

Juzisound Total SOLO Sampler

Owner's Manual

Version: 15.0

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Important Safety Instructions



WARNING

The lightning flash with arrowhead symbol within an equilateral triangle is intended to alert the user to the presence of uninsulated "dangerous voltage" within the product's enclosure that may be of sufficient magnitude to constitute a risk of electric shock to persons.



CAUTION

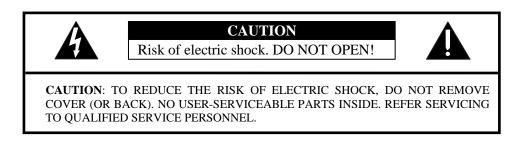
The exclamation point within an equilateral triangle is intended to alert the user to the presence of important operating and maintenance (servicing) instructions in the literature accompanying the product.

INSTRUCTIONS PERTAINING TO A RISK OF FIRE, ELECTRIC SHOCK, OR INJURY TO PERSONS. SAVE THESE INSTRUCTIONS

WARNING - When using electric products, basic precautions should always be followed, including the following:

- Read these instructions.
- Keep these instructions.
- Heed all warnings.
- Follow all instructions.
- Do not use this apparatus near water.
- Clean only with a dry cloth.
- Install in accordance with the manufacturer's instructions.
- Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat.
- Protect the power cord from being walked on or pinched particularly at plugs, convenience receptacles, and the point where they exit from the apparatus.
- Only use attachments/accessories specified by the manufacturer.
- Unplug this apparatus during lightning storms or when unused for long periods of time.
- Refer all servicing to qualified service personnel. Servicing is required when the apparatus has been damaged in any way, such as power-supply cord or plug is damaged, liquid has been spilled or objects have fallen into the apparatus, the apparatus has been exposed to rain or moisture, does not operate normally, or has been dropped.
- Always ensure adequate space for ventilation around the device.
- Do not cover the apparatus with newspapers, tablecloths, curtains, etc., as these will impair ventilation and cause overheating.
- Do not put on the device open flame sources such as candles.
- This unit contains a battery which should not be exposed to excessive heat sources such as sunlight, fire, etc.

Power Supply



- Do not connect this unit to same electrical outlet that is being used by an electrical appliance that is controlled by an inverter or a motor (such as a refrigerator, washing machine, micro-wave oven, or air conditioner). Depending on the way in which the electrical appliance is used, power supply noise may cause this unit to malfunction or may produce audible noise. If it is not practical to use a separate electrical outlet, connect a power supply noise filter between this unit and the electrical outlet.
- To prevent malfunction and equipment failure, always make sure to turn off the power on all your equipment before you make any connections.
- Although the LCD and LEDs are switched off when the unit is turned off, this does not mean that the unit has been completely disconnected from the source of power. If you need to turn off the power completely, first turn off the unit's switch, then unplug the power cord from the power outlet. For this reason, the outlet into which you choose to connect the power cord's plug should be one that is within easy reach and readily accessible.
- Power Switch is only for functional switching of the device. Does not disconnect the device from the mains.
- Due to the risk of electric shock, replacement of fuse and battery are performed only at an authorized dealer of the manufacturer or person with similar qualifications.

Placement

- Using the unit near power amplifiers (or other equipment containing large power transformers) may induce hum. To alleviate the problem, change the orientation of this unit; or move it farther away from the source of interference.
- This device may interfere with radio and television reception. Do not use this device in the vicinity of such receivers.
- Noise may be produced if wireless communications devices, such as cell phones, are operated in the vicinity of this unit. Such noise could occur when receiving or initiating a call, or while conversing. Should you experience such problems, you should relocate such wireless devices so they are at a greater distance from this unit, or switch them off.

- When moved from one location to another where the temperature and/or humidity is very different, water droplets (condensation) may form inside the unit. Damage or malfunction may result if you attempt to use the unit in this condition. Therefore, before using the unit, you must allow it to stand for several hours, until the condensation has completely evaporated.
- When moved from one location to another where the temperature and/or humidity is very different, water droplets (condensation) may form inside the unit. Damage or malfunction may result if you attempt to use the unit in this condition. Therefore, before using the unit, you must allow it to stand for several hours, until the condensation has completely evaporated.
- Do not put anything that contains water on this unit.
- Also, avoid the use of insecticides, perfumes, alcohol, nail polish, spray cans, etc., near the unit. Swiftly wipe away any liquid that spills on the unit using a dry, soft cloth.

Additional Precautions

- Please be aware that the contents of SD card can be irretrievably lost as a result of a malfunction, or the improper operation of the unit. To protect yourself against the risk of losing important data, we recommend that you periodically save a backup copy of SD card.
- Use a reasonable amount of care when using the unit's buttons, sliders, or other controls; and when using its jacks and connectors. Rough handling can lead to malfunctions.
- Never strike or apply strong pressure to the display.
- When disconnecting all cables, grasp the connector itself—never pull on the cable. This way you will avoid causing shorts, or damage to the cable's internal elements.
- A small amount of heat will radiate from the unit during normal operation.
- To avoid disturbing others nearby, try to keep the unit's volume at reasonable levels. You may prefer to use headphones, so you do not need to be concerned about those around you.
- When you need to transport the unit, package it in the box (including padding) that it came in, if possible. Otherwise, you will need to use equivalent packaging materials.

Maintenance

- For everyday cleaning wipe the unit with a soft, dry cloth or one that has been slightly dampened with water. To remove stubborn dirt, use a cloth impregnated with a mild, nonabrasive detergent. Afterwards, be sure to wipe the unit thoroughly with a soft, dry cloth.
- Never use benzine, thinners, alcohol or solvents of any kind, to avoid the possibility of discoloration and/or deformation.

Information for Users on Collection and Disposal of Old Equipment



This symbol on the products, packaging, and/or accompanying documents means that used electrical and electronic products should not be mixed with general household waste.

For proper treatment, recovery and recycling of old products, please take them to applicable collection points, in accordance with your national legislation and the Directives 2002/96/EC.

By disposing of these products correctly, you will help to save valuable resources and prevent any potential negative effects on human health and the environment which could otherwise arise from inappropriate waste handling.

For more information about collection and recycling of old products, please contact your local municipality, your waste disposal service or the point of sale where you purchased the items.

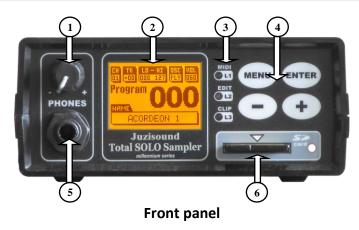
For business users in the European Union

If you wish to discard electrical and electronic equipment, please contact your dealer or supplier for further information.

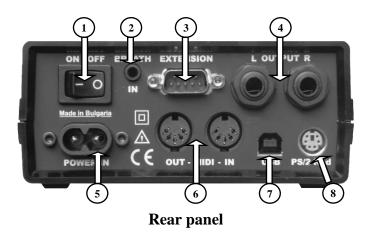
Information on Disposal in other Countries outside the European Union

This symbol is only valid in the European Union. If you wish to discard these items, please contact your local authorities or dealer and ask for the correct method of disposal.

Basic controls and indicators of the sampler



- 1. Potentiometer for adjusting the volume of the headphones and the main outputs.
- 2. LCD.
- 3. LED indicators.
- 4. Menu buttons.
- 5. HEADPHONES output 6.3mm stereo audio jack.
- 6. SD card slot.



- 1. Power switch.
- 2. Analog controller IN usually breath controller.
- 3. Extension port mostly for service functions.
- 4. Main outs -2 separate mono channels -6.3mm mono audio jack.
- 5. Power IN.
- 6. MIDI IN and MIDI OUT.
- 7. Slave USB for connecting to PC.
- 8. To connect PS/2 numeric keyboard numpad.

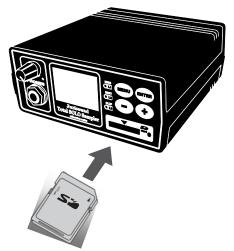
Hardware versions

There are several hardware versions of the sampler:

- Version 1.00 the first one with 2 MIDI INs one real MIDI IN and one from PS/2 because of compatibility to other projects. The MIDI IN was software selectable. DO NOT use PS/2 keyboard with this hardware version of the sampler. This will damage PS/2 keyboard because of high voltage in this port (20V instead of 5V in next versions)!!!
- Version 2.00 this hardware version of the sampler can use PS/2 keyboard. The operating system of the sampler automatically recognizes the hardware version.
- Version 3.00 this is the newest hardware version of the sampler. There is software switch and hardware amplifier for the Analog In.

Hardware Version	rdware Version PS/2 port function Breath In power pin config		Hardware Amplifier
1.00	MIDI In/Out Manual / Jumper		No
2.00	PS2 keyboard	Manual / Jumper	No
3.01	PS2 keyboard	Software / Menu	No
3.10	PS2 keyboard	Manual / Jumper	Yes
3.11	PS2 keyboard	Software / Menu	Yes

Using a SD card



- Carefully insert the SD card all the way in—until it is firmly in place.
- Never touch the terminals of the USB flash drives. Also, avoid getting the terminals dirty.
- SD cards are constructed using precision components; handle the SD cards carefully, paying particular note to the following.
- To prevent damage to the cards from static electricity, be sure to discharge any static electricity from your own body before handling the SD cards.
- Do not touch or allow metal to come into contact with the contact portion of the SD cards.
- Do not bend, drop, or subject SD cards to strong shock or vibration.
- Do not keep SD cards in direct sunlight, in closed vehicles, or other such locations.
- Do not allow SD cards to become wet.
- Do not disassemble or modify the SD cards.

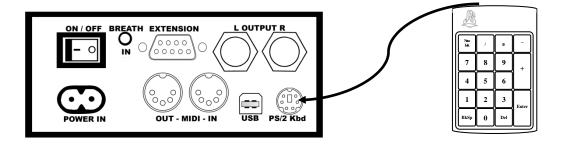
Connecting PS/2 keyboard (numpad)

Warning about using an older hardware version

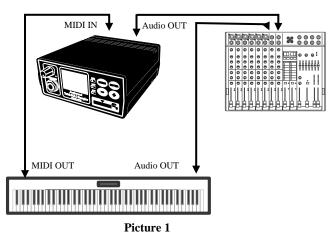


CAUTION! Do not connect PS/2 keyboard to this port, if your sampler's hardware version is **1.00**. This can damage the PS/2 keyboard and/or the sampler.

- With PS/2 numpad (included) you can control the functions of the sampler.
- Use only a numpad purchased or recommended by Juzisound.
- Connect and disconnect the numpad only when the sampler is off.
- When disconnecting the numpad, grasp the connector itself—never pull on the cable. This way you will avoid causing shorts, or damage to the cable's internal elements.



MIDI and AUDIO connections



Connect **MIDI IN** of the sampler to **MIDI OUT** of the synthesizer or MIDI keyboard, which is master to the sampler. You can do this connection by MIDI cable (included). The **MIDI OUT** of the sampler can be configured from MENU:

MIDI connections

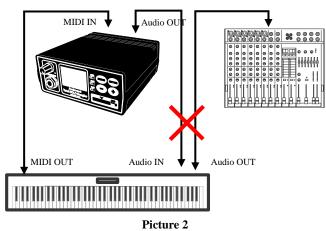
* MAIN MENU *	GLOBAL MENU	MIDI Setup 2/4
Global Setup D	MIDI Setup I	MIDI Out Is.: OUT ThruOutNotes: NO SelitOutNote: LOW TranseLowZn.: +00 TranseHu9hZn: +00 HLimitHi9hZn: OFF

$Menu \rightarrow GLOBAL \rightarrow MIDI$

MIDI OUT Is.:	Function
OFF	No MIDI information going out thru the MIDI OUT
THRU	It simply takes whatever data is received at the input and echoes it back out.
OUT	MIDI OUT - basically output from the breath controller, but can also be used to control the transposing and the scale of an external synthesizer.

AUDIO connections

Always connect the audio output of the sampler DIRECTLY to a mixing console, not through the audio input of the synthesizer.



