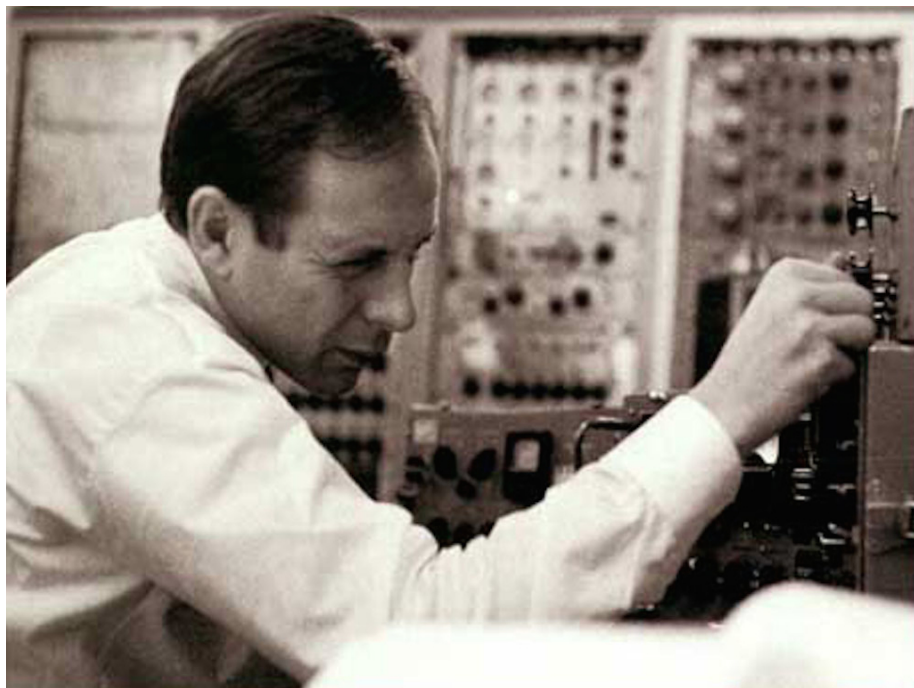


SOLO [Nr.19]

USER'S GUIDE



INTRODUCTION

SOME 'OF HISTORY

"Solo" [Nr.19] *für Melodieninstrument mit Rückkopplung* by Karlheinz Stockhausen, is a piece written for any solo melodic instrument and employs a feedback system to allow the soloist to create polyphonic music. Written in 1966 with its world premiere in Tokyo in the same year, its origin dates back to the same period as the composer's work "Kontakte". "Solo" uses the technique of recording onto tape with a single head together with a moveable arrangement of playback heads, allowing for the creation of complex sounds through accumulation.

The original version (with an analog technical set-up) requires a solo instrumentalist as 4 assistants (technicians): three assistants operating the opening and closing of sliders (L/R channels)—respectively the microphone, delay-line and output—and an assistant for the change of time values of the taps in the delay-line. During a performance a part of what the instrumentalist plays is captured by a two-channel recorder. Through a feedback circuit, the sections are recorded more or less densely stacked and processed with a variable delay time; the result is then played by two sets of speakers and mixed with the direct sound of the soloist.

The score (Universal Edition-Wien, 1969) consists of six pages of music and six pages of FORM-SCHEME (the number six can be considered a key element), in addition to explanatory notes (very detailed) for the performance the composition. "Solo" is one of those compositions by the author employing "controlled alea" as a compositional language. Indeed many choices are left to the soloist and cover almost all the events: from the selection of a version, the positioning of the material, the types of tonal variation requests, the order of pages and more. Really these faculty are provided as input by the composer through the management of macroscopic or microscopic parameters order, restricting the possible versions (virtually infinite) at a predefined or definable set of possibilities. In this piece the role of the interpreter is very important in the creation of a performance that is always unique and unperformed every time.

THE APP SOLO [Nr.19]

SOLO [Nr.19] is an App that aims to implement digitally the original analog set-up (Technical set-up) for the performance (One-man-performance) of six versions of Solo für melodieninstrument mit rückkopplung by Karlheinz Stockhausen. The algorithm is written using Csound.

This project began in 2006 in collaboration with LEMS (Laboratorio Elettronico per la Musica Sperimentale) at the National Music Conservatory "G. Rossini" in Pesaro (Italy). I developed subsequent improvements to the algorithm until arriving at the current version.

