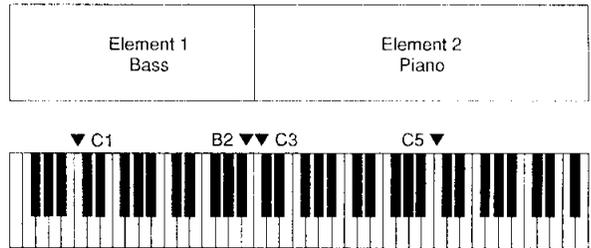
- 1 Low Note Limit (C-2...G8) E1-E4: This specifies the lowest note that will be played by the element.
- 2 High Note Limit (C-2...G8) E1-E4: This specifies the highest note that will be played by the element.
- 3 Pressing F1-F4 will move the cursor to elements 1-4.

Remarks: After moving the cursor to the parameter you want to set, you can modify the data in the usual way or press F8 (Kbd) and then press a key of the SY99's keyboard to enter a note. (The notes of the SY99 keyboard are E0-G6.)

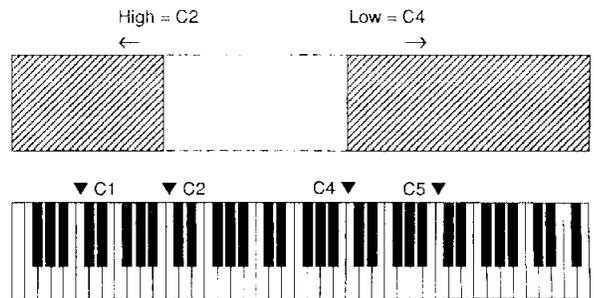
If you want to play an element over the entire range of the keyboard, leave this parameter set at Low=C-2 and High=G8.

This parameter can be used to create keyboard split effects where different elements are played by different keyboard areas. For example in a two-element voice where element 1 is a

bass sound and element 2 is a piano sound, set element 1 to Low=C1 and High=B2 and set element 2 to Low=C3 and High=C5. With these settings, notes below middle C will play bass (element 1) and notes above middle C will play piano (element 2).



It is possible to set the low limit above the high limit. In this case, the element will be played by notes above the low limit and below the high limit. The following diagram shows the keyboard range that would play an element set to Low=C4 and High=C2.



Note: This note limit setting will be ignored if the voice mode is mono (voice mode 1:1AFM mono, 2:2AFM mono, and 3:4AFM mono).

COMMON DATA 5. Element velocity limit JUMP #206

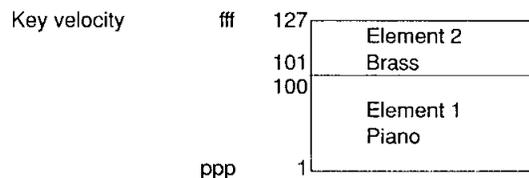
Summary: Specify the range of key-on velocities that will play each element.

Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 05:VILimit (JUMP #206)
- Specify: the range of velocities for which the element will sound

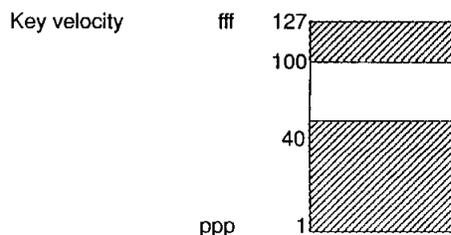
ELEMENT VELOCITY LIMIT				EL	206
VOICE=I -D01(49) ST*Concert				Low	High
Element1	AFM	1	127		
Element2	AFM	1	127		
Element3	AWM	20	127		
Element4	AWM	1	127		
E1	E2	E3	E4		

- ❶ Low Velocity Limit (1...127): This is the lowest key-on velocity for which the element will sound.
- ❷ High Velocity Limit (1...127): This is the highest key-on velocity for which the element will sound.
- ❸ Press F1-F4 to move the cursor to elements 1-4.



Remarks: If you want to play an element over the full range of key velocities, leave this parameter set at Low=1 and High=127. This parameter can be used to make strongly played notes play different elements than softly played notes. For example in a two-element voice where element 1 is a piano sound and element 2 is a brass sound, set element 1 to Low=1 and High=100 and set element 2 to Low=101 and High=127. With these settings, soft notes will play the piano (element 1) and strong notes will play brass (element 2). If desired, you could overlap the velocity limits of the elements, or use more than two elements.

It is possible to set the low limit above the high limit. In this case, the element will be sounded by key-on velocities outside the limits. The following diagram shows the velocity range that would play an element set to Low=100 and High=40.



COMMON DATA

6. Element dynamic pan

JUMP #207

Summary: Select the Dynamic Pan data used by each element. The selected dynamic pan data will determine how the stereo position of the element will change over time.

Procedure:

- From: Voice Common job directory (JUMP #201)
- Select: job 06:ElemPan (JUMP #207)
- Specify: the Dynamic Pan data used by each element

ELEMENT DYNAMIC PAN	EL 1234	207
VOICE=I -D01(49)	ST*Concert	
	Dynamic Pan Select	
Element1 AFM	P- 3 Right	5
Element2 AFM	P- 9 Left	5
Element3 AFM	P- 1 Center	
Element4 AFM	P- 1 Center	
E1	E2	E3
2	1	3

- ❶ Dynamic Pan Select (I1...32, C1...32, P1...64): This specifies the dynamic pan data that will move the stereo position of this element over time. The number and name of the selected pan data will be displayed. For an internal voice, select a pan data memory from Internal or Preset. For a card voice, select a pan data memory from Card or Preset.

- ❷ Pressing F1-F4 will move the cursor to elements 1-4.
- ❸ Press F8 to edit the currently selected pan data. (Only Internal pan data can be edited.)

Remarks: Each Dynamic Pan memory contains the following data.

- a Pan Source which allows the panning movement to be controlled by velocity, note number, or LFO
- EG settings which determine panning movement over time
- a Pan Name

64 preset dynamic pan memories are provided, as explained in the following table. In addition, 32 internal memories are provided for you to store your own pan data, and a RAM card can accommodate 32 more pan data memories. The following section 6.0 Dynamic Pan Edit explains how to edit the Dynamic Pan data.

VOICE EDIT MODE

Preset Dynamic Pan data

#	Name	Description
1	Center	fixed at center
2	Right 6	fixed full right
3	Right 5	...
4	Right 4	...
5	Right 3	...
6	Right 2	...
7	Right 1	fixed slightly right
8	Left 6	fixed full left
9	Left 5	...
10	Left 4	...
11	Left 3	...
12	Left 2	...
13	Left 1	fixed slightly left
14	L>R slow	slowly move L→R
15	L>R	move L→R
16	L>R fast	quickly move L→R
17	R>L slow	slowly move R→L
18	R>L	move R→L
19	R>L fast	quickly move R→L
20	C>R slow	slowly move C→R
21	C>R	move C→R
22	C>R fast	quickly move C→R
23	C→R slow	pause at center then slowly move C→R
24	C→R	pause at center then move C→R
25	C→R fast	pause at center then quickly move C→R
26	C>L slow	slowly move C→L
27	C>L	move C→L
28	C>L fast	quickly move C→L
29	C→L slow	pause at center then slowly move C→L
30	C→L	pause at center then move C→L
31	C→L fast	pause at center then quickly move C→L
32	L<>R slow	start at L then slowly move between LR
33	L<>R	start at L then move between LR
34	L<>R narrow	start at L then move (narrowly) between LR
35	L<>R fast	start at L then move quickly between LR

#	Name	Description
36	R<>L slow	start at R then slowly move between RL
37	R<>L	start at R then move between RL
38	R<>L narrow	start at R then move narrowly between RL
39	R<>L fast	start at R then move quickly between RL
40	C>R<>L slw	start at C then move slowly between RL
41	C>R<>L s&n	start at C then move slowly and narrowly between RL
42	C>R<>L	start at C then move between RL
43	C>R<>L fst	start at C then move quickly between RL
44	C→R<>L sl	pause at C then move slowly between RL
45	C→R<>L	pause at C then move between RL
46	C→R<>L fs	pause at C then move quickly between RL
47	C>L<>R slw	start at C then move slowly between LR
48	C>L<>R s&n	start at C then move slowly and narrowly between LR
49	C>L<>R	start at C then move between LR
50	C>L<>R fst	start at C then move quickly between LR
51	C→L<>R sl	pause at C then move slowly between LR
52	C→L<>R	pause at C then move between LR
53	C→L<>R fs	pause at C then move quickly between LR
54	LFO MWheel	controller regulates the width (initially 0) of LFO panning
55	LFO wide	broad panning by LFO
56	Note wide	broad panning by note number
57	Note narrow	narrow panning by note number
58	Notew+EG n	broad panning by note number + narrow LR movement
59	Noten+EG w	narrow panning by note number + broad LR movement
60	Vel wide	broad panning by key velocity
61	Vel narrow	narrow panning by key velocity
62	Vel w+EG n	broad panning by key velocity + narrow LR movement
63	R&L 1	variation of repeated LR movement
64	R&L 2	variation of repeated LR movement

COMMON DATA / ELEMENT DYNAMIC PAN

6.0 Dynamic pan edit

Summary: This function allows you to edit the currently selected Dynamic Pan data.

Procedure:

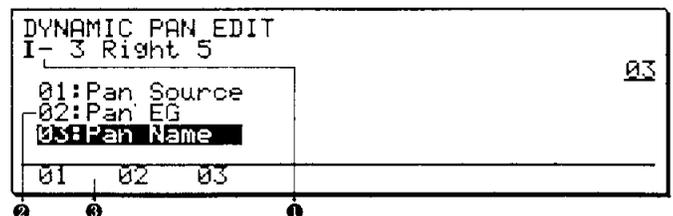
From: Voice Common job 06.ElemPan
(JUMP #207)

When: an Internal dynamic pan memory is selected

Press: F8 (Edit)

Select: the Dynamic Pan parameter you wish to edit

Editing is possible only when an Internal pan memory is selected. If you want to edit one of the preset pan memories, press COPY to copy it to an Internal pan memory as explained in the following section 6.0.1 Copy Pan Data.



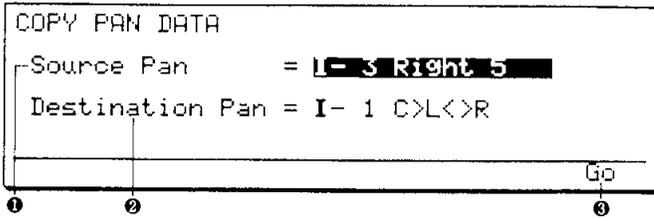
- ❶ The name and number of the currently selected Dynamic Pan data are displayed.
- ❷ Move the cursor in this area to select a job.
 - 01: Pan Source: Select a control source (velocity, note number, or LFO) to affect dynamic panning. (See 6.1 Pan Source)
 - 02: Pan EG: Set the panning EG. (See 6.2 Pan EG)
 - 03: Pan Name: Assign a ten-character name to the pan data. (See 6.3 Pan Name)
- ❸ Pressing F1–F3 will select the corresponding job.

6.0.1 Copy pan data

Summary: This function copies dynamic pan data from another memory into an Internal pan data memory.

Procedure:

- When: editing dynamic pan data
- Press: COPY to get the following display.
- Specify: the source and destination pan.
- To copy: the pan data press F8.
- To quit: without copying press EXIT.



- ❶ Source Pan: Select the dynamic pan data to copy.
- ❷ Destination Pan: Select the dynamic pan data (internal 1–32) into which to copy the Source Pan data.
- ❸ After selecting Source Pan and Destination Pan, press F8 (Go) to copy the data. You will be asked “Are you sure?”. Press YES and the data will be copied.

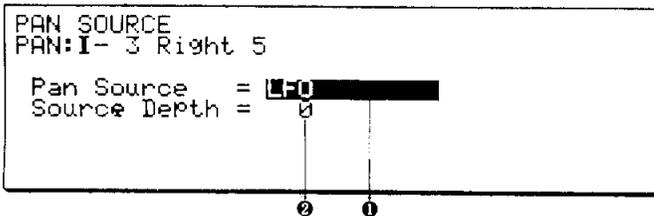
Remarks: Only Internal Pan Data memories can be edited. If you want to edit one of the preset pan tables, use this function to copy it into an Internal pan memory.

6.1 Pan source

Summary: This determines how the dynamic panning will be affected; either by Velocity, or Note Number, or LFO.

Procedure:

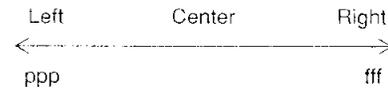
- From: Dynamic Pan Edit job directory
- Select: 01:Pan Source
- Specify: the pan source and depth



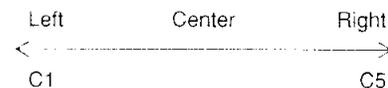
- ❶ Pan Source (Velocity, Key Note Number, LFO): Select the control source which will affect dynamic panning. When velocity is selected, the strength at which each note is played will affect panning. When note number is selected, notes above middle C will be panned increasingly right, and notes below middle C will be panned increasingly left. When LFO is selected, the LFO of the element will continuously pan the sound.
- ❷ Source Depth (0...127): This determines how much the selected Pan Source will affect panning. When this is set to 0, the selected pan

source will have no effect. When this is set to 127, the selected pan source will pan the element from full left to full right.

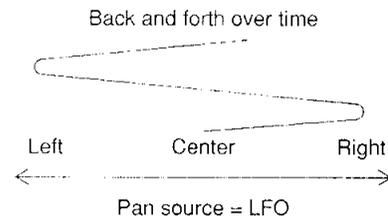
Pan source = Velocity



Pan source = Key note number



Pan source = LFO



Remarks: Dynamic panning is controlled by two factors working together; the Pan Source and the Pan EG. Refer to the diagram in 6.2 Pan EG for an example of this.

6.2 Pan EG

Summary: Specify how the element will be panned over time, starting when each note is played.

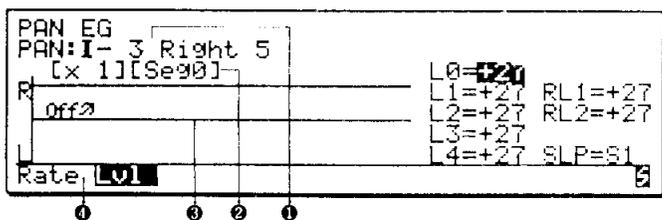
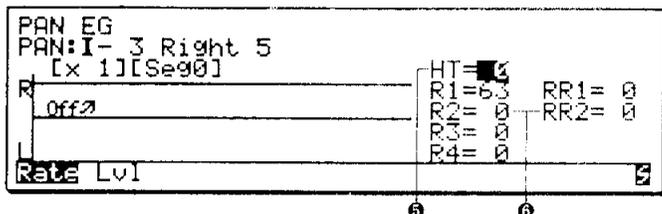
Procedure:

From: Dynamic Pan Edit job directory

Select: 02:Pan EG.

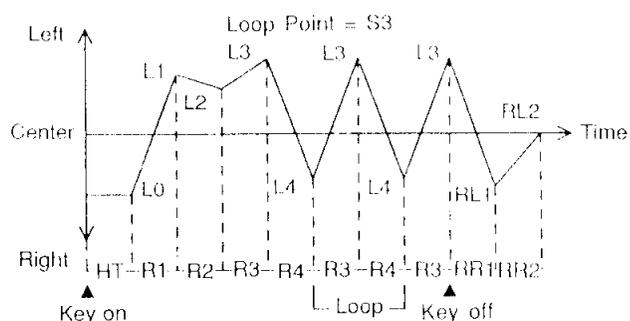
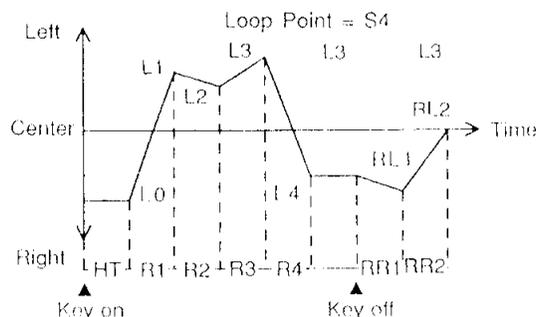
Press: F1 (Rate) to set pan EG rates.

Press: F2 (Lvl) to set pan EG levels.



- ❶ The number and name of the currently selected Dynamic Pan data are displayed.
- ❷ This indicates the displayed segment and range of the EG graphic display. To change the range, hold SHIFT and press F1-F6 (x1, x2, x5, x10, x20, x50). To shift the EG graphic display to a different segment, hold SHIFT and press F7 or F8 to select Seg0-Seg4, Rel1.
- ❸ The pan EG is graphically displayed.
- ❹ Press F1 (Rate) to set EG rates. Press F2 (Lvl) to set EG levels.
- ❺ HT (Keyon Delay Time 63...0): When this is set to 0, the pan EG will begin immediately after a key is pressed. For higher settings, there will be an increasingly longer delay before the pan EG begins.
- ❻ R1-R4, RR1-RR2 (Keyon Rates, Release Rates 0...63): Keyon Rates 1-4 and Release Rates 1-2 determine the speed of the pan EG. Higher settings result in faster change. Refer to the diagram below.
- ❼ L0-L4, RL1-2 (Keyon Levels, Release Levels -32...+32): Keyon Levels 0-4 and Release Levels 1-2 determine the panning direction and distance of the pan EG. Negative settings move left, and positive settings move right. Refer to the following diagram.

- ❸ SLP (Loop Point S1-S4): This specifies the segment from which the EG will continue looping if a key remains depressed after the EG has come to the end. Refer to the following diagram.



When you press a key, the sound will be output at the pan position of L0. When the specified hold time (HT) has elapsed, the pan position will change at the rate of R1 to level L1. When the pan position reaches L1, it will change at the rate of R2 to the position of L2. When the pan position reaches L2, it will change at the rate of R3 to the position of L3. When the pan position reaches L3, it will change at the rate of R4 to the position of L4. When the pan position reaches L4, the EG will begin looping from the specified segment (in the above diagram, SLP=S3).

When you release the key, the pan position will change at the rate of RR1 to the position of RL1. When the pan position reaches RL1, it will change at the rate of RR2 to the position of RL2.

Remarks: Hold Time (HT) is a *time* setting, but the various Rates are *speed*. Higher settings for Hold Time will result in a longer delay before the pan EG begins, but higher settings for Rates will result in faster change. The final result of the Pan EG will depend on the Pan Source settings.

COMMON DATA / ELEMENT DYNAMIC PAN

6.3 Pan name

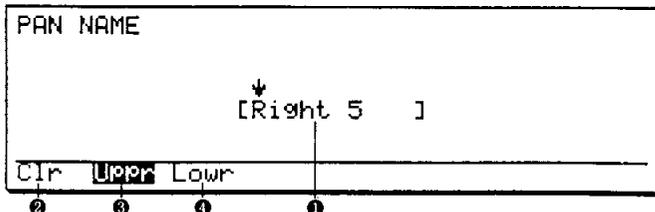
Summary: Specify a ten-character name for the internal Pan data memory being edited.

Procedure:

- From: Dynamic Pan Edit job directory
- Select: 03:Pan Name
- Specify: a name for the pan memory

- 1 Enter a ten-character name for the Pan data.
- 2 To clear the currently entered name press F1 (Clr).
- 3 To switch to upper-case characters press F2 (Uppr).
- 4 To switch to lower-case characters press F3 (Lowr).

Methods of entering character data are explained in *How to enter character data*, on page 30.



COMMON DATA

7. Output group select

JUMP #208

Summary: Specify the output group for each element.

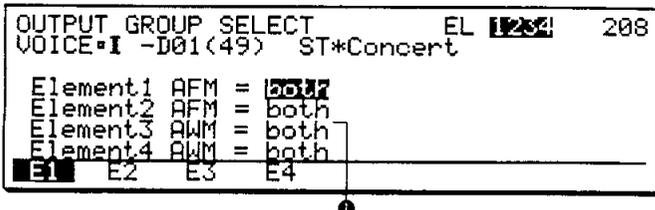
Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 07:OutSel (JUMP #208)
- Specify: the output group for each element

stereo signal is sent to the corresponding output channel.

If you turn off the output group select for an element, that element will not be sent to the effect unit.

Remarks: The effect wet:dry setting determines the mix between the effect processed sound from each effect unit and the stereo signal from the dynamic pan tables. If the wet:dry setting is 100:0, the stereo pan will have no effect, since all the output sound will be coming from the effect unit. If the wet:dry setting is 0:100, the sound from the effect unit will not be heard. Refer to the figure in *10.Effect set*, page 104.



- 1 Element 1-4 (off, grp1, grp2, both): Each of the elements in a voice must be assigned to one or both output groups. The element is then panned as specified by the dynamic pan table, and the

COMMON DATA

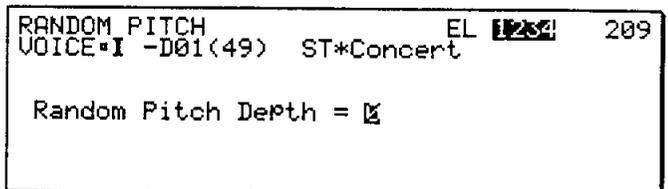
8. Random pitch

JUMP #209

Summary: Specify the amount of random pitch variation for the voice.

Procedure:

- From: voice common job directory (JUMP #201)
- Select: job 08:Random (JUMP #209)
- Specify: the amount of random pitch variation



VOICE EDIT MODE

- 1 Random Pitch Depth (0...7): For a setting of 0, a key will produce the same pitch each time it is pressed. For settings of 1...7, a key will produce a random deviation in pitch. Higher settings result in greater deviation from the standard key pitch.

Remarks: This parameter is helpful when simulating instruments which have a naturally varying pitch.

The random pitch deviation is applied separately to each element in the voice, meaning that pitch differences may appear between elements.

COMMON DATA

9. Portamento

JUMP #210

Summary: Specify the Portamento mode and time.

Portamento creates a smooth glide in pitch between one note and the next.

Procedure:

From: voice common job directory (JUMP #201)

Select: job 09:Porta (JUMP #210)

Specify: the portamento mode and time

```
PORTAMENTO EL 1234 210
VOICE=I -D01(49) ST*Concert
Portamento Mode = follow
Portamento Time = 12
*Portamento affects only AFM elements.*
Mode 12ms
```

- 1 Portamento Mode: If the currently selected voice mode is polyphonic (voice modes 4-10), the portamento mode is fixed at Follow mode. If the currently selected voice mode is monophonic (voice modes 1-3), the portamento mode can be set either to Fingered or Fulltime.

Fingered portamento: Portamento is applied only if you press a note before releasing the previous one; i.e., when you play legato.

Full Time portamento: Portamento is applied between all notes.

- 2 Portamento Time (0...127): This determines the time of the pitch glide between notes. Higher settings result in a longer (slower) glide.

Remarks: If you do not want portamento, set Portamento Time to 0 so that the pitch change between notes is instantaneous.

As noted in the display, portamento applies only to AFM elements. It will have no effect on voices which use only AWM elements (voice modes 6-8). If a voice uses both AFM and AWM elements (voice modes 9 and 10), portamento will apply only to the AFM elements in the voice.

COMMON DATA

10. Effect set

JUMP #212

Summary: Specify how the effect units are connected, how the sound from each element of the voice is sent to the effect units, parameters for each effect unit, and how the effect parameters are controlled in realtime.

Procedure:

From: voice common job directory (JUMP #201)

Select: job 10:Effect (JUMP #212)

Select: the effect parameter you wish to edit

```
EFFECT SET EL 1234 212
VOICE=I -D01(49) ST*Concert
01:Effect Mode Select 01
02:Effect Send
03:Effect 1 Parameter
04:Effect 2 Parameter
05:Effect Control
01 02 03 04 05
```

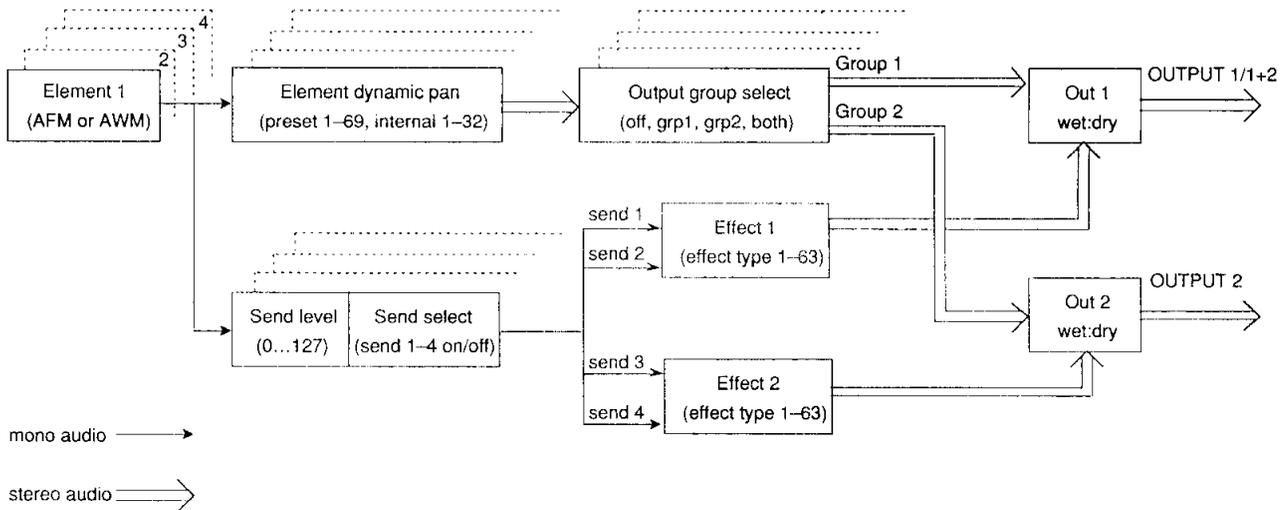
- 1 Move the cursor in this area to select a job.
01: Effect Mode Select: Specify how the two effect units are connected, the effect type, effect output level, and wet/dry balance. Refer to 10.1 Effect Mode Select.

- 02: Effect Send: Specify how the sound of each element of the voice will be sent to the effect units. Refer to 10.2 Effect Send.
- 03: Effect 1 Parameter: Make settings for effect unit 1. Refer to 10.3 Effect 1 Parameter.
- 04: Effect 2 Parameter: Make settings for effect unit 2. These parameters are exactly the same as for Effect 1. Refer to 10.3 Effect 1 Parameter.

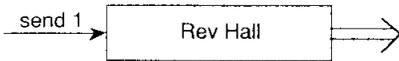
- 05: Effect Control: Specify how effect parameters will be affected by control change messages, aftertouch, velocity, key scaling, and LFO. Refer to 10.5 Effect Control.

② Pressing F1–F5 will select the corresponding job.

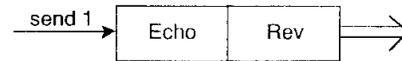
The following figure shows how the signal flow of a normal voice is related to the effect system.



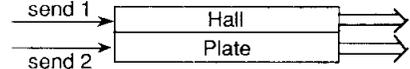
Effect type = 1–29 (single type)
ex., 1. Rev Hall



Effect type = 30–51 (cascade type)
ex., 30. Echo→Rev



Effect type = 52–63 (dual type)
ex., 52. Hall&Plate



COMMON DATA / EFFECT SET

10.0.1 Copy voice effect

Summary: This function copies Effect data from another voice into the effect data of the currently edited voice.

Procedure:

When: editing Effect data (jobs 10.0–10.5)

Press: COPY to get the following display.

Specify: the voice from which to copy the effect data

To copy: the data press F8 (Go)

To quit: without copying press EXIT

```

COPY EFFECT
Source Select
VOICE I -A01(01) Rocks
01:Rocks 06:EP:76St 08:PL:Rock 13:KY:Smok R1
02:HP:Crsk 09:EP:Clas 10:EP:DXis 14:KY:Crsc
03:AP:Conc 07:EP:Nite 11:EP:GrnD 15:KY:Clav
04:AP:Stal 05:EP:Belr 12:EP:UoxL 16:KY:Reso
Voice Mult Go
    
```

- ① Use the voice memory buttons, bank button A–D, and voice select buttons 1–16 to select a source voice from which to copy the Effect data.

VOICE EDIT MODE

- Press F8 (Go) and you will be asked "Are you sure?". If you are sure you want to copy the effect data, press YES and the effect data will be copied from the selected voice to the voice being edited.

Note: The effect send settings (see 10.2 Effect send) will not be copied.

COMMON DATA / EFFECT SET

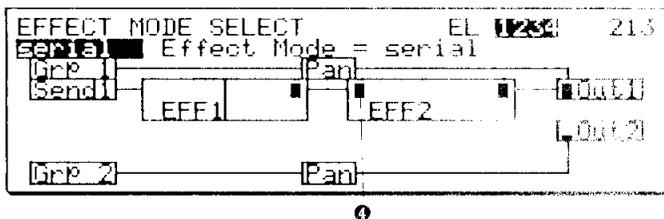
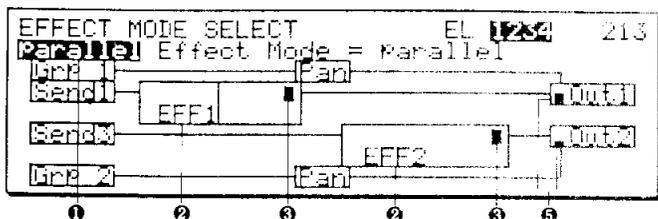
10.1 Effect mode select

JUMP #213

Summary: Specify the effect mode to determine how the two effect units are connected, the effect type for each unit, effect output levels, and the volume balance between processed and unprocessed sound.

Procedure:

- From: Effect Set job directory (JUMP #212)
Select: 01:Effect Mode Select (JUMP #213)
Specify: the effect mode, effect types, and effect levels, and wet:dry balance.



- Effect Mode (off, serial, parallel): This determines how the two effect units will be connected. The effect mode will be graphically shown in the display.

When the Effect Mode is set to Off, the effect units will not be used. No effect processed sound will be added to the sound sent to output groups 1 and 2 from the two stereo groups.

When the Effect Mode is set to Serial, the audio signal from the sends (one, two, or three, depending on the type of effect selected for each effect unit) will be sent to the two effect units as shown in the display. The output of effect 1 will be sent through effect 2; i.e., the effect units will be connected in "series".