Shown in Picture 2 audio connections are INCORRECT and leads to a DRASTIC reduction in sound quality. The correct way is shown in picture 1.

Audio Output Settings



The two audio outputs can be configured in the following way:



 $Menu \rightarrow GLOBAL \rightarrow Audio_OUT_Setup$

Smplr OUT: STEREO Mp3 OUT: STEREO	The sampler and the MP3 player to play in both stereo jacks. Then the sound of the sampler comes out of both jack (dual mono), and MP3 player comes out of both jacks (stereo).
Smplr OUT: mono-L Mp3 OUT: mono-R	The sampler outputs only to the L channel and MP3 Player outputs only to the R channel. This allows the output of the sampler and the output of the MP3 player to be connected to two separate channels of the mix- er for different adjustments and effects.

These settings are stored with GLOBAL and apply globally - for all sounds and MP3.



Volume potentiometer is located on the front panel and controls the level of both the headphones and the main audio outputs located on the rear panel. Headphone is designed to work with low-impedance headphones.



CAUTION!

Excessive sound pressure from earphones and headphones can cause hearing loss.

Connecting Expression pedal

- 1. Connect the Expression pedal to a keyboard or a synthesizer. The keyboard should send MIDI Control Change 11, which is standard MIDI message for expression. If your keyboard is a KORG you do not have to change any additional settings on the keyboard.
- 2. For the sampler enter: <u>Menu→GLOBAL→GL.Expres Setup.</u> On the first page, the first parameter **MIDI** set it to **06**.

G1.Ex	prSetup 1/3
	ExprCrv: 06
Brth ModW	/
JoyD PBUe	
PBDw	

3. The third page, the parameter <u>*Gl.MIDI* \rightarrow *Exp*, change it to **YES**.</u>

G1.ExprSetup	3/3
Gl.MIDI->Exp: Gl.Breath->E: ExpChangeSpd:	N0 N0 008

- 4. On the same page check the last parameter <u>*ExpChangeSpd*</u> should be **008**. This parameter sets the speed/smoothness of the pedal response. Should be **008** but if the expression pedal responds slowly set to **016**. Not good to exceed **032**.
- 5. Save the GLOBAL Setup of the sampler penultimate page on the menu GLOBAL.

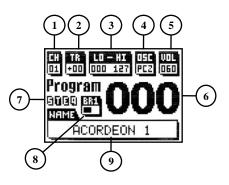
 $\underline{Menu \rightarrow Global \ Setup \rightarrow Save \ Global.}$

Startup screen

Welcome screen - displays the version of the hardware and software. This can be seen later from Menu \rightarrow Device Info.



Main screen



- 1. MIDI channel.
- 2. Transposing of the sampler (global).
- 3. Global limit of the incoming MIDI notes.
- 4. Mode of the oscillators.
- 5. Volume of the current sound.
- 6. Number of the sound. If you change the volume or the transpose of the sound, here will be displayed new value for about 2 seconds.
- 7. Here you can see 4 letters, but they are visible only if the mode that they indicate is ON. Their meaning is as follows:
 - S scale;
 - T terca;
 - E modulating effect;
 - Q equalizer or enhancer;

If the scale or the terca flashes - then they are locked. They can be TURNED OFF and to be still LOCKED. The idea is when they are locked not to change with the change of program. To remain unchanged from the last manual change to the new one.

- 8. Indicates the state of the breath controller (optional). At the top of the indicator is the current preset, and below is the breath pressure. If the lower part depicts the X, this means that the breath controller is in bypass mode.
- 9. The name of the current sound.

Menu structure

You can access the menu by pressing [MENU] on the front panel of the sampler or the combination [Enter + NumLock] from PS/2 keyboard. You can navigate menus by [+], [-] and [Enter]. Pressing [MENU] in any menu level returns back to the home screen. Shortcuts to various menus are available with the combination NumLock + key. If you make any changes in menu GLOBAL, PROGRAM or BREATH, at the end of the menu, there is a page for saving settings. Otherwise changes will be lost when you turn off the power.

MAIN MENU	Global setup	 MIDI Setup MIDI Filter 1 MIDI Filter 1 MIDI PG Setup MIDI PG Table Gl. Modif. Setup Gl. Scala Setup Gl. Expres Setup Other Gl. Setup Effects Setup Audio Out Setup Display Setup Save Global Return to MAIN
	Program Setup	 Oscillator Setup Zone Setup Samples Offset Modificat. Setup Expression Setup Amplifier Setup Filter Setup Scala Setup Terca Setup LFO Setup EffSound Setup EQ Enhancer Setup ModEffectSetup Delay Setup Program Name Save Program Return to MAIN
	Breath Setup	 Analog In Setup Breath MIDI Setup Br. Preset Select -1- Message Setup -2- Message Setup -3- Message Setup -4- Message Setup Breath Vibrato Breath Out Summary Save Breath Return to MAIN
	Device Info	
	License Control	 Enter License Code Check Main Use License Check SAX Use License Return to MAIN
	Util Function	MIDI AnalyzerSysex Buf ViewerReturn to MAIN

Display



 $Menu \rightarrow GLOBAL \rightarrow Display_Setup$

Contrast	0 - 99
Mode of numbering the programs	000-998 or 001-999.
Show errors	Service function

Any changes in these parameters should be saved from page $Menu \rightarrow GLOBAL \rightarrow Save_Global$. Otherwise changes will be lost when you turn off the power.

MIDI channe	
MIDI Setup 1/4	
MIDI Channel: 01 Gl.Transpose: +00	
TranspInNote: YES Zone: >=000-127<=	
SendTrnspS9x: NO SendScalaS9x: NO	

 $Menu \rightarrow GLOBAL \rightarrow MIDI_Setup \rightarrow MIDI Channel$

MIDI messages received on this channel are implemented by the sampler. There are separate MIDI channels for MIDI Notes and for Program Change - i.e. changing the programs can be done from separate MIDI channel.

Velocity Remap

In the menu GLOBAL you can REMAP the velocity of the incoming midi notes according to defined curves and limits. This is very useful. There is also an analyzer of input and output notes velocity. The idea basically is to play the keyboard and by monitoring the values of the analyzer to select such parameters that regardless of the incoming notes velocity to use the full range of MIDI dynamics. The goal is the best possible control over the sampler with respect to notes velocity coming from this MIDI keyboard.

These settings are global for the sampler and can be found on third page of MIDI Setup

MIDI Set	JP	3/	4
VeloCurve	1	IHP Dut	001
	1	CU 01 <	(D1
	···	000	127
~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~~		000	127

 $Menu \rightarrow GLOBAL \rightarrow MIDI_Setup$ 

## **Changing Programs**

#### The Programs can be changed in 3 different ways:

#### By Sampler's buttons

With the buttons on the front panel of the sampler [+] and [-] - move +1 and -1. Pressing [+] and [-] simultaneously sets program to 000.



There is also a variation of program change using the buttons on the front of the sampler. It is used if there is no any other way to change programs - for example a broken PS/2 keyboard or something. By pressing [+] or [-] you can only select the next program without activating it. Activating is done by pressing ENTER. To do this the menu

* MAIN MENU *	global menu	Other GlbStp 1/3
Global Setup D	Other Gl Setup <b>C D</b>	Fast PG Mode: (YES) Auto Volume.: NO SyxSclCorect: NO Scala +00 =>: 000 FilterSpeedF: 016 Trc.UseSNote: NO

 $Menu \rightarrow GLOBAL \rightarrow OtherGL \ Setup \rightarrow Fast \ PG \ MODE$ 

should be changed to NO. Then [+] and [-] only selects the next program. Program number flashes on the screen but the program does not change until you press ENTER.

#### By PS/2 keyboard

By selecting the number from PS/2 keyboard. If you enter three digits the program will be activated immediately after pressing the third digit. If you enter one or two digits should press ENTER to activate the program. If you start entering a number and you have to cancel, press NUMLOCK. There is also locking range of 10 programs. This can be done by pressing [Del]. When the range is locked, the first two numbers are shown enclosed top and bottom on the display, and the press of a single key change programs from 0 to 9 in the current range. This allows for quick-change programs. If the range is locked L2 LEDs lit.

#### **By MIDI**

You can change sampler's programs via MIDI Program Change messages. You can select only 128 of entire 999 sampler's programs.

In sampler's GLOBAL there is a links table. In this table, are stored the 3 possible messages calling program. When the sampler receives Program Change message it checks a links table for

Num Lock	/	*	-
7	8	9	+
4	5	6	
1	2	3	
BkS p	0	Del	Enter

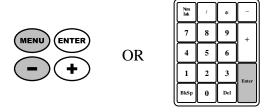
match. If there is a match then sampler loads the program that is found in a table. If there is not a match – there are two possibilities:

- To stay on current program.
- To load a pre-specified program. Usually this program is empty. In other words if you send to sampler unknown Program Change message, then the sampler loads an empty program and keep silence.

By MIDI you can call any sampler's program from any synthesizer's program simply by linking these programs.

#### How to make these links

The sampler always remembers the last received MIDI messages for bank 0, bank32 and Program Change. Therefore, to make a link, you should FIRST send the message from the keyboard which makes the link. Sending MIDI message is always BEFORE making a link. After that press and hold for about 3 seconds the buttons MENU and [-] (or press and hold down for 3 seconds ENTER on PS/2 numpad)



Then on the display starts blinking the number of the program that is linked. Now you can release the buttons. You can get error messages in the following cases:

- This MIDI combination is already linked to other sampler's program.

- This sampler's program is already linked to other MIDI combination.

All 128 links are stored in GLOBAL. Remember to save GLOBAL before turning power off. Table links can be viewed and edited manually.

MIDI	l PG Table	
MIDI: last:	CC01C321PR 0001000100	G
LinkNu	umbr:001/12 ON PRG:00	8
midi: [+]Lin	0001000100	ō.

 $Menu \rightarrow GLOBAL \rightarrow MIDI_PG_Table$ 

- On the upper two rows are the last received MIDI messages for Program Change.
- Next is the number of currently active connection. PLEASE NOTE that the number of the link is not the number of the calling program.
- Next is the link status
  - Active or not active.
  - To which number of program points.
  - With which MIDI messages are called.
- At the bottom there are two functions that are activated when the marker is on them (i.e. blinking) by pressing the [+] button.
  - LINK this function is for making a link between the MIDI messages shown on the second row (last :) and the program shown on the fourth row (PRG:).
- RESET this function RESETS the entire link table and you can start from scratch.

Once again – REMEMBER to save GLOBAL after changing something.

#### **MIDI Program Change Settings**

MIDI PGSetue	1/2
Rx PG Change: MIDI Channel: PG BANK Mode: IfUncownBank: GoTo Program: Mp3 PG Lock.:	

 $Menu \rightarrow GLOBAL \rightarrow MIDI_PG_Setup.$ 

- First is the global status whether to respond to a received MIDI Program Change messages or not.
- Next is the MIDI channel, used for changing programs. It's important that this channel can be different from channel for notes.
- Next is Program Change BANK mode:
  - Only 0.
  - Only 32.
  - Both.
- Next is how to react if unknown Program Change is received:
  - [NN] do nothing.
  - [GP] go to default program, defined in the next parameter
- Next is default program (if unknown message is received).
- With next parameter you can LOCK sampler's programs during playing MP3s. For example, the sampler can play MP3s and at the same time you can play on the external keyboard or synthesizer without interrupting the sampler by MIDI messages.

MIDI PGSetup	2/2
SendPGChange: SendPGChanel:	OFF Ø1
sendrochanel.	61

MIDI Program Change Setup – Second page:

- The first parameter is whether to send messages to change the program from the sampler when the change is made by the sampler itself.
- The second parameter defines the MIDI channel used if the sending is enabled. For this purpose MIDI OUT function defining how to work MIDI output of the sampler should be in position OUT.

*Menu*→*GLOBAL*→*MIDI_Setup*→*MIDI_Out_Is*: OUT

## Save GLOBAL Setup

Enter **MENU** and then enter **GLOBAL**. Scroll to **Save Global**. Enter **Save Global**, change confirmation to **YES** and press button ENTER. Then exit with button MENU.