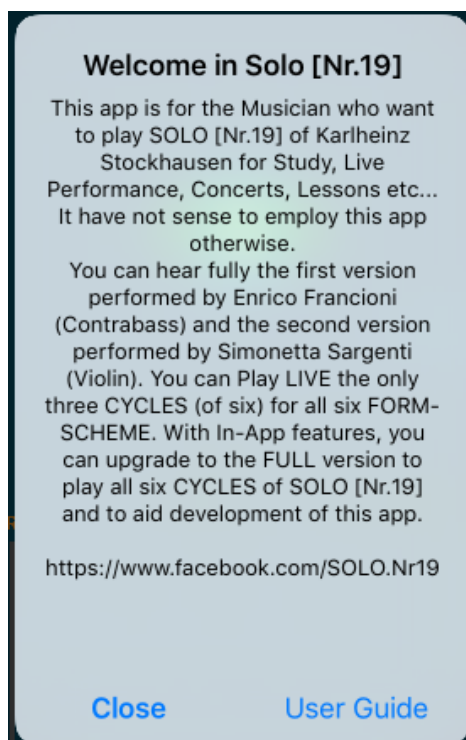
The algorithm aspires:

- *to overcome the practical limitations that the original analog layout posed the interpreter. It is now possible to perform the composition as an instrumental solo accompanied by iPad, iPadMini and iPhone;*
- *to develop an application iOS which is a valid support for real-time performance and a means to realize the requirements of Stockhausen's original setup (Technical set-up) in a digital environment*
- *to enrich the piece—tonally speaking—in both the instrumental and the algorithmic domains.*

N.B.

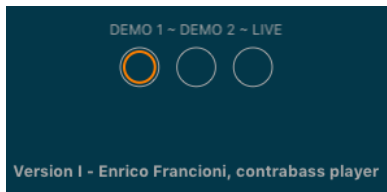
SOLO [Nr.19] in versions for iPad, iPhone and iPod can be **free** downloaded from the App Store and on the website <http://www.densitygs.com/>.

Just launch the app, it displays this message:

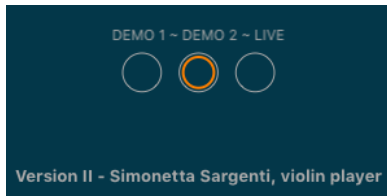


SOLO [Nr.19] is active in mode **DEMO 1**, **DEMO 2** or **LIVE**.

In **DEMO** mode:
tap on **DEMO 1**



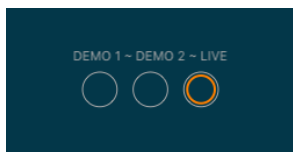
or **DEMO 2**



In **DEMO 1** you will hear a performance of the Version I for contrabbass¹, instead in **DEMO 2** you will hear a performance of the Version II for violin².

After the **START** observing the movement of the interface on the recorded audio signal; all controls are active.

In **LIVE**³ mode



you can interact with its own microphone signal limited to the first three *CYCLES* (A, B, C) of each *FORM-SCHEME* (Version).

To perform the other three *CYCLES* (D, E, F) have to purchase the app **SOLO [Nr.19]**.

In fact, *tapping* one of the buttons D, E or F, and after giving the **START**, the following message appears:

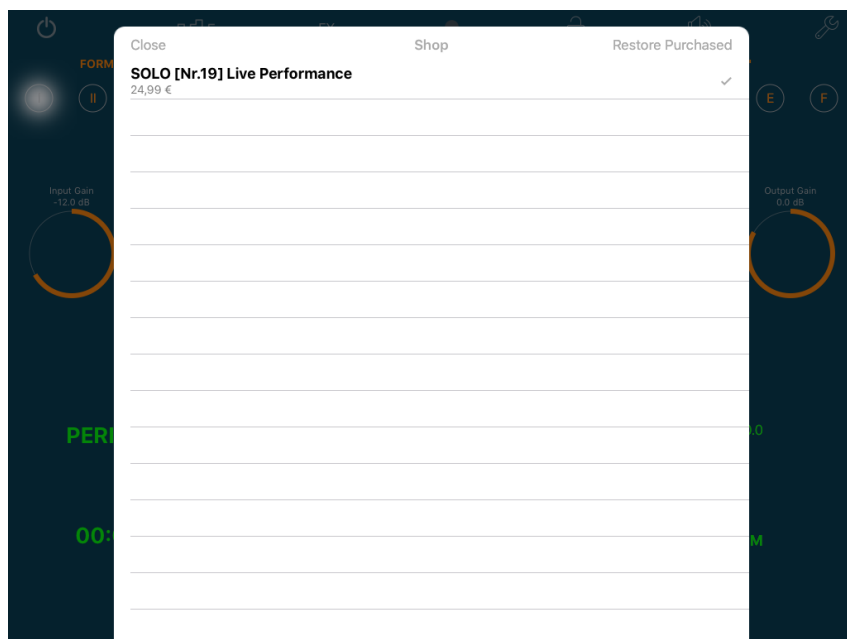
¹ Enrico Francioni, contrabass player.