When the Effect Mode is set to Parallel, the audio signal from the sends (one, two, three, or four, depending on the type of effect selected for each effect unit) will be sent to the two effect units as shown in the display.

- Effect 1 Type, Effect 2 Type (1...63): These determine the type of effect for each of the two effect units. A list of the 63 effects and their parameters is given in the following section 10.3 *Effect 1 parameter* (page 108).
- EFF1/EFF2 Output level 1, 2 (0...100%): These determine the output level of each effect. The level is indicated as a number and also as a bar graph. If a dual-type effect (52-63) is selected, the output level can be set independently for each effect. At a setting of 0% the effect unit will output no sound, and at a setting of 100% the output of the effect unit will be at full volume.
- Effect 2 Mix Level (0...100%): This will appear only if the Effect Mode is set to Serial. It determines the amount of Effect 2 that is added to the effect chain.
- Out1/Out2 wet:dry (100:0...0:100): These determine the mix between the processed (wet) and unprocessed (dry) sounds for each output 1 and 2. When set to 100:0, only the processed sound from the effect unit will be heard. When set to 0:100, only the unprocessed sound from the dynamic pan will be heard.

If the wet:dry balance is set to 100:0, the sound from the stereo pan will not be heard, meaning that the element dynamic pan settings will have no effect.

Single type effects

- 01. Rev Hall
- 02. Rev Room 1
- 03. Rev Room 2
- 04. Rev Room 3
- 05. Rev Stage 1
- 06. Rev Stage 2
- 07. Rev Plate
- 08. Rev White Room
- 09. Rev Tunnel
- 10. Rev Canyon
- 11. Rev Basement
- 12. Early Reflection 1
- 13. Early Reflection 2
- 14. Gate Rev
- 15. Reverse
- 16. Delay L,R
- 17. Delay L,C,R
- 18. Stereo Echo
- 19. Pitch Change 1
- 20. Pitch Change 2
- 21. Pitch Change 3
- 22. Aural Exciter®*
- 23. EG Flange
- 24. EG Chorus
- 25. EG Symphonic
- 26. EG Phasing
- 27. Rotary SP
- 28. Ring Modulator
- 29. D. Filter (Wah)

- 33. Chorus → Rev
- 34. Sympho → Rev
- 35. Phase → Rev
- 36. Aural Exciter®* → Rev
- 37. Dist → Rev
- 38. Dist → Dly L,R
- 39. Dist → St.Echo
- 40. EQ → Rev 1 (Hall)
- 41. EQ → Rev 2 (Room)
- 42. EQ → Delay L,R
- 43. EQ → St.Echo
- 44. EQ → St.Flange
- 45. EQ → St.Chorus
- 46. EQ → Symphonic
- 47. EQ → St.Phasing
- 48. St.Flange → Delay L,R
- 49. St.Chorus → Delay L,R
- 50. Symphonic → Delay L,R
- 51. St.Phasing → Delay L,R

Cascade type effects

- 30. Echo → Rev
- 31. Delay L,R → Rev
- 32. Flange → Rev

Dual type effects

- 52. Hall & Plate
- 53. Echo & Rev
- 54. Delay & Rev
- 55. St.Flange & St.Chorus
- 56. St.Flange & Symphonic
- 57. Symphonic & St.Chorus
- 58. St.Flange & Rev
- 59. St.Chorus & Rev
- 60. Symphonic & Rev
- 61. St.Flange & Delay L,R
- 62. St.Chorus & Delay L,R
- 63. Symphonic & Delay L,R

* Aural Exciter® is a registered trademark and is manufactured under license from APHEX Systems Ltd.

COMMON DATA / EFFECT SET

10.2 Effect send

JUMP #219

Summary: Specify how the sound from each element of the voice will sent to each of the one, two, three, or four effect sends.

Procedure:

- From: Effect Set job directory (JUMP #212)
- Select: job 02:Effect Send (JUMP #219)
- Specify: the effect send settings for each element

EFFECT SEND						EL	219
VOICE=I -D01(49) ST*Concert							
	Send	Sel	Level	Vel	Sens	Scaling	
E1:AFM	1	-	127		+0	+0	
E2:AFM	1	-	127		+0	+0	
E3:AWM	1	-	127		+0	+0	
E4:AWM	1	-	127		+0	+0	

VOICE EDIT MODE

- ❶ Send Sel (send select 1–4): Specify whether or not to send the sound from the element to each effect send 1–4. The number of effect sends available will depend on the effect mode and the effect type. Non-existent sends will be indicated by a dash (-), and cannot be selected. In the above display, effect sends 1 and 2 are available. If an effect send is turned on, its number will be displayed, and the sound of the element will be sent to that effect send. If an effect send is turned off, a period (.) will be displayed, and the sound of the element will not be sent to that effect send. In the above display, element 1 is being sent to effect sends 1 and 3 only.
- ❷ Level (0...127): Specify how much sound will be sent from the element to the effect unit.
- ❸ Vel Sens (-7...+7): Specify how key velocity will affect the effect send level of the element. For positive settings (+1...+7), more sound will be sent to the effect unit for strongly played notes. For negative settings (-1...-7), less sound will be sent to the effect unit for strongly played notes.
- ❹ Scaling (-7...+7): Specify how key position will affect the effect send level of the element. For positive settings (+1...+7), higher notes will send more sound to the effect unit. For negative settings (-1...-7), higher notes will send less sound to the effect unit.
- ❺ Pressing F1–F4 will move the cursor to elements 1–4.

COMMON DATA / EFFECT SET

10.3 Effect 1 parameter

JUMP #214

Summary: Set effect parameters for the effect type selected for effect 1. This explanation also applies to effect 1.

Procedure:

- From: Effect Set job directory (JUMP #212)
- Select: job 03:Effect 1 Parameter (JUMP #214)
- Specify: parameter settings for the selected effect.

EFFECT 1 PARAMETER		EL	214
Parallel EFF1 Type =	46:EQ	->	Symphonic
01:Low Frequency	=	315	Hz
02:Low Gain	=	+02	dB
03:Mid Frequency	=	2.0	KHz
04:Mid Gain	=	-10	dB
05:High Frequency	=	12.0	KHz

- ❶ The number and type of effect parameters will depend on the selected Effect Type. Refer to the following effect parameter tables. Parameter numbers which do not exist for the selected effect type will be displayed as a dash (-).
- ❷ Press F1 or F2 to move between effect parameters 1–5 and effect parameters 6–10.

Effect parameters: Many of the effect types have similar parameters. Commonly-appearing parameters are explained below. Parameters unique to a specific effect type will be explained after the parameter list for that effect type.

- AM Depth: the amount of amplitude modulation
- Delay Time: the delay before the processed sound is heard
- Density: the density of reverberation
- Diffusion: the spatial spread of the reverberant sound
- Feedback Delay (FB Delay Time): the interval between repeated echoes
- Feedback Gain: the volume ratio of each echo to the previous one (negative settings will invert the phase)
- Fine (Pitch Fine): fine adjustment in cents (1/100th of a semitone) to the pitch change
- High (Rev High): the proportion at which high frequency reverb will decay in relation to the overall reverb time
- High Control: the proportion at which high frequencies will be retained in each successive feedback echo
- High Gain: boost or cut of the high frequencies
- HPF (high-pass filter): frequencies above this setting will be allowed to pass
- Initial Delay: the delay before the processed sound will be heard
- Low Gain: boost or cut of the low frequencies
- LPF (low-pass filter): frequencies below this setting will be allowed to pass
- PM Depth: the amount of pitch modulation
- Modulation Delay: the time delay before the modulation begins to occur

Modulation Depth: the amount of modulation (flanging, chorusing, etc.)

Modulation Frequency: the speed of modulation (flanging, chorusing, etc.)

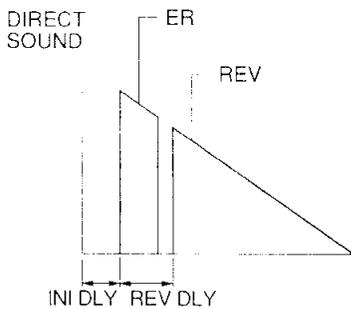
Pitch: the amount of pitch change in semi-tones

Reverb Time: the time during which the level of the reverb sound will decrease by 60 dB

Single type effects

1:Rev Hall, 2:Rev Room 1, 3:Rev Room 2, 4:Rev Room 3, 5:Rev Stage 1, 6:Rev Stage 2, 7:Rev Plate

No.	Parameter Name	Range
1	Reverb Time	0.3 ... 30.0 sec
2	High	0.1 ... 1.0
3	Diffusion	0 ... 10
4	Initial Delay	0.1 ... 200.0 ms
5	Rev Delay	0.1 ... 200.0 ms
6	Density	0 ... 4
7	ER/Rev Balance	0 ... 100 %
8	Low Gain	-12 ... +12 dB
9	High Gain	-12 ... +12 dB
10	LPF	1 ... 16 kHz, Thru



Initial Delay: This parameter sets the time delay before the early reflections begin.

Rev Delay: This parameter sets the delay between the beginning of the early reflections and the beginning of the reverb.

ER/Rev Balance: This determines the level balance between early reflections and reverb. At a setting of 0%, only the reverb will be heard. At a setting of 100%, only the early reflections will be heard.

8:Rev White Room, 9:Rev Tunnel, 10:Rev Canyon, 11:Rev Basement

No.	Parameter Name	Range
1	Reverb Time	0.3 ... 30.0 sec
2	High	0.1 ... 1.0
3	Diffusion	0 ... 10
4	Initial Delay	0.1 ... 200.0 ms
5	Width	0.5 ... 30.2 m
6	Height	0.5 ... 30.2 m
7	Depth	0.5 ... 30.2 m
8	Wall Vary	0 ... 30
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

Width, Height, Depth: These parameters determine the size of the simulated room in units of meters.

Wall Vary: This parameter determines the irregularity of the wall surfaces. At a setting of 0, the walls will be perfectly regular. Higher settings will simulate more irregular walls, creating more complex reverberation.

12:Early Ref. 1, 13:Early Ref. 2

No.	Parameter Name	Range
1	Type	S-Hall, L-Hall, Random, Reverse, Plate, Spring
2	Room Size	0.1 ... 20.0
3	Liveness	0 ... 10
4	Diffusion	0 ... 10
5	Initial Delay	0.1 ... 400.0 ms
6	ER Number	1 ... 19
7	Feedback Delay	0.1 ... 800.0 ms
8	Feedback Gain	-99 ... +99 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

Room Size: This parameter determines the size of the simulated room in arbitrary units. Higher settings will result in a larger room with more space between reflections.

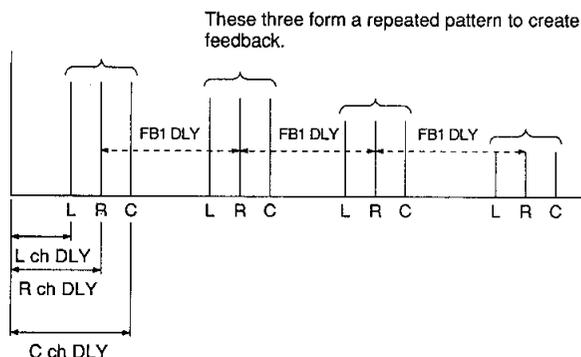
Liveness: This parameter determines how early reflections will decay. Higher settings will result in a more live sound.

ER Number: This parameter determines the number of early reflections that will occur.

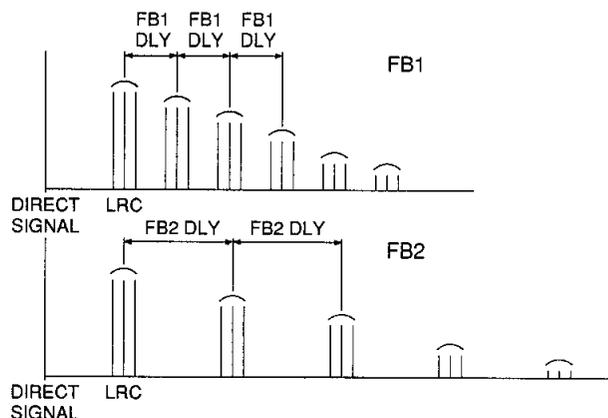
VOICE EDIT MODE

14:Gate Reverb, 15: Reverse Gate

No.	Parameter Name	Range
1	Type	Type A, Type B
2	Room Size	0.1 ... 20.0
3	Liveness	0 ... 10
4	Diffusion	0 ... 10
5	Initial Delay	0.1 ... 400.0 ms
6	ER Number	1 ... 19
7	Feedback Delay	0.1 ... 800.0 ms
8	Feedback Gain	-99 ... +99 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

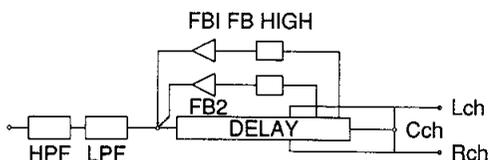


The following diagram shows how the pattern of the two/three delayed sounds (Lch Delay Time, Rch Delay Time, Center Delay Time) is repeated at intervals of the feedback time values FB1 and FB2.



16:Delay L,R

No.	Parameter Name	Range
1	Lch Delay Time	0.1 ... 1360.0 ms
2	Rch Delay Time	0.1 ... 1360.0 ms
3	—	—
4	FB1 Delay Time	0.1 ... 1360.0 ms
5	FB2 Delay Time	0.1 ... 1360.0 ms
6	FB Gain	-99 ... +99 %
7	FB1 High Control	0.1 ... 1.0
8	FB2 High Control	0.1 ... 1.0
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru



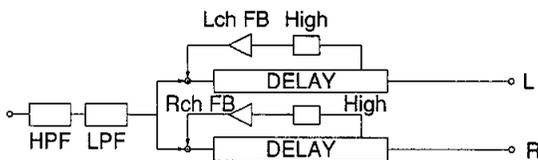
These delay programs provide a dual delay (16.Delay L,R) or triple delay (17.Delay L,C,R) with two feedback loops. The following diagram shows the pattern created by the two/three delay times, and the effect of FB1.

17:Delay L,C,R

No.	Parameter Name	Range
1	Lch Delay Time	0.1 ... 1360.0 ms
2	Rch Delay Time	0.1 ... 1360.0 ms
3	Center Delay Time	0.1 ... 1360.0 ms
4	FB1 Delay Time	0.1 ... 1360.0 ms
5	FB2 Delay Time	0.1 ... 1360.0 ms
6	FB Gain	-99 ... +99 %
7	FB1 High Control	0.1 ... 1.0
8	FB2 High Control	0.1 ... 1.0
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

18: Stereo Echo

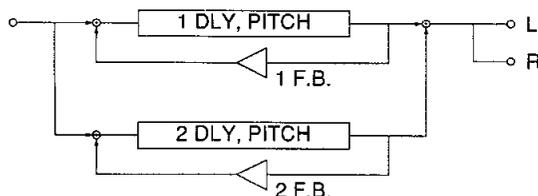
No.	Parameter Name	Range
1	Lch Init Dly Time	0.1 ... 680.0 ms
2	Lch FB Delay Time	0.1 ... 680.0 ms
3	Lch FB Gain	-99 ... +99 %
4	Rch Init Dly Time	0.1 ... 680.0 ms
5	Rch FB Delay Time	0.1 ... 680.0 ms
6	Rch FB Gain	-99 ... +99 %
7	Lch FB High con.	0.1 ... 1.0
8	Rch FB High con.	0.1 ... 1.0
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru



This effect type provides two independent delays with independent feedback.

19: Pitch Change 1

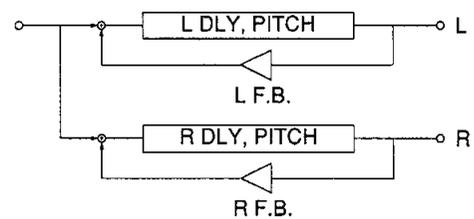
No.	Parameter Name	Range
1	1 Pitch	-24 ... +24
2	1 Fine	-100 ... +100
3	1 Delay	0.1 ... 650.0 ms
4	1 FB Gain	-99 ... +99 %
5	1 Level	0 ... 100 %
6	2 Pitch	-24 ... +24
7	2 Fine	-100 ... +100
8	2 Delay	0.1 ... 650.0 ms
9	2 FB Gain	-99 ... +99 %
10	2 Level	0 ... 100 %



This effect type provides two independent pitch changes, each with delay and feedback. The sound of both pitch changes is combined into both the left and right outputs.

20: Pitch Change 2

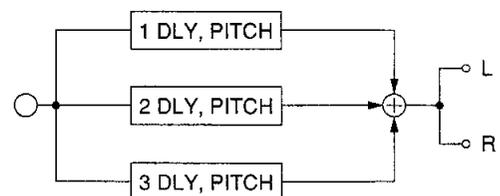
No.	Parameter Name	Range
1	L Pitch	-24 ... +24
2	L Fine	-100 ... +100
3	L Delay	0.1 ... 650.0 ms
4	L FB Gain	-99 ... +99 %
5	R Pitch	-24 ... +24
6	R Fine	-100 ... +100
7	R Delay	0.1 ... 650.0 ms
8	R FB Gain	-99 ... +99 %
9	—	—
10	—	—



This effect type provides two independent pitch changes, each with delay and feedback. The sound of each pitch change is sent independently from the left and right outputs.

21: Pitch Change 3

No.	Parameter Name	Range
1	1 Pitch	-24 ... +24
2	1 Fine	-100 ... +100
3	1 Delay	0.1 ... 1300.0 ms
4	2 Pitch	-24 ... +24
5	2 Fine	-100 ... +100
6	2 Delay	0.1 ... 1300.0 ms
7	3 Pitch	-24 ... +24
8	3 Fine	-100 ... +100
9	3 Delay	0.1 ... 1300.0 ms
10	—	—



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This effect type provides three independent pitch changes, each with delay. The sound of all three pitch changes is combined into both left and right outputs.

22: Aural Exciter®*

No.	Parameter Name	Range
1	HPF	500 Hz ... 16 kHz
2	Enhance	0 ... 100 %
3	Mix Level	0 ... 100 %
4	Delay Time	0.1 ... 650 ms
5	—	—
6	—	—
7	—	—
8	—	—
9	—	—
10	—	—

This effect adds new harmonics to a sound to make it “stand out” in the mix.

HPF: Overtones will be added to frequencies above this frequency.

Enhance: Higher settings of this value will increase the level of the newly generated harmonics.

Mix Level: This determines the mix of the direct sound and the sound produced by the exciter. Higher settings will increase the amount of sound from the exciter.

Delay: This determines the delay which will be applied to the sound produced by the exciter.

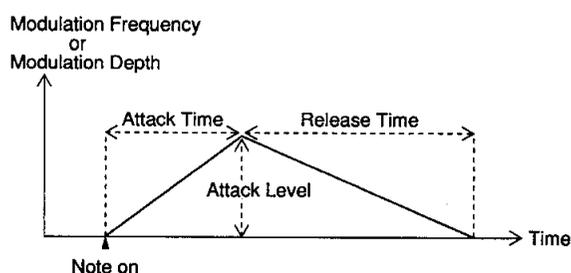
* *Aural Exciter® is a registered trademark and is manufactured under license from APHEX Systems Ltd.*

23:EG Flange

No.	Parameter Name	Range
1	Attack Time	2.0 ... 22000 ms
2	Attack Level	0 ... 100 %
3	Release Time	2.0 ... 22000 ms
4	EG Target	0:M.Freq, 1:M.Dep
5	Low Gain	-12 ... +12 dB
6	High Gain	-12 ... +12 dB
7	Modulation Freq.	0.05 ... 40 Hz
8	Modulation Depth	0 ... 100 %
9	Modulation Delay	0.1 ... 100.0 ms
10	Mod. FB Gain	0 ... 99 %

This effect provides a Flanger whose modulation frequency or modulation depth can be controlled over time by a simple envelope.

EG Target: This determines which parameter will be controlled by the envelope. If “M.Freq” is selected, the envelope will increase and then decrease the modulation frequency over time. If “M.Dep” is selected, the envelope will increase and then decrease the modulation depth over time. **Attack Time, Attack Level, Release Time:** These parameters determine the shape of the envelope. This envelope is re-started each time a note is played, and will go through its full contour regardless of whether the note is held or released.



24:EG Chorus

No.	Parameter Name	Range
1	Attack Time	2.0 ... 22000 ms
2	Attack Level	0 ... 100 %
3	Release Time	2.0 ... 22000 ms
4	EG Target	0:M.Freq, 1:M.Dep
5	Low Gain	-12 ... +12 dB
6	High Gain	-12 ... +12 dB
7	Chorus Mod. Freq.	0.05 ... 40 Hz
8	Chorus PM Depth	0 ... 100 %
9	Chorus AM Depth	0 ... 100 %
10	—	—

This effect provides a Chorus whose modulation frequency or modulation depth can be controlled over time by a simple envelope. For details, refer to the explanation of *23.EG Flange*.

25:EG Symphonic

No.	Parameter Name	Range
1	Attack Time	2.0 ... 22000 ms
2	Attack Level	0 ... 100 %
3	Release Time	2.0 ... 22000 ms
4	EG Target	0:M.Freq, 1:M.Dep
5	Low Gain	-12 ... +12 dB
6	High Gain	-12 ... +12 dB
7	Modulation Freq.	0.05 ... 40.0 Hz
8	Modulation Depth	0 ... 100 %
9	—	—
10	—	—

This effect provides a Symphonic effect whose modulation frequency or modulation depth can be controlled over time by a simple envelope. For details, refer to the explanation of *23.EG Flange*. Symphonic is an effect that is similar to chorusing, but with multiple modulation and more delay time modulation.

26:EG Phase

No.	Parameter Name	Range
1	Attack Time	3.0 ... 22000 ms
2	Attack Level	0 ... 100 %
3	Release Time	3.0 ... 22000 ms
4	EG Target	0:M.Freq, 1:M.Dep
5	Low Gain	-12 ... +12 dB
6	High Gain	-12 ... +12 dB
7	Modulation Freq.	0.05 ... 40 Hz
8	Modulation Depth	0 ... 100 %
9	Modulation Delay	0.1 ... 5.0 ms
10	—	—

This effect provides a Phasing effect whose modulation frequency or modulation depth can be controlled over time by a simple envelope. For details, refer to *23.EG Flange* (page 112).

27:Rotary Speaker

No.	Parameter Name	Range
1	Mid Speed	0.05 ... 40 Hz
2	Depth	0 ... 100 %
3	Transition Time	2 ... 22000 ms
4	L/M/H Speed Diff.	0.05 ... 5.80 Hz
5	Switch L/M/H	0:Low, 1:Mid, 2:High
6	Low Gain	-12 ... +12 dB
7	High Gain	-12 ... +12 dB
8	—	—
9	—	—
10	—	—

This effect simulates a rotary speaker that can be made to change smoothly between high, mid and low speeds. By assigning a controller to affect parameter 5 (Switch L/M/H) of this effect, you can switch between slow and fast rotor speeds as you play.

Mid Speed: This determines the mid rotor speed.

Transition time: This determines the time over which the rotor will speed up or slow down when switched between low, mid, and high speed.

L/M/H Speed Diff.: The high speed is faster than mid speed, and the low speed is slower than mid speed, by this amount.

Switch L/M/H: This switch is used to select low, mid, or high rotor speed. It may be useful musically to assign a controller to affect this parameter.

28:Ring Modulator

No.	Parameter Name	Range
1	Wave PM Depth	0 ... 100 %
2	Wave PM Frq.	0.05 ... 40.0 Hz
3	Wave AM Depth	0 ... 100 %
4	Wave AM Frq.	0.05 ... 40.0 Hz
5	Low Gain	-12 ... +12 dB
6	High Gain	-12 ... +12 dB
7	—	—
8	—	—
9	—	—
10	—	—

VOICE EDIT MODE

This effect modifies the input waveform by multiplying it with a sine wave. The pitch and amplitude characteristics of this sine wave may be altered freely.

Wave PM Depth: This determines the depth of sine wave pitch modulation depth.

Wave PM Frq.: This determines the depth of sine wave pitch modulation frequency.

Wave AM Depth: This determines the depth of sine wave amplitude modulation depth.

Wave AM Frq.: This determines the depth of sine wave amplitude modulation frequency.

29:D. Filter (Wah)

No.	Parameter Name	Range
1	Filter Frq.	125 Hz ... 14 kHz
2	Filter 1 Q	1.0 ... 5.0
3	Filter 1 Gain	0 ... +12 dB
4	Filter 2 Q	0.1 ... 0.7
5	---	---
6	---	---
7	---	---
8	---	---
9	---	---
10	---	---

This effect applies a wah-wah effect to the input sound using two filters. Filter 1 is a presence filter; Filter 2 is a band pass filter (BPF).

Filter Frq.: This determines the frequencies of Filters 1 and 2.

Filter 1 Q: This determines the width characteristic of Filter 1.

Filter 1 Gain: This determines the gain of Filter 1.

Filter 2 Q: This determines the width characteristic of Filter 2.

30:Echo → Rev

No.	Parameter Name	Range
1	Echo Lch Delay	0.1 ... 320.0 ms
2	Echo Lch FB Gain	-99 ... +99 %
3	Echo Rch Delay	0.1 ... 320.0 ms
4	Echo Rch FB Gain	-99 ... +99 %
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev ER/Rev Balance	0 ... 100 %
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides an echo effect which is then fed into a reverb effect. For details refer to 18.Stereo Echo (page 111) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between echo and reverb sound. At a setting of 0%, only the echo sound will be heard. At a setting of 100%, only the reverb sound will be heard.

31:Delay L,R → Rev

No.	Parameter Name	Range
1	Dly Lch Delay	0.1 ... 640.0 ms
2	Dly Rch Delay	0.1 ... 640.0 ms
3	Dly FB Gain	-99 ... +99 %
4	Dly FB High	0.1 ... 1.0
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev ER/Rev Balance	0 ... 100 %
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides an delay effect which is then fed into a reverb effect. For details refer to 16.Delay L,R (page 110) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between delay and reverb sound. At a setting of 0%, only the delay sound will be heard. At a setting of 100%, only the reverb sound will be heard.

32:Flange → Rev

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40 Hz
2	Modulation Depth	0 ... 100 %
3	Modulation Delay	0.1 ... 100.0 ms
4	Mod. FB Gain	0 ... 99 %
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev Initial Delay	0.1 ... 200.0 ms
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a flange effect which is then fed into a reverb effect. For details refer to 23.EG Flange (page 112) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between flange and reverb sound. At a setting of 0%, only the flange sound will be heard. At a setting of 100%, only the reverb sound will be heard.

33:Chorus → Rev

No.	Parameter Name	Range
1	Chorus Mod. Freq.	0.05 ... 40 Hz
2	Chorus PM Depth	0 ... 100 %
3	Chorus AM Depth	0 ... 100 %
4	—	—
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev Initial Delay	0.1 ... 200.0 ms
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a chorus effect which is then fed into a reverb effect. For details refer to 24.EG Chorus (page 112) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between chorus and reverb sound. At a setting of 0%, only the chorus sound will be heard. At a setting of 100%, only the reverb sound will be heard.

34:Sympo → Rev

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40 Hz
2	Modulation Depth	0 ... 100 %
3	—	—
4	—	—
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev Initial Delay	0.1 ... 200.0 ms
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a symphonic effect which is then fed into a reverb effect. For details refer to 25.EG Symphonic (page 113) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between symphonic and reverb sound. At a setting of 0%, only the symphonic sound will be heard. At a setting of 100%, only the reverb sound will be heard.

35:Phaser → Rev

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	Modulation Delay	0.1 ... 5.0 ms
4	—	—
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev Initial Delay	0.1 ... 200.0 ms
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a phase effect which is then fed into a reverb effect. For details refer to 26.EG Phase (page 113) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between phase and reverb sound. At a setting of 0%, only the phase sound will be heard. At a setting of 100%, only the reverb sound will be heard.

VOICE EDIT MODE

36:Aural Exc → Rev (Aural Exciter®*)

No.	Parameter Name	Range
1	Ex. HPF	500 Hz ... 16 kHz
2	Ex. Enhance	0 ... 100 %
3	Ex. Mix Level	0 ... 100 %
4	Reverb Time	0.3 ... 30.0 sec
5	Rev High	0.1 ... 1.0
6	Rev Diffusion	0 ... 10
7	Rev Initial Delay	0.1 ... 400.0 ms
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides an exciter effect which is then fed into a reverb effect. For details refer to 22.Aural Exciter (page 112) and 1.Rev Hall (page 109).

Rev Mix Level: This determines the proportion between exciter and reverb sound. At a setting of 0%, only the exciter sound will be heard. At a setting of 100%, only the reverb sound will be heard.

* *Aural Exciter® is a registered trademark and is manufactured under license from APHEX Systems Ltd.*

37:Dist → Rev

No.	Parameter Name	Range
1	Distortion Level	0 ... 100 %
2	Middle Frequency	315 Hz ... 6.3 kHz
3	Mid Gain	-12 ... +12 dB
4	Treble Gain	-12 ... +12 dB
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Rev Initial Delay	0.1 ... 400.0 ms
8	Rev Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a distortion effect which is then fed into a reverb effect. For details of the reverb parameters, refer to 1.Rev Hall (page 109).

Distortion Level: This determines the degree of distortion. At higher settings, the sound will be strongly distorted.

Rev Mix Level: This determines the proportion between distortion and reverb sound. At a setting of 0%, only the distortion sound will be heard. At a setting of 100%, only the reverb sound will be heard.

38:Dist → Dly L,R

No.	Parameter Name	Range
1	Distortion Level	0 ... 100 %
2	Middle Frequency	315 Hz ... 6.3 kHz
3	Mid Gain	-12 ... +12 dB
4	Treble Gain	-12 ... +12 dB
5	Dly Lch Delay	0.1 ... 1360.0 ms
6	Dly Rch Delay	0.1 ... 1360.0 ms
7	Dly FB Gain	-99 ... +99 %
8	Delay Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a distortion effect which is then fed into a delay effect. For details refer to 36.Dist→Rev above and 16.Delay L,R (page 110).

Delay Mix Level: This determines the proportion between distortion and delay sound. At a setting of 0%, only the distortion sound will be heard. At a setting of 100%, only the delay sound will be heard.