From front panel: [MENU] > [ENTER] > 12x[+] > [ENTER] > [+] > [ENTER] > [MENU]

From PS/2 numpad: [ENTER + NUM LOCK] > [ENTER] > 12x[+] > [ENTER] > [+] > [ENTER] > [NUM LOCK]

## Transposing

There are two type transposing and they can work simultaneously.

- *Global transposing* for entire module. Can be done with [+] and [–] from PS/2 numpad Global transposing range is +/-24 semitones. For global transpose RESET press [+] and [–] simultaneously. Displayed on the screen for a moment with large numbers but otherwise permanently in the upper left corner of the main screen.
- *Program transposing.* It is in the menu **oscillator** and applies only to this program. It is stored with the program. The sampler can SEND messages to control transpose of external synthesizers. This is useful when you need to transpose multiple synthesizers simultaneously. For this purpose, the output should be set to OUT and the sending should be enabled from the menu GLOBAL.

The sampler may NOT ACTUALLY TRANSPOSE incoming MIDI notes, but to INTERPRET them as transposed (if master keyboard already sends them transposed). This is in order to correctly interpret the scale and terca when there is transpose but not to transpose the notes twice - once from the master keyboard and once more from the sampler. This parameter is located in

*Menu*→*GLOBAL*→*MIDI_Setup*→*TranspInNote:* YES/NO

## Scale

The sampler separately memorizes scale inside each program, plus 3 separate independent global scales. Global scales are stored in menu Global. The scales are fully programmable - each tone can be changed by  $\pm$ -16 degrees. The scale is called with the ENTER button on the front panel of the sampler or with the [/] of PS/2 numpad. NumLock  $\pm$  [/] enter directly into the control menu of the scale.

The scale can be locked and unlocked.

- If IS NOT LOCKED each change of program calls the stored in the program scale.
- If IS LOCKED the type of the scale does not change when changing programs. If the scale is locked the sign "scale" flashes, whether the scale is ON or OFF. When the scale is ON the flashing sign is normal and when the scale is OFF the sign is inverted.

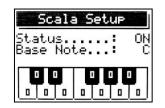
For locking and unlocking press ENTER + [/]. Global scales 1, 2 and 3 can be called at any time with a combination of keys ENTER + [7], [8] or [9]. Calling of the global scale AUTOMATICALLY lock them.

The scale can be controlled with SYSEX messages from KORG keyboard. KORG On the keyboard you can change scales in 3 ways:

- by using STS
- by pressing the key
- by changing numbers on the KORG's display.

The first two ways send SYSEX message, and can be used to modify the scale of the sampler. The third method does not change the scale of the sampler.

The Scale settings are in menu



*Menu*→*Program_Setup*→*Scala_Setup* 

- The status indicates whether the scale is ON when the program is called or should be turned ON additionally.
- Base note shows the basic tone of the scale in which tonality should be played to be altered right tones of the mode.
- At the bottom of the screen shows which tones (relative to base note) have been changed and by how much. The change can be +/- 16. Zero indicates that the specific tone is unchanged.

The scale has SCANNING mode. The idea is as KORG keyboard. This can be done with a key BackSpace (BkSp) from PS/2 numpad and for keyboards that do not have it, can be done with key [.].

- Keep the BackSpace (BkSp) from PS/2 numpad pressed. The keyboard will not play at this time. Press the notes you want to lower a quarter tone. Release the BackSpace.
- Play with your new scale. Notes you pressed on step above are now lowered of a quarter tone. If the BackSpace is pressed and released without playing a note, the scale is turned OFF. Any scanning automatically locks the scales.

If you wish to use the key [.] you should make changes in menu

Global→OtherGobalSetup→ScalaScan[.]: YES/NO

## Terca

**IMPORTANT!** Not every oscillatory mode may use terca!

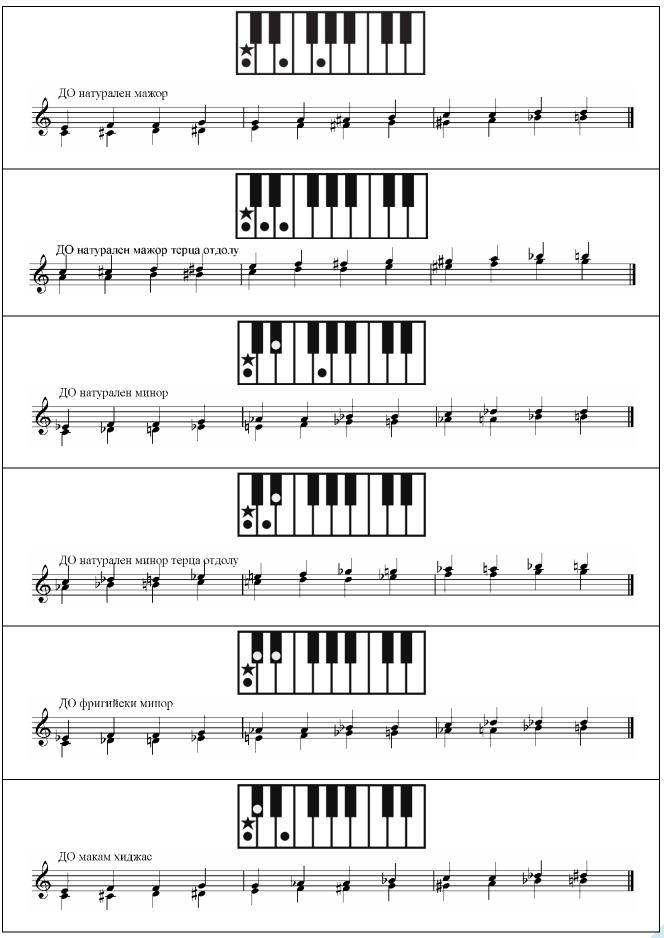
The terca is a doubling of the played tone at a certain interval with respect to set basic tone (tonic) and mode (Latin *modus*). The most common interval is terca up but may also be terca down or octave. Tercas are set by mode and basic tone. By pressing and holding the [*] of PS/2 numpad enters scanning mode for terca (on the display flashes quickly letter [T]). By pressed simultaneously three keys (which exactly is explained below in a table), is set a base note and mode for the terca. These three tones **not heard**. They are just to set basic tone and mode. If the sampler detects a valid combination after the three tones are pressed, the display shows letter [T] and you can play the selected terca.

The terca can be switched on/off and lock/unlock - just like the scale. After scanning the terca locked itself! Button for switching on/off is [*]. Combination for switching lock/unlock is ENTER + [*]. Combination NumLock + [*] enters menu for Terca Setup.



Stat:	Terca status – on/off		
	Terca mode $- 0/1$ . In mode 0 the sampler plays the terca note and the main note. In mode		
Mode:	1 the sampler plays ONLY TERCA NOTE. Mc	ode 1 is available in all modes of oscilla-	
	tors - so the sampler can plays terca tone and the keyboard to play the main tone.		
Vid:	The type of mode that is used for the terca. (See	table below)	
Base:	Basic tone for the terca		
Oct:	Octave for the terca +/- 2 octaves (24 semitones)	).	
Vol:	Volume of the terca with respect to the main voi	ice. (Terca/main voice mix)	
	Special note. It changes only one tone of the te	rca - and only in the octave in which the	
	terca is selected. This tone is different for terca u	up and terca down.	
	For terca up this special note is the fifth only i	in the octave below the terca is selected.	
	The difference: On that tone instead of playing	g terca up as expected, the sampler plays	
	perfect forth up.		
	<b>Example:</b> C Major, terca up. For the lower G	• •	
	phrase (if this function is OFF) the terca tone v	will be B. If this function is ON the terca	
	tone will be C.		
	When terca down mode is selected (i.e. you pl		
	Вие свирите:	note) the special note is the basic note	
		only in the octave below the terca is	
		selected. <b>Example:</b> C Major, terca	
SNt:		down. For C (if this function is OFF) the	
		terca tone will be lower A. If this func-	
	С изключена специална нота звучи така:	tion is ON the terca tone will be lower	
		G.	
		The special note can easily be managed	
		during play with a combination of keys ENTER + [6] and ENTER + [3].	
		ENTER + [6] and $ENTER + [5]$ . ENTER + [6] turns ON the function for	
	С включена специална нота звучи така:	special note. Terca sign becomes a low-	
		er case letter [t].	
	048888	ENTER $+$ [3] turns OFF the function for	
		special note. Terca sign becomes a capi-	
		tal letter [T]. This is on the main screen	
	of the sampler.		
	The zone is used to be able to play terca with an	nother simple, different from the original.	
	If this parameter is ORG the terca plays the same multisamples as the original tone. If the		
	parameter is ZNx (where x is the zone number) the terca plays with the multisample		
7	which is chosen for this zone from the multisample menu. This makes it possible the		
Zna:	multisample for the terca to be different from the multisample for the main voice.		
	The multisample can be set from the menu		
	Menu $\rightarrow$ Program $\rightarrow$ Zone Setup		
Ms:	Reserved		

Terca scanning table



## **Program volume**

Each program has its own volume which is stored along with it. It can easily and quickly be changed during playback by pressing a combination NumLock + [+] and NumLock + [-].The change appears on the screen in large digits. On main screen (if the volume of the program was changed from the original) volume parameter start flashes. To save the changed volume you have to save the program. This is most easily done with a shortcut NumLock + ENTER.



There is PEAK indicator - the lower LED on the front panel. During normal operation this peak indicator should NOT BE constantly ON. If the indicator flashes frequently (e.g. when playing chords) volume should be reduced slightly until the indicator stops

flashes or flashes very rarely. Thereafter you need to align the other programs with respect to that one.

There is also an AUTOMATIC volume reduction. The idea is the sampler automatically to reduce volume until there is no audio clip. This function is available from the menu

 $Menu \rightarrow GLOBAL \rightarrow Other_Gl_Setup \rightarrow Auto_Volume$ 

If this function is used, and it has changed the volume of the program, you have to save the program to save the changes.

## **Pitch Bend range (global)**

Each program has its own parameters for Pitch Bend range. They can be +/-2 or +/-1. These parameters are individual for each program and are saved with it. Very often in practice a keyboard player works with the same PB range for all programs. For this purpose, there is a possibility that PB range can be set globally. The global PB range (if enabled) HAS PRIORITY over the individual program PB range. The global PB range is available from menu

Other	GlbStp	2/3
Terca PB Up	Unlock: Fix To:	NO
PB Dou	nFixTo: Parser:	ÖFF
Scalas	can[.]: D to 4:	NŎ

 $Menu \rightarrow GLOBAL \rightarrow Other_Gl_Setup$ 

Second page, parameters *PB Up Fix to:* and *PB DownFixTo:*. Values can be +/-1 or +/-2. These parameters are saved with GLOBAL.

**IMPORTANT!** When changing the PB range activation occurs at the next change of program. The PB range of the currently selected program will not change until the next change of program.

## **Ribbon Controller – fixing of Control Change 18**

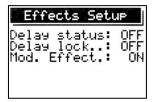


 $Menu \rightarrow GLOBAL \rightarrow Other_Gl_Setup$ 

This parameter GLOBALLY sets received messages from Ribbons controller (Control Change 18) to be interpreted for all programs as Pitch Bend message +/-1 or 2 +/-. Each program has its own parameter for this, but the global one (if enabled) has priority.

## **Effects setup**

Sampler has no REVERB effect, but there are modules DELAY and MODULATION EFFECT. These modules are DISABLED globally by default because is recommended to use external effects from the mixer to obtain the best result. If you need to use internal effects, they should be enabled globally. This is the menu:



 $Menu \rightarrow GLOBAL \rightarrow Effect_Setup$ 

	Enables globally the DELAY module. If it's enabled globally - now each one
Delay Status	program has its own parameters for DELAY. These parameters are saved with
	the program.
	LOCKS the delay parameters at the time of activation. Once they are locked,
	they are no longer change when changing programs. The idea is to be able to
	adjust some delay parameters that are suitable for the current environment, and
	they to remains the same for all sampler's programs. Usually if you use this func-
	tion you should do the following:
Delay Lock	Each time you turn ON the sampler the program with number 000 is loading. The
	DELAY module automatically loads its parameters from program 000. You need
	to set the Delay parameters of program 000 for the specific room and to set the
	Delay Lock to ON. Now every time you start Sampler the Delay module loads its
	parameters from program 000, locks itself and you have the same delay for all
	programs.
	Enables or disables globally the Modulation effect. (Not all oscillator modes sup-
Mod. Effect	port Modulation effect).