² Simonetta Sargenti, violin player.

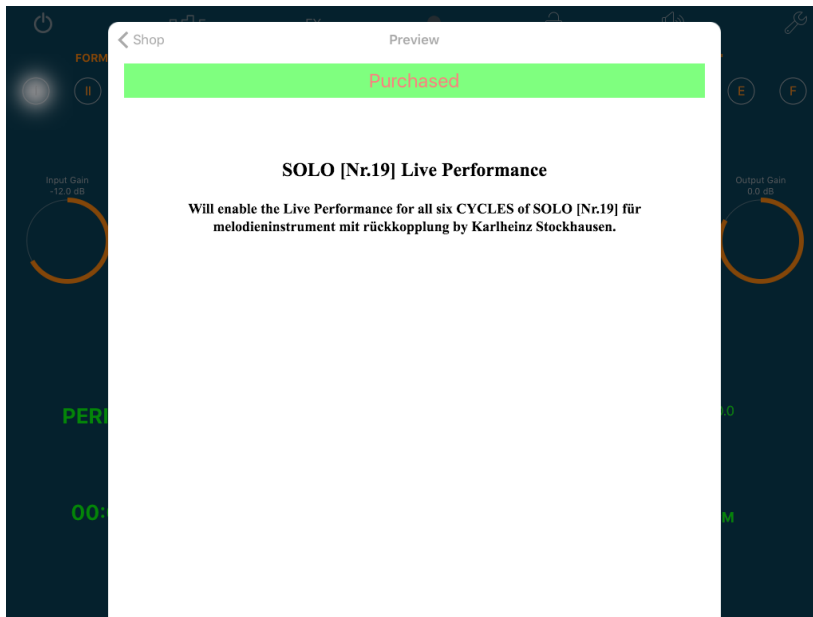
³ Mic (microphone) - AB (AudioBus) - IAA (Inter App Audio)



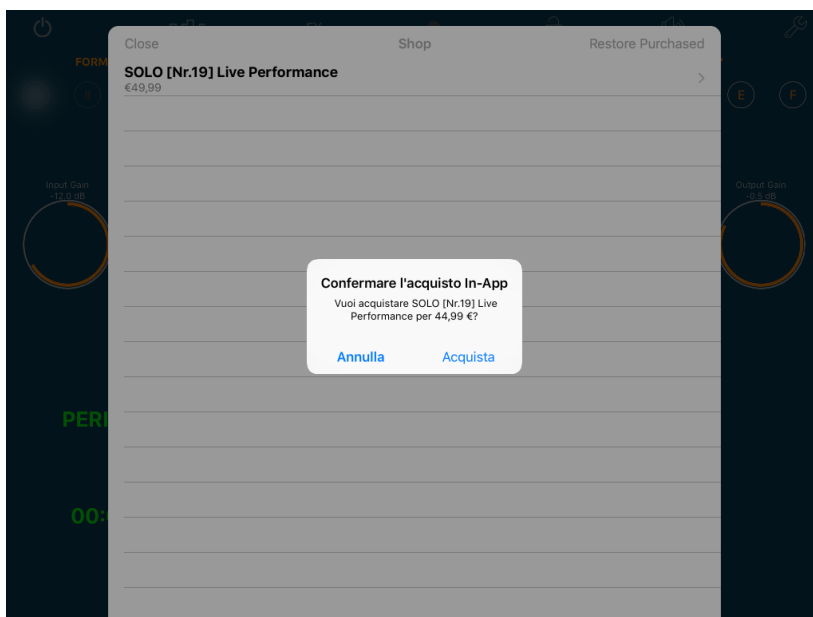
Tapping a toggle **OK** will appear where you can choose whether or not to purchase the app **SOLO [Nr.19]**:



after you click *Buy* and after you have registered with your ID and password,



you will be routed to the App Store where you can buy and download the *app* for a fee:



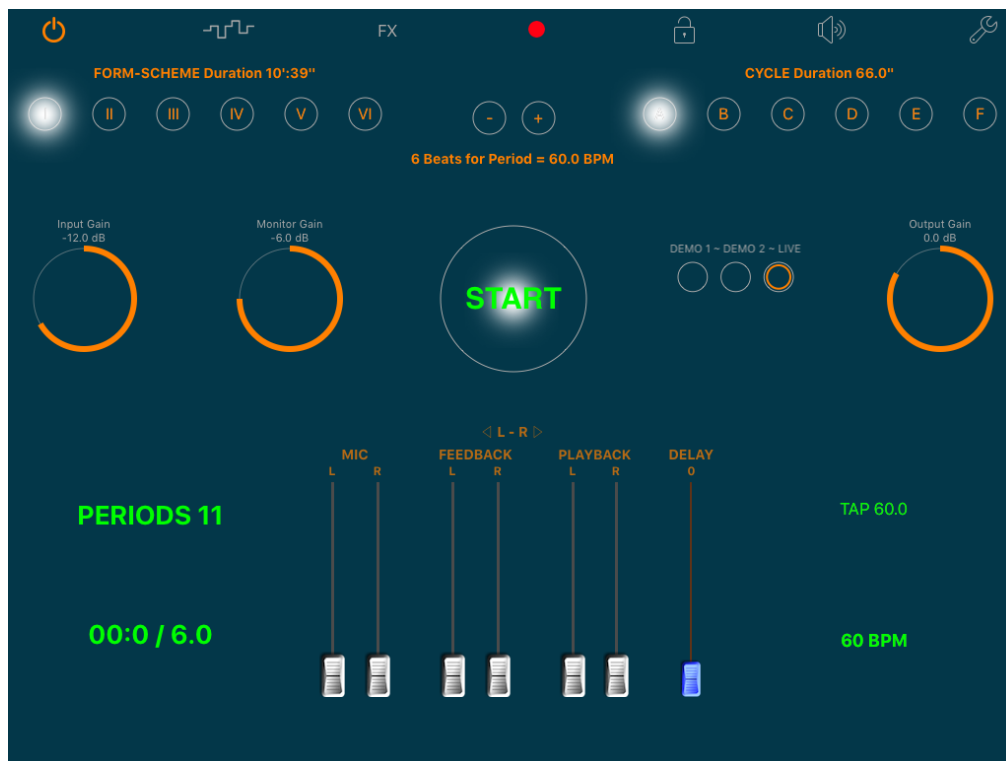
In the paid version of the app all the functions are active; will be able to exploit the full potential of the application for the full implementation of all *CYCLES*, of six *Versions* **SOLO [Nr.19]**, which are described below in detail.

GETTING STARTED

After giving a *tap* on the icon of **SOLO [Nr.19]**:



appears immediately screen interface **SOLO [Nr.19]**:



A *refresh* and the center button prepares the **START** position:



The power button:



is automatically set in the standby mode (*standby*);



(default is on) on/off DSP (Digital Signal Processing). **SOLO [Nr.19]** supports mix audio with other applications, you can also disable DSP in order to save battery and CPU consumption. You should not toggle off when is connected to audiob.us or Inter-App Audio!

However, if it is given another *tap* on the power button:



the algorithm switches in the *off* mode and the center button, it automatically goes into the position **DSP is OFF**: the algorithm is now *off*.

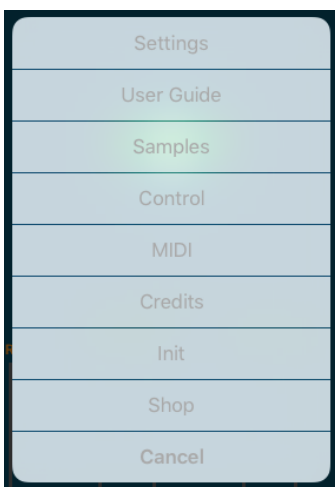


If you exit the *app* (two *tap* on the Home button) and you keep on your operating system, but even if you shut down the operating system for iPad (iPadMini), or iPhone, to re-boot the iOS and then **SOLO [Nr.19]**, all the latest *app* settings will be restored.