39:Dist → Echo

No.	Parameter Name	Range
1	Distortion Level	0 ... 100 %
2	Middle Frequency	315 Hz ... 6.3 kHz
3	Mid Gain	-12 ... +12 dB
4	Treble Gain	-12 ... +12 dB
5	Echo Lch Delay	0.1 ... 680.0 ms
6	Echo Rch Delay	0.1 ... 680.0 ms
7	Echo FB Gain	-99 ... +99 %
8	Echo Mix Level	0 ... 100 %
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect type provides a distortion effect which is then fed into a stereo echo effect. For details refer to 36.Dist→Rev above and 18.Stereo Echo (page 111).

Echo Mix Level: This determines the proportion between distortion and echo sound. At a setting of 0%, only the distortion sound will be heard. At a setting of 100%, only the echo sound will be heard.

40:EQ → Rev 1, 41:EQ → Rev 2

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Reverb Time	0.3 ... 30.0 sec
8	Rev Initial Delay	0.1 ... 400.0 ms
9	Rev ER/Rev Balance	0 ... 100 %
10	Rev Mix Level	0 ... 100 %

These effect types provide a three-band sweepable equalizer which is then fed into a reverb effect. For details refer to 1.Hall Rev (page 109).

42:EQ → Dly L,R

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Dly Lch Delay	0.1 ... 1360.0 ms
8	Dly Rch Delay	0.1 ... 1360.0 ms
9	Dly FB Gain	-99 ... +99 %
10	Delay Mix Level	0 ... 100 %

This effect type provides a three-band sweepable equalizer which is then fed into a delay effect. For details refer to 16.Delay L,R (page 110).

43:EQ → Echo

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Echo Lch Delay	0.1 ... 680.0 ms
8	Echo Rch Delay	0.1 ... 680.0 ms
9	Echo FB Gain	-99 ... +99 %
10	Echo Mix Level	0 ... 100 %

This effect type provides a three-band sweepable equalizer which is then fed into a stereo echo effect. For details refer to 18.Stereo Echo (page 111).

44:EQ → Flange

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Modulation Freq.	0.05 ... 40.0 Hz
8	Modulation Depth	0 ... 100 %
9	Mod. FB Gain	0 ... 99 %
10	Flange Mix Level	0 ... 100 %

This effect type provides a three-band sweepable equalizer which is then fed into a flange effect. For details refer to 23.EG Flange (page 112).

VOICE EDIT MODE

45:EQ → Chorus

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Chorus Mod. Freq.	0.05 ... 40.0 Hz
8	Chorus PM Depth	0 ... 100 %
9	Chorus AM Depth	0 ... 100 %
10	Chorus Mix Level	0 ... 100 %

This effect type provides a three-band sweepable equalizer which is then fed into a chorus effect. For details refer to 24.EG Chorus (page 112).

46:EQ → Symphonic

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Modulation Freq.	0.05 ... 40.0 Hz
8	Modulation Depth	0 ... 100 %
9	—	—
10	Sympho Mix Level	0 ... 100 %

This effect type provides a three-band sweepable equalizer which is then fed into a symphonic effect. For details refer to 25.EG Symphonic (page 113).

47:EQ → Phaser

No.	Parameter Name	Range
1	Low Frequency	32 Hz ... 2.0 kHz
2	Low Gain	-12 ... +12 dB
3	Middle Frequency	315 Hz ... 6.3 kHz
4	Mid Gain	-12 ... +12 dB
5	High Frequency	500 Hz ... 16 kHz
6	High Gain	-12 ... +12 dB
7	Modulation Freq.	0.05 ... 40.0 Hz
8	Modulation Depth	0 ... 100 %
9	Modulation Delay	0.1 ... 5.0 ms
10	Phase Mix Level	0 ... 100 %

This effect type provides a three-band sweepable equalizer which is then fed into a phaser effect. For details refer to 26.EG Phase (page 113).

48:St.Flange → Delay L,R

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	Modulation Delay	0.1 ... 100.0 ms
4	Mod. FB Gain	0 ... 99 %
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	Delay Mix Level	0 ... 100 %

This effect type provides a flange effect which is then fed into a stereo delay. For details refer to 23.EG Flange (page 112) and 16.Delay L,R (page 110).

49:St.Chorus → Delay L,R

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Chorus PM Depth	0 ... 100 %
3	Chorus AM Depth	0 ... 100 %
4	—	—
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	Delay Mix Level	0 ... 100 %

This effect type provides a chorus effect which is then fed into a stereo delay. For details refer to 24.EG Chorus (page 112) and 16.Delay L,R (page 110)

50:Symphonic → Delay L,R

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	—	—
4	—	—
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	Delay Mix Level	0 ... 100 %

This effect type provides a symphonic effect which is then fed into a stereo delay. For details refer to 25.EG Symphonic (page 113) and 16.Delay L,R (page 110)

51:St.Phasing → Delay L,R

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	Modulation Delay	0.1 ... 5.0 ms
4	—	—
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	Delay Mix Level	0 ... 100 %

This effect type provides a phaser effect which is then fed into a stereo delay. For details refer to 26.EG Phase (page 113) and 16.Delay L,R (page 110)

Dual type effects**52:Hall & Plate**

No.	Parameter Name	Range
1	Hall Reverb Time	0.3 ... 30.0 sec
2	Hall High	0.1 ... 1.0
3	Hall Diffusion	0 ... 10
4	Hall Initial Delay	0.1 ... 200.0 ms
5	Hall LPF	1 ... 16 kHz, Thru
6	Plate Reverb Time	0.3 ... 30.0 sec
7	Plate High	0.1 ... 1.0
8	Plate Diffusion	0 ... 10
9	Plate Initial Delay	0.1 ... 200.0 ms
10	Plate LPF	1 ... 16 kHz, Thru

This effect provides plate reverb and hall reverb in parallel. For details of the parameters, refer to 1.Rev Hall (page 109).

VOICE EDIT MODE

53:Echo & Rev

No.	Parameter Name	Range
1	Echo Lch Delay Time	0.1 ... 320.0 ms
2	Echo Rch Delay Time	0.1 ... 320.0 ms
3	Echo FB Gain	-99 ... +99 %
4	Reverb Time	0.3 ... 30.0 sec
5	Rev High	0.1 ... 1.0
6	Rev Diffusion	0 ... 10
7	Rev Initial Delay	0.1 ... 100.0 ms
8	Rev ER/Rev Balance	0 ... 100 %
9	Rev HPF	Thru, 32 ... 1000 Hz
10	Rev LPF	1 ... 16 kHz, Thru

This effect provides echo and reverb in parallel. For details refer to 18.Stereo Echo (page 111) and 1.Rev Hall (page 109).

54:Delay & Rev

No.	Parameter Name	Range
1	Dly Lch Delay Time	0.1 ... 640.0 ms
2	Dly Rch Delay Time	0.1 ... 640.0 ms
3	Dly FB Gain	-99 ... +99 %
4	Reverb Time	0.3 ... 30.0 sec
5	Rev High	0.1 ... 1.0
6	Rev Diffusion	0 ... 10
7	Rev Initial Delay	0.1 ... 100.0 ms
8	Rev ER/Rev Balance	0 ... 100 %
9	Rev HPF	Thru, 32 ... 1000 Hz
10	Rev LPF	1 ... 16 kHz, Thru

This effect provides delay and reverb in parallel. For details refer to 16.Delay L,R (page 110) and 1.Rev Hall (page 109).

55:Flange & Chorus

No.	Parameter Name	Range
1	Flg Mod. Freq.	0.05 ... 40.0 Hz
2	Flg Mod. Depth	0 ... 100 %
3	Flg Mod. Delay	0.1 ... 100.0 ms
4	Flg Mod. FB Gain	0 ... 99 %
5	---	---
6	Chorus Mod. Freq.	0.05 ... 40.0 Hz
7	Chorus PM Depth	0 ... 100 %
8	Chorus AM Depth	0 ... 100 %
9	---	---
10	---	---

This effect provides flange and chorus in parallel. For details refer to 23.EG Flange (page 112) and 24.EG Chorus (page 112).

56:Flange & Sympho

No.	Parameter Name	Range
1	Flg Mod. Freq.	0.05 ... 40.0 Hz
2	Flg Mod. Depth	0 ... 100 %
3	Flg Mod. Delay	0.1 ... 100.0 ms
4	Flg Mod. FB Gain	0 ... 99 %
5	---	---
6	Sym Mod. Freq.	0.05 ... 40.0 Hz
7	Sym Mod. Depth	0 ... 100 %
8	---	---
9	---	---
10	---	---

This effect provides flange and symphonic in parallel. For details refer to 23.EG Flange (page 112) and 25.EG Symphonic (page 113).

57:Sympho & Chorus

No.	Parameter Name	Range
1	Sym Mod. Freq.	0.05 ... 40.0 Hz
2	Sym Mod. Depth	0 ... 100 %
3	—	—
4	—	—
5	—	—
6	Chorus Mod. Freq.	0.05 ... 40.0 Hz
7	Chorus PM Depth	0 ... 100 %
8	Chorus AM Depth	0 ... 100 %
9	—	—
10	—	—

This effect provides chorus and symphonic in parallel. For details refer to 24.EG Chorus (page 112) and 25.EG Symphonic (page 113).

58:Flange & Rev

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	Modulation Delay	0.1 ... 100.0 ms
4	Mod. FB Gain	0 ... 99 %
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Diffusion	0 ... 10
8	Initial Delay	0.1 ... 200.0 ms
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect provides flange and reverb in parallel. For details refer to 23.EG Flange (page 112) and 1.Rev Hall (page 109).

59:Chorus & Rev

No.	Parameter Name	Range
1	Chorus Mod. Freq.	0.05 ... 40.0 Hz
2	Chorus PM Depth	0 ... 100 %
3	Chorus AM Depth	0 ... 100 %
4	—	—
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Diffusion	0 ... 10
8	Initial Delay	0.1 ... 200.0 ms
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect provides chorus and reverb in parallel. For details refer to 24.EG Chorus (page 112) and 1.Rev Hall (page 109).

60:Sympho & Rev

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	—	—
4	—	—
5	Reverb Time	0.3 ... 30.0 sec
6	Rev High	0.1 ... 1.0
7	Diffusion	0 ... 10
8	Initial Delay	0.1 ... 200.0 ms
9	HPF	Thru, 32 ... 1000 Hz
10	LPF	1 ... 16 kHz, Thru

This effect provides symphonic and reverb in parallel. For details refer to 25.EG Symphonic (page 104) and 1.Rev Hall (page 109).

VOICE EDIT MODE

61:Flange & Dly L,R

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	Modulation Delay	0.1 ... 100.0 ms
4	Mod. FB Gain	0 ... 99 %
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	LPF	1 ... 16 kHz, Thru

This effect provides flange and delay in parallel. For details refer to 23.EG Flange (page 112) and 16.Delay L,R (page 110).

62:Chorus & Dly L,R

No.	Parameter Name	Range
1	Chorus Mod. Freq.	0.05 ... 40.0 Hz
2	Chorus PM Depth	0 ... 100 %
3	Chorus AM Depth	0 ... 100 %
4	—	—
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	LPF	1 ... 16 kHz, Thru

This effect provides chorus and delay in parallel. For details refer to 24.EG Chorus (page 112) and 16.Delay L,R (page 110).

63:Sympho & Dly L,R

No.	Parameter Name	Range
1	Modulation Freq.	0.05 ... 40.0 Hz
2	Modulation Depth	0 ... 100 %
3	—	—
4	—	—
5	Delay Lch Delay	0.1 ... 800.0 ms
6	Delay Rch Delay	0.1 ... 800.0 ms
7	Delay FB1 Delay	0.1 ... 800.0 ms
8	Delay FB2 Delay	0.1 ... 800.0 ms
9	Delay FB Gain	-99 ... +99 %
10	LPF	1 ... 16 kHz, Thru

This effect provides symphonic and delay in parallel. For details refer to 25.EG Symphonic (page 113) and 16.Delay L,R (page 110).

COMMON DATA / EFFECT SET

10.4 Effect 2 parameter

JUMP #216

Summary: Specify effect parameters for the effect type selected for Effect 2.

Procedure:

From: Effect Set job directory (JUMP #212)

Select: job 04:Effect 2 Parameter (JUMP #216)

Specify: parameter settings for the selected effect.

Remark: Effect selection and parameter settings for Effect 2 are made in the same manner as those for Effect 1. Please refer to the explanation in the preceding section, *10.3 Effect 1 parameter* (page 108).

10.5 Effect control

Summary: Specify how two selected parameters of the effect unit will be controlled in realtime.

Procedure:

- From: Effect Set job directory (JUMP #212)
 Select: job 05:Effect Control (JUMP #218)
 Specify: the parameters to be controlled and the controller for each

Param	Ctrl# & Device	Mi	Ma
Unt1 EF1 Par1	off	0	99
Unt2 EF1 Par1	off	0	99
EF1	Overriable	Sp=55	DI=0 Ph=free
Unt1	Unt2 LFO		

- ① Param (see below): Select any two of the following effect parameters to be controlled. The selected controller will affect ...EF1 Par1-10: a parameter 1-10 of effect unit 1.
 EF1 Level1/2: the effect level 1 or 2 of effect unit 1.
 EF2 Mix Level: the amount of Effect 2 that is present in the effect chain. (This parameter is meaningful only when the Effect Mode is Serial.)
 EF2 Par1-10: a parameter 1-10 of effect unit 2.
 EF2 Level1/2: the effect level 1 or 2 of effect unit 2.
 Out1/2 w:d: the wet:dry balance of output 1 or 2.
 Cnt1 MIN: the percentage of a parameter's full range that will result when a controller moves to its minimum position (see ③).
 Cnt1 MAX: the percentage of a parameter's full range that will result when a controller moves to its maximum position (see ④).
 LFO Wave: the waveform of the Effect LFO (see ⑤).
 LFO Spd: Control the speed of the Effect LFO (see ⑤).
 LFO Dly: Control the delay time of the Effect LFO (see ⑤).
 LFO Phs: Control the initial phase of the Effect LFO (see ⑤).
- ② Ctrl# & Device (001...120, Aftertouch, Velocity, Key Scale, LFO): Select the control source which will affect the selected effect parameter. 001 Modulation ... 120 Non-Assign: This control change message will affect the effect

parameter selected in ①. (Control numbers 000 and 032 are reserved for the Bank Select message, and are not available.) For more details on controllers, refer to *Common data 12.(F2) Controller set (Modulation)*.

Aftertouch: Pressing down on the keyboard after playing a note will affect the effect parameter selected in ①.

Velocity: The key-on velocity of the most recently played note will affect the effect parameter selected in ①.

Key Scale: The keyboard position of the most recently played note will affect the effect parameter selected in ①. If "Mi" ③ is below "Ma" ④, higher notes will result in a higher parameter setting.

LFO: The Effect LFO will affect the effect parameter selected in ①. See ⑤ below.

- ③ Mi (0...99): When the controller is at minimum position (e.g., the modulation wheel is moved fully toward you, no aftertouch pressure is being applied, etc.), the parameter will be set to this percentage of its maximum value. (See also "Ma" ④.)
- ④ Ma (0...99): When the controller is at maximum position (e.g., the modulation wheel is moved fully away from you, the maximum aftertouch pressure is being applied, etc.), the parameter will be set to this percentage of its maximum value.

You can set the "Mi" value above the "Ma" value to reverse the effect of the controller. For example if "Mi" is 99 and "Ma" is 50, and you have set the "Ctrl# & Device" to be "001 Modulation", the selected parameter will be at 100% of its full range when the modulation wheel is moved fully toward you, and at 50% of its full range when the modulation wheel is moved fully away from you.

- ⑤ LFO: A special Effect LFO is provided for controlling the effect parameters you select in ①. You can set the waveform (WV= triangle, saw down, saw up, square, sine, sample&hold), the speed (Sp= 0...99), the delay (DI= 0...99), and the initial phase (Ph= 0...99, free) of the Effect LFO. For a detailed explanation of these parameters, refer to *AFM element data 6.(F1) AFM*

LFO (Main) (page 146), which is essentially identical to this Effect LFO.

The initial phase parameter determines the point of the LFO waveform from which the LFO will begin each time a key is pressed. When this is set to "free", the Effect LFO will not be restarted for each note.

This Effect LFO is independent of the LFO which provides the "modulation" for some effect types. For example, you could even use this Effect LFO to modulate the modulation frequency of a chorus effect.

- ⑥ Pressing F1–F3 will move the cursor to the Control 1, Control 2, or LFO area.

Remark: The assignment of some effect parameters to realtime control can result in noise. Should this happen, try modifying the settings.

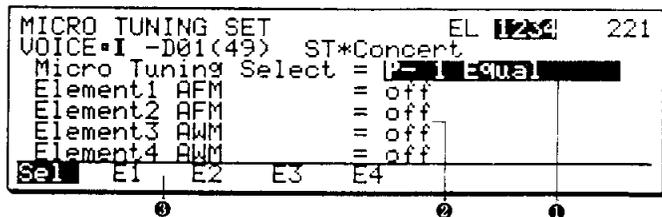
COMMON DATA JUMP #221

11. Micro tuning set

Summary: Select a micro tuning for the entire voice, and specify whether or not each element will use this micro tuning.

Procedure:

- From: Voice Common job directory (JUMP #201)
- Select: job 11:Mcrtune (JUMP #221)
- Specify: the micro tuning, and element micro tuning on/off



Remarks: As with Pan data, Micro Tuning data is not part of the voice. This Micro Tuning Select setting merely specifies which micro tuning will be used.

The sixty-four micro tunings preset inside the SY99 are as follows.

01 Equal temperament: The "compromise" tuning used for most of the last 200 years of Western music, and found on most electronic keyboards. Each half step is exactly 1/12th of an octave, and music can be played in any key with equal ease. However, none of the intervals are perfectly in tune.

02–13 Pure major (C...B): This tuning is designed so that most of the intervals (especially the major third and perfect fifth) in the major scale are pure. This means that other intervals will be correspondingly out of tune. You need to specify the key (C...B) you will be playing in.

14–25 Pure minor (A...G#): The same as Pure Major, but designed for the minor scale.

26–37 Mean tone (C...B): This is an adjustment of the Pure and Pythagorean tunings. The interval between the root and fifth is tuned slightly flat, so that the interval between the root and second degree is exactly halfway between a major and minor pure second; i.e., an average or "mean".

38–49 Pythagorean (C...B): This scale is derived by tuning pure perfect fifths upward from the root. This causes the octave to be flat, so one of the fourths is mistuned to compensate. (In the key of C, the Ab – Eb interval.)

- ① Micro Tuning Select (I-1, I-2, C-1, C-2, P-1... P-64): Select a micro tuning to be used by the voice. 64 micro tuning memories are preset inside the SY99 (see the following remarks). Two of your own micro tunings can be stored in Internal memory, and a card can accommodate another two micro tunings. Internal voices cannot use card micro tunings, nor can card voices use internal micro tunings.
- ② Element 1–4 (off, on): When this is turned on, the element will use the micro tuning selected by Micro Tuning Select. When this is turned off, the element will use equal temperament scale instead of the selected micro tuning.
- ③ Pressing F1 (Sel) will move the cursor to Micro Tuning Select. Pressing F2–F5 will move the cursor to Element 1–4.

50 Werckmeister: Andreas Werckmeister, a contemporary of Bach, designed this tuning so that keyboard instruments could be played in any key. Each key has a unique character.

51 Kirnberger: Johan Philipp Kirnberber was also concerned with tempering the scale to allow performances in any key.

52 Vallotti & Young: Francescantonio Vallotti and Thomas Young (both mid-1700s) devised this adjustment to the Pythagorean tuning in which the first six fifths are lower by the same amount.

53 1/4 shifted equal: This is the normal equal tempered scale shifted up 50 cents.

54 1/4 tone: Twenty-four equally spaced notes per octave. (Play twenty-four notes to move one octave.)

55 1/8 tone: Forty-eight equally spaced notes per octave. (Play forty-eight notes to move one octave.)

56 JustAdjust: This is a special tuning used to make fine adjustments in the pitch of an AWM waveform.

57 Big Chord: The pitch is adjusted down one octave for every 8 half notes, allowing chords to be played anywhere without becoming overly heavy.

58 Log Equal: A variation of conventional equal temperament.

59 1/4 Tonelo: The same as micro tuning 54, but lower in pitch.

60 Harmonic A: The white keys will play the harmonic series beginning on A, and the black keys will play the harmonic series beginning on E.

61 Reverse: The conventional equal tempered scale is inverted.

62 Far East: The black keys and white keys will each play a different eastern scale.

63 Blue: The white keys will play the blues scale. The black keys add a blue flavor

64 Question-1: Play the white keys consecutively upwards starting from C2.

Play the black keys consecutively upwards starting from C#1.

Editing: To edit the currently selected Internal micro tuning, press F8 (Edit). Preset or Card micro tunings cannot be edited. If you want to edit a preset or card micro tuning, you must first copy it to an internal micro tuning memory.

COMMON DATA / MICRO TUNING SET

11.0 Micro tuning edit

JUMP #222

Summary: This function allows you to edit the currently selected Micro Tuning data.

Procedure:

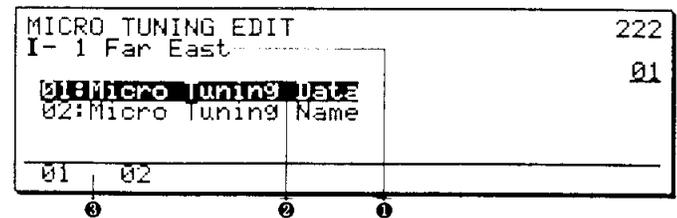
From: Voice Common job 11. Micro Tuning Set
(JUMP #221)

When: an Internal micro tuning is selected

Press: F8 (Edit) (JUMP #222)

Select: the micro tuning data you wish to edit

Editing is possible only when an Internal micro tuning memory is selected. If you want to edit a Preset or Card micro tuning memory, press COPY to copy it to an Internal micro tuning memory as explained in the following section *11.0.1 Copy Micro Tuning Data*.



- ❶ The number and name of the currently selected Micro Tuning data are displayed.
- ❷ Move the cursor in this area to select a job, and press ENTER.
 - 01: Micro Tuning Data: Edit the tuning for each note of the scale. See 11.1 Micro Tuning Data
 - 02: Micro Tuning Name: Assign a ten-character name to the micro tuning data. See 11.2 Micro Tuning Name.
- ❸ Pressing F1 or F2 will select the corresponding job.

11.0.1 Copy micro tuning

Summary: This function copies micro tuning data from another memory into an Internal micro tuning memory.

Procedure:

When: editing micro tuning data
(JUMP #223, #224)

Press: COPY to get the following display.

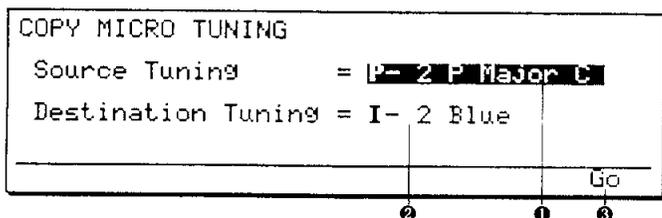
Select: the micro tuning copy source and destination

To copy: the data press F8

To quit: without copying press EXIT

- ❶ Source Tuning (I-1, I-2, C-1, C-2, P-1...P-64): Select the micro tuning data to copy. C-1 and C-2 (card) can be selected only if a VOICE card is
- ❷ Destination Tuning (I-1, I-2): Select the micro tuning (internal 1 or 2) into which to copy the Source Tuning data.
- ❸ After selecting Source Tuning and Destination Tuning, press F8 (Go) to copy the data. You will be asked "Are you sure?". Press YES and the data will be copied.

Remarks: Only Internal micro tuning data can be edited. If you want to edit one of the preset or card micro tunings, use this function to copy it into an Internal micro tuning memory.



11.1 Micro tuning data

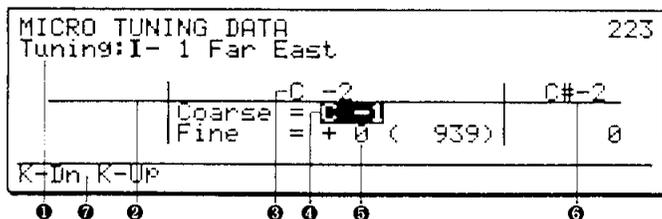
Summary: Edit the tuning for each note of the selected micro tuning data.

Procedure:

From: Micro Tuning Edit job (JUMP #222)
directory

Select: 01:Micro Tuning Data (JUMP #223)

Specify: the tuning for each note



- ❶ The number and name of the micro tuning data being edited are displayed.
- ❷ The previous note and its absolute tuning value.
- ❸ The note whose tuning you are editing.
- ❹ Coarse Tuning (C#-1...G8): With the cursor located at Coarse, adjust the tuning of the currently edited note in half steps.
- ❺ Fine Tuning (-43 or -42...+42): With the cursor located at Fine, adjust the tuning of the currently

edited note in fine steps of 1.171875 cents. The absolute tuning value displayed in parentheses indicates the number of these steps starting from 0 steps at C#-2. The lowest setting of this parameter will be either -43 or -42 depending on the Coarse Tuning value.

- ❻ The next note name and its absolute tuning value.
- ❼ Pressing F1 (K-Dn) or F2 (K-Up) will move to the previous or next note. You can also play a note on the keyboard to select a note at any time.

Remarks: First use F1 (K-Dn) and F2 (K-Up) to select the note you want to edit. You can also use the SY99 keyboard to select the note. The currently edited note will appear in the center of the display, with the previous note shown at left and the next note shown at right. Next set the Coarse and Tune tuning for the selected note. If you adjust Fine Tuning beyond the range of ±42, the Coarse Tuning will be moved up or down as appropriate.

COMMON DATA / MICRO TUNING SET

11.2 Micro tuning name

JUMP #224

Summary: Specify a ten-character name for the internal Micro Tuning memory being edited.

Procedure:

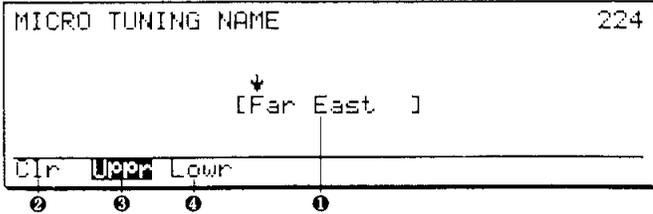
From: Micro Tuning Edit job directory (JUMP #222)

Select: 02:Micro Tuning Name (JUMP #224)

Specify: a name for the micro tuning data

- ❶ Enter a ten-character name for the Micro Tuning data.
- ❷ To clear the currently entered name press F1 (Clr).
- ❸ To switch to upper-case characters press F2 (Uppr).
- ❹ To switch to lower-case characters press F3 (Lowr).

Remarks: Methods of entering character data are explained in *How to enter character data*, on page 30.



COMMON DATA

12. (F1) Controller set (Pitch bend)

JUMP #225

Summary: Specify the range over which the PITCH wheel will affect the pitch.

Procedure:

From: Common Data job directory (JUMP #201)

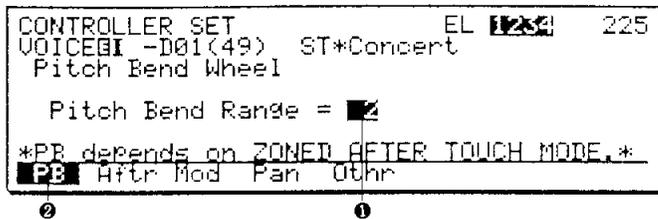
Select: job 12:Ctrlr

Press: F1 (PB)

Specify: the range of the pitch bend wheel

- ❶ Pitch Bend Range (0...12): This determines the range (0...12 half steps) over which the PITCH wheel will affect the pitch. When this is set to 12, the PITCH wheel will move the pitch one octave up or down. When this is set to 0, the PITCH wheel will have no effect.
- ❷ To make controller settings for Aftertouch, Modulation, Pan, or Other, press F2, F3, F4, or F5. Refer to sections 12.(F2), 12.(F3), 12.(F4), or 12.(F2).

Note: The action of the PITCH wheel may be modified using the Zoned Aftertouch settings, described in the following section. This allows for more sophisticated usage of the PITCH wheel.



COMMON DATA

12. (F2) Controller set (Aftertouch)

JUMP #276

Summary: Specify how aftertouch will affect the pitch and any other parameters to which it is assigned.

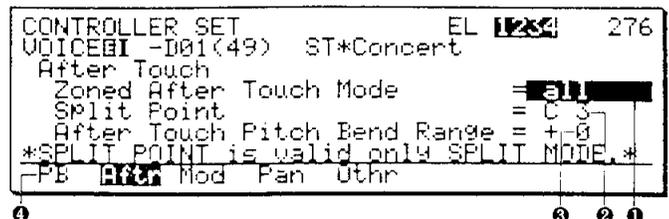
Procedure:

From: Common Data job directory (JUMP #201)

Select: job 12:Ctrlr

Press: F2 (Aft)

Specify: the aftertouch mode, split point, and bend range



VOICE EDIT MODE

① Zoned After Touch Mode (all, top, bottom, split hi, split lo): This setting determines which notes of the voice will be affected by aftertouch. Note that this includes aftertouch pitch bend set in ③ and also any Modulation, Pan, or Other parameter controlled by aftertouch. For example, if you have assigned pitch modulation to be affected by aftertouch, pressure on the keyboard will increase the pitch modulation only for the note(s) determined by this Zoned After Touch Mode setting. Refer to the following sections 12.(F3)-12.(F5).

All: Aftertouch will affect all notes of the voice.

Top: Aftertouch will affect only the highest of the currently played notes.

Bottom: Aftertouch will affect only the lowest of the currently played notes.

Split hi: Aftertouch will affect only notes at or above the split point specified in ②.

Split lo: Aftertouch will affect only notes below the split point specified in ②.

② Split Point (C-2...G8): This determine the split point for Zoned After Touch Mode settings of "split hi" or "split lo". If the Zoned After Touch Mode is set to "all", "top", or "bottom, this Split Point setting will have no effect.