#### Program parameters for Delay setup



*Menu*→*Program_Setup*→*Delay_Setup* 

Shortcut for this menu: NumLock + [7]

Delay Mix	Wet/Dry signal mix
Downsampling	Speed range of the delay (fast, medium or slow) by changing the frequency. The slower speed (lower sample rate) cuts the higher frequencies, which in many cases is a desirable effect.
Delay TimeMs	Delay time in milliseconds.
Delay Feedback	Delay Feedback

#### **EQ/Enhancer**

In the sampler there is a module that can be set to one of two effects - EQ or Enhancer. Each program has its own parameters for this module. What is the fundamental difference between EQ and enhancer? The EQ usually divides the entire audio spectrum in to 3 bands (low, middle and high) and combines these bands in different ratios. This is in other words, frequency-specific volume knobs. The ENHANCER on the other hand, passes the entire audio spectrum to the outputs and adds or subtracts only one frequency band. The Enhancer makes certain instrument more prominent, enhance particular aspects of instrument's tone. It is most commonly used for adjusting a high frequency of acoustic instruments.

Enhancer Pros.: The original audio signal passes untouched. The module only adds or sub-tracts specific band.

Enhancer Cons.: Only one frequency band to modify.

In the sampler each program has separate parameters for this module. They are located in the menu

EQ/E	nhancer	• Stp
Statu	s: Equa	alizer
Selit	Freq.	100
Q Fac	tor	18d
LOW	MID	HIGH
+00	+00	+00

 $Menu \rightarrow Program_Setup \rightarrow EQ_Enhanser_Setup$ 

Status: Defines da state of the module:

- DISABLED;
- Enhancer;
- EQ.

Other parameters vary according to the selected mode.

- Enhancer mode:
  - Enhance Band: LOW; MID; HIG frequency bands;
  - <u>Q Factor:</u> 12d; 18d, 24d slope of the filter in dB;
  - <u>Mix Polarity:</u> + or - to add or to subtracts from original signal ;

- <u>Frequency</u>: 000-127 the exactly frequency of the modified band (usually about 090-100);
- Intensity: Intensity of adding or subtracting.
- EQ mode:
  - <u>Split Freq:</u> frequency separation of medium and high frequencies;
  - <u>Q Factor:</u> slope of the filter in dB;
  - <u>Gain</u>- filters gain;

## **Samples Offset**

Each program has its own parameters for this Sample Offset and they are located in the menu

Samples Ofst	1/2	Samples Ofst	2/2
GlobalOffset: AutoMaxSectr: AutoTimeBase: Velo->Offset: Porta->Offst: Terca->GlbOf:	002 0FF 360 0FF 0FF 0FF	MonoRetrnOfs:	OFF

Menu→Program_Setup→Samples_Offset

#### Shortcut: NUMLOCK + [2]

The offset is one of the most important parameters that are typical for the sampler. There is an AUTOMATIC sample offset. For example, sounds with a long attack (such as saxophone) usually does not sound good for short notes because there is no time to develop the attack and to sounds the actual tone. Such clumsy sounds can't handle fast passages accurately. Automatic offset works on the principle of measuring the speed of incoming notes. As the notes are closer to each other, the greater is the offset of the new note. The goal is to skip the attack of the sample where it is not yet stabilized.

stabilizeu.	
Global Offset	Defines global and permanent offset for all samples from a current program.
AutoMaxSector	Defines the value of the greatest automatic offset. Such greatest offset would
Automassector	occur if the note is played IMMEDIATELY after another.
	Defines the maximum time interval that is tracked AFTER a note is played. As
	new note is closer to the old one, so it's offset is closer to the defined in [Au-
AutoTimeBase	toMaxSector]. Accordingly, as the new note is closer to the end of the specified
AutoThileDase	interval, so its offset is less, the decrease is with respect to the defined maxi-
	mum. And lastly - if the time distance between two notes is equal to or greater
	than the specified tracked distance, the auto-offset is NOT done.
Velo→Offset	An extra offset that is applied to the notes with higher velocity. The idea is to
veio→Offset	omit part of a sample attack for notes with high velocity.
The	e meaning of the following parameters varying in different modes.
Dorto Offect	When in MONO program with portamento we have a transition from tone to
Porta→Offset	tone BY PORTAMENTO, the new sample is played with a specified offset.
Lagata Officiat	When in MONO program with or without portamento we have a transition from
Legato→Offset	tone to tone BY LEGATO, the new sample is played with a specified offset.
	When in MONO program we have a transition from tone to tone with porta-
MonoRetrnOfs	mento or legato, and we return to the first tone, the last sample is played with a
	specified offset.

## **Amplifier Setup**

Here are defined volume, dynamics, attack and release of the program.

Amplifie	er Setup
Program /	vol.: 060
Offerla	sen: 000 : 0000 ms
Release. JS2 Lega	0015 ms toXf: 010
Release	Mode: LOG

*Menu*→*Program_Setup*→*Amplifier_Setup* 

#### Shortcut: NUMLOCK + [5]

Program Volume	Volume of the program.		
Velocity Sen	This is the dynamics of the program (velocity sensitivity). Value 000 meaning		
velocity Self	no dynamics at all, value 127 is the greatest dynamics.		
Attack	Attack in milliseconds – from 0000 to 3999		
Release	Release in milliseconds – from 0000 to 3999		
	This parameter applies only to the oscillatory mode JS2. Mean mixing time		
JS2 LegatoXf	(crossfading) between the samples of two tones in milliseconds when doing		
	legato.		
	Release Mode:		
	- LOG – logarithmic - closer to the acoustic nature of the sound, but		
Release Mode	very difficult to calculate. At low values is indistinguishable from		
	the linear release. LOG is useful for sounds with short release such		
	as all SOLO sounds.		
	- LIN – linear release – easy to calculate. LIN is useful for sounds		
	with long release such as pianos, guitars and more.		

## **Oscillators Setup**

Oscilator 1/6	Oscilator 2/6	Oscilator 3/6
Porta speed.: 032 Velo->Porta.: OFF Note->Porta.: OFF Velo->Legato: OFF	UseLastVelo.: NO Mono HOLD: OFF JS2 FadeCoef: 00%	Transpose: -12 PB Up Range.: +02 PB DownRange: -02 Ribbon PB Up: OFF Ribbon PB Dw: OFF
Oscilator 4/6	Oscilator 5/6	Oscilator 6/6
Mod1+2=JBack: NO LegatoIfIntr: OFF LegatoIfIntr: OFF MonoRetrnLEG: NO	GrrExprTresh: OFF GrrExprDepth: OFF GrrVeloTresh: OFF GrrVeloDepth: OFF	Octaver Stat: OFF Oct Filter: RES Oct Cuttof: 000 Oct Rsonance: 000 Oct 1 Volume: 000 Oct 2 Volume: 000

*Menu*→*Program_Setup*→*Oscillator_Setup* 

#### Shortcut: NUMLOCK + [0].

Here are the most important parameters for a program. They also vary in different modes - so we will focus only on the most important:

ModeThe operating mode of the program - polyphonic, monophonic, etc. This is the most important parameter for a program. It defines several things: 
Porta speed       - For what kind of instrument shall be used current program;         - Be polyphonic or monophonic;       - To have portamento or not;         - To have portamento or not;       - If there is portamento what type to be, etc.         Possible modes of oscillators are:       -         - Normal POLY       -         - Bass MODE 1       -         - Mono RETRIG       -         - Mono+PRT CZ       -         - Mono+PRT DX       -         - V. Bouzouki       -         - LPC Gibson       -         - Bass MODE 2       -         - Mono+PRT KG       -         - Mono+PRT JS       -         - StradivariV       -         - Poly 4 NOTE       -         - Strad Vio 2       -         - JS Alt Sax       -         - JS Bouzouki       -         Porta speed       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
Porta speed       - Be polyphonic or monophonic;         - To have portamento or not;       - To have portamento what type to be, etc.         Possible modes of oscillators are:       - Normal POLY         - Bass MODE 1       - Mono RETRIG         - Mono LEGATO       - Mono+PRT CZ         - Mono+PRT DX       - V. Bouzouki         - LPC Gibson       - Bass MODE 2         - Mono+PRT KG       - Mono+PRT JS         - Mono+PRT JS       - MonoPRT JS2         - StradivariV       - Poly 4 NOTE         - JS Clarinet       - JS Clarinet         - JS Bouzouki       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
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-Mono+PRT KG-Mono+PRT JS-MonoPRT JS2-StradivariV-Poly 4 NOTE-Strad Vio 2-JS TenorSax-JS Alt Sax-JS Clarinet-JS BouzoukiPorta speedDefines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
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-StradivariV-Poly 4 NOTE-Strad Vio 2-JS TenorSax-JS Alt Sax-JS Clarinet-JS BouzoukiPorta speedDefines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
-       Poly 4 NOTE         -       Strad Vio 2         -       JS TenorSax         -       JS Alt Sax         -       JS Clarinet         -       JS Bouzouki         Porta speed       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
-       Strad Vio 2         -       JS TenorSax         -       JS Alt Sax         -       JS Clarinet         -       JS Bouzouki         Porta speed       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
-       JS TenorSax         -       JS Alt Sax         -       JS Clarinet         -       JS Bouzouki         Porta speed       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
-       JS Alt Sax         -       JS Clarinet         -       JS Bouzouki         Porta speed       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
- JS Clarinet         - JS Bouzouki         Porta speed         Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
- JS Bouzouki         Porta speed       Defines the MAIN portamento time. In some cases, this is the shortest time of the portamento, which then can be increased under the influence of other parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
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Porta speed       parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
parameters. In certain cases, it is best that parameter to be considered as a BASE time of the portamento.         Defines how much to increase portamento time for low velocity. High velocities have portamento time, which is defined in parameter [Porta speed]. The
Defines how much to increase portamento time for low velocity. High veloc- ities have portamento time, which is defined in parameter [Porta speed]. The
ities have portamento time, which is defined in parameter [Porta speed]. The
larger is the value of parameter [Velo $\rightarrow$ Porta], the bigger is the portamento
Velo→Porta time for low velocities relative to high velocities. Note that portamento time
can only INCREASE, not decrease. If the value of parameter [Velo $\rightarrow$ Porta]
is 000 the portamento time will be EQUAL to [Porta speed] in entire range
of velocities.
Determine how much to increase portamento time when the interval between
the notes is bigger. This parameter is used if you want portamento time to
Note $\rightarrow$ Porta depend of the interval between the notes. Usually this parameter is set to
Defines at what velocity values of a new note in mono program, the new
$Velo \rightarrow Legato$ note is played with legato, not portamento. In other words, over what veloci-
ty value to disable portamento and switch to legato.
Defines the smallest value of the portamento. If for some reason portamento
Port-Legato must be less than the set value - automatically switching to legato. Because
at very short values portamento does not sounds good and is better to do
transition between tones with legato.
Aft $\rightarrow$ PitchAfterTouch to Pitch conversion. Can be +/-1 or +/-2.
For MONO sounds, when we return from MONO situation, two velocity
Use LastVelo values can be used for calculating the portamento time – the velocity of the
NOTE OFF message or the velocity of the last received NOTE ON message.