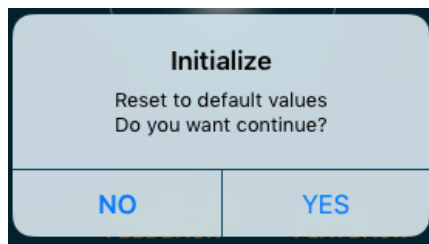
To restore the *default* settings, *tap* on the button:



tap on *Init*



At this point you open the toggle:

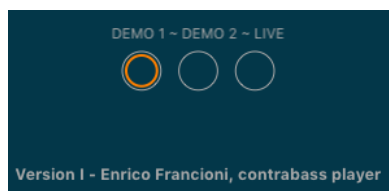


Here you can choose whether (or not) a *reset* of the live performance (with the exception of *drywet* which remains open).

At the bottom of this Users'Manual shows the *default* settings.

SELECTING THE INPUT SIGNAL

The choice of the signal is accomplished via the buttons:



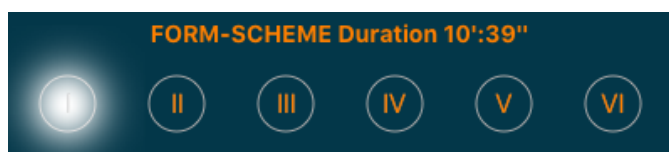
you can choose the type of input signal you want to import:

- **DEMO 1** or **DEMO 2** for the files (mp3) demo, or
- **LIVE** for the microphone signal

Moving from **DEMO** mode to mode **Mic** (or vice-versa: from **Mic** to **DEMO**) algorithm adjusts itself on *CYCLE A*.

CHOICE OF FORM-SCHEME (VERSION)

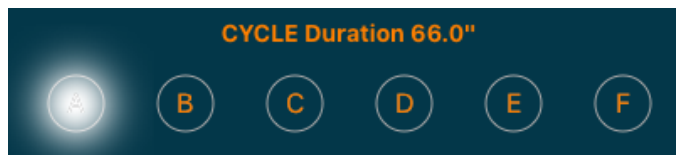
Mode **LIVE** now choose the *FORM-SCHEME (Version)* to activate (from I to VI) by tapping the buttonhole:



lights the button with the *FORM-SCHEME (Version)* chosen and it shows the *Duration* in minutes and seconds.

CYCLES AND SKIPTIME

In setting phase is also possible to decide to start the execution from the beginning of a given *CYCLE* of the piece (A, B, C, D, E, F).

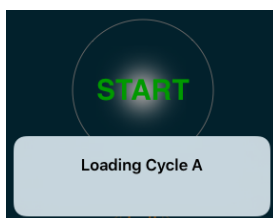
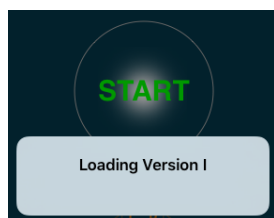


With a *tap* on the button corresponding to the desired *CYCLE* as well as to start the execution of the shown *CYCLE*, is the time value of the *CYCLE Duration* (in seconds).

This feature is especially useful in the study phase to avoid having to repeat the whole piece again.

Changing the *FORM-SCHEME (Version)* algorithm automatically adjusts itself on *CYCLE A*.

Choosing a new *FORM-SCHEME (Version)*, or a new *CYCLE*, the warning messages to appear:



CHOOSING THE NUMBER OF BEATS BY PERIOD AND TAP

If the values specified in the string of *beats*:



(set by *default*) do not meet the personal music choices (made through the creation of their Pages taken from the six pages that the author makes available in printed material for the execution, Universal Edition - UE14789), with the algorithm we set the value on number of *beats* per *PERIOD* selected in each of the six *CYCLES*, with the help of the keys to decrease/increase the number of *beats*:



the BPM value is calculated and displayed automatically.

Remember that in the original paper score (score) the *beats* are indicated by dashed vertical lines that run through all staves (each staff is a *PERIOD*) of the Page and each Page is a *CYCLE*.

In addition, the interpreter has provided another valuable tool primarily used in the study phase, when, **while wanting to keep the same number of beats per PERIOD**, he wishes to study the piece to a different speed; it is the button:

TAP 60.0

(or *Master-time*) where *tapping* on a regular basis at least 2/3 times the speed you choose, you can set the desired value of BPM on the current *CYCLE* and (in a consequent and proportional mode) in all remaining *CYCLES*; now the selected value appears next to TAP.