③ After Touch Pitch Bend Range (-12...+12): This determines how aftertouch will affect the pitch. Pressing strongly down on the keyboard after playing a note will move the pitch down one octave (for a setting of -12) or up one octave (for a setting of +12).

④ To make controller settings for Pitch Bend, Modulation, Pan, or Other, press F1, F3, F4, or F5. Refer to sections 12.(F1), 12.(F3), 12.(F4), or 12.(F5).

Keyboard: When the cursor is located at Split Point, you can press F8 (Kbd) and then press a note to set the split point.

COMMON DATA

12. (F3) Controller set (Modulation)

JUMP #226

Summary: Specify the controller device that will add vibrato (pitch modulation), tremolo (amplitude modulation), and wah-wah (filter modulation).

Procedure:

From: Common Data job directory (JUMP #201)

Select: job 12:Ctrlr

Press: F3 (Mod) (JUMP #226)

Specify: the controller and depth for each parameter

CONTROLLER SET		EL	1226	226
VOICE#1 -D01(49)		ST*Concert		
Modulation Depth				
Pitch	32	001	Modulation	
Amplitude	0		off	
Filter	0		off	
PB	After	Mod	Pan	Othr

① Pitch Modulation Depth (0...127): This setting determines the range over which the specified device will add vibrato (pitch modulation). For a setting of 127, the selected controller will be able to add the maximum amount of vibrato. For a setting of 0, the selected controller will not be able to add vibrato.

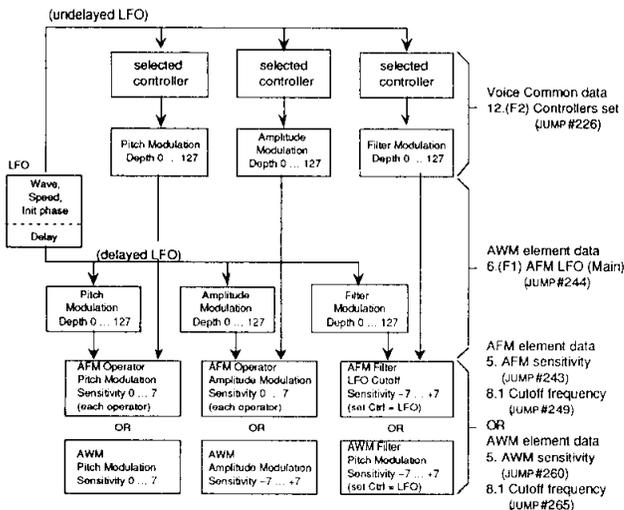
② Amplitude Modulation Depth (0...127): This setting determines the range over which the specified device will add tremolo (amplitude modulation). Details are the same as in ①.

③ Filter Modulation Depth (0...127): This setting determines the range over which the specified device will add wah-wah (filter modulation). Details are the same as in ①.

④ MIDI Ctrl No. & Device (1...120, After Touch): These settings determine the controller devices that will add Pitch modulation, Amplitude modulation, and Filter modulation. The selected MIDI control number (1...120) is displayed at left, and the function which is defined for that number is displayed at right. Control numbers 0 and 32 are reserved for the Bank Select message, and cannot be selected here.

⑤ To make controller settings for Pitch Bend, Aftertouch, Pan, or Other, press F1, F3, F4, or F5. Refer to sections 12.(F1), 12.(F3), 12.(F4) or 12.(F5).

Remarks: As shown by the following diagram, the controllers selected here will be able to *add* pitch, amplitude, and filter modulation *in addition to* the pitch, amplitude, and filter modulation specified by the PModDepth, AModDepth, and FModDepth settings of each element LFO. The effect of the resulting modulation will depend on the *sensitivity* settings of each element.



Fixed controllers: The control numbers transmitted by the following five controllers built into the SY99 are fixed, and cannot be changed. When these controllers are moved, they will transmit MIDI control messages of the corresponding number. When the following control numbers are selected, these built-in controllers will regulate the assigned function. The official MIDI standard defines Aftertouch not as a control number but as a different type of message, so it is not given a control number.

- 001: The MODULATION 1 wheel located at the left of the keyboard
- 002: An optional breath controller connected to the rear panel BREATH jack
- 004: An optional foot controller connected to the rear panel FOOT CONTROLLER jack
- 006: Data entry slider
- 007: Foot volume
- 064: An optional foot switch connected to the rear panel SUSTAIN jack
- Aftertouch: (i.e., pressing down on the keyboard after playing a note)

Assignable controllers: In addition to these five controllers which have fixed functions, the SY99 has the following two assignable controllers. With the factory settings, the MODULATION 2 wheel is assigned MIDI control number 13, and the FOOT SWITCH is assigned control number 65. The control number transmitted by these controllers can be changed by the setting in *Utility mode System Utility 3. Controllers (JUMP #803)*.

- The MODULATION 2 wheel located at the left of the keyboard (initially set to transmit control number 013:Non-assigned)
- An optional foot switch connected to the rear panel FOOT SWITCH jack (initially set to transmit control number 065:Portamento Switch)

MIDI controllers: The specified MIDI Ctrl No.& Device applies both to the built-in controllers of the SY99 and to incoming MIDI control data received at MIDI IN. For example if you have specified that Amplitude Modulation Depth be regulated by 001:Modulation Wheel, it will be regulated by incoming MIDI Control Change 001 messages in addition to movements of the SY99's own MODULATION 1 wheel.

The official MIDI standard does not define the purpose of all of the MIDI Control Change messages 1–120. If the selected control number has not been defined, the LCD will show “Non-assigned no.”. You can use these control numbers just like any other control number. For example you might assign an assignable controller (see *Utility mode System Utility 3. Controllers*, page 254) to one of these numbers, and then select that control number to regulate Pitch Modulation depth. “Non-assigned no.” simply means that there is no official agreement as to the use of that control number.

COMMON DATA

12. (F4) Controller set (Pan)

JUMP #227

Summary: Specify the controller device that will regulate the depth of the cyclical panning movement (Pan LFO), and the controller that will directly adjust the pan position (Pan Bias).

Procedure:

- From: Voice Common job directory (JUMP #201)
- Select: job 12:Ctrlr
- Press: F4 (Pan) (JUMP #227)
- Specify: the controller and depth for each parameter

CONTROLLER SET		EL 1234	227
VOICEBI -D01(49) ST*Concert			
Pan Control			
Pan LFO	Depth	MIDI Ctrl No. & Device	
Pan Bias	0	off	
	0	off	
PB	Aftr Mod	Pan	Othr

- ❶ Pan LFO Depth (0...127): This determines the range over which the specified controller will regulate the depth of the Pan LFO. When this is set to 127, the selected controller will regulate LFO panning over the full range from no LFO panning to maximum LFO panning. When this is set to 0, the selected controller will have no effect on LFO panning.
- ❷ Pan Bias Depth (0...127): This determines the range over which the specified controller will affect pan position.
- ❸ MIDI Ctrl No. & Device (0...120, After Touch): These settings determine which controllers will regulate the depth of LFO panning and Pan Bias.

- ❹ To make controller settings for Pitch Bend, Aftertouch, Modulation, or Other, press F1, F2, F3, or F5. Refer to sections 12. (F1), 12. (F2), 12. (F3), or 12. (F5).

MIDI Ctrl No. & Device: For details refer to 12. (F3) Controller set (Modulation).

Remarks: Since the MODULATION 2 wheel is detented at the center position, it is especially useful for controlling pan. Since the MODULATION 2 wheel is assignable, to use it you must select the MIDI Ctrl No. to which it has been assigned. Check the *Utility mode System Utility 3. Controllers* (JUMP #803) settings to see the MIDI Control Number to which the assignable wheel has been set. For example if the assignable wheel has been set to its initial setting of MIDI Control Number 13, you would select "013:Non-assigned no." for the Pan Bias control number in order to use WHEEL 2 to control panning.

Note: When a voice is used in Multi Play mode, these Pan Control settings will be effective only if the Static Pan is set to Voice. Refer to Multi Edit 5. Voice static pan (JUMP #408, #409).

COMMON DATA

12. (F5) Controller set (Other)

JUMP #228

Summary: Specify controller devices that will regulate the volume, the EG bias and the Cutoff Frequency of the filters in each element of the voice.

Procedure:

- From: Voice Common job directory (JUMP #201)
- Select: job 12:Ctrlr
- Press: F5 (Othr) (JUMP #228)
- Specify: the controller and depth for each parameter

CONTROLLER SET		EL 1234	228
VOICEBI -D01(49) ST*Concert			
Volume, EG bias, & Filter Cutoff Freq.			
VolLowLimit	Value	MIDI Ctrl No. & Device	
EGbiasDepth	0	off	
CutoffDepth	127	013 Non-assigned no.	
PB	Aftr Mod	Pan	Othr

- ❶ VolLowLimit (Volume Low Limit 0...127): This determines the lowest volume that can be set by the specified controller. For example when this is set to 80, the controller will reduce the volume

no lower than 80. When this is set to 0, the controller can reduce the volume to silence. When this is set to 127, the controller will have no effect on the volume. In addition to the controller specified here, the volume can always be controlled over its full range by an optional foot controller connected to the rear panel VOLUME jack.

- ② EGbiasDepth (Eg Bias Depth 0...127): This determines the range over which the specified controller will control the EG bias. The result of controlling EG Bias for an AFM element will depend on the AModSens setting for each operator. If AModSens is set for carrier operators, the controller assigned to EG Bias will affect the volume of an AFM element. If AModSens is set for modulator operators, the controller assigned to EG Bias will affect the tone of an AFM element. For details refer to *Voice AFM Element job 5. Sensitivity* (page 145). For an AWM element, this setting will affect the Volume.
- ③ CutoffDepth (Filter Cutoff Depth 0...127): This determines the range over which the specified controller will increase the cutoff frequency specified for the filters of each element. Higher settings will allow the controller to brighten the tone. If the filter cutoff frequencies are already at maximum, this will have no effect.
- ④ MIDI Ctrl No. & Device (0...120, Aftertouch): These settings determine which controllers will regulate each parameter.

- ⑤ To make controller settings for Pitch Bend, Modulation, or Pan, press F1, F2, or F3. Refer to sections 12. (F1), 12. (F2), or 12. (F3).

MIDI Ctrl No. & Device: For details refer to 12. (F2) *Controllers set (Modulation)*.

Filter Cutoff Depth: The controller assigned to CutoffDepth can be used in two ways to affect the filter, depending on the Control Source setting of each filter. Refer to *Voice AFM Element 8.3 Cutoff EG* (page 153) or *Voice AWM Element 8.3 Cutoff EG* (page 169).

- Continuously control the filter cutoff: If the Control Source of a filter is set to LFO then the controller assigned to CutoffDepth can be used to continuously control the cutoff frequency even while a note is sounding.
- Control the filter cutoff at key-on: If the Control Source of a filter is set to EG or EG-VA then the controller assigned to CutoffDepth will be used only at the instant the note is played; i.e., after playing a note you can move the controller without affecting the sound. This can be used to give different filter cutoffs to individual notes as you play them.

COMMON DATA

13. Voice name

JUMP #229

Summary: Specify a ten-character name for the voice being edited. In voice play mode, this voice name will be displayed in large characters.

Procedure:

- From: Voice Common job (JUMP #201)
directory
- Select: job 13:Name (JUMP #229)
- Specify: the voice name

- ① Enter a ten-character name for the voice.
- ② To clear the currently entered name press F1 (Clr).
- ③ To switch to upper-case characters press F2 (Uppr).
- ④ To switch to lower-case characters press F3 (Lowr).

Remarks: Methods of entering character data are explained in *How to enter character data*, on page 30.

```

VOICE NAME                               229
VOICEBI -I01(49)
↓
[ST*Concert]
Clr  Uppr  Lowr
②   ③   ④   ①

```

COMMON DATA

15. Initialize voice

Summary: Initialize the Voice Common data being edited to a set of standard values. The Voice Mode will not change.

Procedure:

From: Voice Common job (JUMP #201)
directory

Select: job 15:Init

To execute: initialization press YES

To quit: without initializing press NO

INITIALIZE VOICE

ARE YOU SURE ?

(Yes or No)

This function sets all voice common data values to the minimum or simplest possible settings. When creating your own new voices, it is usually best to begin by editing an existing voice. However if you want to start from scratch, this Initialize function can often be helpful.

If you are sure you want to initialize the voice data, press YES and the voice common data of the voice being edited will be set to the values shown below. If you decide not to initialize, press NO.

This function initializes only Voice Common data. Other initialize functions are provided for initializing AFM Element or AWM Element data. Refer to *Voice AFM Element 15. Initialize* (page 154) or *Voice AWM Element 15. Initialize* (page 170).

Initialized settings for Voice Common Data

01 Element Levels

Voice Volume = 127

Element level = 127 (all elements)

02 Element Detune

Element detune = ±0 (all elements)

03 Element Note Shift

Shift = ±0 (all elements)

04 Element Note Limit

Low Limit = C-2 (all elements)

High Limit = G8 (all elements)

05 Element Velocity Limit

Low Limit = 1

High Limit = 127

06 Element Dynamic Pan

Element Preset 1-01 "Center" (all elements)

(Pan Source = velocity, Source Depth = 0,

Pan EG; HT=0, R1-RR2=63, L0-RL2=0,

SLP=S1)

07 Output Select

Output Group = Both (all elements)

08 Random Pitch

Random Pitch Depth = 0

09 Portamento

Mode = Follow (poly)/ Fingered
(mono)

Speed = 0

10 Effect Set

Effect Mode = off

Effect Type = 01:Rev.Hall (both effect units)

Output Level = 100% (both effect units)

Effect 2 Mix Level = 100%

wet:dry = 50:50 (both effect units)

Control Parameter = EFF1 par1

(both effect units)

Control Device = off (both effect units)

Control MIN = 0% (both effect units)

Control MAX = 99% (both effect units)

Effect LFO Wave = triangle

Effect LFO Speed = 65

Effect LFO Delay Time = 0

Effect LFO Initial Phase = free

Effect send select = all on (all elements)

Effect send level = 127

(both effect units)

Effect send vel sense = 0 (both effect units)

Effect send level scale = 0 (both effect units)

- 11 Micro Tuning Set
 - Preset-01 Equal Temperament
 - Element = off (all elements)
- 12 Controller Set
 - Pitch Bend Wheel Depth = 2
 - Zoned Aftertouch Mode = all
 - Zoned Aftertouch Split Point = C3
 - Aftertouch Pitch Bend Depth = 0
 - Pitch Modulation Depth = 64
 - Pitch Modulation Device = 1
 - Amplitude Modulation Depth = 64
 - Amplitude Modulation Device = 12
 - Filter Modulation Depth = 0
 - Filter Modulation Device = 1

- Pan LFO Depth = 64
- Pan LFO Device = 13
- Pan Bias Depth = 0
- Pan Bias Device = 10
- Volume Low Limit = 0
- Volume Low Device = 14
- EG Bias Depth = 0
- EG Bias Device = 2
- Filter Cutoff Frequency Depth = 0
- Filter Cutoff Frequency Device = 12
- 13 Name Voice
 - Name = INIT VOICE

COMMON DATA	<h2>16. Recall voice</h2>
-------------	---------------------------

Summary: Recall the previously edited voice data.

Procedure:

- From: Voice Common job (JUMP #201) directory
- Select: job 16:Recall
- To recall: the data press YES
- To quit: without recalling press NO

If after editing a voice you exit voice edit mode without storing, the edited voice data will be lost. In such cases, you can use this function to recall the previously edited voice data into the editing buffer.

If you are sure you want to recall, press YES and the previously edited voice data will be recalled into the editing buffer. If you decide not to recall, press NO.

This function recalls all voice data; element data as well as common data. The same function is also available when editing AFM Element or AWM Element data.

RECALL VOICE
<h1>ARE YOU SURE ?</h1>
(Yes or No)

AFM element data

AFM ELEMENT DATA

AFM element job directory

JUMP #230

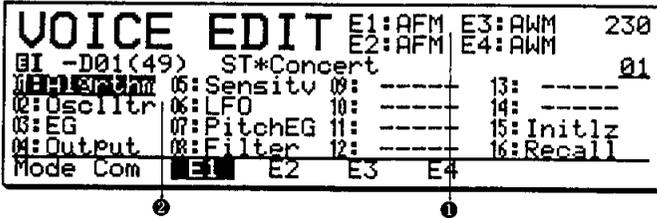
Summary: This job directory shows the editing jobs for an AFM element.

Procedure:

From: voice edit mode (JUMP #200 or #201)

When: editing a normal voice that contains AFM elements

Select: an AFM element F3-F6 (JUMP #230) (E1-E4).



- ❶ This area shows the number (1-4) and type (AFM or AWM) of elements in the selected voice mode.
- ❷ Move the cursor in this area to select a job and press ENTER to go to the selected job.

01:Algrthm (AFM algorithm):

F1; The algorithm determines how the six operators are connected. Three feedback sources can be selected and sent to other operators.

F2; Each operator can be modulated from an external source such as an AWM waveform or the noise generator.

F3; Each operator has two inputs In1 and In2 with input level settings for each input.

02:Osciltr (AFM oscillator): The frequency produced by each operator can either be fixed or made to change according to the note played.

03:EG (AFM operator EG):

F1; Make operator EG settings for an individual operator while viewing a graphic display.

F2; Make operator EG settings for all operators.

04:Output (AFM operator output):

F1; The output level of each operator can be made to vary across the keyboard.

F2; The output level of each operator can be set.

05:Sensitiv (AFM sensitivity): The output level and frequency of each operator can be affected by key-on velocity or the LFO, and the EG rates of each operator can also be affected by key-on velocity.

06:LFO (AFM LFO):

F1; The Main LFO is used to create tremolo (amplitude modulation), vibrato (pitch modulation), or wah-wah (filter modulation).

F2; The Sub LFO is used to create vibrato (pitch modulation).

07:PitchEG (AFM pitch EG): The pitch EG creates a fixed shape of pitch change over time for each note, and can be switched on/off for each operator.

08:Filter (AFM filter): The two filters of each element can be used to control the tone in various ways. The filter EG creates a fixed pattern of tonal change over time, and a periodic signal from the LFO can be applied to the filter to create wah-wah.

15:Initlz (Initialize AFM element): The AFM element data being edited can be set to the minimum or simplest possible setting as a convenience when creating an element from scratch.

16:Recall (Recall voice): All data of the previously edited voice.

AFM ELEMENT DATA

Operator on/off

Summary: Any time while editing an AFM element, you can turn the output of each operator off/on. This is useful when you want to hear how each operator affects the others, or when you want to hear only certain operators.

Procedure:

From: any job in the AFM job directory
 Press: the OPERATOR ON/OFF buttons (memory select buttons 9–14) to turn operators 1–6 on/off.

The on/off condition of each operator is shown in the upper right of the LCD when editing an AFM ele-

ment, and also indicated by the OPERATOR ON/OFF LEDs. Operators that are on are displayed in inverse. In the following LCD all operators are on.

Remarks: If you turn off all the carrier operators there will be no sound.

This function is provided as a help in editing. Operator on/off settings are not stored as part of voice data.

When you select a different AFM or AWM element or exit element editing, all operators you turned off will be turned back on.

AFM ELEMENT DATA

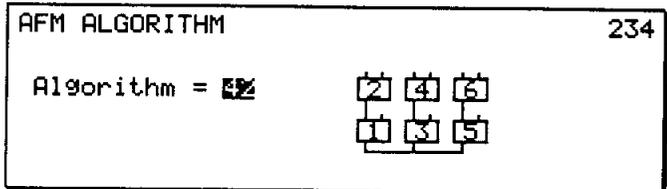
AFM algorithm

JUMP #234

Summary: You can view a graphic display of the current algorithm at any time while editing an AFM element and select a different algorithm if you wish. Since the algorithm determines how each operator functions, you should always be aware of the algorithm when editing AFM operator data.

Procedure:

From: any job in the AFM (JUMP #231–#255) job directory
 Press: F8 (Alg) (JUMP #234)
 To exit: to the previous editing job press EXIT



To return to the previous display, press EXIT.

Note: When you select an algorithm, all settings which modify the routings in the algorithm (feedback, input, etc.) will be cleared to their initial settings.

AFM ELEMENT DATA

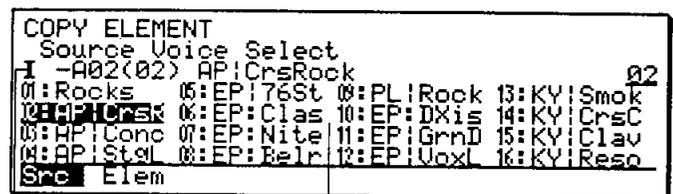
Copy element

Summary: While editing AFM element parameters other than EG, Output, Filter, or Effect, you can copy data from an element of another voice to the element you are now editing.

Procedure:

From: AFM element job 1, 2, 6, or 7
 Press: COPY
 Press: F1 (Src) and select the source voice
 Press: F2 (Elem) and select the source element

To execute: the copy operation press F8 (Go)



- ① Source Voice Select: Specify the memory (internal, card, preset 1 or preset 2), the bank A–D, and the number 1–16 of the voice from which you want to copy element data.
- ② Source Voice: The number and name of the selected source voice is displayed.

VOICE EDIT MODE

Source Element Type: The type (AWM or AFM) of each element in the selected voice is displayed.

Destination Element Number: Specify the element 1–4 of the selected Source Voice which you want to copy into the currently edited element. The type (AWM or AFM) of the selected element will be displayed in the Element Type line above. The selected source element must be the

same type as the currently edited element. If not, the bottom line will show “Element type mismatch!”

After specifying the source voice and element, press F8 (Go). The display will ask “Are you sure?”. If you are sure you want to copy the element data then press YES, and the data will be copied.

AFM ELEMENT DATA

Copy operator

Summary: While editing the parameters for operator EG or Output, you can copy EG and Output data from one operator to another.

Procedure:

From: AFM element job 3 or 4
(JUMP #236–242)

Press: COPY

Select: the data type, source operator, and destination operator

To execute: the copy operation press F8 (Go)

```

COPY OPERATOR
Data Type = EG&OUTPUT
Source    = OP 4
Destination = OP 1
    
```

Go

① ② ③

- ① **Data Type:** Select one of the following types of data to be copied.
EG&OUTPUT: Envelope generator, output level and scaling
EG: Envelope generator
OUTPUT: Output level and scaling
- ② **Source:** The specified data will be copied from this operator.
- ③ **Destination:** The specified data will be copied to this operator.

Remarks: It is often the case that many or all operators in a voice have similar settings, especially for EG parameters. In such cases you can save time by setting the average EG for the voice on one operator, and then copying it to the others.

AFM ELEMENT DATA

1. (F1) AFM algorithm (Form)

JUMP #231

Summary: Select the Algorithm and specify feedback routings between operators.

Procedure:

From: AFM Element job directory (JUMP #230)

Select: job 01:Algrth., and press (JUMP #231)
F1 (Form)

Specify: the algorithm number, and feedback sources and destinations

```

AFM ALGORITHM          OP 123456 231
VOICEB1 -D01(49) ST*Concert(E1/AFM)
Algorithm Number = 42
FB  Src  Dst  OP1  OP2  OP3  OP4  OP5  OP6
FB1  OP5  |  in2  use  use  off  use  in1
FB2  OP4  |  use  use  use  in1  in2  off
FB3  OP2  |  use  in1  in2  off  use  off
Form  Extn  Inpt  |  |  |  |  |  |  |
    
```

② ③ ④ ①

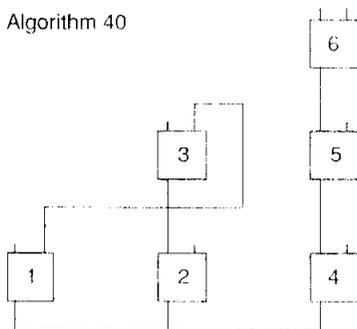
- ① **Algorithm Number (1...45):** Select the algorithm to determine the “arrangement” of the six operators in an AFM element. Refer to the chart of the 45 algorithms on page 302. When you change the Algorithm, the Src ③ and Dst ④ settings explained below and the external input settings explained in the following section will be initialized.
- ② **FB1–FB3:** Feedback can be drawn from three of the operators in the algorithm and applied to any operator that has an unused input.
- ③ **Src 1–3 (OP1...OP6):** Select the source of feedback for the three feedback routings. Any operator can be selected as the source of feedback. In some algorithms, one or more feedback

sources may be fixed by the choice of algorithm, and cannot be changed. In such cases, an “F” will be displayed after the Source (e.g., OP3F) and the Destination operator explained in 4 will be displayed in uppercase characters (e.g., IN1).

- 4 Dst OP1...OP6 (in1, in2): Select the destination of each feedback source. Each feedback source can be sent to as many destinations as you like. Each operator has two inputs, and an operator can be selected as a feedback destination only if at least one of its inputs is free. It makes no difference whether in1 or in2 is used, but remember that the input levels of each operator are set independently for in1 and in2. Refer to the following section 1.3 Algorithm Input.

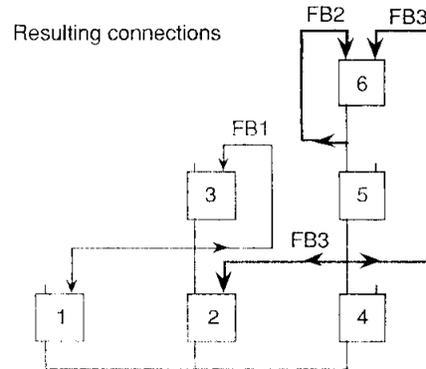
If both inputs of an operator are already used by the algorithm connection, “-” will be displayed. If both inputs are already used because of a feedback assignment, “use” will be displayed. If the algorithm has a fixed feedback loop, the feedback destination operator will be displayed in uppercase characters (e.g., IN1). The cursor cannot be moved to the Dst setting for such operators.

Having three selectable feedback sources which can be sent to any or all other operators allows you to connect the operators in very complex ways. The following diagrams show how the operators would be connected for algorithm 40 when feedback sources and destinations are set as shown in the following table. Thin lines indicate the connections defined by the algorithm, and heavy lines indicate the feedback connections. Whether a connection between two operators is the result of the algorithm or the result of feedback routing has no influence on the sound.



Feedback settings

FB	SRC	DST	OP1	OP2	OP3	OP4	OP5	OP6
FB1	OP3F		off	use	IN2	off	off	use
FB2	OP6		off	use	off	off	off	in1
FB3	OP5		off	in2	off	off	off	in2



Remarks: In previous Yamaha 6-operator FM synthesizers, you had a choice of 32 algorithms each of which included one feedback loop. The SY99 provides broader possibilities with 45 algorithms, and three feedback loops that can be sent to more than one operator. In addition, operators can be modulated by external waveforms.

Feedback from a carrier: The output level of carrier operators is automatically adjusted according to the number of carriers in the algorithm, in order to keep the resulting output volume consistent. If the nominal output of a one-carrier algorithm is considered to be 0 dB, each carrier in a two-carrier algorithm will be reduced -3.010 dB, in a three-carrier algorithm -8.278 dB, in a four-carrier algorithm -9.783 dB, in a five-carrier algorithm -12.041 dB, and in a six-carrier algorithm -14.299 dB.

If you use a carrier operator as a feedback source, be aware that even if the output level of the carrier operator is the same, the actual output (and therefore the amount of feedback produced by the carrier operator) will depend on the number of carriers in the algorithm.

Free Algorithm: By sending exclusive messages from a computer with the appropriate software to the SY99, it is possible to select a "free algorithm" in which all algorithm routings can be specified arbitrarily.

If the free algorithm has been selected, algorithm-related parameters can be edited only by system exclusive parameter change messages from MIDI IN, and cannot be modified from the SY99 front panel.

It is not possible to select or edit the free algorithm from the SY99 itself. This display will never appear unless you are programming the SY99 from a personal computer with editing software that takes advantage of this capability.

AFM ELEMENT DATA

1. (F2) AFM algorithm (External input)

JUMP #232

Summary: Modulate an operator from an external source such as an AWM waveform or the noise generator.

Procedure:

From: AFM Element job directory (JUMP #230)

Select: job 01:Algrthm., and press (JUMP #232)
F2 (Extn)

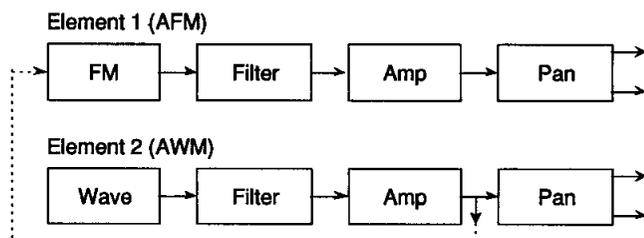
Specify: noise and/or AWM input for each operator

AFM ALGORITHM		OP 123456		232	
VOICEBI	-D01(49)	ST*Concert(E1/HFM)			
Input	OP1	OP2	OP3	OP4	OP5
Noise	use	use	use	off	use
AWM	use	in2	use	off	use
Form	Ext	InPt	HI9		

- ❶ Noise (off, in1, in2): The SY99 contains a noise generator which produces a type of white noise. This can be sent to any free operator input to modulate the operator.
- ❷ AWM (off, in1, in2): If the voice contains both AFM and AWM elements (ie., if the voice mode is either 9:1AFM&1AWM or 10:2AFM&2AWM), an AWM waveform can be received at any free operator input to modulate the operator.

For voice mode 9:1AFM&1AWM, the waveform of AWM element 2 will be used to modulate AFM element 1. For voice mode 10:2AFM&2AWM, the waveform of AWM element 3 will be used to modulate AFM element 1, and the waveform of AWM element 4 will be used to modulate AFM element 2.

❶ and ❷: If both inputs of an operator are already used by the algorithm connection, "-" will be displayed. If both inputs are already used because of a feedback assignment, "use" will be displayed. The cursor cannot be moved to the Dst setting for such operators.



Remarks: Noise modulation makes it possible to create sounds that were difficult for previous FM synthesizers.

By using an AWM waveform to modulate one or more AFM operators, new harmonics can be added to the AWM sample. A simple example of this is given in *Using RCM Hybrid Synthesis* in the appendix.

AFM ELEMENT DATA

1. (F3) AFM algorithm (Input level)

JUMP #233

Summary: Set input levels In1 and In2 for each operator.