	<u> </u>
	If your MIDI keyboard sends velocity values for NOTE OFF messages you should set this parameter to [NO] otherwise set this parameter to [YES]. Regular MIDI keyboards does not send NOTE OFF velocity, therefore, the default value for this parameter is [YES]. When return from MONO porta- mento the velocity of the last received note is used.
Mono HOLD	If this parameter is enabled, the last released tone does not stop sounds. This makes it easier playing strings instruments in mono mode.
JS2 Fade coefficient	Defines the fading between tones in a moment of portamento, when oscilla- tor mode is JS2. The higher the value of the parameter, the fading between the tones during the portamento is greater. Used for "automatic" separation of tones.
Porta VCMode	Defines what curve to be used to calculate the portamento time in relation with note velocity. The curve can be linear, logarithmic or as in saxophone modes.
JS1 XFadeMode	Applies to portamento mode JS (JS1). If it is on, the portamento between tones is highlighted as during portamento the sound is slightly boosted.
Transpose	Transpose that is applied only to this program.
PB Up Range	Pitch Bend Up range
PB Down Range	Pitch Bend Down range
Ribbon Up Range	Ribbon to Pitch Up range
Ribbon Down Range	Ribbon to Pitch Down range
Mono Ret Legato	Defines when returning from mono how to get back - with portamento or legato. If this parameter is enabled obligatory come back with legato. Go to the new note with portamento, but always go back to the old note with lega- to. This is typical for wind instruments with reed.
Mono Priority	MONO priority - lowest note, highest note or the last note. It is normal to use the last note.

## **Zone Setup**

Zone Setur	
Zn: Mz1 MSmp:0001 ACORDEON 1	1
Transpose: +00	
Volume: 127 Velo : 000 to 127	
Key: 000 to 127	1

*Menu*→*Program_Setup*→*Zone_Setup* 

Shortcut: NUMLOCK + [1]

This menu defines which multisamples to which zones will be located and optionally with what parameters. Note that we always have ONLY ONE selected area and all parameters in the edit window are for this SELECTED area.

Each sampler program can use up to 14 zones simultaneously. In some modes multisamples location in the zones is specified, in other modes is not specified (free).

The zones names:

Mz1Mz8	(Main Zone)– from 1 to 8. Here are usually the VELO zones.
Lzn	LEGATO zone
Rzn	RELEASE zone
Md1Md4	Additional four areas – for effects, modulators, etc.

The parameter [Zn:] is to select the zone. Thereafter, all displayed parameters are for this zone only.

MSmp	This is the multisample which is defined for that zone. Below it is the name of the multisample.
Transpose	e This transpose is only for that zone - it is not currently active.
Volume	This is the volume of that zone.
Velo	Note velocity – from/to – where the zone is active.
Key	These are the keys which works the zone - not valid for now.

## **Expression Setup**

Expression 1/2	Expression 2/2
	FJOY -Y->Exp: OFF FRibbon->Exp: OFF FRibbon->Mde: FUL FRibbon->Mde: FUL FF

 $Menu \rightarrow Program_Setup \rightarrow Expression_Setup$ 

Shortcut: NUMLOCK + [4]

Expression - that is generally the LOUDNESS of the program. It can be controlled (modulated) from multiple locations simultaneously. Within this menu is the matrix that defines which modulator in what proportion and in what direction to affect the expression. It is important to know that all modulators can operate simultaneously, but always apply the most recent one. Therefore it is better to enable only one of them.

Each modulator has a value of -127 OFF +127 setting the direction and intensity.

First parameter [InitialExpr] is the value for an initial EXPRESSION when the program is loaded. If this parameter is OFF, the expression value remains the same as from the last program.

## **Filter Setup**

Filter Setup 1/6	Filter Setup 2/6	Filter Setup 3/6
		NtVelo->Cutf: OFF NtVelo->Curv: 01 JDwCC2->Cutf: OFF JDwCC2->Curv: 01
Real Cutof: 65536	Real Cutof: 65536	Real Cutof: 65536
Filter Setup 4/6	Filter Setup 5/6	Filter Setup 6/6
PB Up->Curve: 01 PB Dwn->Cutf: 0FF PB Dwn->Curv: 01	CC74 Up->Crv: 01 CC74 Dw->Ctf: OFF	Whel->LFO->F: OFF Auto->LFO->F: OFF AftT->LFO->F: OFF LFO>Ctf>Mode: NRM Real Cutof: 65536

*Menu*→*Program_Setup*→*Filter_Setup* 

#### Shortcut: NUMLOCK + [6]

Here you can configure the filter of the program. There are many modulators, but difference is that they all can work together. All modulators together generate a value that is passed to the filter. Each modulator has a direction, intensity, and its own curve that has its influence on the filter. At the bottom is always displayed the resulting cutoff which is shown graphically and numerically.

Parameters:		
Filter Type	The type of the filter. It is not altered by the modulators.	
Cutoff	This is the BASIC cutoff, upon which affect all modulators in a certain direc-	
Cuton	tion. If there are no activated modulators - this would be the actual cutoff.	
Resonance	Resonance - it also is not affected by the modulators and is constant.	
NtVelo→Mode	How the Note Velocity to affect the filter mode - as an absolute value that over-	
INT V EIO→MOUE	rides all modulators - or as a modulator.	
	How CC74 to affect the filter - either directly to the cutoff - or thru the modula-	
CC74→Mode	tor matrix. (* This parameter and the parameter above are reserved for compati-	
	bility with older hardware versions of the sampler)	
NtVelo→Cutf	How the Note Velocity to affect the filter cutoff.	
JDwCC2→Cutf	How the joystick of the keyboard, pulled down to affect the filters cutoff. (On	
JDwCC2→Cuti	KORG he gives CC2).	
PB Up→Cutof	Pitch Bend Up to cutoff.	
PB Dwn→Cutf	Pitch Bend Down to cutoff.	
CC74 Up→Ctf	CC74 Up to cutoff.	
CC74 Dw→Ctf	CC74 Down to cutoff.	
Whel→LFO→F	If there is a vibrato from the modulation wheel, how it affects the filter.	
Auto→LFO→F	If the auto vibrato, which occurs after a specified time period is enabled - how it	
	affects the filter.	
AftT→LFO→F	If the vibrato on AfterTouch is enabled, how it affects the filter.	
	IMPORTANT! There are several ways in which vibrato to affect the filter:	
	- NRM - the filter is moving in the same direction as the vibrato;	
	- INV - the filter is moving in the opposite direction of the vibrato;	
LFO>Ctf>Mode	- + NR - Only up - in the normal phase;	
	- + IR - Only up - in inverted phase;	
	NR - Only down - in the normal phase;	
	IR - Only down - in inverted phase;	

## LFO Setup (Low Frequency Oscillator)

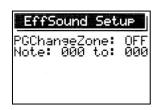
LFO Setup 1/2	LFO Setup 2/	′2
LFO Waveform: ^/ LFO PulseWdt: 064 LFO Speed: 064 Wheel->Pitch: 016 AutoModDelay:0000 AutoModDepth: 0FF	AftrT->Pitch:	OFF

 $Menu \rightarrow Program_Setup \rightarrow LFO_Setup$ 

Shortcut: NUMLOCK + [7]

LFO Waveform	Waveform of the LFO – different types.	
LFO PulseWdt	Duty cycle (for some waveforms).	
LFO Speed	Speed (frequency).	
Wheel→Pitch	Modulation wheel to Pitch Band.	
AutoModDelay	Automatic vibrato – after what period of time to trigger.	
AutoModDepth	The intensity of the Auto Vibrato.	
AftrT→Pitch	AfterTouch to Vibrato.	

## **EffSound Setup**



*Menu*→*Program_Setup*→*EffSound_Setup* 

Shortcut: --- NONE

This menu defines the sound (sound effect) which is played WHEN CALLING this program. The most common example of this is the change of register of accordion. If the specified program is a timbre of an accordion, then through this menu can be set exactly at the time WHEN CALLING program to hear a sound like a changing register.

PGChangeZone: this is the main parameter. It defines whether the function is active, and if so, in which zone are located the multisamples of the sound to be played.

Next to this parameter we have two parameters to set the note range, where we have placed sound effects. The idea is that if you have several sounds, you can use them all. The sound is selected automatically and randomly by the sampler. If the two values are equal, will always played the same sample (one to which values point). With these parameters we choose the note or notes to which we have located sound in the relevant multisample.

IMPORTANT!!! The volume is adjusted by the VOLUME OF THE ZONE where it is located - menu zones (NumLock + 1).

## **Preset Name (renaming of a program)**

	Pre	eset	NF	IME	
A0	ORD	EON	1		
;	<=>	<u>7alA</u>	BC	DEFG	1

*Menu*→*Program_Setup*→*Preset_NAME* 

Shortcut: NUMLOCK + [9]

Here you can set the name of the program. By repeatedly pressing the ENTER button we go through all 16 positions of the program's name. Moreover the key [.] jumps between characters '', '0', 'A', 'a' for faster movement through the character table.

With buttons on the front panel of the sampler:

[MENU] > [+] > [ENTER] > 15x[+] > [ENTER] > (choice program with +/-) > [ENTER] > [+] > [ENTER] > [MENU]

From PS/2 numpad: [NUM LOCK + ENTER] > (choice program with +/-) > [ENTER] > [+] > [ENTER] > [NUM LOCK]

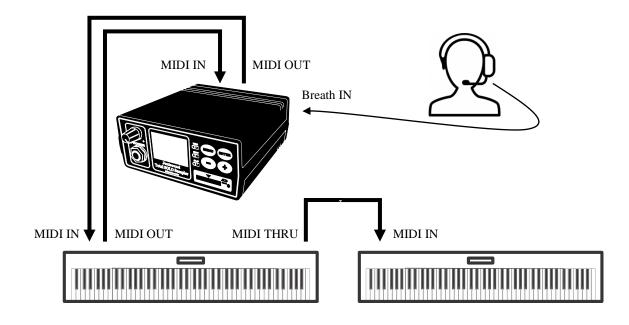
## **Breath controller**





[Breath IN] connector on the rear panel of the sampler is used to connect an analog controller. Analog controller means the following: controller, pedal, potentiometer or any other type of device that produces an analog signal. This connector provides +5 volts direct current (DC) and the input signal must also be in a range from 0 to +5 volts. In this input are most often plugged breath controllers from Juzisound or Yamaha model BC-3.

As a result of the input analog value, the sampler can generate simultaneously up to 4 different types of MIDI messages. They can be targeted both inward (to the engine of the sampler) and externally (via MIDI output of the sampler). When messages are sent out, they can be simultaneously sent in 3 different MIDI channels to be used to control other external MIDI device. An example for this is a breath controller, which controls simultaneously the sampler and the keyboard that controls the sampler. MIDI output of the sampler along with the analog input can be considered as separate and independent of the sampler device for converting analog to MIDI signal. To use the builtin analog converter to drive an external device, you should use another MIDI cable that is connected from MIDI OUT of the sampler to MIDI IN of the synthesizer you want to control. If you want to control more than one external MIDI synthesizer, then you need to connect MIDI THRU of the first controlled MIDI synthesizer with MIDI IN on the second. If the external synthesizers are more than two, then you need to connect MIDI OUT of the second to MIDI IN of the third and so to the end of the chain.



#### **Breath Setup**

There's a whole separate section in the main menu for managing and performance tuning of breath controller. This section is called "Breath Setup". To get to this menu first enter the main menu and then repeatedly press [+] until you see the screen "Breath Setup". Now you have to press [ENTER], to enter the menu "Breath Setup".



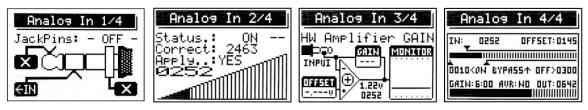
*Menu→Breath_Setup* 

Once you enter the menu "Breath Setup" inside also has several sub-menus, each oriented to a particular aspect of the analog input settings.