To re-set the BPM value to the situation before a *tap* on:

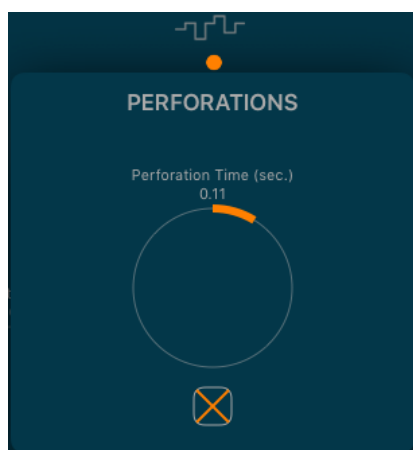
60 BPM

PERFORATIONS

To activate the *Perforations*, a *tap* on:



is a *toggle* that gives us the ability to enable (*on*) or not activated (*off*) *Perforations* provided in *FORM-SCHEME* you want to perform;



inside the *toggle* is also possible to establish the *Perforation Time*, ie the time value of the ramps of the envelope *irise-decay* that will make the sliders of the first and second Assistant.

While the number of *Perforations* to be made for each *PERIOD* is set by default based on the *FORM-SCHEME*, the *action time* "ad libitum" of *Perforations*, within

the *PERIOD*, is calculated randomly and it is always different for each performance.

COUNT-DOWN

To set the value of *Countdown* cheaper, not more than a countdown before the start of the performance, we go into *Settings* and we set the numerical value expressed in *beats* with the buttons increment/decrement:



The speed of execution of the *Countdown* is aligned with that of the *BPM* of the *CYCLE* that we are going to perform. The *default* is 5 seconds.

ACTIVATE THE EXECUTION

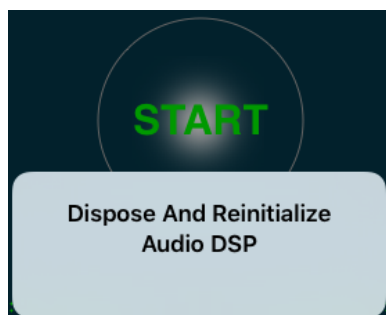
With a *tap* on the **START** button starts the *Countdown*, with decreasing numerical values that appear within the button itself, over which the algorithm is to perform reactions on the active input (**DEMO 1** or **DEMO 2** or **LIVE**) based on the chosen *FORM-SCHEME*:



The button is now in the position:



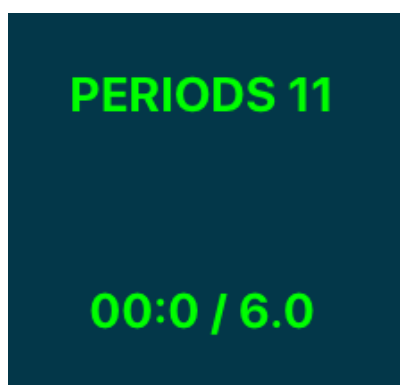
To turn a *tap* on and the **STOP**, appears this message:



CONTROLS USEFUL TO "ONE-MAN-PERFORMANCE"

METRONOMES AND CONTROLS

In the lower left of the graphical **SOLO [Nr.19]** are displayed in two tests

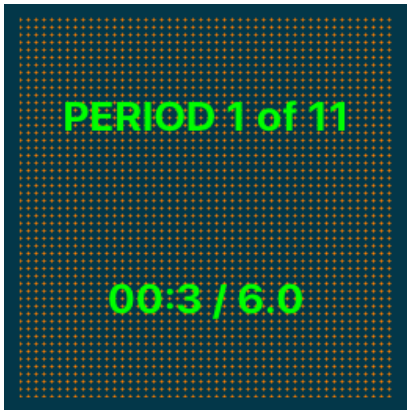


before pressing the **START** key, the first informs us about the number of *PERIODS* (*CYCLE* within the current and the current *FORM-SCHEME*) while the second is waiting to scan the time value of the instantaneous current *PERIOD*.