Procedure:

- From: AFM Element job directory (JUMP #230)
- Select: job 01:Algrthm. and press (JUMP #233)
- F3 (Inpt)
- Specify: the input level for each operator input

AFM ALGORITHM		OP <u>OP4</u> 233					
VOICEBI -D01(49) ST*Concert(E1/HFM)		OP1	OP2	OP3	OP4	OP5	OP6
In1 Src		OP2 FB#2		OP4 FB#4		OP6 FB#6	
Level		0	7	0	7	0	7
In2 Src		FB#6	AWM FB#2	off	FB#4	off	
Level		7	3	7		7	
Form Extn		Inpt					H19

- ① In1 Src, In2 Src: This displays the input sources for input In1 and In2 of each operator, as determined by Algorithm and Feedback settings (F1) and External input settings (F2). The input sources cannot be changed from this job.

- ② In1 Level, In2 Level (0...7): Adjust the input level of In1 and In2 for each operator. If an operator input is not used, the Src will display "off" and the "Level will display "-". The cursor cannot be moved to the Level setting for such operators.

Remarks: Previous FM synthesizers allowed you to set only the output level of each operator. However on the SY99, the input levels In1 and In2 of each operator can also be set. If the input source is the feedback from another operator, the input level setting functions as the feedback level.

Correct adjustment of input levels is especially important when bringing in AWM to AFM as a modulator.

AFM ELEMENT DATA

2. AFM oscillator

JUMP #235

Summary: Set frequency-related parameters for each operator.

Procedure:

- From: AFM Element job directory (JUMP #230)
- Select: job 02:Osclltr (JUMP #235)

AFM OSCILLATOR		OP <u>OP4</u> 235	
VOICEBI -D01(49) ST*Concert(E1/HFM)			
Freq Mode =	<u>2356</u>	Waveform =	1
Coarse =	1.00	Phase Sync =	on
Fine =		Init. Phase =	60
Detune =	+ 0		

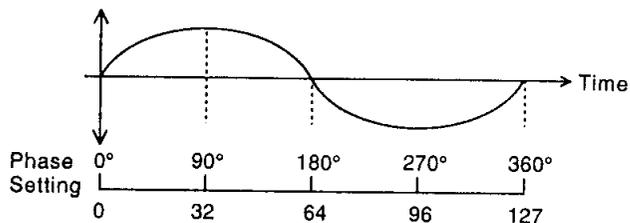
- ① Operator (1-6): This displays the operator being edited. To move to another operator, use the OPERATOR SELECT buttons OP1...OP6.
- ② Freq Mode (fixed, ratio): When "fixed" is selected the operator will produce the same pitch regardless of what note is played. When "ratio" is selected the operator pitch will depend on the note that is played
- ③ Coarse/Fine (0 Hz...9762 Hz in Fixed Frequency mode, 0.5...61.69 in Ratio Frequency

mode): This setting specifies the pitch produced by the operator. By moving the cursor to coarse or fine you can adjust the pitch in large steps or in small steps. When the Freq Mode is set to "fixed" the range is 0 Hz...9762 Hz. When the Freq Mode is set to "ratio" the range is 0.5...61.69. (In "ratio" mode with a Coarse/Fine setting of 1.0 the A3 key will produce the standard pitch of 440 Hz.)

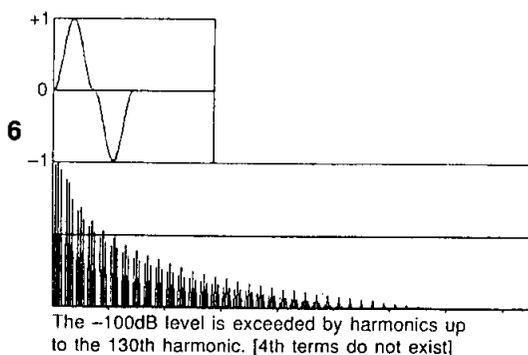
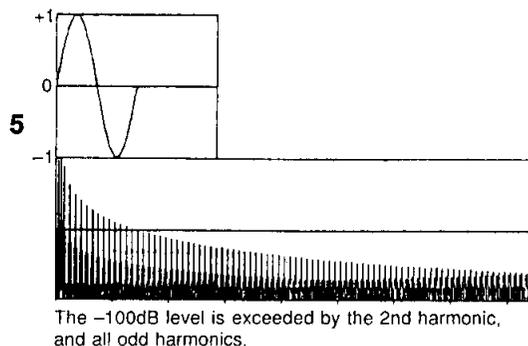
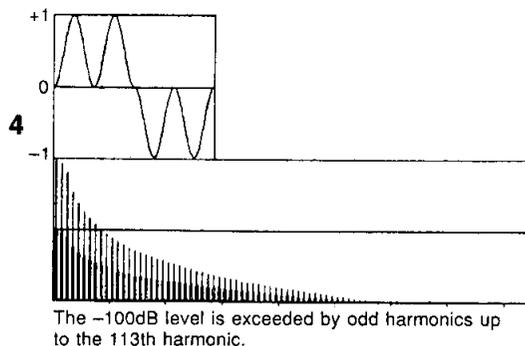
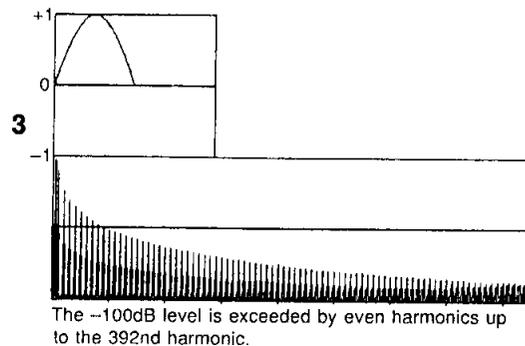
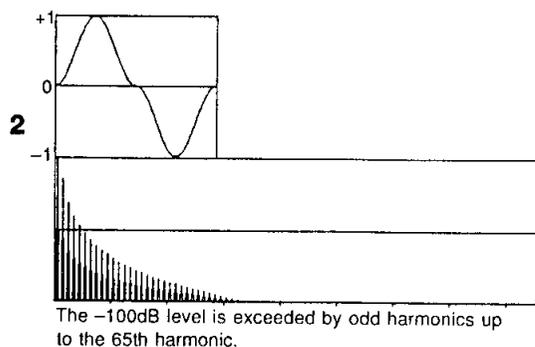
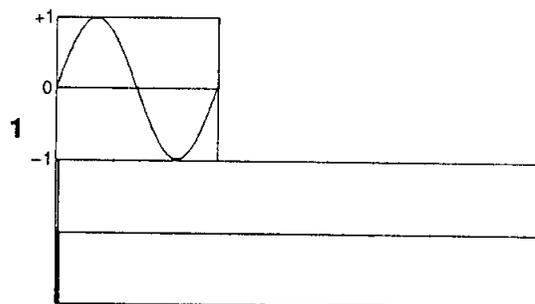
- ④ Detune (-15...+15): The pitch of each operator can be adjusted in fine steps of 1.171875 cents.
- ⑤ Waveform (1...16): Each operator can produce sixteen different waveforms; a sine wave with no harmonics, and fifteen other more complex waveforms containing additional harmonics. This allows you to create complex waveforms using fewer operators. A graphic display of the selected waveform is shown below the waveform number. The table below shows the harmonic content of each waveform.
- ⑥ Phase Sync (on, off): When phase sync is on, the selected waveform will be re-started each time a key is pressed.

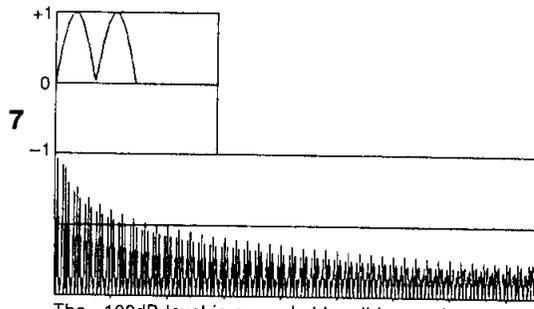
VOICE EDIT MODE

- 7 Init Phase (0...127): This determines the initial phase position from which the selected waveform will be re-started when phase sync is on. The init phase range of 0...127 corresponds to a range of 0...360 degree starting phase. This setting is effective only when Phase Sync is on.

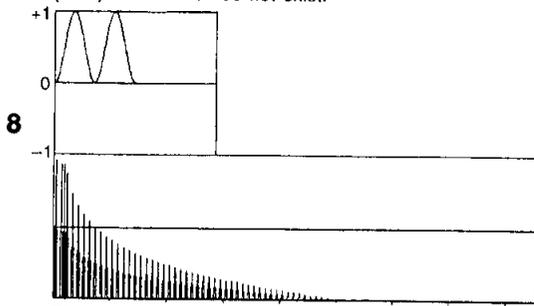


Waveform: The sixteen waveforms that can be produced by each operator are not modeled after any "real" instrument but are mathematical transformations of sinewaves. The following chart shows the harmonic content of each waveform. The amplitude of each harmonic partial is given as a percentage of the fundamental.

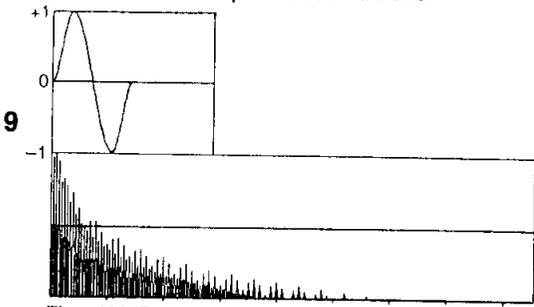




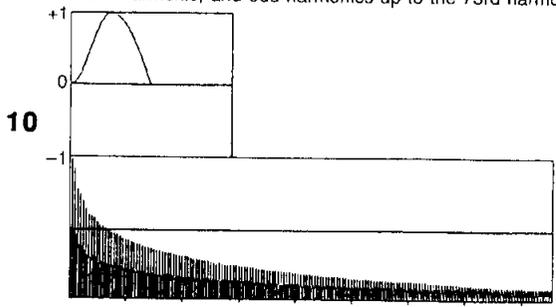
The -100dB level is exceeded by all harmonics except $(4n-2)$ terms which do not exist.



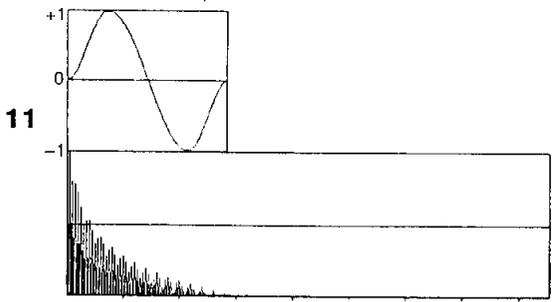
The -100dB level is exceeded by the 4th harmonic, and odd harmonics up to the 99th harmonic.



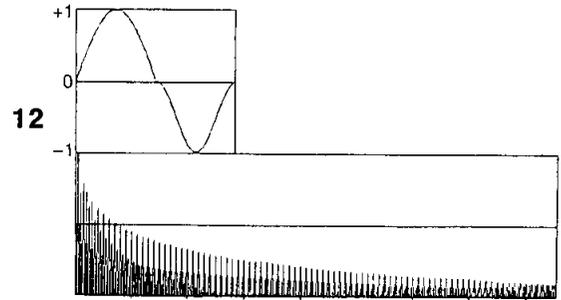
The -100dB level is exceeded by even harmonics up to the 112th harmonic, and odd harmonics up to the 73rd harmonic.



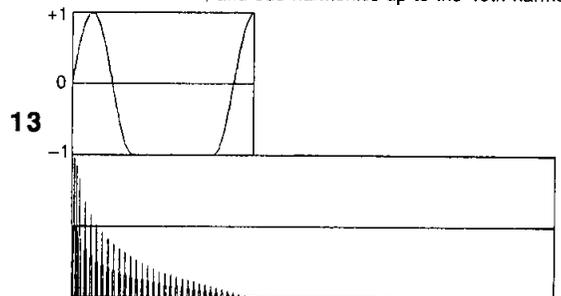
The -100dB level is exceeded by all harmonics up to the 270th harmonic.



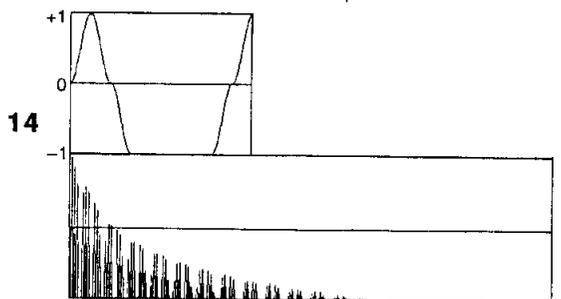
The -100dB level is exceeded by harmonics up to the 66th harmonic.



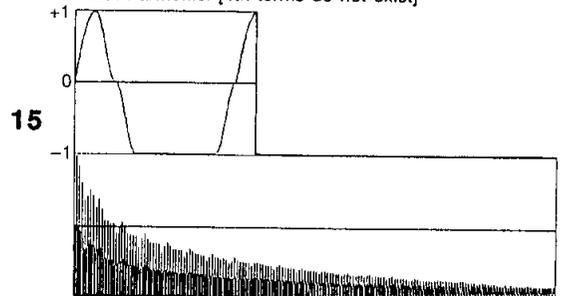
The -100dB level is exceeded by even harmonics up to the 310th harmonic, and odd harmonics up to the 49th harmonic.



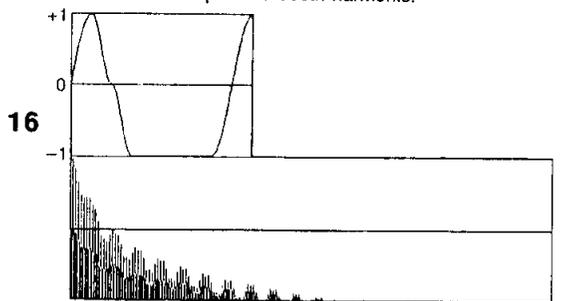
The -100dB level is exceeded by the 21st harmonic, and odd numbered harmonics up to the 63rd harmonic.



The -100dB level is exceeded by harmonics up to the 105th harmonic. [4th terms do not exist]



The -100dB level is exceeded by all harmonics up to the 300th harmonic.



The -100dB level is exceeded by harmonics up to the 91st harmonic.

3. (F1) AFM operator EG (Each operator)

JUMP #236

Summary: Make EG settings for a single operator while viewing a graphic display of the operator envelope.

Procedure:

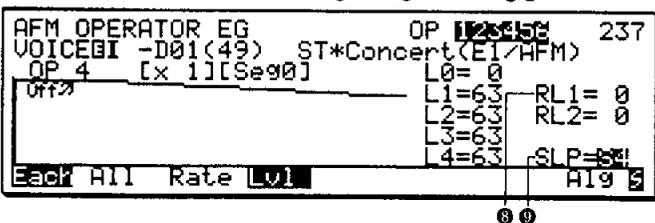
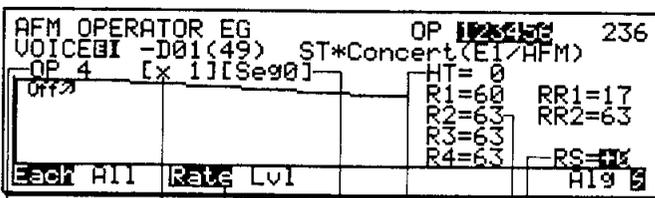
From: AFM Element job directory (JUMP #230)

Select: job 03:EG and press F1 (Each)

EG rates press F3 (Rate) (JUMP #236)

EG levels press F4 (Lvl) (JUMP #237)

Specify: envelope parameters for the selected operator



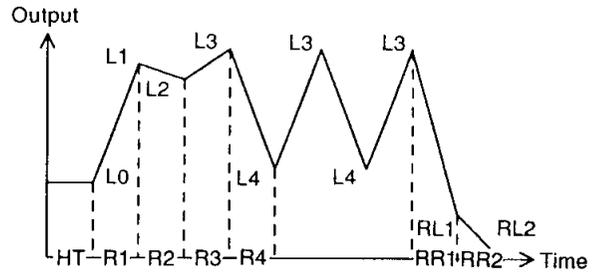
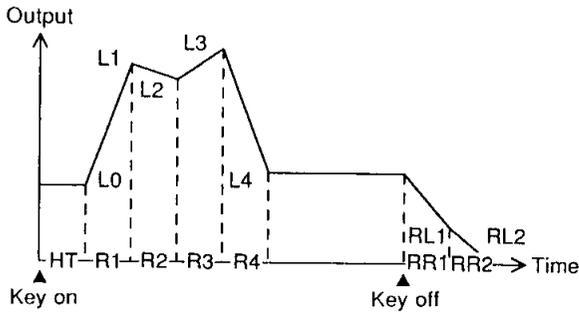
- ① The number of the currently selected operator is displayed. To move to another operator, use the OPERATOR SELECT buttons OP1...OP6.
- ② This indicates the time range of the EG graphic display. A range of "x1" shows the shortest time and gives the greatest detail. To change the range, hold SHIFT and press F1-F6 (x1, x2, x5, x10, x20, x50).
- ③ This indicates the segment from which the EG graphic display begins. To shift the display to a different segment, hold SHIFT and press F7 or F8 to select Seg0-Seg4 or Rel1.
- ④ Press F3 (Rate) to set EG rates. Press F4 (Lvl) to set EG levels.
- ⑤ HT (Keyon Delay Time 63...0): When this is set to 0, the operator EG will begin immediately after a key is pressed. For higher settings, there will be an increasingly longer delay before the operator EG begins.

- ⑥ R1-R4, RR1-RR2 (Keyon Rates, Release Rates 0...63): Keyon Rates 1-4 and Release Rates 1-2 determine the speed of the operator EG. Higher settings result in faster change.
- ⑦ RS (Rate Scaling -7...+7): Rate Scaling allows the operator EG rates to be increased or decreased depending on the key that is played. For positive settings the EG rates will increase as you play higher notes, resulting in shorter envelopes. For negative settings the EG rates will decrease as you play higher notes, resulting in longer envelopes.
- ⑧ L0-L4, RL1-2 (Keyon Levels, Release Levels 0...63): Keyon Levels 0-4 and Release Levels 1-2 determine the levels of the operator EG.
- ⑨ SLP (Segment Loop Point S1...S4): This specifies the segment from which the EG will continue looping if a key remains depressed after the EG has arrived at level L4.

Rates and Levels: When you press a key, the operator output will be at the level of L0. When the specified hold time (HT) has elapsed, the level will change at the rate of R1 to level L1. When the level reaches L1, it will change at the rate of R2 to the level of L2. When the level reaches L2, it will change at the rate of R3 to the level of L3. When the level reaches L3, it will change at the rate of R4 to the level of L4. When the level reaches RL1, the EG will begin looping from the specified segment.

When you release the key, the level will change at the rate of RR1 to the level of RL1. When the level reaches RL1, it will change at the rate of RR2 to the level of RL2.

Remember that Hold Time (HT) is a *time* setting, but the various Rates are *speed* settings. Higher settings for Hold Time will result in a longer delay before the operator EG begins, but higher settings for Rates will result in faster change.



Segment Loop Point: The SLP setting determines the Level from which the EG will loop. If you continue holding a note after Level 4 is reached, when SLP is set to ...

SLP=S1 the level will change L4 → L1 → L2 → L3 → L4 → L1 → ...

SLP=S2 the level will change L4 → L2 → L3 → L4 → L2 → ...

SLP=S3 the level will change L4 → L3 → L4 → L3 → ...

SLP=S4 the level will remain at L4

The following diagram shows how EG levels would change when SLP=S3

Rate Scaling: On most acoustic instruments, high notes have a naturally shorter attack and decay. This can be simulated by setting rate scaling to a positive value (+1...+7). The following diagram shows how higher notes will have faster rates (shorter EGs). Negative settings will have the opposite effect.



AFM ELEMENT DATA

3. (F2) AFM operator EG (All operators)

JUMP #238

Summary: Make operator EG settings for all operators while viewing the data for all operator EGs in a single screen.

Procedure:

From: AFM Element job directory (JUMP #230)

Select: job 03:EG and press F2 (All)

Specify: EG key-on rates (R1-R4) (JUMP #238)

EG key-on levels (L1-L4) (JUMP #239)

EG key-off rates and levels (RR1-2, RL1-2) (JUMP #240)

AFM OPERATOR EG													OP	238
VOICEBI -D01(49)													ST*Concert(EI/HFM)	
Keyon Rates & Rate Scaling														
	HT	R1	R2	R3	R4	RS	HT	R1	R2	R3	R4	RS		
1	0	37	37	0	21	+2	4	0	60	63	63	63	+0	
2	0	60	63	63	63	+0	5	0	37	30	0	21	+2	
3	0	37	30	0	21	+2	6	0	60	63	63	63	+0	
Each All UnR UnL K-of													H19	

AFM OPERATOR EG													OP	239
VOICEBI -D01(49)													ST*Concert(EI/HFM)	
Keyon Levels & Loop Point														
	L0	L1	L2	L3	L4	LP	L0	L1	L2	L3	L4	LP		
1	0	63	63	55	55	0	0	63	63	63	63	64		
2	0	63	63	55	55	0	0	63	63	55	55	64		
3	0	63	63	55	55	0	0	63	63	63	63	64		
Each All UnR UnL K-of													H19	

AFM OPERATOR EG													OP	240
VOICEBI -D01(49)													ST*Concert(EI/HFM)	
Keyoff Rates & Levels														
	RR1	RR2	RL1	RL2	RR1	RR2	RL1	RL2						
1	0	63	0	0	4	0	63	0	0					
2	17	63	0	0	5	0	63	0	0					
3	30	63	0	0	6	17	63	0	0					
Each All UnR UnL K-of													H19	

- ① HT (Keyon Delay Time 63...0): This specifies the time by which the beginning of the EG will be delayed after a key is pressed.
- ② R1-R4 (Keyon Rates 0...63): Keyon Rates 1-4 determine the speed of the operator EG while a key is being pressed.
- ③ RS (Rate Scaling -7...+7): Rate Scaling determines how the key position will affect the operator EG rates.

VOICE EDIT MODE

- ④ L0-L4 (Keyon Levels 0...63): These determine the levels to which the operator EG will move while a key is being pressed.
- ⑤ LP (Segment Loop Point S1...S4): This specifies the segment from which the EG will continue looping if a key remains depressed after the EG has arrived at level L4.
- ⑥ R1, R2 (KeyOff Rates 0...63): These determine the speed with which the operator EG will change levels after a key is released.

- ⑦ L1, L2 (KeyOff Levels 0...63): These determine the levels to which the operator EG will change after a key is released.

The meaning of these EG parameters is explained in the previous section 3.1 *Operator EG*.

In this *AFM operator EG (All)* display, the OPERATOR SELECT buttons cannot be used to select operators.

AFM ELEMENT DATA

4. (F1) AFM operator output (Each)

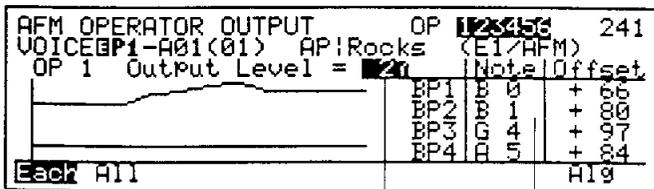
JUMP #241

Summary: Set output level and scaling for a single operator while viewing a graphic display of the scaling.

Procedure:

From: AFM Element job directory (JUMP #230)
 Select: job 04:Output and press (JUMP #241)
 F1 (Each)

Specify: the output level and scaling for the selected operator

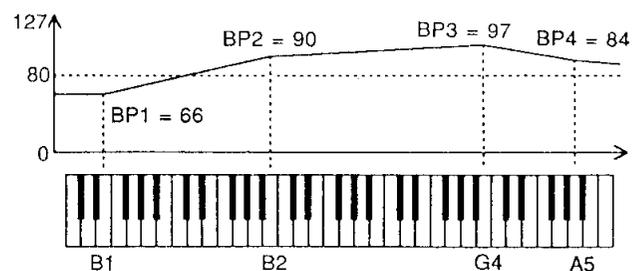


Break Point: The operator output level can be made to vary depending on the note that is played. On most acoustic instruments, notes differ in volume and tone depending on the range in which they are played. For example the low notes of a piano are more tonally complex and louder than the high notes.

Use the four break points to specify how the operator output level will be adjusted across the keyboard.

- Offset (-127...+127) determines the output level adjustment for each of the four points specified by Note.
- The four note settings must be in ascending order. It is not possible to set a break point to a note lower or higher than the note settings of the neighboring break points.

The following diagram shows how the operator output level would be adjusted across the keyboard for the settings shown in the above LCD.



Each offset is added to the overall operator output level of 80. For example the offset at break point 1 (E1) is -4, so the resulting operator output level at E1 is 76. The resulting operator output level is limited to the range of 0...127.

- ① Output Level (0...127): The output level of the operator
- ② BP1-4 (Break Point 1-4): Note (C-2...G-8) and Offset (-127...#127) of each Break Point determine how the output level of each operator will vary across the keyboard. When the cursor is located at note, you can press F7 (Kbd) and press a key to enter the new note setting.

Output Level: This sets the output level of each operator. The output level of a carrier operator will affect the volume, and the output level of a modulator operator will affect the tone. Remember that the input levels of each operator input In1 and In2 can also be adjusted as explained in 1. (F3) *Algorithm (Input level)* (JUMP #233). Even if the output level of an operator is raised, it will have no effect on another operator to which it is connected if the corresponding input level of the operator is set at 0.

AFM ELEMENT DATA

4. (F2) AFM operator output (All)

JUMP #242

Summary: Set operator output level while viewing output levels for all operators. (Output scaling cannot be set in this job.)

Procedure:

From: AFM Element job directory (JUMP #230)

Select: job 04:Output and press (JUMP #242)

F2 (All)

Specify: the output level of each operator

AFM OPERATOR OUTPUT							OP	123456	242	
VOICEBI -D01(49)							ST*Concert(EI/AFM)			
Output Level										
OP1	OP2	OP3	OP4	OP5	OP6					
127	104	127	108	127	108					
Each All							All			

1

- 1 Output Level OP1-OP6 (0...127): Set the output level of each operator. This is the same setting as explained in 1 of the previous section 4. (F1) AFM operator output. The difference is that here you can view and set the output level for all six operators at once. However, breakpoint Levels and Offsets cannot be set here.

Remarks: Refer to 4. (F1) AFM operator output for details.

Note: In algorithms with two or more carriers, some Velocity Sensitivity settings may cause distortion. In this case, reduce carrier levels.

AFM ELEMENT DATA

5. AFM sensitivity

JUMP #243

Summary: These settings determine how each operator will be affected by key-on velocity and by the LFO.

Procedure:

From: AFM Element job directory (JUMP #230)

Select: job 05:Sensitiv (JUMP #243)

Specify: the sensitivity of each operator

AFM SENSITIVITY							OP	123456	243	
VOICEBI -D01(49)							ST*Concert(EI/AFM)			
	OP1	OP2	OP3	OP4	OP5	OP6				
Velocity	+3	+1	+3	+1	+3	+1				
Rate Vel	on	on	on	on	on	on				
AModSens	0	0	0	0	0	0				
PModSens	0	0	3	3	3	0				
KVS	Rate	AMS	PMS				All			

1 2 3 4

5

- 1 Velocity (-7...+7): This determines how the output level of each operator will be affected by key-on velocity. For positive settings (+1...+7) the output level will increase as you play more strongly. For negative settings (-1...-7) the output level will decrease as you play more strongly.
- 2 Rate Velocity (on/off): When the Rate Velocity switch is "on", key-on velocity will affect the operator EG R1. The result will depend on the Velocity setting.

Velocity = +1...+7: If Rate Velocity is on, strongly played notes will cause the operator EG R1 to increase, resulting in a faster attack. For notes played with maximum velocity, R1 will be at the value specified by the EG settings.

Velocity = -1...-7: If Rate Velocity is on, strongly played notes will cause the operator EG R1 to decrease, resulting in a slower attack. To hear the effect of negative settings you will need to lower the operator output level.

When the Rate Velocity switch is "off", the operator EG R1 will not be affected by key-on velocity.

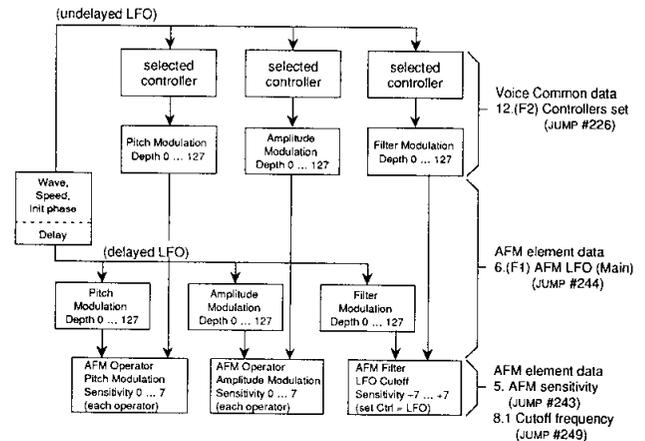
- 3 AModSens (0...7): Amplitude Modulation Sensitivity determines how greatly the output level of each operator will be affected by Amplitude Modulation from the LFO.
- 4 PModSens (0...7): Pitch Modulation Sensitivity determines how greatly the pitch of each operator will be affected by Pitch Modulation from the LFO.
- 5 Pressing F1 (KVS), F2 (Rate), F3 (AMS), or F4 (PMS) will move the cursor to Velocity, Rate Vel, AModSens, or PModSens.

VOICE EDIT MODE

AModSens and PModSens: These settings determine the *sensitivity* of each operator to the Amplitude Modulation Depth (AMD) and/or Pitch Modulation Depth (PMD) produced by the LFO. Refer to 6.(F1) AFM LFO (Main) (JUMP #244). If the LFO settings for AMD and/or PMD are set to 0, these AModSens and PModSens settings will have no effect.

PModSens determines the sensitivity of each operator to PMD from the Main LFO. Independently of this, the pitch of an AFM element can also be affected by the Sub LFO. Refer to 6.(F2) AFM LFO (Sub) (JUMP #245).

In this *AFM sensitivity* display, the OPERATOR SELECT buttons cannot be used to select operators.