BREATH MENU	BREATH MENU	BREATH MENU	BREATH MENU
Analog In Setup	Brth MIDI Setup	Br.Preset Sellect	-1- Messages Setup
Ð	α Β	α Β	α β
BREATH MENU	BREATH MENU	Breath Menu	BREATH MENU
-2- Messages Setup	-3- Messages Setup	-4- Messages Setup	Breath Vibrato
a d	α Β	α Β	a d
BREATH MENU	BREATH MENU	BREATH MENU	
BreathOut Summary	Save Breath	Return to MAIN	
a d	a b	α	

Let's look at each sub-menu separately:

#### **Breath IN Setup**



 $Menu \rightarrow Breath_Setup \rightarrow Breath_In_Setup$ 

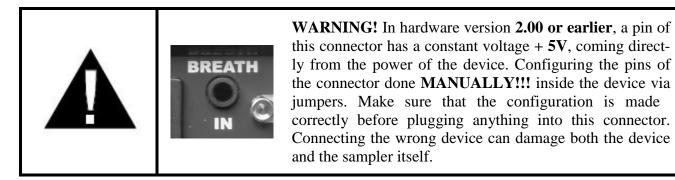
The section of the menu Breath IN Setup is used to configure the analog input. It sets the polarity of the pins of the connector, and gain and offset of the input signal. There are 4 pages. Let us look at the parameters of each of them. The first page is for configuring (polarity) of the power of the analog controller. The different hardware versions have different capabilities and correspondingly different menu that we can see here. The difference is that in the old hardware versions (up to version 2.00) controlling the polarity of the power supply to the analog input was manually - using the jumpers inside the sampler. Since version 3.00 is a software configuration that is selected from the menu and is displayed on the screen. Since version 3.00 there is another improvement. If there is a problem with the power supply to the analog controller, or cable short-circuit or something else fails, the sampler will automatically shut off power to the analog controller and will alert you with an appropriate message. After removing the problem, you can restore the power again - from the same menu. Due to the differences in the different hardware versions, let's look at the contents of this menu separately:

#### Hardware version up to 2.00

In hardware version 2.00 the setting is done manually with jumpers inside the sampler casing. The menu screen looks like this:

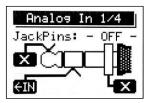
Analog	In	1/4	
JackPins			-
		gure	
l 1 ma	nua	with	
J	ump	ers!	

## Warnings when using older hardware versions

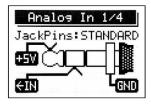


#### Hardware version 3.00 and later

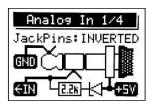
In hardware versions after 3.00, if the last digit of the version number is a 1, it means that you have a software module for configuring the pins on this connector. The same module also has short circuit protection. In this case the configuration of the pins of the connector can be done through the menu of the operating system. There are three possible configurations.



OFF! In this configuration, the supply voltage on the jack connector is turned off. This is the default configuration and the lack of power both protects the connected external device and the sampler itself.



This is the STANDARD configuration. In this configuration, the TIP of the inserted jack is supplied with a positive voltage + 5V, while the SLEEVE of the jack is the ground of the supply voltage. The RING of the jack is the input signal to the sampler. This configuration is used when working with breath controller Juzisound Ltd or with an external pedal or potentiometer.



This is the so-called INVERTED configuration. In this configuration, the TIP of the inserted jack is ground, while the SLEEVE of the jack fed by +5V. The RING of the jack again is input signal to the sampler, but in addition to it is connected Pull Up resistor value 2.2k to the positive pole of the supply voltage. This configuration is used when operating with a breath controller Yamaha BC-3 or with an external Expression pedal.

#### Troubleshooting the power supply to Breath IN

If there is a problem with the power to the Breath IN connector, immediately displays an error message and the power goes off. The most common reasons:

- A damaged cable.
- Connecting inappropriate external equipment.
- Incorrect configuration of power to the port and respectively to the external device.

In any case, if a problem is detected, the following happens:

- Automatically shuts down power to the connector.
- The first of the above-described configuration is activated.
- The screen displays an error message as shown below.

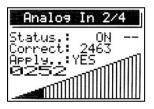
Error found!
Found 01 flags!
IN SHORT CIRCUIT

## Precautions when there is a problem with the power of Breath IN



**WARNING**! If you see this error message, immediately turn off whatever is plugged in the Breath IN connector and try to identify the reason for the fault. If you are sure that the port is configured correctly, but nevertheless the message persists, do not connect this device again, as this may cause damage to the connector or the connected device. Contact Juzisound Ltd to resolve the problem.

On the second page there are the following parameters:



Status	ON/OFF – This enables or disables the overall operation of the analog input.
Correct	This is the correction value of the input that you should not change. It is ad-
Contect	justed automatically.
Annly	This parameter allows you to specify whether the automatic correction of the
Apply	input to be applied or not. The default is YES (the correction is applied).
	At the bottom with large numbers 0252 and graphically is shown the current
	value of the analog input in real time. The range is from 0000 to 1023.

On the third page you have full control of the hardware amplifier to form a signal from the analog input. (*** This amplifier may be missing in some hardware versions. In this case, instead of the values of the parameters there are displayed signs [---]).

Analos In 3/4
HW Amelifier GAIN
0775530 (D) 1.220

Here you can set:

- The gain of the input amplifier,
- Output polarity (it can set the output to be in the same or opposite direction of the signal from the controller)
- Displacement amplification (which set the equilibrium point of the input amplifier hard-ware)

There is also a graphical representation of each value and the output (in the "Monitor"). In this page you can set the analog input so that it meets your requirements. For example: if you use breath controller you can adjust the sensitivity, making a lighter breath produce a bigger value change. Or to define the start and end point of pedal movement, etc.

On the last page are the parameters for the final shaping of the output signal.

Analog	In	4/4	
IN: (+) 0132	OFF	SET: D	45
0010<0N BYPE	11111111 155小	0FF>03	11111 300
GAIN:6:00 AVI	:: ND	OUT: 00	100

On this page you can see all parameters and both the input and the output signal from the controller. The parameters are as follows:

IN	<ul> <li>With this parameter you can reverse the polarity of the input signal by software.</li> <li>(+) The signal is NOT inverted</li> <li>(-) The signal IS inverted</li> <li>(X) The signal is interpreted as a signal with a midpoint. Example - imagine accordion bellows. When the bellows is in the middle position deviation in any direction makes the accordion to sounds, i.e. the volume changes from silence (zero) to full sound. Similarly, the deviation of the input signal from the middle point in any direction results in an increase of the output signal.</li> </ul>
Offset	With this parameter, you should set the value that is considered zero (e.g. the value which gives the breath controller when you <b>NOT</b> breathe in it). This parameter has no effect when the above parameter is the interpretation of type $(X)$ .
Bypass ON	Threshold for switching to Bypass mode. When the input level drops below this value the analog input is switched to Bypass Mode. Take for example a breath controller. Suction (creating a partial vacuum in the mouthpiece) drives the breath controller to a value below the threshold and the analog in module of the sampler enters BYPASS mode. Until the moment that you blow into the mouthpiece and the value produced by the controller jumps over another threshold, set by the Bypass OFF parameter.
Bypass direction	This parameter defines where to go and to stay the output value of the analog section, while Bypass mode is active. Arrow Up means that the value is set to its maximum position (like we apply maximum pressure). Arrow Down means that the value is set to its minimum value (like we do not blow in to the mouth-piece at all). The standard setting is UP. This is the default setting when breath controller is not used.
Bypass OFF	Threshold for leaving (exit) Bypass mode. If the controller sends a value above the set in this parameter, the "analog IN" module exits bypass mode. Using a separate and higher value prevents the controller from activation in case of ac- cidental lighter breath. The idea is that in order to activate the controller again, you have to blow once a little bit harder.
GAIN	This is a software amplification of the signal from the controller. Using a step of 0.25 in range from 1.00 to 10.00. This gain is duplicated as a function of the hardware gain from the third page, and if your hardware configuration have hardware gain, it is better to use it instead of this one. If you do not have hard- ware gain is not a problem set this parameter high enough until you reach the needed input sensitivity.
AVR	Averaging. This is a parameter that defines the speed of change of the output signal. With its help, you can apply some smoothing of the output signal from the controller if necessary. Degrees of smoothing are NO - off; 2:1, 4:1, 8:1, and FULL - completely smooth.
	At the bottom of the screen you see the final output value from the analog sec- tion.

### **Breath MIDI Setup**

Brth	MIDI	Set	UP
MIDI   MIDI	Chnnl Chnnl Chnnl s Mode	A: B: C: D: D:	01 02 03 FLT

 $Menu \rightarrow Breath_Setup \rightarrow Breath_MIDI_Setup$ 

Here you can define the output channels through which to send MIDI messages, resulting from the operation of the analog input. And what happens to the values already sent via MIDI when the controller enters Bypass mode.

It is possible to send MIDI messages via 3 MIDI channels simultaneously. These MIDI channels are labeled A, B and C. For each generated MIDI message you may choose a MIDI channels to be sent.

The last parameter (as mentioned above) defines what to happen to the values of the sent messages when the controller enters into BYPASS mode. There are two options:

- DFLT to sent the default values for these MIDI controllers;
- STOP Interrupt further transmission of these MIDI messages without sending the default values. Values remain as they were at the time of entry into BYPASS mode.

### **Breath Preset Select**

This menu selects which of the six different presets to be used for generating MIDI messages. For each preset you can remember up to 4 MIDI messages with all their parameters.

BrP	reset	Sell	ect
Use	Pres:	set:	01

### PRESET'S MIDI MESSAGES:

There are four separate and identical menus. Each of these four menus defines the generation of a single MIDI message, together with its associated parameters. We will discuss only the first menu - the others are the same.

Message	Setup	1-2-3-4
---------	-------	---------

BrControl	1-4
Expression:	011
	ME AI
	$01 \ll 01$
	000 127
	000 127

### $Menu \rightarrow Breath_Setup \rightarrow -1$ - Message Setup

This menu determines what to be the generated MIDI message and what parameters to be used for generation.

- The first parameter defines the MIDI message which MIDI controller to be. If there is no MIDI controller selected, this section is not used and does not generate any MIDI message.
- Next you have 4 characters (ABCI). They define the target of already generated MIDI message. Characters A, B and C are to direct MIDI message to the MIDI output on the rear panel of the sampler. Each letter corresponds to one of the already selected MIDI channels. Thus, activation of any or all of these letters, you can send the generated MIDI message to the 3 MIDI channels simultaneously. Letter [I] sends the same MIDI information internally to the sampler itself. From now on, the program settings define how will be interpret the signal from the analog input.
- Next you have 2 numbers for choice the curves that shape a MIDI message, and an opportunity to make a mix of the two selected curves. Curves themselves represent pre-adjusted relationships that shape the output value relative to the input value. The use of different curves for different parameters gives many great features. You can try it.
- Next parameter sets the lowest level of input that is accepted as valid. Values above this threshold starts to generate a MIDI signal. All values of the input signal below this minimum value is considered zero and the output does not change. The next number on the same line sets the highest value of the input signal to be considered valid. All values of the input signal higher than this threshold are ignored and the output does not change.
- The last line is defined minimum and respectively maximum value of the output MIDI controller (signal) that can be generated. These are the lowest and highest output value generated as a result of the input signal.

The combination of all these parameters gives huge opportunities to achieve what you want in the best possible way. The remaining three sections are exactly the same as the first. Each section is independent of the others and they all can work together. There are a total of four sections, and it is possible to generate 4 different MIDI messages simultaneously from the same input analog signal.

	3rea	911	Levels
011	AI	080	
074	A	080	
OFF		000	I
OFF		000	I
OFF			

**Breath Out Summary** 

 $Menu \rightarrow Breath_Setup \rightarrow BreathOutSummary$ 

In this menu, there is nothing to change. Here you can watch all generated MIDI signals on one screen simultaneously. You can see all the numbers of the generated MIDI messages, and routes that are sent to (internally to the sampler or externally via MIDI output).