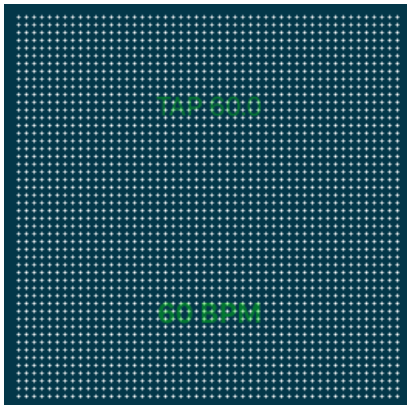
After the **START** the first scans the number of the current *PERIOD* on the total of *PERIODS*, while the second times the time value of current *PERIOD* (*CYCLE* within the current and the current *FORM-SCHEME*) in the form ss:d / s.d



Running in this space is also shown the first metronome (orange) that lights marking the beginning of each *PERIOD*:



After the **START** space in the lower right is displayed the second metronome (white) that lights marking the *beats* of the current *PERIOD*.



CYCLES & PERIODS PROGRESSION

The interface (bottom) is also equipped with two *meters* that display in a linear manner (scrolling from left to right):

a) the passage of time the current *PERIOD* (white):



at the end of the race the color of display changes (yellow) warning us of the impending end of the current *PERIOD*:



and similarly

b) of current *CYCLE* (orange):



Also this *meter* at the end of the race will change color (yellow) warning us of the impending end of the current *CYCLE*:

ASSISTANTS

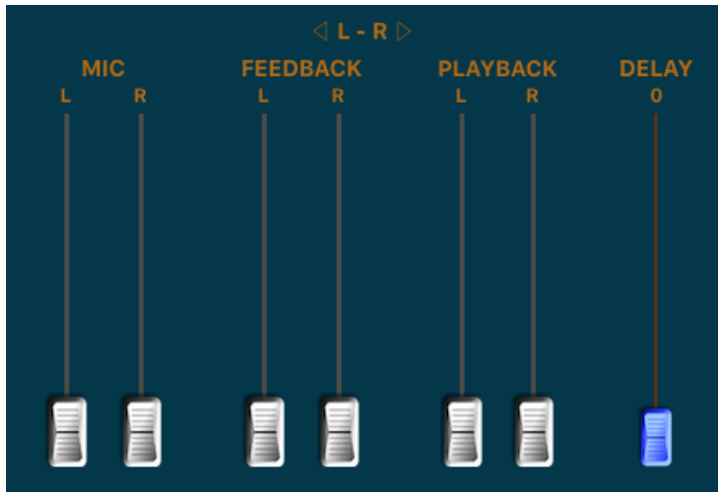
At the center (bottom) are the sliders Assistants:

MIC (*Assistant 1*)

FEEDBACK (*Assistant 2*)

PLAYBACK (*Assistant 3*)

DELAY (*Assistant 4*)



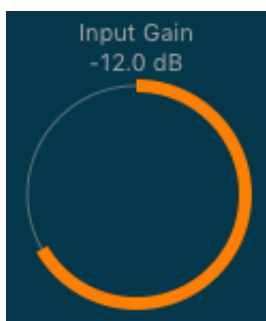
six sliders for the first three Assistants (one per channel) and one slider for the fourth Assistant.

The last slider (in blue) to the right (**DELAY** - *Assistant 4*) marks the value in seconds of the delay-line tap in use. For example: the succession of values of time delay of the *taps* (in sec) for *FORM-SCHEME (Version I)* is the following: 6, 14.2, 19, 25.3, 10.6, 8.

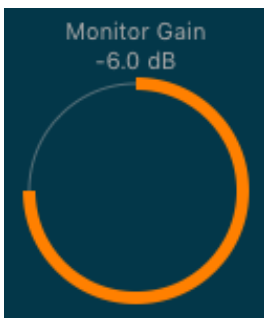
MANAGEMENT OF SIGNAL

In the interface we also have the ability to:

- Increase the mic input with Input Gain (with a range of customizable default values that range from -60dB to +12 dB):