AFM ELEMENT DATA

6. (F1) AFM LFO (Main)

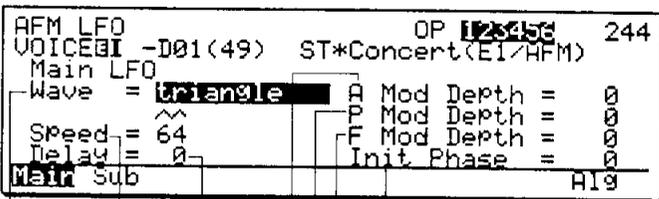
JUMP #244

Summary: The Main LFO creates a periodic control signal that can be used to create tremolo (amplitude modulation), vibrato (pitch modulation), and wah-wah (filter modulation).

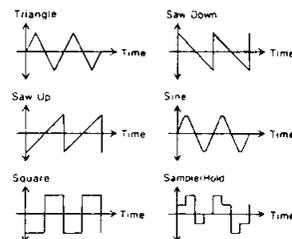
Procedure:

From: AFM Element job directory (JUMP #230)
 Select: job 06:LFO and press (JUMP #244)
 F1 (Main)

Specify: parameters for the main LFO

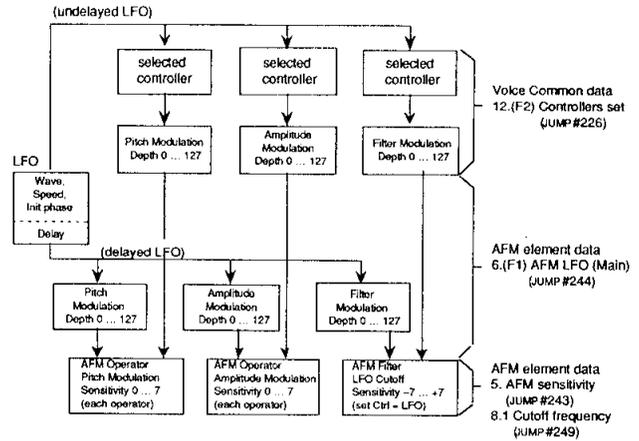


① Wave (triangle, saw down, saw up, square, sine, sample&hold): This selects the wave (shape of modulation) produced by the Main LFO. The selected wave is graphically displayed in the LCD. When sample&hold is selected, the LFO will produce a control signal whose level will change randomly at intervals of time determined by the Speed setting.



- ② Speed (0...99): The speed of the LFO modulation. Higher settings result in faster modulation. At a setting of 0, one cycle will be approximately 34 seconds.
- ③ Delay (0...99): The time delay before the LFO modulation begins.
- ④ Init Phase (0...99): Initial Phase determines the point of the waveform from which the LFO will begin when a key is pressed. The LFO waveform always starts over again from this initial phase point when each note is played. An initial phase setting of 0...99 corresponds to a phase of 0...360 degrees.
- ⑤ A Mod Depth (0...127): Amplitude Modulation Depth determines how much the LFO will affect the output level (amplitude) of the operators. For this setting to have an effect, the AModSens (amplitude modulation sensitivity) of an operator must be set above 0. Refer to 5. *AFM sensitivity*.

- ⑥ P Mod Depth (0...127): Pitch Modulation Depth determines how much the LFO will affect the pitch of the operators. For this setting to have an effect, the PModSens (pitch modulation sensitivity) of an operator must be set above 0. Refer to 5. AFM sensitivity.
- ⑦ F Mod Depth (0...127): Filter Modulation Depth determines how much the LFO will affect the cutoff frequency of the filter. For this setting to have an effect, the Ctrl setting of a filter must be set to "LFO", and the LFO Cutoff Sens setting must not be 0. Refer to 8.1 Cutoff frequency.



AFM ELEMENT DATA

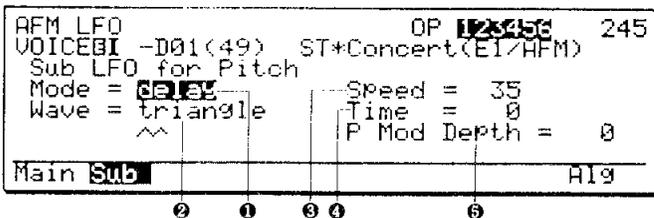
6. (F2) AFM LFO (Sub)

JUMP #245

Summary: The Sub LFO is completely independent of the Main LFO, but can be used only to create vibrato (pitch modulation). This will apply equally to all operators, and is not affected by pitch modulation sensitivity.

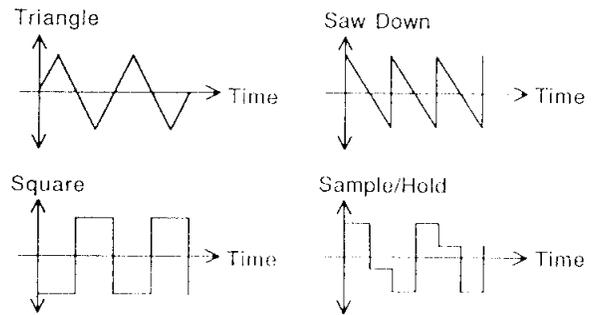
Procedure:

- From: AFM Element job directory (JUMP #230)
- Select: job 06:LFO and press (JUMP #245)
- F2 (Sub)
- Specify: parameters for the sub LFO

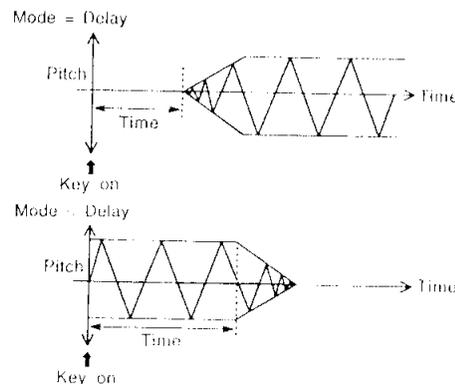


- ① Mode (delay, decay): When this is set to "delay", the Sub LFO will begin after the time delay specified by ④ Time. When this is set to "decay", the Sub LFO will begin fading out after the time specified by ④ Time.
- ② Wave (triangle, saw down, square, sample& hold): The wave produced by the Sub LFO.
- ③ Speed (0...99): The speed of the LFO modulation. Higher settings result in faster modulation. The speed of the sample&hold wave will be faster than the other waves.
- ④ Time (0...99): The time length used for Sub LFO delay or decay.
- ⑤ P Mod Depth (0...127): The depth of pitch modulation produced by the Sub LFO.

Wave: The following four waveforms can be selected for the Sub LFO.



Mode and Time: The mode and time settings work together to determine how the Sub LFO will begin or end. When Mode=delay the Sub LFO will begin after the time delay specified by ④ Time. When Mode=decay the Sub LFO will affect the sound beginning immediately from when the key is pressed, but will gradually die out after the time delay specified by ④ Time.

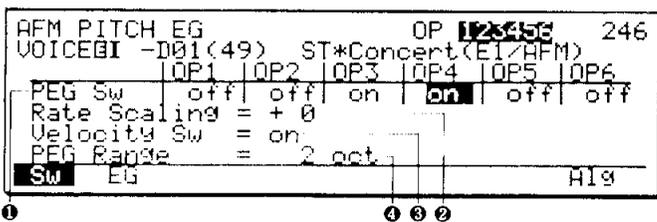


AFM ELEMENT DATA **7. (F1) AFM pitch EG (Switch)** JUMP #246

Summary: The pitch change over time created by the pitch EG can be switched on/off for each operator. To set the shape of the pitch EG, see 7. (F2) AFM pitch EG (EG settings).

Procedure:

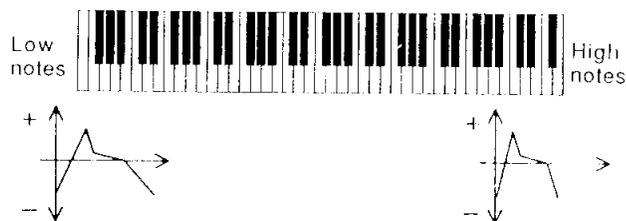
- From: AFM Element job directory (JUMP #230)
- Select: job 07:PitchEG and press (JUMP #246)
- F1 (Sw)
- Specify: pitch EG switches, scaling, and range



- 1 PEG Sw (off, on): When the Pitch EG Switch is "off" for an operator, it will not be affected by the pitch EG.
- 2 Rate Scaling (-7...+7): Pitch EG Rate Scaling determines how pitch EG rates will change according to the note played. When this is set to +1...+7, the pitch EG will be faster for higher notes. When this is set to -1...-7, the pitch EG will be slower for higher notes. When this is set to 0, the pitch EG will be the same rate for all notes.
- 3 Velocity Sw (off, on): When this is on, strongly played notes will cause the pitch EG to change over a greater range.
- 4 Range (1/2 oct, 1 oct, 2 oct, 8 oct): This determines the maximum range of the AFM pitch EG, from 1/2 octave to 8 octaves.

PEG Sw: When using the Pitch EG to make the pitch of a sound change over time, you will normally turn the PEG switch on for *all* operators. If the pitch of a modulator operator changes while the pitch of another operator it is modulating remains constant (or vice versa), the *carrier:modulator* ratio will shift during the duration of the sound, changing the overtone structure. This can be an interesting effect in its own right.

Rate Scaling: This setting determines how Pitch EG Rates (the speed of pitch change) will be affected by the key number of each note. The following diagram shows the result when Pitch EG Rate Scaling is set to +7. Notice that high notes have a shorter pitch EG (faster EG rates) than lower notes.



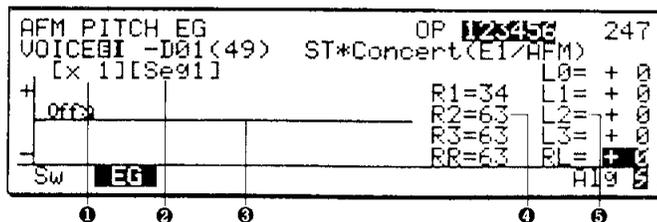
Velocity Sw (velocity switch): When this is "on", strongly played notes will cause the pitch EG to change over a greater range.

AFM ELEMENT DATA **7. (F2) AFM pitch EG (EG settings)** JUMP #247

Summary: The pitch EG creates a fixed shape of pitch change over time for each note. To switch the pitch EG on/off for each operator, see 7. (F1) AFM pitch EG (Switch).

Procedure:

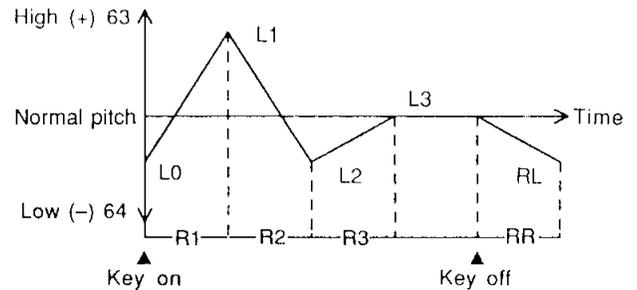
- From: AFM Element job directory (JUMP #230)
- Select: job 07:PitchEG and press (JUMP #247)
- F2 (EG)
- Specify: pitch EG parameters



- ❶ This indicates the time length shown by the graphic display. To change this, hold SHIFT and press F1–F6 (x1, x2, x5, x10, x20, x50). The exact length of time will depend on the range. When the pitch EG range is 1 octave, the graphic display will cover approximately 0.5 seconds at “x1” and approximately 5 seconds at “x50”.
- ❷ This indicates the EG segment (“seg1–3” or “rel1”) from which the pitch EG graphic display begins. If the EG is too long to be fully shown in the LCD, hold SHIFT and press F7 or F8 to move the pitch EG graphic display to a different segment.
- ❸ The pitch EG is graphically displayed.
- ❹ R1–R3, RR1 (0...63): Keyon Rates 1–3 and the Release Rate determine the speed of the pitch EG. Higher settings result in faster change. A rate of 63 will make the pitch jump instantly to the next level.
- ❺ L0–L3, RL (-64...+63): Keyon Levels 0–3 and the Release Level determine the levels of the pitch EG. Positive settings raise the pitch and negative settings lower the pitch.

Rates and Levels: When you press a key, the pitch will begin at the level of L0, and will change at the rate of R1 to level L1. When the level reaches L1, the pitch will change at the rate of R2 to the level of L2. When the pitch reaches L2, it will change at the rate of R3 to the level of L3 and will stay at L3 as long as the key is pressed.

When the key is released, the pitch will change at the rate of RR to the level of RL.



Note: Even if the AFM pitch EG and the AWM pitch EG have identical Rate settings, there will be slight differences in the timing of the pitch change.

AFM ELEMENT DATA

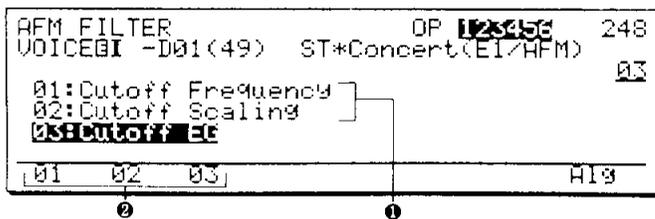
8. AFM filter

JUMP #248

Summary: The two filters of each element can be used to control the tone in various ways.

Procedure:

- From: AFM Element job directory (JUMP #230)
- Select: job 08:Filter (JUMP #248)
- Specify: the desired job and press ENTER



- ❶ Move the cursor in this area to select a job and press ENTER to move to the selected job.
 - 01: Cutoff Frequency: Make overall settings for the filters.
 - 02: Cutoff Scaling: Specify how each filter will be adjusted across the keyboard.
 - 03: Cutoff EG: Specify how each filter will change over time.
- ❷ Pressing F1–F3 will select the corresponding job.

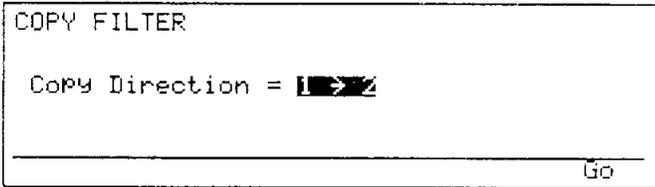
AFM ELEMENT DATA / AFM FILTER

8.0 Copy filter

Summary: Any time while editing a filter, you can copy the data from one filter to the other filter.

Procedure:

- From: 8.1 Cutoff Frequency (JUMP #249)
- 8.2 Cutoff Scaling (JUMP #250)
- 8.3 Cutoff EG (JUMP #252-#255)
- Press: COPY
- Select: the copy direction (1 → 2 or 2 → 1)
- To execute: the copy operation press F8 (Go)
- To quit: without copying press EXIT



Specify whether to copy the data from filter 1 to filter 2 (1 → 2) or from filter 2 to filter 1 (2 → 1). Press F8 (Go) and the data will be copied. If you decide not to copy the data, press EXIT to exit without copying.

The filter type (HPF/LPF/THRU) will not be copied.

AFM ELEMENT DATA / AFM FILTER

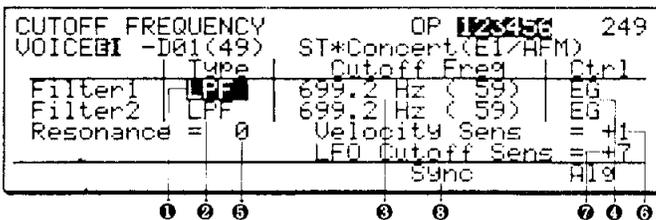
8.1 Cutoff frequency

JUMP #249

Summary: Each filter can be set to a different type, cutoff frequency, and control source. Overall resonance, velocity sensitivity, and LFO Cutoff Sensitivity can also be specified.

Procedure:

- From: AFM Element job (JUMP #248)
- directory 8. AFM filter
- Select: 01:Cutoff Frequency (JUMP #249)
- Specify: parameters for filters 1 and 2



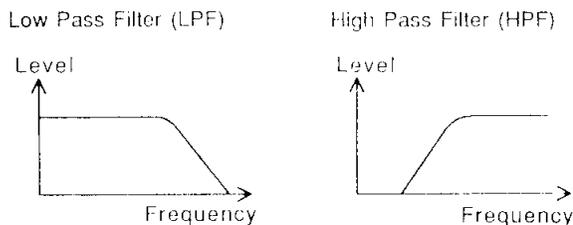
- ❶ Filter 1 Type (Thru, LPF, HPF): Filter 1 can be used either as a Low Pass Filter (LPF) or as a High Pass Filter (HPF). When “Thru” is selected the filter will have no effect.
- ❷ Filter 2 Type (Thru, LPF): Filter 2 can be used only as a LPF.
- ❸ Cutoff Freq (HPF = 0 Hz ... 11.66 kHz (0...114); LPF = 0 Hz ... 22.43 kHz (0...127): The cutoff frequency of each filter can be adjusted independently. The number 0...127 displayed in parentheses indicates the data value input when using the numeric keypad. Note that the highest HPF setting is 11.66 kHz.

- ❹ Ctrl (EG, LFO, EG-VA): Each of the two filters can be controlled in a different way. For details, see the explanations below for *Ctrl = EG*, *Ctrl = LFO*, *Ctrl = EG-VA*.
- ❺ Resonance (0...99): Higher settings of resonance will result in a more pronounced peak of emphasis at the cutoff frequency. This setting will apply to both filters 1 and 2.
- ❻ Velocity Sens (-7...+7): This determines how the cutoff frequency of both filters will be affected by key-on velocity. For positive settings (+1...+7) the cutoff frequency will increase as you play more strongly, resulting in a brighter sound. For negative settings (-1...-7) the cutoff frequency will decrease as you play more strongly, resulting in a darker sound.
- ❼ LFO Cutoff Sens (-7...+7): This determines how Filter Modulation from the Main LFO will affect the filters. This setting also determines how sensitive the filters will be to the controller assigned to Filter Bias in *Voice common data 12. (F5) Controller set (Other)* (JUMP #228) (page 130). Negative settings will reverse the effect of the assigned controller.
- ❽ Sync: This is not a parameter, but a convenience that allows you to adjust both filters together. When you press F8 (Sync), the “Sync” display will be inverted, and all adjustments you make to either filter 1 or filter 2 will be automatically

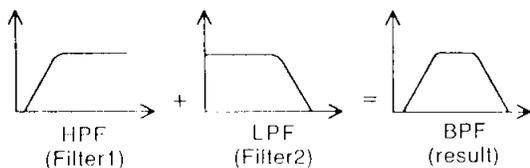
set for the same parameter of the other filter. This is useful when you are using the two filters in tandem to act as a steep 24 dB/octave filter or as a 12 dB/octave band pass filter. See the following explanation *Type and Cutoff Freq.* Press F6 once again to cancel filter sync editing mode.

Type and Cutoff Freq: Filter 1 can be used either as a Low Pass Filter (LPF) or as a High Pass Filter (HPF), and filter 2 can be used only as a LPF.

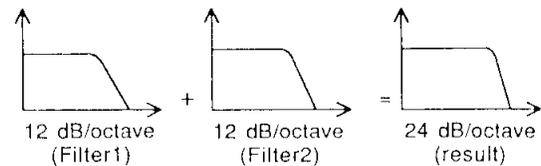
When set to LPF, filters 1 and 2 will allow sound lower than the cutoff frequency to pass unchanged, and will diminish the sound above the cutoff frequency. When set to HPF, filter 1 will allow sound higher than the cutoff frequency to pass, and will diminish the sound below the cutoff frequency.



By setting filter 1 to HPF and filter 2 to LPF, you can create a Band Pass Filter that passes only a central band of frequencies.

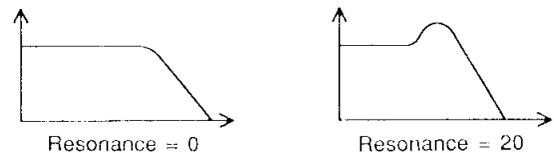


Each of the SY99's filters has a slope of 12 dB/octave. This means that if the cutoff frequency of an LPF is 1 kHz, frequencies at 2 kHz will be reduced by 12 dB and frequencies at 4 kHz will be reduced by 24 dB. If you set both filters 1 and 2 to LPF, set both to the same cutoff frequency, and set both filter EGs in the same way, the result will be the equivalent of a single 24 dB/octave filter. The filter copy function explained in *8.0 Copy filter* is a quick way to give both filters the same settings.



24 dB/octave filtering creates a sharp cutoff which is quite obvious, while 12 dB/octave filtering is a more subtle effect. Analog synthesizers of the past have used both types. 12 dB/octave filtering was considered especially suitable for strings, and 24 dB/octave filtering was commonly used for brass or synth bass sounds.

Resonance: Resonance lowers the level of sound below the cutoff frequency, creating an increased peak of emphasis. (This may reduce the overall volume.) High settings of resonance will make changes in cutoff frequency quite easy to notice. When the two filters are being used in tandem as a Band Pass Filter (i.e., when filter 1 is set to HPF), resonance will have no effect.

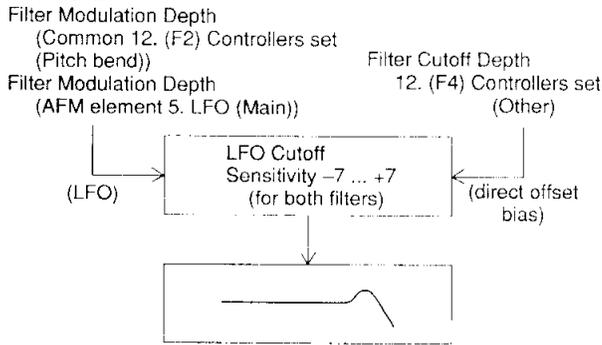


Extremely high settings of filter resonance will make the filter oscillate so that it produces a pitch of its own. This is a technique often used on analog synthesizers of the past.

Ctrl = LFO: When Ctrl is set to LFO, the filter will be controlled both by the Main LFO and by the controller which has been assigned to Filter Cutoff Depth. (Refer to *Voice common data 12. (F4) Controller set (Pan)*, page 130.) Key velocity will shift the cutoff frequency.

VOICE EDIT MODE

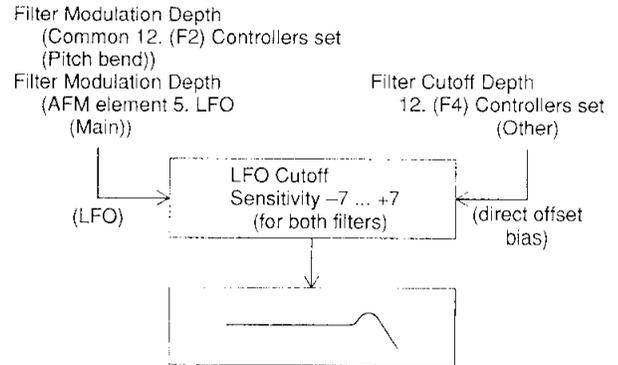
Ctrl = LFO



Ctrl = EG: When Ctrl is set to EG, the filter will be controlled by its own filter EG as explained in the following section 8.3 *Filter EG*. If Velocity Sense is set to a value other than 0, key velocity will shift the overall offset of the EG. The position of the controllers assigned to Filter Modulation Depth and Filter Cutoff Depth will be sampled at the beginning of the note (key on), but will have no effect *during* the note.

Ctrl = EG-VA: When Ctrl is set to EG-VA (EG voice attack), the filter will be controlled by its own filter EG as explained in the following section 8.3 *Filter EG*. If Velocity Sense is set to a value other than 0, key velocity will modify L1 (level 1) and R1 (rate 1) of the filter EG.

Ctrl = EG or Ctrl = EG - VA



Note: When Ctrl=EG or Ctrl=EG-VA, the effect of the controller on the cutoff frequency will be fixed when the note is played. Moving the controller after playing the note will have no effect.

AFM ELEMENT DATA / AFM FILTER

8.2 Cutoff scaling

JUMP #250

Summary: The cutoff frequency of each filter can be adjusted across the keyboard.

Procedure:

- From: AFM element job (JUMP #248)
directory 8. AFM filter
- Select: 02:Cutoff Scaling
- filter 1 press F1 (Flt1) (JUMP #250)
- filter 2 press F2 (Flt2) (JUMP #251)
- Specify: filter scaling parameters

CUTOFF SCALING		OP	250
VOICE BP1-A01(01)	APIRocks	(E1/AFM)	
LPF COF = 2.047kHz(80)	Note	Offset	
BP1	B 1	- 4	
BP2	B 2	+ 10	
BP3	G 4	+ 17	
BP4	A 5	+ 4	
Flt1	Flt2	Sync Kbd	H19

- 1 This indicates the type of the filter being edited, and its cutoff frequency. The cutoff frequency can be modified from this job, but to modify the type of filter you must use job 8.1 *Filter cutoff*.
- 2 BP1-4 (Break Point): Note (C-2...G-8) and Offset (-127...+127) of each Break Point determine

how the cutoff frequency level of the filter will vary across the keyboard. When the cursor is located at note, you can press F7 (Kbd) and press a key to enter the new note setting.

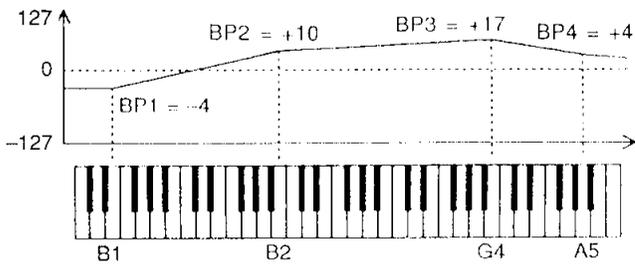
- 3 Sync: Press F6 to turn filter sync editing mode on or off. For details, refer to 3 Sync on page 150.

Break Point: The filter cutoff frequency can be made to vary depending on the note that is played. On most acoustic instruments, notes differ in tone depending on the range in which they are played.

Use the four break points to specify how the filter cutoff frequency will be adjusted across the keyboard. Offset (-127...+127) determines how the cutoff frequency will be adjusted at each of the four points specified by Note (C-2...G-8).

The four note settings must be in ascending order. It is not possible to set a break point to a note lower or higher than the note settings of the neighboring break points.

The following diagram shows how the filter cutoff frequency would be adjusted across the keyboard for the values in the LCD above.



The offset at each break point is added to the cutoff frequency of 80. For example the offset at break point 1 (E1) is -4, so the resulting cutoff frequency at E1 is 76. The resulting cutoff frequency is limited to the range of 0...127.

AFM ELEMENT DATA / AFM FILTER

8.3 Cutoff EG

JUMP #252

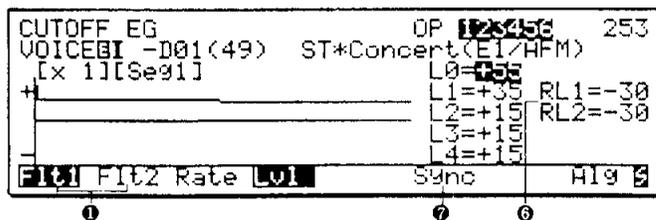
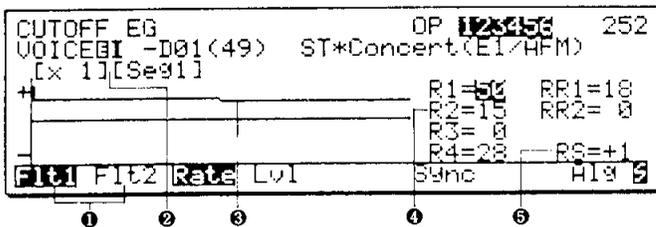
Summary: The cutoff frequency of each filter can be moved over time by its own EG to make the tone change.

Procedure:

From: AFM element job directory (JUMP #248)
8. AFM filter

- Select: 03:Cutoff EG
- filter 1 rates press (JUMP #252)
F1 (Flt1), F3 (Rate)
- filter 1 levels press (JUMP #253)
F1 (Flt1), F4 (Lvl)
- filter 2 rates press (JUMP #254)
F2 (Flt2), F3 (Rate)
- filter 2 levels press (JUMP #255)
F2 (Flt2), F4 (Lvl)

Specify: filter EG parameters



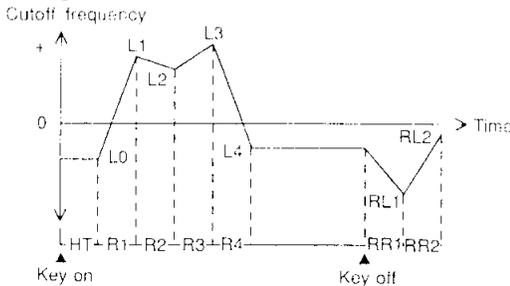
- ❶ This indicates whether you are editing the EG of filter 1 or 2.
- ❷ This indicates the displayed segment and range of the EG graphic display. To change the display range, hold SHIFT and press F1-F6 (x1, x2, x5, x10, x20, x50). To shift the display to a different segment, hold SHIFT and press F7 or F8 (Seg1...Seg4, Rel1).
- ❸ The filter EG is graphically displayed.
- ❹ R1-R4, RR1-RR2 (Keyon Rates, Release Rates 0...63): Keyon Rates 1-4 and Release Rates 1-2 determine the speed of the filter EG. Higher settings result in faster change.
- ❺ RS (Rate Scaling -7...+7): Rate Scaling allows the filter EG rates to be increased or decreased depending on the key that is played. For positive settings the EG rates will increase as you play higher notes, resulting in shorter envelopes. For negative settings the EG rates will decrease as you play higher notes, resulting in longer envelopes.
- ❻ L0-L4, RL1-2 (Keyon Levels, Release Levels -64...+63): Keyon Levels 0-4 and Release Levels 1-2 determine how the filter EG will increase or decrease the cutoff frequency specified for the filter.
- ❼ Sync: Press F6 to turn filter sync editing mode on or off. For details, refer to ❸ Sync on page 150.

Rates and Levels: The levels of the filter EG do not directly determine the cutoff frequency of the filter, but rather *adjust* the filter cutoff frequency you set in 8.1 Filter cutoff.

VOICE EDIT MODE

When a note is played, the filter cutoff will be adjusted by the amount of L0, and will change at the rate of R1 to level L1. When the level reaches L1, it will change at the rate of R2 to the level of L2. When the level reaches L2, it will change at the rate of R3 to the level of L3. When the level reaches L3, it will change at the rate of R4 to the level of L4. The filter cutoff frequency will remain at the level of L4 as long as you continue pressing the key.