### Save BREATH settings

With buttons on the front panel of the sampler:  $[MENU] \rightarrow [+] \rightarrow [+] \rightarrow [ENTER] \rightarrow 9 \text{ пьти } [+] \rightarrow [ENTER] \rightarrow [+] \rightarrow [ENTER] \rightarrow [MENU]$ 

From PS/2 numpad: [ENTER + NUM LOCK]  $\rightarrow$  [+]  $\rightarrow$  [+]  $\rightarrow$  [ENTER]  $\rightarrow$  9 пъти [+]  $\rightarrow$  [ENTER]  $\rightarrow$  [+]  $\rightarrow$ [ENTER]  $\rightarrow$  [NUM LOCK]

# **Example of breath controller settings**

Common settings in the menu Global Setup

1. Enter: Menu->Global Setup->MIDI Setup->MIDI Out is (on the second page), and set it to OUT.



2. Enter: Menu->Global Setup->GL.Expres Setup->Gl.Breath->E (on the third page), and set it to YES.

* MAIN MENU *	GLOBAL MENU	Gl.ExprSetup 1/3	Gl.ExprSetup 3/3
Global Setup	Gl.Expres Setup	MIDI▶ ExerCrv: 06 Brth ModW JoyD PRUs	Gl.MIDI->Exp: YES Gl.Breath->E: YES ExpChangeSpd: 016
Ð	<b>a</b> D	PBDw	

3. Save Global - penultimate point in the menu Global Setup - Menu->Global_Setup->Save_Global.

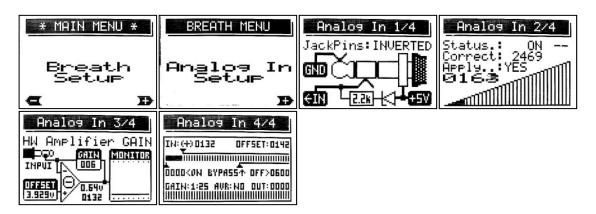


### **Using Juzisound Breath Controller**

Enter: Menu->Breath Setup->Analog In Setup and set all pages as shown below:

* MAIN MENU *	BREATH MENU	Analos In 1/4	Analos In 2/4
Breath Setup	Analog In Setup		Status.: ON Correct: 2444 Apply.:YES 0141
a b			
Analos In 3/4	Analog In 4/4		
HW Amp. DC.Offset	IN: 0137 OFFSET:0151		
INPUT IN DOI			
INPUT L	0030<0W BYPASS+ DFF>0350		
0.66v 2.065v + 0137	GAIN: 6:00 AVR: NO OUT: 0000		

Using Yamaha BC-3 Breath Controller Enter: Menu->Breath Setup->Analog In Setup and set all pages as shown below:



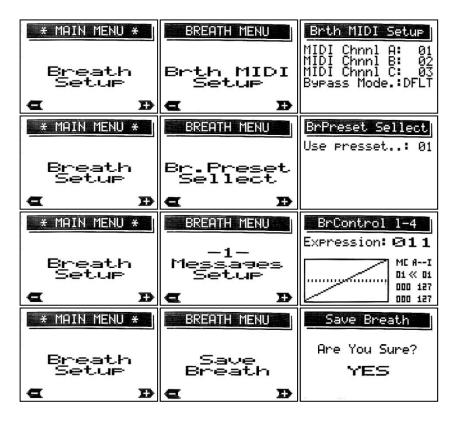


Additionally, on the breath controller make the following settings:

- Increase GAIN to end.
- Do not inflate.
- With OFFSET knob find the position from which the screen "Analog In 2/4" start a change in the reading on the display the one with the big numbers.
- Slightly return OFFSET knob backwards so that it stands in the position just before start changing.

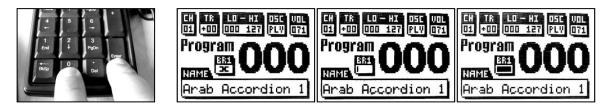
### Common settings in the menu Breath Setup

Enter: Menu->Breath Setup and set all submenus as shown below:



#### Enabling and disabling the Breath Controller - Bypass mode

When working with breath controller, usually there is no sound if you do not blowing into the mouthpiece. To be able to work without breath controller you need to disable the module "Analog IN". In this case although you do not blowing, the sampler works like if you apply maximum breath pressure or like if there is no breath controller at all. Activation and deactivation of the breath controller can be controlled by the PS/2 numpad keys [ENTER] + [0]. When using breath controller Juzisound, the control can be done from the breath controller itself. In this case to disable the breath controller (to enter BYPASS mode) you should shortly suck the mouthpiece (creating a partial vacuum). To enable it again (to exit BYPASS mode) you should blow a little harder for a moment. The current state of the breath controller is displayed on the main screen of the sampler. If the breath controller is displayed a flashing sign [X]. If the breath controller is active - this field displays the strength of breath pressure. These two methods for managing the state of the breath controller (by mouth or by PS/2 numpad) can be used together and simultaneously, without interfering.



#### **IMPORTANT!**

Value **OFFSET**, shown on page "**Analog In 4/4**" may vary from breath controllers. For this purpose, with plugged breath controller and **without blowing**, you should note the value of the previous parameter (shown on the left in the same row and on the same page). Then, the value of OFFSET is adjusted with 10 units more than the previous value. The value of GAIN on the same page defines the necessary breath pressure to fill the entire scale. Increasing GAIN makes lighter breath produce biggest signal changes. Set GAIN high enough until you become comfortable. Do not set the value of GAIN too high, because it leads to loss of control in the middle range. This happens because the breath controller at the lighter breath "sticks" to the highest value and thus its use loses its meaning. For this purpose, look at the indicator on the main screen from time to time. If it stays constantly at a minimum or maximum, probably you set GAIN too high. In this case reduce it to a few units and try again.

# Key commands at startup

By pressing a combination of buttons on the front panel of the sampler during power-up, you can run different service functions, and to control various parameters relating to the sampler's subroutines. Scanning the buttons is done immediately after power and once you see signs of the screen, you can release the buttons. The following combinations of keys have associated functions:

Button pressed	Function
No button is pressed	Normal start
	Sampler starts its main operating system.
[ENTER]	<b>Detailed scanning of all files of the bank for fragmentation.</b> The sampler launches its main operating system and performs
	DETAILED check of files on the SD card. Each file is checked for fragmentation sector by sector (in contrast to normal startup in which only checks the location of the files).
[+]	Search for the beginning of the bank-file - by sectors.
	The sampler starts its main operating system, but the search for bank file on the SD card is not by FAT table as usual but by searching the first sector of the file. This first sector must have a specific content. The search starts at sector 0 and continues until the first sector of the bank-file is found. This function is provided in case of failure of the FAT table on SD card. This is possible because during normal opera- tion the sampler NOT uses FAT table but direct offsets from the be- ginning of the bank file. FAT table is only necessary for Mp3 player.
[+] II [-]	<b>SD Card speed test</b> Sampler starts its main operating system, but before doing this tests the speed of the SD card. After completion of the test sampler shows the result and wait for button pressed.
[-]	Skipping the initialization of the MP3 player
	Sampler starts its main operating system, but skips the initialization of the MP3 player. This function is provided for cases when during the initialization of the MP3 player a fatal error occurs, (a usually the rea- son for this is an attempt to update the list of Mp3 files and on the SD card there is no place for it), however, can continue to work with the sampler without Mp3 player.

[MENU] и [ENTER]	Subroutine "Firmware Loader"			
	Sampler starts subroutine Firmware Loader, which is designed to up- date another subroutine called Bootloader. Firmware Loader is stored in the main memory of the sampler by the manufacturer and is the only software in the device, which CANNOT be changed or updated by the user. The other main feature of this subroutine is that it cannot be damaged by the user, and therefore through it is always possible to recover the rest of the sampler's software. For proper operation of the Firmware Loader is needed a PC with a serial port and a serial cable to connect the computer and sampler. During operation of the Firmware Loader its version is displayed on the screen. To exit this subroutine			
	you should restart the sampler (power off and power on).			
	Running the subroutine Bootloader			
[MENU] и [-]	Sampler starts subroutine <b>Bootloader</b> , which provides the following			
	<ul> <li>functions: <ul> <li>Updating sampler's OS using a PC with serial port.</li> <li>Updating sampler's OS directly from SD Card.</li> <li>Using the sampler as a computer USB to SD Card reader.</li> <li>Displaying the serial number of the sampler (to be able to check it, even if there is no main operating system).</li> </ul> </li> <li>Bootloader – serves mainly to update the core OS of the sampler. Most often, the update is done by copying the new OS (BIN file) to</li> </ul>			
	the root directory of the SD card and load directly from it. PC is ne- cessary only for downloading of the new OS and to copy it on the SD card. During operation of the Bootloader its version is displayed on the screen. To exit Bootloader should execute a command to launch the main OS, or simply to restart the sampler by power OFF and power ON.			

# Software Update

After the initial programming (done by the manufacturer) only two subroutines can be updated - **Bootloader** and **Operation System**. Each of them is updated through the previous subroutine (located on a lower level). Respectively the **Bootloader** should be updated through the **Firmware Loader**, and core operating system **OS** should be updated through the **Bootloader**.



Note that each update **Bootloader** ALWAYS destroys the installed operating system. So after each update of the Bootloader, you should always install the **Operating System**, whether a newer version or the same that was installed until the update of the **Bootloader**.



**WARNING!** The process of updating of the **Bootloader** and the **OS** should NOT be interrupted. This can cause damage to the device and it had to be returned to the manufacturer for reprogramming!

# **Update of the Bootloader**

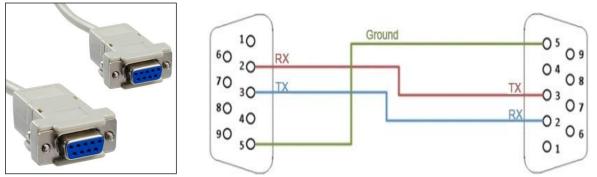
Bootloader is updatable through the subroutine Firmware Loader. Firmware Loader only works via PC connection. The connection is made through a serial interface (RS232) and the computer needs to have hardware serial port or have USB-> RS232 serial cable.



PC COM port (RS232)



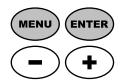
USB to SERIAL (RS232) cable



Picture and circuit diagram of the serial cable required for connection between a computer and a sampler.

To upgrade the Bootloader from the computer you need to have the software Hyper Terminal, and it can be configured as follows:

🔜 1 - HyperTerminal		
File Edit View Call Transfer Help		
0603000		
	COM1 Properties       Port Settings       Bits per second:       115200       Data bits:       8       Partyr:       None       Stop bits:       1       Flow control:       None       Restore Defaults       0K	
Disconnected Auto detect	Auto detect SCROLL CAPS NUM Capture Print echo	



Then you should start the sampler in a special way to start the subroutine Firmware Loader, which helps to upgrade Bootloader. For this purpose, during power-up you have to press and hold simultaneously the buttons MENU and ENTER on the front panel of the sampler.

If Firmware Loader was launched successfully on a computer screen you should see a message similar to the one shown below:

🔲 IAP - HyperTerminal	
File Edit View Call Transfer Help	
DEBUG: Power Fault Detect	
= = (C) COPYRIGHT 2011 Computer Sound Studio - *Juzisound* = = Juzisound Firmware Loader = = Dev ID: 33FFE5F53143333720651157 Version 1.0.6 =	
Option bytes: 0x1FFFF804:[000104 Valid ] 0x1FFFF806:[000255] Dusplay[EA DOGS102] Contrast[000050] JackPIN[NO] Amp[NO] Download firmware to device flash memory	
Connected 00:01:48 ANSI 115200 8-N-1 SCROLL CAPS NUM Capture Print etho	

On the screen of the sampler you can see the version of the Firmware Loader.



Then you need to press the key [1] Windows PC keyboard. This selects the loading function from the menu of Hyper Terminal "**Download firmware to device flash memory**". Once the function is selected, on the sampler should light the lamp EDIT, rather than Hyper Terminal you have to select "**Transfer**" and "**Send File ..."**. On the computer screen should appear a window for selecting the file to be sent to the sampler and to select the protocol for file transfer. For a file, you should select the BIN file with the new version of the **Bootloader**, and for the protocol type from the drop down menu you should select the type **Ymodem**.

Folder: D:\Myl Filename:	Data\IAR Projects\Sample	r -
Simpler BOOT	LOADER 2.4.bin	Browse
Protocol:		
Ymodem		v

After selecting the file type and the protocol, you should press [Send] which starts loading of the **Bootloader**. On the computer screen should shows up another window that shows the status and progress of the loading. On the screen of the sampler will shows a message "**Loading ...**" and progress indicator that also shows the progress of the update. During loading the uppermost lamp L1 of the front panel of the sampler flashes.

Sending:	D:\MyData\IAR Projects\Sampler - TotalBass\project\EWARMv5\Relea				
Packet:	167	Error checking:	CRC	File size:	502K
Retries:	0	Total retries:	0	Files:	1 of 1
Last error:					
	( 			164K of 5	02K
File:					

If the loading of the Bootloader is successful, on the computer screen in the window of Hyper Terminal should display a message similar to the following:

Programming OK! Name: Sampler BOOTLOADER 2.4.bin Size: 514048 Bytes

If you see the message "Programming OK", this means that the loading was successful. In all other cases, you will have a different error message, which describes in detail the type of error that occurred.

# Update of the main OS

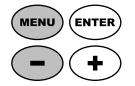
As already mentioned above, the update of the operating system is done with the subroutine, which is one level below it. In our case, this is Bootloader. Bootloader can update the operating system using a PC (via a serial port connection) or directly (by reading the operating system from files stored on an SD card). In this case, we will discuss the procedure of update using the SD card, as it is much easier. What you will need:

**1. Computer**. Computer will only be required to download the latest OS from the Juzisound Ltd website, and to save this file to a standard SD card. Note that the downloaded from the website OS is archived in a ZIP archive. So after you download the ZIP file, you must first unzip the file that contains the OS. This is a file with extension BIN and name like TSSxxx, where xxx is the version number of the OS. Exactly this file you need for updating of OS. In addition, make sure that your computer can read and write SD cards. If you do not have a built-in SD card slot, then you can use an external USB SD card reader.

**2. SD Card.** For the upgrade, you can use the SD card with which the sampler operates or any other SD card, which does not even need to have a valid sound bank to work with the sampler. The card should be in working condition and to have at least 1MB free space for the file with the OS. After the update is complete, the OS will be permanently stored in the internal FLASH memory of the sampler. From here on, for the normal operation of the sampler the file with the OS is no longer necessary. You can delete it from the SD card or leave it there - it does not matter.

### The procedure:

- Download the latest version of the OS from <u>www.juzisound.com</u>. The operating system is a single file that is archived in a ZIP archive. *Example: Assume that you update with version 2.60. File you need to download from the site will be named "TSS260.ZIP".*
- 2. Unzip the downloaded ZIP file and extract the BIN file that's inside. *Example: The file you should have finally after unzipping the downloaded archive in the pre-vious paragraph will be named "TSS260.BIN". The name is formed as follows: TSS is "Total SOLO Sampler", and 260 is the version number of the operating system in this case 2.60.*
- 3. Insert an SD card into the Card reader of your computer and copy the extracted BIN file with the OS in the root directory of the card. IMPORTANT: The file must not be in any folder, but directly in the root directory of the card. Otherwise sampler cannot recognize it.
- 4. Turn OFF the power of the sampler.
- 5. Remove the card from your computer and insert it into the sampler.
- 6. Press and hold the [MENU] and [-] on the front panel of the sampler and turn the power on.



7. Sampler's screen should display the start page of the Bootloader. Once you see it, you can release the buttons.



8. Wait about 3 seconds until the screen displays this menu.

Bootloader:	2.4
Load from R	S232
▶Load from S	DCard
USB Card Re	ader
Show Device	Info
Show Device Start Progr	ham ->

9. With the [+] button on the front panel of the sampler select the second menu item named "Load from SDCard".

Load Fr	om SDCard			
File:	003/003			
TSS260.BIN				
[-]/[+] [ENTER]	Prev/Next Load file			

- 10. Press [ENTER].
- 11. Wait until the sampler detects the card. You should see a screen with operating systems files. If you have more than one operating system on the card, you can select one of them with the keys [+] or [-].
- 12. Once you select the file operating system, press [ENTER]. This starts the update. You will see an indicator that shows the process of update. Wait for the procedure to finish completely.





If there are no error messages after the update is completed the screen will display a message: "**Flash memory write OK!**". This means that the OS is stored successfully in the program memory of the sampler, which means that the procedure is completed.



Press any key on the front panel of the sampler or restart the power. Both actions will reset the device and now at the starting sequence of the sampler you should see the version of the new operating system. It shows for a moment during loading.



## **IMPORTANT!**

Do not interrupt the power of the sampler until the operating system is updated. This could damage the device and may have to start the procedure again. Or even have to return it to the manufacturer for a new complete reprogramming.

# **MIDI IMPLEMENTATION CHART**

#### Juzisound Total SOLO Sampler OS Version 2.60 – Mart 2014

		1		OS Version 2.60 – Mart 2014
	Function	Transmitted	Recognized	Remarks
Basic	Default	1-16	1-16	Memorized *1
Channel	Changed	1-16	1-16	
Mode	Default Messages Altered	X X ******	1 X	
Note Number	True voice	0-127	0-127	
	Note ON	O 1-127	O 1-127	Remapped *2
Velocity	Note OFF	O 1-127	O 1-127	
	Poly (Key)	Х	Х	
Aftertouch	Mono (Channel)	0	0	*3
Pitch Bend		0	0	
Control Change	0, 32 1, 2, 6, 38 7, 11, 12, 13, 16, 71, 74, 98, 99, 100,101, 120,123 ALL 0 < 120 3,9,14,15,20,21,22,23,24,25,26,27,28,29,30,31	000000000000000000000000000000000000000	000000000000000000000000000000000000000	Bank Select (Msb, Lsb) Modulation Breath controller Data Entry MSB / LSB Volume Expression Fx control 12 / Terca control *4 Fx control 13 / Terca control *4 Ribbon Controller Resonance / Filter Resonance Brightness / Filter Cutoff Non registered parameter LSB, MSB Registered parameter LSB, MSB Registered parameter LSB, MSB All sound off, All note off Send from Analog IN>MIDI converter*5 Received when control mode is ON *6
Program Change	True #	O 0-127	O 0-127 0-127	
System Exclusive		0	0	Control scale function
System Common	Song Position Song Select Tune	X X X	X X X	
System Real Time	Clock Commands	X X	X X	
Aux Messages	Local ON/OFF Reset All Controllers All Notes Off Active Sense Reset	X X X X X	X X O O X	
Notes	<ul> <li>*1: It is possible that different MIDI channels to be used for input and for output. Stored with the global settings.</li> <li>*2: The volume of the input notes can be recalculated according to 16 different curves. The settings are stored along with the global settings.</li> <li>*3: AfterTouch can be sent to MIDI only from the converter of the analog input.</li> <li>*4: These controllers are used outside of the standard, when are input. They are used to control the start and stop of the terca, and for start the scanning mode for terca.</li> <li>*5: All controller numbers from 1 to 119 are available for send from the converter to the MIDI from the analog input.</li> <li>*6: These controllers are accepted when is activated one of direct control modes: Control Mode = "Real Time", "EQ" or "Modulation Effect". These are controllers that send potentiometers of the Behringer UMX61, and may be used to directly control certain functions of the sampler.</li> </ul>			

Mode 1: OMNI ON, POLY Mode 3: OMNI OFF, POLY Mode 2: OMNI ON, MONO Mode 3: OMNI ON, MONO O: Yes X: No

# **Specifications**

#### CPU

ARM CPU STM32F103RET6 Frequency 73.728 MHz

**Digital Audio Processing** 32 bit / integer / 48 kHz

**Sample Rate** 48 kHz

Wave memory Unlimited - depending on the size of SD card used

**SD** Card slot SD Cards 2.7-3.6V

Supported SD cards All models 16MB - 32GB

Audio file format WAV / 48000Hz / 16 bit / Mono

**Sampling method** Full Length – entire length of the sample Not use Loop Point Single WAV file for each note

**Multisamples** 2048 multisamples in a bank 14 multisamples for a program

**Multisample files** 128 WAV files, single file for each note

Maximum polyphony 8 notes, (depends on the mode of the oscillator)

**Oscillator mode** Normal POLY, Bass MODE 1, Bass MODE 2, Mono RETRIGER, Mono LEGATO, Mono Portamento CZ, Mono Portamento DX, Mono Portamento KORG, Bouzouki, Gibson Guitar, Mono Portamento JS, Mono portamento JS2, Violin, Violin 2, Poly 4 notes, JS Tenor Sax, JS Alt Sax, JS Clarinet, JS Bouzouki

**Programs** 999 user programs for a bank Scale / comma Resolution +/- 16 units per semitone One USER scale for each program, plus 3 separate programmable global scales with option for instant recall at any time. There is ability to control the scales via MIDI directly from a synthesizer (with Sysex messages).

#### Terca

6 presets for terca plus modes for scanning tercas. Ability for terca with a timbre, that differs from the basic timbre. There is a possibility to control the tercas via MIDI directly from a synthesizer (with CC12 and CC13).

#### Effects

Delay, EQ/Enhancer, Modulation effect (Chorus, Flanger, Phaser) (All effects are mono)

### DAC

Texas Instruments PCM1791A Sampling frequency 48 kHz Audio data 32bit

### **Operational amplifiers**

NE5532P, TL072

#### Master volume

Analog, ALPS stereo potentiometer

### Audio out – main

2 separate mono 1/4" TRS connectors, mono, unbalanced, rear panel Impedance 140  $\Omega$ Signal-to-noise ratio 112 dB Stereo crosstalk< -80 dBu/1 kHz Maximum output level +20 dBu

**Audio out – phones** 1 stereo 1/4" TRS connector, unbalanced, located on the front panel Impedance 75  $\Omega$ Signal-to-noise ratio 112 dB Stereo crosstalk < -80 dBu/1 kHzMaximum output level +20 dBu

### **MIDI** interface

Standard 5-pin DIN connector MIDI IN ungrounded through opto-isolator MIDI OUT configurable as MIDI OUT or as MIDI THRU

#### **Remap of note velocity**

Yes, using 16 different velocity curves, plus the limitation and mixing of 2 curves.

#### **MIDI** Analyzer

Yes, shows the 6 most recent MIDI events

**Sysex Analyzer** Yes, shows up to 128 Sysex bytes

### **Input for Analog Controller**

3.5mm stereo jack Software configuration of the power pins Power +5V Direct current (DC) Short circuit protection Input signal level from 0 to +5V Software controlled Pull Up resistor 2.2 kOhm Analog amplifier with software-controlled GAIN and OFFSET

### MIDI converter from Analog IN

Ability to generate up to 4 different MIDI messages simultaneously, each with independent parameters. Send up to 3 different MIDI channels simultaneously. Messages are sent via MIDI OUT.

#### Other external controllers

Two independent inputs of the EXTENSION connector for connecting the controller type switch (pedal). The switch controller should connect the active signal to the ground.

### Ability to power external devices

(Through the EXTENSION port) +5V / 300 mA +3.3V / 100 mA

#### **Additional interfaces**

Serial interface RS232 with TX and RX, connected to the EXTENSION port.

#### External keyboard

Standard PS/2 numeric keyboard (numpad) with native PS/2 interface.

Front panel controls

4 buttons: [MENU], [ENTER], [+] and [-] Master Volume ALPS potentiometer

### Display

101 x 64 pixels monochrome display with LED backlight

#### **LED Indicators** MIDI In, Audio CLIP, Edit, SD card

### Package contents

Sampler, 32GB SD Card, PS/2 numpad, MIDI cable, power cord

**Optional items (sold separately)** 

Breath Controller, second MIDI cable

#### **Operating conditions**

Temperature from 0° to 40° C Humidity < 90 %

### **Power Supply**

AC 110V/60Hz or 220V/50Hz Selected by the manufacturer without the possibility of change by the user.

#### **Build-in fuse**

20mm fast-blow glass fuse, rated 250 V / 315 mA  $\,$ 

# **Power consumption** <12 W

**Dimensions** 147 x 129 x 57 mm

### Weight

1.000 kg



All specifications are subject to change without notice



# Address:

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