When you release the key, the filter cutoff frequency will change at the rate of RR1 to the level of RL1. When the level reaches RL1, it will change at the rate of RR2 to the level of RL2.



Rate Scaling: On most acoustic instruments, high notes have a naturally shorter attack and decay. This can be simulated by setting rate scaling to a positive value (+1...+7). The following diagram shows how higher notes will have faster rates (shorter EGs). Negative settings will have the opposite effect.



AFM ELEMENT DATA

15. Initialize AFM element

Summary: Initialize the AFM element data being edited to a set of basic values.

Procedure:

From: AFM Element job (JUMP #230)
directory

Select: job 15:Initlz

To execute: the initialize operation press YES

To quit: without executing press NO or EXIT

INITIALIZE AFM ELEMENT

ARE YOU SURE ?

(Yes or No)

This function sets all AFM element data values to the minimum or simplest possible settings. When creating your own new voices it is usually best to begin by editing an existing voice. However if you want to start from scratch, this Initialize function can be helpful.

If you are sure you want to initialize the AFM element data, press YES. The AFM element data being edited will be set to the values shown below. If you decide not to initialize, press NO or EXIT.

This function initializes only AFM element data. Other initialize functions are provided for initializing Voice Common data and AWM element data. Refer to *Voice Common 15* (page 132). *Initialize voice or AWM element 15. Initialize AWM element* (page 170).

Initialized settings for AFM element data

01 Algorithm Set

Algorithm number	= 30
Feedback 1	= none (free)
Feedback 2	= none (free)
Feedback 3	= none (free)
Input Level 1	= 7 (operator 1-5)
Input Level 1	= 0 (operator 6)
Input Level 2	= 0 (all operators)
Noise	= Off (all operators)
AWM Wave	= Off (all operators)

02 Operator Oscillator (all operators)

Freq.Mode = Ratio
 Freq = 1.00
 Detune = ± 0
 Waveform = 1 (sine)
 Phase Sync = On
 Init Phase = 0

03 Operator EG (all operators)

Keyon Hold Time = 0
 Keyon Rates 1-4 = 63
 Keyoff Rates 1-2 = 63
 Rate Scaling = ± 0
 Keyon Level 0 = 0
 Keyon Levels 1-4 = 63
 Keyoff Levels 1-2 = 0
 Loop Point = S4

04 Operator Output

Output Level = 127 (operator 1)
 Output Level = 0 (operators 2-6)
 Break Point 1 Note = C1
 Break Point 2 Note = G2
 Break Point 3 Note = E4
 Break Point 4 Note = C6
 Break Point Levels = 0 (break points 1-4)

05 Operator Sensitivity (all operators)

Keyon Velocity Sens = 0
 Rate Velocity Switch = off
 AMS = 0
 PMS = 3

06 LFO

Main LFO
 Wave = triangle
 Speed = 35
 Delay Time = 0
 AMD, PMD, FMD = 0
 Init Phase = 0

Sub LFO

Mode = delay
 Wave = triangle
 Speed = 80
 Time = 0
 PMD = 0

07 Pitch EG

Operator On/Off = on (all operators)
 Rate Scaling = ± 0
 Velocity Switch = off
 Range = 8 oct
 Keyon Rates 1-3 = 63
 Keyoff Rate 1 = 63
 Keyon Levels 1-3 = ± 0
 Keyoff Level 1 = ± 0

08 Filter

Resonance = 0
 Cutoff Mod Sens = ± 0
 Keyon Velocity Sens = ± 0
 *** following data is same for both filters ***
 Filter Type = thru
 Filter Control = LFO
 Cutoff Frequency = 127
 Break Point 1 Note = C1
 Break Point 2 Note = G2
 Break Point 3 Note = E4
 Break Point 4 Note = C6
 Break Point Offset = 0 (BP 1-4)
 Keyon Rates 1-4 = 63
 Keyoff Rates 1-2 = 63
 Rate Scaling = ± 0
 Keyon Levels 0-4 = ± 0
 Keyoff Levels 1-2 = ± 0

AFM ELEMENT DATA

16. Recall voice

Summary: Recall all data of the previously edited voice.

Procedure:

From: AFM Element job (JUMP #230)
 directory

Select: job 16:Recall voice

To execute: the recall operation press YES

To quit: without executing press NO or EXIT.

Note: This operation recalls all voice data, not just AFM element data, and is also available while editing Common data, AWM Element data, or Drum Set data. For details refer to Voice Common 16. Recall voice, page 133.

AWM element data

AWM ELEMENT DATA

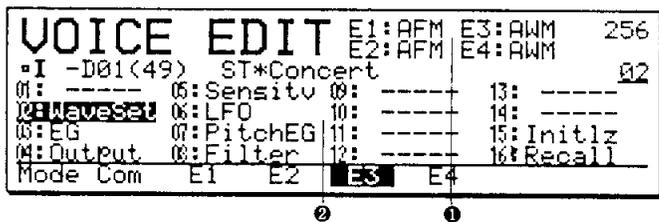
AWM element job directory

JUMP #256

Summary: This directory shows the jobs which edit AWM element data.

Procedure:

- From: voice edit mode (JUMP #200 or #201)
- When: editing a normal voice that contains AWM elements
- Select: an AWM element F3-F6 (JUMP #256) (E1-E4)



- ❶ This area shows the number (E1-E4) and type (AFM or AWM) of elements in the selected voice mode.
- ❷ Move the cursor in this area to select a job and press ENTER to go to the selected job.

- 02: WaveSet (AWM waveform set):** Select an AWM sampled waveform and specify the pitch at which it will sound.
- 03: EG (AWM EG):** The AWM Amplitude EG determines how the volume of each note will change over time.

04: Output (AWM output): The output level of an AWM element can be adjusted across the keyboard.

05: Sensitiv (AWM sensitivity): Key-on velocity can affect the volume or the speed of attack and decay. The control signal from the AWM LFO can create vibrato, tremolo, or wah-wah.

06: LFO (AWM LFO): The AWM element LFO creates a cyclically changing control signal that can be used for tremolo, vibrato, or wah-wah.

07: PitchEG (AWM pitch EG): The pitch of each note can be made to change in a fixed way over time.

08: Filter (AWM filter): The tone of an AWM element can be made to change in a fixed way over time, or can be controlled by a controller or the LFO.

15: Initlz (Initialize AWM element): When you are creating a voice from scratch, it is sometimes convenient to set all AWM element data to the basic or minimum values.

16: Recall (Recall voice): All data of the previously edited voice can be recalled.

AWM ELEMENT DATA

Copy element

Summary: While editing any AWM parameter (except for 8. AWM filter), you can copy data from an AWM element of another voice into the AWM element you are now editing.

Procedure:

- From: AWM element job 2, 3, 4, 5, 6, or 7
- Press: COPY
- Press: F1 (Src) and select the source voice
- Press: F2 (Elem) and select the source element

To execute: the copy operation press F8 (Go).

VOICE EDIT MODE

Preset 1

No.	WaveForm
1	Piano
2	Trumpet
3	MuteTp 1
4	MuteTp 2
5	Horn
6	Flugel
7	Trombone
8	Tuba
9	Brass
10	BrsFall
11	Tenor1
12	Tenor2
13	Alto Sax
14	Baritone
15	Soprano
16	Tenors
17	Flute
18	Clarinet
19	Piccolo
20	Reed Wv
21	Basoon
22	Recorder
23	MtReedWv
24	PanFlute
25	Violin
26	Cello
27	ContraBs
28	Pizz
29	SectPizz
30	Strings1
31	Strings2
32	StringsL
33	StringsR
34	Organ 1
35	Organ 2
36	E.P.Wv1
37	E.P.Wv2
38	Clavi 1
39	Clavi 2

No.	WaveForm
40	Celesta
41	Harpsi
42	Pipe Wv
43	AnlgBrs1
44	AnlgBrs2
45	AnlgBrs3
46	Pad 1
47	Pad 2
48	AnlgBass
49	FrtlsSyn
50	Chorus
51	Chorus L
52	Chorus R
53	Itopia
54	Choir
55	OohChoir
56	Vibe
57	Marimba
58	Tubular
59	Xylophon
60	Glocken
61	SteelDrm
62	HandBell
63	Shamisen
64	Koto
65	Harp
66	Sitar
67	E.Bass 1
68	E.Bass 2
69	E.Bass 3
70	ThmpBass
71	SlapBass
72	Fretless
73	WoodBass
74	GtrSteel
75	GtrNylon
76	12string
77	EG Sng1
78	EG Humbk

No.	WaveForm
79	EG Harm1
80	EG Harm2
81	EG Mute
82	EG Comp
83	EG Dist
84	EG Pluk1
85	EG Pluk2
86	BD 1
87	BD 2
88	BD 3
89	BD 4
90	BD 5
91	BD 6
92	BD 7
93	BD 8
94	SD 1
95	SD 2
96	SD 3
97	SD 4
98	SD 5
99	SD 6
100	SD 7
101	SD 8
102	SD 9
103	SD Side
104	Tom 1
105	Tom 2
106	Tom 3
107	Tom 4
108	Tom 5
109	Tom 6
110	HH foot
111	HH light
112	HH mid
113	HH heavy
114	HH open
115	HHclAnlg
116	HHopAnrg
117	Crash

No.	WaveForm
118	Ride
119	Choke
120	Claps 1
121	Claps 2
122	Cowbell1
123	Cowbell2
124	Tambrn
125	Shaker
126	FngrSnap
127	AnlgPerc
128	NoisePrc
129	Scratch
130	Agogo
131	Berimbau
132	Bongo
133	Cabasa
134	Cga Hi
135	CgaHiSlp
136	Cga Lo
137	CgaLoSlp
138	Clave
139	Guiro 1
140	Guiro 2
141	Maracas
142	SD roll
143	Tabla Hi
144	Tabla Lo
145	Temple
146	Timbale1
147	Timbale2
148	Timpani
149	Whisle
150	Belltree
151	BDs Wv
152	SDs Wv
153	Toms Wv
154	CymbalWv
155	Drums Wv

Preset 2

No.	WaveForm
1	Piano Np
2	E.P. Np
3	Vibe Np
4	DmpPiano
5	Bottle 1
6	Bottle 2
7	Bottle 3
8	Tube
9	Vocal Ga
10	Vocal Ba
11	Sax tran
12	Bow tran
13	Blub
14	Tear
15	Bamboo
16	Cup Echo
17	Digi Atk
18	Temp Ra
19	Giri
20	Water
21	Steam
22	Narrow
23	Airy
24	Styroll
25	Noise
26	Bell Mix
27	Haaa
28	OhAttack

No.	WaveForm
29	Typist
30	BellRing
31	SeqLatin
32	EleMagic
33	Vox Bell
34	Mellow
35	BigSyn L
36	BigSyn R
37	VoxGrace
38	Cry Bell
39	Voices
40	AnlgSaw1
41	AnlgSaw2
42	CS Saw
43	CS Sqr
44	Digital1
45	Digital2
46	Digital3
47	Digital4
48	Digital5
49	Digital6
50	Digital7
51	Digital8
52	Digital9
53	Digitl10
54	Digitl11
55	Digitl12
56	DigiVox1

No.	WaveForm
57	DigiVox2
58	DigiVox3
59	DigiVox4
60	DigiVox5
61	Pluse 10
62	Pluse 25
63	Pluse 50
64	Tri
65	DigiWild
66	Stuff 1
67	Stuff 2
68	Stuff 3
69	Stuff 4
70	Stuff 5
71	Stuff 6
72	Stuff 7
73	Stuff 8
74	Stuff 9
75	Stuff 10
76	Stuff 11
77	Stuff 12
78	Stuff 13
79	Stuff 14
80	Stuff 15
81	Stuff 16
82	Stuff 17
83	Stuff 18
84	Stuff 19

No.	WaveForm
85	Stuff 20
86	Stuff 21
87	Stuff 22
88	Stuff 23
89	Stuff 24
90	Stuff 25
91	Stuff 26
92	Stuff 27
93	Stuff 28
94	Stuff 29
95	Stuff 30
96	Stuff 31
97	Stuff 32
98	Stuff 33
99	Stuff 34
100	Stuff 35
101	Stuff 36
102	Stuff 37
103	Stuff 38
104	Stuff 39
105	Stuff 40
106	Stuff 41
107	Stuff 42
108	Stuff 43
109	Stuff 44
110	Stuff 45
111	Stuff 46
112	Stuff 47

2.0 Waveform edit

Summary: Using these functions, you can edit the samples that have been loaded into the internal sample memory, and assign them as waveforms.

Procedure:

When: internal waveforms have been selected by pressing F4 from waveform set

(JUMP #257)

Press: F8 (Edit).

Select: a waveform to edit.

Specify: the first and last samples used by the waveform.

To clear: the sample assignments for the waveform, press F5 (Init).

To rename: the waveform, press F7 (Name).

To edit: sample data, press F8 (Smpl).

WAVEFORM EDIT			01
Waveform name	From	To	
01: New Wave	01	03	
02: New Wave	03	03	
03: INT-WAVE			
04: INT-WAVE			
	Init	Name	Smpl

1
2
4
5
6

- 1 This area displays the names and numbers of internal waveforms (1-64). Move the cursor or use the numeric keys to select a waveform. If necessary, press F1 or F2 to scroll the display.
- 2 Move the cursor to the "From" column to specify the first sample used by the selected waveform.
- 3 Move the cursor to the "To" column to specify the first sample used by the selected waveform. (The samples used by a waveform must be consecutive.)
- 4 To initialize the currently selected waveform, press F7 (Init). For details refer to the following section *Initialize waveform*.
- 5 To rename the currently selected waveform, press F7 (Name). For details refer to the following section *Waveform name*.
- 6 To edit samples or change the sample assignment for the currently selected waveform, press F8 (Smpl). Refer to the following section *Sample assign*.

To copy samples: Samples may be copied within the sample memory using the sample utility function (JUMP #827) described in *Sample utility 1. Sample directory*, page 275.

To load samples: To load samples via MIDI sample dump, simply connect the MIDI OUT of the transmitting device to the MIDI IN of the SY99, and make the transmitting device transmit a sample dump. The SY99 can receive sample dumps at any time, regardless of the mode it is in.

To load sample data from a floppy disk, use the operation described in *Sample utility 3. Load from disk*, page 277. To load waveform data from a waveform card, use the operation described in *Card utility 4. Load from wave card*, page 264. Samples from disks and waveform cards may be edited; keep in mind, however, that sample data loaded to the SY99 from waveform cards and copy-protected disks cannot be saved to disk or output via MIDI sample dump.

Before samples can be used in an AWM element, they must be assigned to a waveform. Refer to the following section *Sample assign*.

Remarks: When referring to this and the following sections, keep in mind the difference between samples and waveforms:

A sample is a piece of digitally recorded sound. You can load up to 99 user samples from disk or via MIDI sample dump, and adjust the pitch, volume, and loop mode of each sample. (It is not possible to edit the samples of a preset or card waveform.)

A waveform consists of one or more samples assigned across the keyboard. You can create 64 user waveforms by specifying a range of samples to assign to the waveform, the key range sounded by each sample, and the key at which each sample will sound its original recorded pitch. (It is not possible to edit preset or card waveforms.) Refer to the following section 2. (F8) *Sample assign*.

AWM ELEMENT DATA / WAVEFORM EDIT

2. (F5) Initialize waveform

Summary: You can initialize the currently selected waveform using this function.

Procedure:

From: Waveform edit

Press: F7 (Name)

To execute: initialization press YES.

To quit: without initializing press NO.

INITIALIZE WAVEFORM

ARE YOU SURE ?

(Yes or No)

This function initializes all sample assignments for the currently selected waveform.

If you are sure you want to initialize the waveform, press YES.

When a waveform is initialized, it is renamed "INT-WAVE" automatically. Dashes are displayed in the "From" and "To" columns for initialized waveforms when the Waveform edit function is selected.

Remark: This function initializes the settings for a single waveform only. It does not initialize the SY99's internal sample memory. To initialize sample memory, use the initialize sample procedure described in *Sample utility 5.Initialize sample*, page 278.

AWM ELEMENT DATA / WAVEFORM EDIT

2. (F7) Waveform name

Summary: You can use this function to change the names of internal waveforms.

Procedure:

From: Waveform edit

Press: F7 (Name)

Specify: the waveform name.

WAVEFORM NAME

↓
[My Piano]

Clr Uppr Lowr

② ③ ④ ①

- ① Enter an eight-character name for the waveform.
- ② To clear the currently entered name press F1 (Clr).
- ③ To switch to upper-case letters press F2 (Uppr).
- ④ To switch to lower-case letters press F3 (Lowr).

Remarks: Methods of entering character data are explained in *How to enter character data*, page 30.

AWM ELEMENT DATA / WAVEFORM EDIT

2. (F8) Sample assign

Summary: Here you can edit the original key, key range, volume, pitch, and loop settings for each sample.

Procedure:

From: waveform edit

Press: F8 (Smpl)

Select: one of the samples assigned to the currently selected waveform.

Specify: the original, low key, and high key assignments for the selected sample.

To rename: the sample, press F7 (Name).

To edit: the volume, pitch and loop settings for the sample, press F8.

- 5 Loop mode. This parameter will appear on the display only if a "...loop" mode was selected in 4. If "normal" looping is selected, the sample will play back continuously in the specified direction. If "alter" looping is selected, the sample will play back alternately forward and backward.

For a sample of "abc", the loop type and mode settings would produce the following results:

Forward only:	abc
Backward only:	cba
Forward loop (normal):	abcabcabca...
Backward loop (normal):	cbacbacbac...
Forward loop (alter):	abccbaabccba...
Backward loop (alter):	cbaabccbaabc...

- 6 To rename the selected sample press F6 (Name). For details, refer to *How to enter character data*, page 30.
- 7 To return to *Sample assign* to change the key assignments for the currently selected sample, press F7 (Asgn). For details refer to the preceding section, *Sample assign*.
When you finish editing sample data, press EXIT to return to the *Waveform assign* display.

Remark: Some samples loaded from external devices may produce noise when played using the SY99. Try changing the parameter settings of such samples using the external devices.

AWM ELEMENT DATA

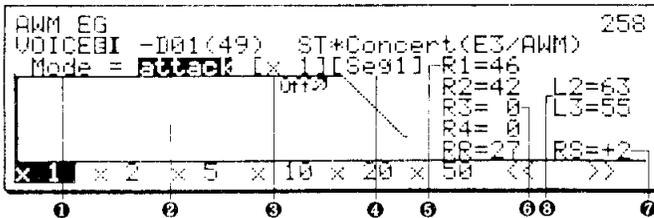
3. AWM EG

JUMP #258

Summary: This determines how the volume of an AWM element will change over time.

Procedure:

- From: AWM Element job directory (JUMP #256)
- Select: job 03:EG. (JUMP #258)
- Specify: volume EG parameters



- 1 EG Mode (Mode = hold, attack): This setting determines whether the first segment of the AWM EG will begin from level 0 (attack mode) or from maximum level (hold mode).
- 2 The AWM EG is graphically displayed.
- 3 This indicates the time range of the EG graphic display; "x1" displays the shortest time with the greatest detail. To change the time range, press F1-F6 (x1, x2, x5, x10, x20, x50).
- 4 This indicates the segment from which the EG is displayed. To begin the graphic display from a different segment, press F7 or F8 to select Seg1...Seg4 or Rell.
- 5 Hold Time or Rate 1 (HT=63...0 or R1=0...63): If the EG Mode is set to "hold" this will deter-

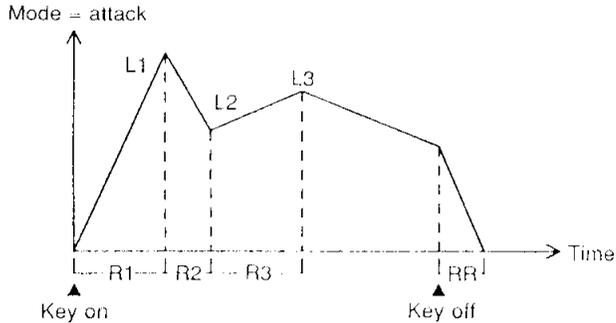
mine the Hold Time for which the level of the waveform is held at maximum. A setting of HT=63 results in the longest time. If the EG Mode is set to "attack" this will determine Rate 1 of the EG. A R1 setting of 63 results in the fastest attack.

- 6 Keyon Rate 2-4, Release Rate (R2-R4 = 0...63, RR = 0...63): These settings determine the speed of the operator EG. Higher values result in faster change.
- 7 Rate Scaling (RS = -7...+7): Rate Scaling allows the operator EG rates to be increased or decreased depending on the key that is played. For positive settings the EG rates will increase as you play higher notes, resulting in shorter envelopes. For negative settings the EG rates will decrease as you play higher notes, resulting in longer envelopes.
- 8 Keyon Levels 2-3 (L2-L3 = 0...63): These determine the levels of the AWM EG. There is no L1 setting since the AWM EG either begins from 0 and moves toward maximum level (in attack mode), or begins at maximum level and stays there until the hold time has elapsed (hold mode). Nor is there a L4 setting since the level of the AWM EG immediately begins to move toward 0 after reaching L3. If you want the sound to continue sustaining as long as you

VOICE EDIT MODE

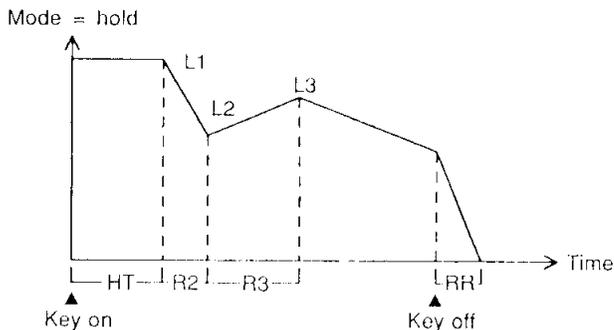
press a key, set R4 to 0. However some AWM waveforms naturally decay to zero, so setting R4 to 0 will not make these waveforms sustain.

Rates and Levels: The AWM EG will function in two ways depending on the Mode setting.



In normal mode the AWM EG level will begin from 0 and rise at the rate of R1 to maximum level. When maximum level is reached it will move at the rate of R2 to level L2. When level L2 is reached it will move at the rate of R3 to level L3. When level L3 is reached it will begin moving at the rate of R4 to 0. (If rate R4 is 0, the sound will move at an infinitely slow rate toward zero; i.e., it will sustain at level L3 as long as the key is pressed.)

When you release the key, the level will move at the rate of RR to a level of 0.



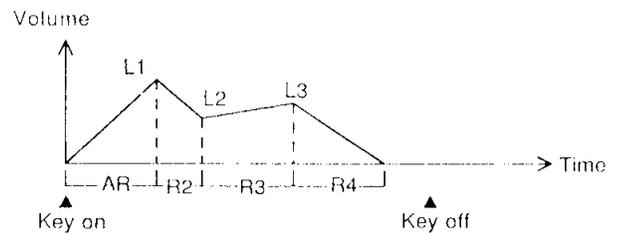
In hold mode the AWM EG level will begin at maximum and stay there for the duration of the specified hold time HT. When the hold time has elapsed, the level will change at the rate of R2 to level L2. The rest of the EG is the same as for normal mode.

Using the AWM EG in hold mode is especially effective when you are using an AWM waveform which includes a definite attack. Keeping the level at maximum for a while allows the natural attack of the AWM sample to be heard. After the natural sampled attack is over the AWM waveform will continue sustaining,

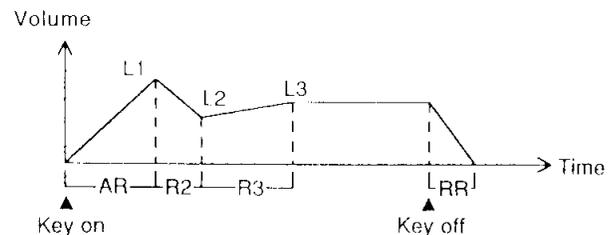
and you can use the remaining AWM EG parameters to create an appropriate decay and release.

Rate 4 and Release Rate: Rate 4 (R4) and Release Rate (RR) can be used in conjunction to create a variety of AWM EG shapes.

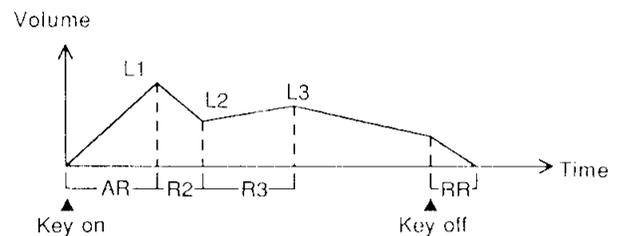
- If R4 is greater than 0 and you continue holding a note, after the level reaches L3 it will decrease at the rate R4 and will move to 0 even though you continue holding the note.



- If R4=0 and you continue holding a note, after the level reaches L3 it will stay at L3 as long as you hold the note. When you release the note, the level will decrease at the rate of RR to a level of 0.



- After reaching L3 the level will decrease at the rate R4, but when you release the note the level will begin decreasing at the rate RR.



Rate Scaling: On most acoustic instruments, high notes have a naturally shorter attack and decay. This can be simulated by setting rate scaling to a positive value (+1...+7). The following diagram shows how higher notes will have faster rates (shorter EGs). Negative settings will have the opposite effect.



AWM ELEMENT DATA

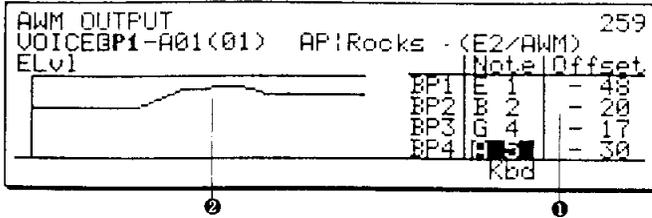
4. AWM output

JUMP #259

Summary: The output level of an AWM element can be adjusted across the keyboard.

Procedure:

- From: AWM Element job directory (JUMP #256)
- Select: job 04:Output. (JUMP #259)
- Specify: the output level scaling



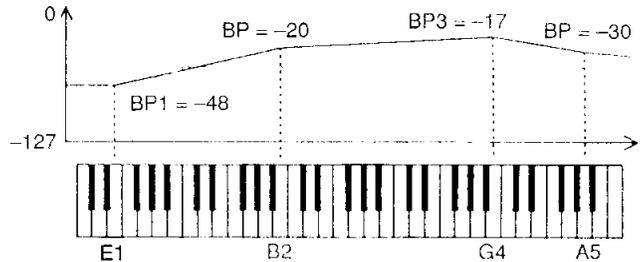
- ❶ Break Point 1-4 (BP1-4): Note (C-2...G8) and Offset (-127...+127) of each break point determine how the level specified in *Voice common data, 1. Element level* (JUMP #202) (page 96) will be adjusted across the keyboard. When the cursor is located at note, you can press F7 (Kbd) and press a key to enter it as the new note setting.
- ❷ The keyboard level scaling is graphically displayed.

Break Point: The AWM output level can be adjusted according to the note that is played. On most acoustic instruments, notes differ in volume and tone depending on the range in which they are played. For example, the low notes of a piano are louder than the high notes.

Use the four break points to specify how the AWM output level will be adjusted across the keyboard. Offset (-127...+127) determines the output level adjustment for each of the four points specified by the note (C-2...G8).

The four note settings must be in ascending order. It is not possible to set a break point to a note lower or higher than the note settings of the neighboring break points.

The following diagram shows how the AWM output level would be adjusted across the keyboard for the settings shown in the above LCD.



Each offset is added to the element level (80 in this example). For example the offset at break point 1 (E1) is -48, so the resulting element level at E1 is 32. The resulting element level is limited to the range of 0...127.

5. AWM sensitivity

Summary: These settings determine how the AWM element will be affected by key-on velocity and by the LFO.

Procedure:

- From: AWM Element job directory (JUMP #256)
- Select: job 05:Sensitiv (JUMP #260)
- Specify: sensitivity to velocity and modulation

AWM SENSITIVITY 260
 VOICEBI -D01(49) ST*Concert(E3/AWM)

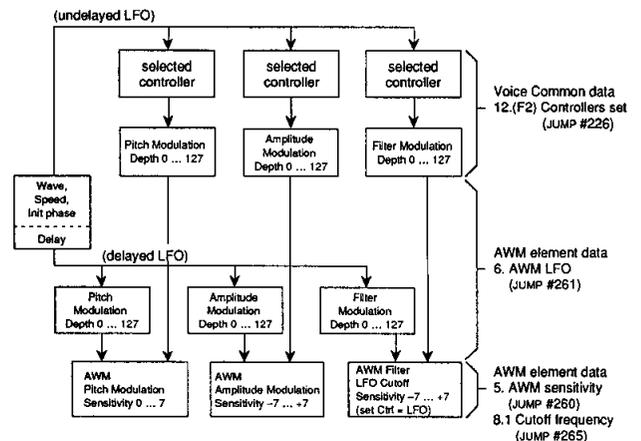
Velocity Sens = **12**
 Rate Vel Switch = on
 Amp Mod Sens = +0
 Pitch Mod Sens = 3

- ① Velocity Sensitivity (-7...+7): This determines how the output level of the AWM element will be affected by key-on velocity. For positive settings (+1...+7) the output level will increase as you play more strongly. For negative settings (-1...-7) the output level will decrease as you play more strongly. For negative settings to have an effect the element level must be lowered.
- ② Rate Velocity Switch (on, off): When the Rate Velocity switch is on, key-on velocity will affect the AWM EG attack rate (R1). The effect will depend on the Velocity Sensitivity setting.
 Velocity = +1...+7: If Rate Velocity is on, strongly played notes will cause the AWM R1 to increase, resulting in a faster attack. For the strongest possible velocity, the EG attack will change at the speed specified by the EG R1 setting.
 Velocity = -1...-7: If Rate Velocity is on, strongly played notes will cause the AWM R1 to decrease, resulting in a slower attack.
 When the Rate Velocity switch is off, the AWM EG attack rate will not be affected by key-on velocity.
- ③ Amplitude Modulation Sensitivity (Amp Mod Sens = -7...+7): Amplitude Modulation Sensitivity determines how greatly the output level of the AWM element will be affected by Amplitude Modulation from the LFO. Increasingly higher positive settings (+1...+7) will allow the LFO to have a greater effect.

Negative settings (-1...-7) are effective only for EG Bias. When Amplitude Modulation Sensitivity is set to a negative value, the controller assigned to EG Bias by the setting in *Voice common data, 12. (F4) Controller set* (JUMP #228) (page 130) will decrease the amplitude of the AWM element, and the LFO will have no effect. For example, two AWM elements in a voice might be given opposite Amplitude Modulation Sensitivity settings, so that the controller assigned to *EGbiasDepth* would crossfade between the two elements.

- ④ Pitch Modulation Sensitivity (Pitch Mod Sens = 0...7): Pitch Modulation Sensitivity determines how greatly the pitch of the AWM element will be affected by Pitch Modulation from the LFO.

Amplitude Modulation Sensitivity and Pitch Modulation Sensitivity: These settings determine the *sensitivity* of the AWM element to the Amplitude Modulation Depth (AMD) and/or Pitch Modulation Depth (PMD) produced by the AWM element LFO. If the LFO settings for AMD and/or PMD are set to 0, these settings will have no effect.



6. AWM LFO

JUMP #261

Summary: The AWM element LFO creates a cyclically changing control signal that can be used to create tremolo (Amplitude modulation), vibrato (pitch modulation), and wah-wah (filter modulation).

Procedure:

From: AWM Element job directory (JUMP #256)
 Select: job 06:LFO (JUMP #261)
 Specify: the LFO parameters

AWM LFO		VOICE#1 -D01(49) ST*Concert(E3/AWM)		261	
Wave =	triangle	A Mod Depth =	0		
Speed =	58	P Mod Depth =	12		
Delay =	46	F Mod Depth =	0		
		Init Phase =	0		

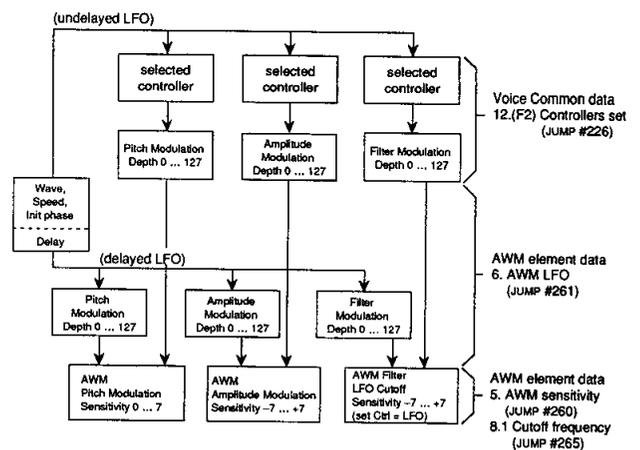
③ ② ① ④ ⑤ ⑥ ⑦

- ① Wave (triangle, saw down, saw up, square, sine, sample&hold): This selects the wave (shape of modulation) produced by the AWM LFO. The selected wave is graphically displayed in the LCD. When sample&hold is selected, the LFO will produce a control signal whose level will change randomly at intervals of time determined by the Speed setting.
- ② Speed (0...99): The speed of the LFO modulation. Higher settings result in faster modulation. At a setting of 0, one cycle will be approximately 34 seconds.
- ③ Delay (0...99): The time delay before the LFO modulation begins.
- ④ Amplitude Modulation Depth (0...127): This determines how greatly the LFO will affect the output level (amplitude) of the operators.
- ⑤ Pitch Modulation Depth (0...127): This determines how greatly the LFO will affect the pitch of the operators.
- ⑥ Filter Modulation Depth (0...127): This determines how greatly the LFO will affect the cutoff frequency of the filter.
- ⑦ Initial Phase (0...99): This determines the point of the LFO waveform from which the LFO will start each time a key is pressed.

Wave, Speed, Delay, Initial Phase: Detailed explanations and diagrams of these parameters are given in *AFM element job 6.1 LFO (Main)* (page 146).

Amplitude Modulation Depth and Pitch Modulation Depth: For these settings to have an effect, the AModSens (amplitude modulation sensitivity) or PModSens (pitch modulation sensitivity) of the AWM element must be set above 0. Make these settings in *AWM element job 5. AWM sensitivity* (JUMP #260).

Filter Modulation Depth: For this setting to have an effect, the Ctrl setting of a filter must be set to "LFO", and the LFO Cutoff Sens setting must not be 0. Make these settings in *AWM element data, 8.1 Cutoff frequency* (JUMP #265).

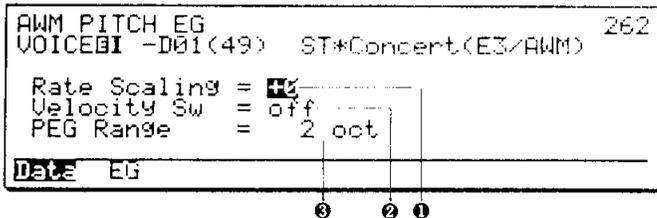


7. (F1) AWM pitch EG (Data)

Summary: The pitch change over time created by the pitch EG can be affected by key-on velocity and the speed of pitch change can be adjusted across the keyboard. To set the shape of the pitch EG, see 7. (F2) AWM pitch EG (EG).

Procedure:

- From: AWM Element job directory (JUMP #256)
- Select: job 07:PitchEG and press (JUMP #262)
- F1 (Data)
- Specify: pitch EG scaling, velocity, and range



① Rate Scaling (-7...+7): Pitch EG Rate Scaling determines how pitch EG rates will change according to the note played. When this is set to +1...+7, the pitch EG will be faster for higher notes. When this is set to -1...-7, the pitch EG will be slower for higher notes. When this is set to 0, the pitch EG will be the same rate for all notes.

- ② Velocity Sw (off, on): When this is on, strongly played notes will change in pitch more than softly played notes.
- ③ Range (1/2 oct, 1 oct, 2 oct): This determines the maximum range of the AWM pitch EG, from 1/2 octave to 2 octaves. (Note that the 8 octave range of the AFM pitch EG is not available for the AWM pitch EG.)

Rate Scaling: This setting determines how Pitch EG Rates (the speed of pitch change) will be affected by the key number of each note. The following diagram shows the result when Pitch EG Rate Scaling is set to +7. Notice that high notes have a shorter pitch EG (faster EG rates) than lower notes.



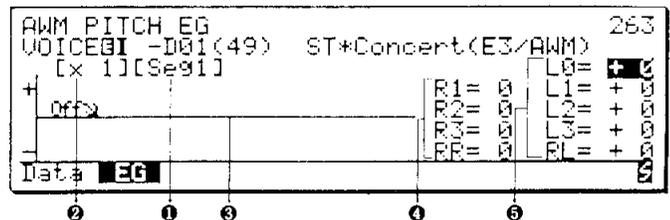
Velocity Sw (velocity switch): When this is on, strongly played notes will change in pitch more than softly played notes.

7. (F2) AWM pitch EG (EG settings)

Summary: The pitch EG creates a fixed shape of pitch change over time for each note. To adjust speed of pitch change across the keyboard, see 7. (F1) AWM pitch EG (Data).

Procedure:

- From: AWM Element job directory (JUMP #256)
- Select: job 07:PitchEG and press (JUMP #263)
- F2 (EG)
- Specify: pitch EG parameters

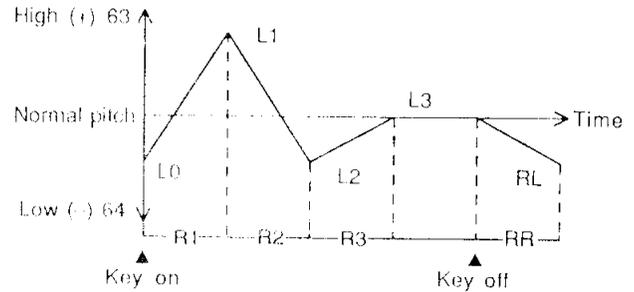


① This indicates the time length shown by the graphic display. To change this, hold SHIFT and press F1-F6 (x1, x2, x5, x10, x20, x50). The exact length of time will depend on the range. When the pitch EG range is 1 octave, the graphic display will cover approximately 0.5 seconds at "x1" and approximately 5 seconds at "x50".

- ② This indicates the EG segment (“seg1–3” or “rel1”) from which the pitch EG graphic display begins. If the EG is too long to be fully shown in the LCD, hold SHIFT and press F7 or F8 to change the segment from which the display begins.
- ③ The pitch EG is graphically displayed.
- ④ R1–R3, RR1 (0...63): Keyon Rates 1–3 and the Release Rate determine the speed of the pitch EG. Higher settings result in faster change. A rate of 63 will jump immediately to the following level.
- ⑤ L0–L3, RL (–64...+63): Keyon Levels 0–3 and the Release Level determine the levels of the pitch EG. Positive settings raise the pitch and negative settings lower the pitch.

Rates and Levels: When you press a key, the pitch will begin at the level of L0, and will change at the rate of R1 to level L1. When the level reaches L1, the pitch will change at the rate of R2 to the level of L2. When the pitch reaches L2, it will change at the rate of R3 to the level of L3 and will stay at L3 as long as the key is pressed.

When the key is released, the pitch will change at the rate of RR to the level of RL.



Note: Even if the AWM pitch EG and the AFM pitch EG have identical Rate settings, there will be slight differences in the timing of the pitch change.

AWM ELEMENT DATA

8. AWM filter

JUMP #264

Summary: The two filters of each element can be used to control the tone in various ways.

Procedure:

- From: AWM Element job directory (JUMP #256)
- Select: job 08:Filter (JUMP #264)
- Specify: the desired filter edit job and press ENTER

```
AWM FILTER                               264
VOICEBI -D01(49) ST*Concert(E3/AWM)
01:Cutoff Frequency-
02:Cutoff Scaling
03:Cutoff EG
01 02 03
```

- ① Move the cursor in this area to select a job and press ENTER to move to the selected job.

- 01: Cutoff Frequency: Make overall settings for the filters. (JUMP #265)
- 02: Cutoff Scaling: Specify how each filter will be adjusted across the keyboard. (JUMP #266, #267)
- 03: Cutoff EG: Specify how each filter will change over time. (JUMP #268, #269, #270, #271)

- ② Pressing F1–F3 will select the corresponding job.

Note: Filter settings for an AWM element are exactly the same as for an AFM element. For details, refer to AFM element job 8. AFM filter, on pages 149–153.

AWM ELEMENT DATA

15. Initialize AWM element

Summary: Initialize the AWM Element data being edited to a set of basic values.

Procedure:

From: AWM Element job (JUMP #256)

Select: job 15:Initz

To execute: the initialize operation press YES

To quit: without initializing press NO or EXIT

INITIALIZE AWM ELEMENT

ARE YOU SURE ?

(Yes or No)

This function sets all AWM element data values to the minimum or simplest possible settings. When creating your own new voices it is usually best to begin by editing an existing voice. However if you want to start from scratch, it is often useful to start from an initialized setting rather than having to re-set all the parameters.

If you are sure you want to initialize the AWM element data, press YES. The AWM element data being edited will be set to the values shown below. If you decide not to initialize, press NO.

This function initializes only AWM element data. Other initialize functions are provided for initializing Voice Common data and AFM element data. Refer to *Voice Common 15. Initialize* (page 132) and *AFM Element 15. Initialize* (page 154).

Initialized settings for AWM Element data

02 AWM Waveform Select

Waveform = Preset 65
(triangle wave)

Frequency Mode = normal

Fixed Mode Note # = C3

Frequency Fine = ±0

03 AWM Amplitude EG

Mode = normal

Keyon Rates 1, 2, 3 = 63

Keyon Rate 4 = 0

Keyoff Rate 1 = 63

Rate Scaling = ±0

Keyon Level 2, 3 = 63

04 AWM Output

Break Point 1 Note = C1

Break Point 2 Note = G2

Break Point 3 Note = E4

Break Point 4 Note = C6

BP1-4 Offset = ±0

05 AWM Sensitivity

Velocity Sens = ±0

Rate Velocity Switch = off

AMS = 0

PMS = 3

06 AWM LFO

Wave = Triangle

Speed = 65

Delay Time = 0

AMD, PMD, FMD = 0

Init Phase = 0

07 AWM Pitch EG

Rate Scaling = ±0

Velocity Switch = off

Range = 2 octaves

Keyon Rates 1-3 = 63

Keyon Levels 0-3 = ±0

Keyoff Rate 1 = 63

Keyoff Level 1 = ±0

08 AWM Filter

Resonance = 0

Cutoff Mod Sens = ±0

Keyon Velocity Sens = ±0

*** following data is same for both filters ***

Filter Type = thru

Filter Control = LFO

Cutoff Frequency = 127

Break Point 1 Note = C1

Break Point 2 Note = G2

Break Point 3 Note = E4

Break Point 4 Note = C6

Break Point Offset = 0 (BP 1-4)

Keyon Rates 1-4 = 63

Keyoff Rates 1-2 = 63

Rate Scaling = ±0

Keyon Levels 0-4 = ±0

Keyoff Levels 1-2 = ±0

AWM ELEMENT DATA

16. Recall voice

Summary: Recall all data of the previously edited voice.

Procedure:

From: AWM Element job (JUMP #256)
directory

Select: job 16:Recall

To execute: the recall operation press YES

To quit: without executing press NO or EXIT.

Note: This operation recalls all voice data, not just AWM element data, and is also available while editing Common data, AFM element data, or Drum Set data. For details refer to Voice common data, 16. Recall voice.

Drum set data

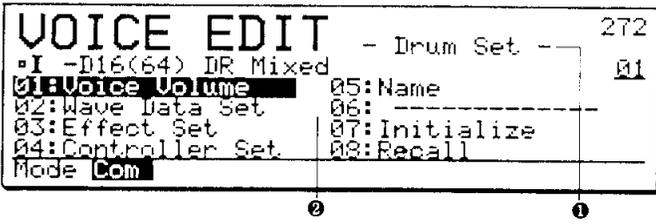
DRUM SET DATA JUMP #272

Drum set job directory

Summary: This job directory shows the jobs containing data for a drum voice.

Procedure:

- From: voice edit mode (JUMP #200)
- When: the Voice Mode is set to 11:Drum Set
- Select: the drum set job directory (JUMP #272)
- F2 (Com)



- ❶ This area indicates that “Drum Set” is the selected voice mode.
- ❷ Move the cursor in this area to select a job and press ENTER to go to the selected job.

- 01: Voice Volume: Adjust the overall volume of the entire drum voice.
- 02: Wave Data Set: Select a waveform for each key of the SY99’s 76-note keyboard, and specify tuning and pan for each.
- 03: Effect Set: Specify how the effect units are connected, how the sound from each note of the drum voice is sent to the effect units, parameter settings for each effect unit, and how the effect parameters are controlled in realtime.
- 04: Controller Set: The overall volume of a drum set voice can be adjusted using a specified controller.
- 05: Name: Specify a ten-character name for the voice being edited.
- 07: Initialize: Initialize the drum set data being edited to the basic or minimum settings.
- 08: Recall: Recall the previously edited voice into the editing buffer.

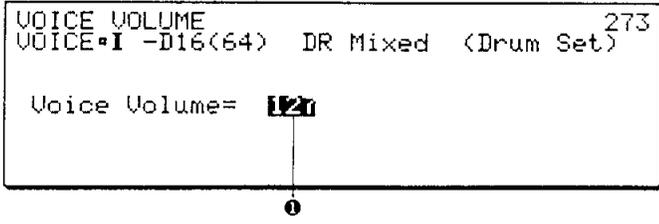
DRUM SET DATA JUMP #273

1. Voice volume

Summary: Adjust the overall volume of the entire drum voice.

Procedure:

- From: drum set job directory (JUMP #272)
- Select: 01:Voice Volume (JUMP #273)
- Specify: the volume of the entire drum set



- ❶ Voice Volume (0...127): This determines the overall volume of the entire drum voice.

DRUM SET DATA

2. Wave data set

JUMP #274

Summary: Select a waveform for each key of the SY99's 76-note keyboard, and specify tuning and pan for each.

Procedure:

From: drum set job directory (JUMP #272)
 Select: 02:Wave Data Set (JUMP #274)
 Specify: parameters for each key note number

```

WAVE DATA SET                               274
VOICE=I -D15(64) DR Mixed (Drum Set)
Key Note Number : E 0
Waveform       = Preset1 [08] rom 5
Level          = 120   Note Shift = -2
Alternate      = off   Fine Tune  = +24
Output Group   = both  Static Pan = +30
K-Dn K-Up Pres1 Pres2 Card Intr
  
```

- ① Key Note Number (E0...G6): Press a key on the SY99 keyboard to select a key note number, and it will be displayed here. (It is not possible to move the cursor here.) You can also press F1 (K-Dn) or F2 (K-Up) to select key note numbers.
- ② Waveform (Preset 1: 1...155, Preset 2: 1...112): Select the AWM sample that will be played by the corresponding Key Note Number. A list of the preset waveforms is given in *AWM element data, 2. AWM Waveform Set* (page 157). (Internal and card waveforms may also be selected. The SY99 may contain a maximum of 64 internal waveforms. The number of waveforms available on a waveform card will vary with the card being used.)
- ③ Level (0...127): This determines the volume of the waveform.
- ④ Alternate (on, off): When this is "on" for two or more key note numbers, the last-played key will take priority and the waveform of the previously played key will be turned off.
- ⑤ Output Group (off, grp1, grp2, both): Select the output group from which the waveform selected for this key note number will be output.
- ⑥ Note Shift (-48...+36 in half steps): Adjust the tuning of the waveform in half steps.
- ⑦ Fine Tuning (-64...+63 in units of 1.171875 cents): Adjust the tuning of the waveform in fine steps.
- ⑧ Static Pan (-31...+31 = Left...Right): Specify the stereo position for each key note number.

Use the following two steps to make settings ②–⑧ for each note of the keyboard. Repeat the two steps as necessary.

1. Press a key on the SY99 keyboard to select a key note number. The selected key note number will be displayed in ①.
2. Make settings ②–⑧ for the selected key note number.

Alternate: If two or more waveforms would sound unnatural if they were played at the same time, select alternate "on" for each of these waveforms. For example it is impossible for a real drum set to sound the closed hi-hat and open hi-hat at the same time. By selecting alternate "on" for the two key note numbers that play the closed hi-hat and open hi-hat waveforms, playing the closed hi-hat will make the open hi-hat stop, and vice versa.

There is only one alternate group for the entire drum voice; i.e., it is not possible to specify two or more *pairs* of key note numbers to play alternately. You may select alternate "on" for as many key note numbers as you like but they will all be in the same alternate group, and only one of them will sound at any time.

Static Pan: The stereo position of the waveform played by each key note number is determined by the static pan setting. The "dynamic" pan of AFM or AWM element can be moved over time by an EG or LFO, but the "static" pan for each key note number of a drum voice cannot be moved over time.

When using a drum set voice in a Multi, these static pan settings will be used if the multi static pan is set to "VC" (voice). Refer to *Multi edit, 5. Voice static pan*, page 191.

Output Group: The stereo signal from the static pan of each key note number is sent to output group 1, 2, or both. If output group is set "off" the waveform for that key note number will not be heard. The diagram in the following section 3.2 *Effect send* shows how the Output Group setting is part of the signal flow in a drum voice.

3. Effect set

JUMP #212

Summary: Specify how the effect units are connected, how the sound from each note of the drum voice is sent to the effect units, parameters for each effect unit, and how the effect parameters are controlled in realtime.

Procedure:

From: drum set job directory (JUMP #272)
 Select: job 03:Effect Set (JUMP #212)
 Select: the effect job you wish to edit

```

EFFECT SET                               212
VOICE: I -D16(64) IR Mixed (Drum Set)
01:Effect Mode Select                     01
02:Effect Send
03:Effect 1 Parameter
04:Effect 2 Parameter
05:Effect Control
01 02 03 04 05
  
```

① Move the cursor in this area and press ENTER to select a job.

01: Effect Mode Select: Specify how the two effect units are connected, the effect type, and volume balance. This is exactly the same as for a normal voice. Refer to *Common Data job 10.1 Effect Mode Select*, page 106.

02: Effect Send: Specify how the sound of each element of the voice will be sent to the effect units. Refer to the following section *Drum set data 3.2 Effect Send*.

03: Effect 1 Parameter: Make settings for effect unit 1. This is exactly the same as for a normal voice. Refer to *Common Data job 10.3 Effect 1 Parameter*, page 108.

04: Effect 2 Parameter: Make settings for effect unit 2. These parameters are exactly the same as for Effect 1. Refer to *Common Data job 10.3 Effect 1 Parameter*, page 108.

05: Effect Control: Specify how effect parameters will be affected by control change messages. This is exactly the same as for a normal voice. Refer to *Common Data job 10.5 Effect Control*, page 123.

② Pressing F1–F5 will select the corresponding job.

Note: Drum voice effect settings differ from normal voice effect settings only in job 2. Effect Send. A normal voice allows you to specify the effect send level for each of the one, two, or four elements. A drum voice allows you to specify the effect send level for each of the 76 keys.

3.1 Effect mode select

JUMP #213

Summary: Specify the effect mode to determine how the two effect units are connected, the effect type for each unit, effect output levels, and the volume balance between processed and unprocessed sound.

Procedure:

From: Effect Set job directory (JUMP #212)
 Select: 01:Effect Mode Select (JUMP #213)

Specify: the effect mode, effect types, and effect levels, and wet:dry balance.

Remarks: This function is exactly the same as that described for normal voices. For details, refer to *Common Data job 10.1 Effect Mode Select*, page 106.

3.2 Effect send

Summary: Specify how the sound from each key of the drum voice will be sent to each of the one, two, three, or four effect sends.

Procedure:

- From: Effect Set job directory (JUMP #212)
- When: the voice mode is "Drum"
- Select: job 02:Effect Send (JUMP #220)
- Specify: the effect send settings for each of the 76 keys

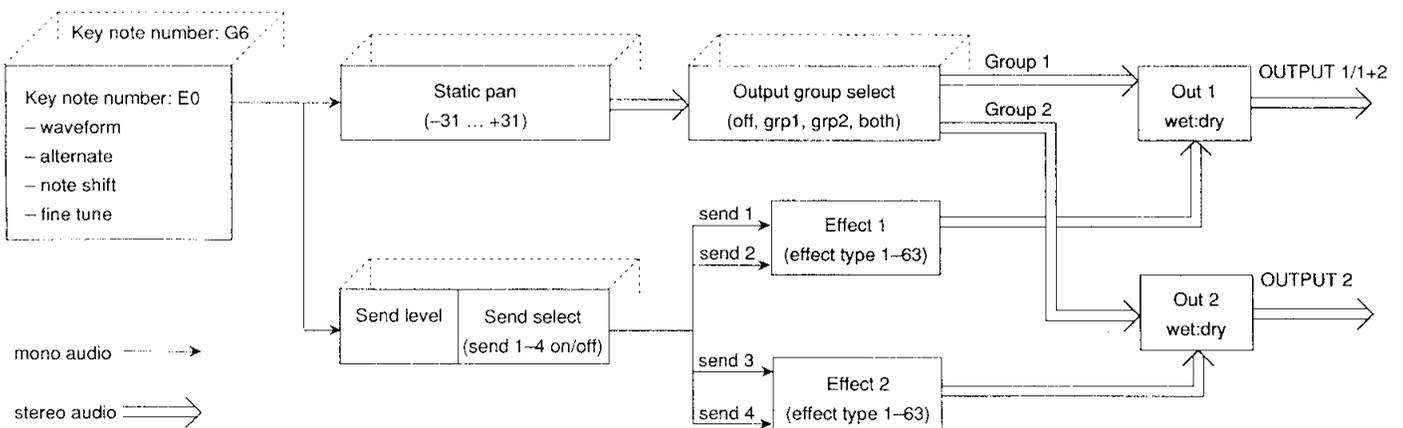
EFFECT SEND	VOICE=I	-D16(64)	DR Mixed	(Drum Set)	220
			Send Sel	Level	Vel Sens
G#5:Guiro 2	1	-	-	110	+3
A 5: Cabasa	1	-	-	110	+3
A#5: Shaker	1	-	-	100	+5
E 5: Whistle	1	-	-	127	+5

- ❶ This area displays the key names (E0–G6), and the waveform assigned to each note. (To set the waveform for each key, see *Drum set data 2.Wave data set*, page 157.) To select a key note number, you can press a key on the SY99 keyboard or press F1 (K-Dn) and F2 (K-Up).
- ❷ Send Sel (send select 1–4): Specify whether or not to send the sound from the key to each effect send 1–4. The number of effect sends available will depend on the effect mode and the effect type. Non-existent sends will be indicated by a dash (–), and cannot be selected. In the above display, effect sends 1 and 2 are available.

If an effect send is turned on, its number will be displayed, and the sound of the waveform assigned to that key will be sent to that effect send. If an effect send is turned off, a period (.) will be displayed, and the sound of the waveform assigned to that key will not be sent to that effect send. In the above display, C3 (SD1) is being sent to effect send 1 only, and C#3 (Tom1) is being sent to effect sends 1 and 3.

- ❸ Level (0–127): Specify how much sound will be sent from the element to the effect unit.
- ❹ Vel Sens (–7...+7): Specify how key velocity will affect the effect send level of the element. For positive settings (+1...+7), more sound will be sent to the effect unit for strongly played notes. For negative settings (–1...–7), less sound will be sent to the effect unit for strongly played notes.
- ❺ You can scroll through the 76 keys E0–G6 by pressing F1 (K-Dn) or F2 (K-Up), or by pressing a key on the SY99 keyboard.

Remark: The following diagram shows the signal flow for a drum voice.



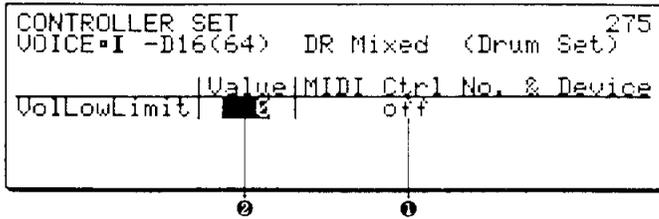
DRUM SET DATA JUMP #275

4. Controller set

Summary: The overall volume of a drum set voice can be adjusted using a specified controller.

Procedure:

- From: drum set job directory (JUMP #272)
- Select: 04:Controller Set (JUMP #275)
- Specify: the minimum level and controller for drum voice volume



- ① Device (MIDI Control #): Select a controller number 1–120 or aftertouch. For example when “001 Modulation” is selected, the MODULATION 1 wheel will regulate the volume of the drum voice. For a detailed explanation of controller numbers, refer to *Voice Common job 12. (F2) Controllers*, page 127.

- ② Vol Low Limit (Value 0...127): This determines the lowest volume that can be set by the selected controller. For example when this is set to 80, the lowest position of the controller will set the volume of the drum voice to 80. When this is set to 0 the lowest position of the controller will reduce the volume of the drum voice to silence. When this is set to 127 the controller will have no effect on the volume.

Remarks: In addition to the controller specified here, the volume of the entire SY99 can always be controlled over its full range by an optional foot controller connected to the rear panel VOLUME jack.

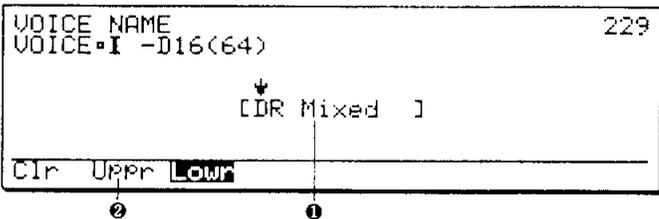
DRUM SET DATA JUMP #229

5. Voice name

Summary: Specify a ten-character name for the voice being edited. In voice play mode this voice name will be displayed in large characters.

Procedure:

- From: drum set job directory (JUMP #272)
- Select: 05:Drum Set Name (JUMP #229)
- Specify: the drum voice name



- ① Enter a ten-character name for the drum voice.
- ② To clear the currently entered name press F1 (Clr). To switch to upper-case characters press F2 (Uppr). To switch to lower-case characters press F3 (Lowr).

For a detailed explanation of how to enter character data, refer to *How to enter character data*.

DRUM SET DATA

7. Initialize voice

Summary: Initialize the drum set data being edited to the basic or minimum settings.

Procedure:

From: drum set job directory (JUMP #271)

Select: 07:Initialize

To execute: the initialize operation press YES

To quit: without executing press NO or EXIT

INITIALIZE VOICE

ARE YOU SURE ?

(Yes or No)

This function sets all drum set data to the basic or minimum settings. If you are creating an entirely new voice, using this initialize function may be faster than resetting all the parameters by hand.

If you are sure you want to initialize the drum set data press YES and the data will be set to the values shown below. If you decide not to initialize, press NO.

This function initializes only Drum Set data. Other initialize functions are provided for initializing Voice Common data, AFM Element data, or AWM Element data.

Initialized settings for Drum Set data

- 01 Voice volume
Voice volume = 127
- 02 Wave data set (for each Key Note Number)
 - Level = 127
 - Waveform = preset 1 86 BD1
 - Note shift = 0
 - Fine tuning = 0
 - Alternate = off
 - Output group = both
 - Static pan = 0
- 03 Effect set
*** same as normal voice initial data except for
Effect Send *** (for each Key Note Number)
 - Effect send sel = all on
 - Effect send level = 127
 - Effect send velocity sense = 0
- 04 Controller set
 - Volume Low Limit = 0
 - Device = 14
- 05 Drum set name
Name = INT VOICE

DRUM SET DATA

8. Recall voice

Summary: Recall all data of the previously edited voice.

Procedure:

From: drum set job directory (JUMP #271)

Select: 08:Recall

To execute: the recall operation press YES

To quit: without executing press NO or EXIT.

RECALL VOICE

ARE YOU SURE ?

(Yes or No)

If after editing a voice you exit voice edit mode without storing, the edited voice data will be lost. In such cases you can use this function to recall the previously edited data into the editing buffer.

Note: This operation recalls voice data, not just Drum Voice data, and is also available while editing Common data, AFM Element data, or AWM Element data. For details refer to Voice Common 16. Recall voice, page 133.

VOICE EDIT MODE