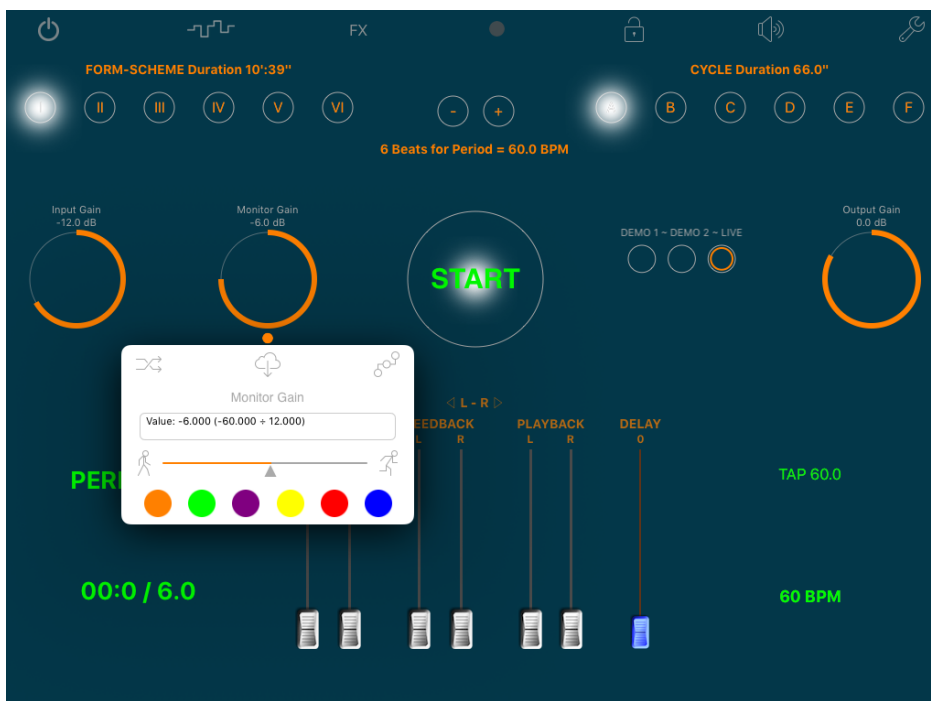
- *Bypass a certain amount of the input signal directly output with Monitor Gain (with a range of customizable default values that range from -60dB to +12 dB):*



- *increase the output signal with the Output Gain (with a range of customizable default values that range from -60dB to +12 dB):*



for these three controls is also provided for the internal management of their operations (such as the example shows the Monitor Gain):

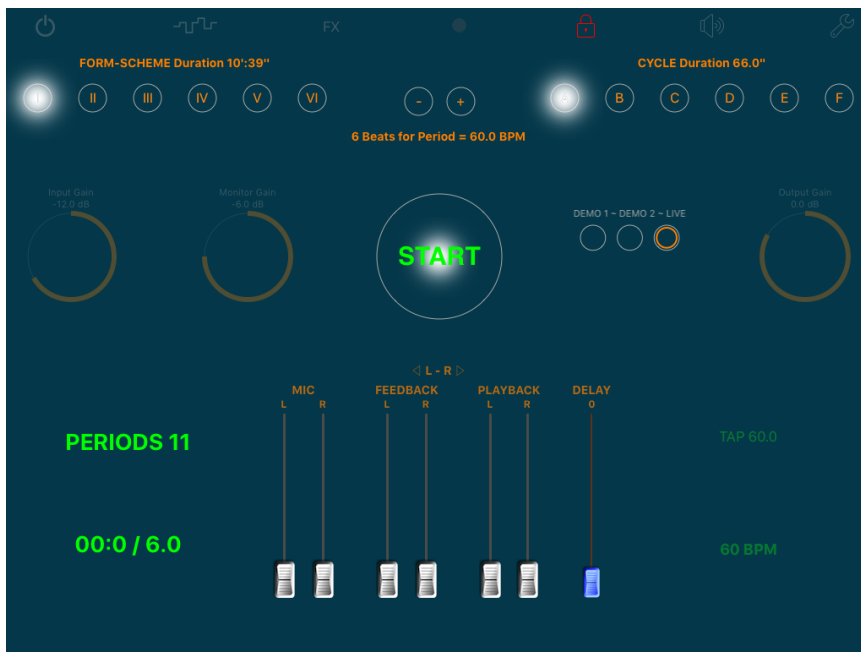


In the setup phase, but also after the **START**, you can **lock/unlock** the three controls (*Input Gain, Output Gain and Gain Monitor*), the controls in the *top bar*, in addition to TAP and BPM, through the button:

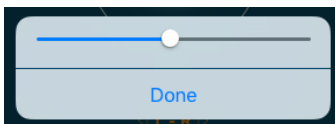


This feature is very useful to prevent accidental *tapping* (**DEMO 1** or **DEMO 2** or **LIVE** mode).

Here is how the interface **locked**:



-  Sets the *iDevice* hardware volume:



- check the overall level of output (L and R) through a *vu-meter*:



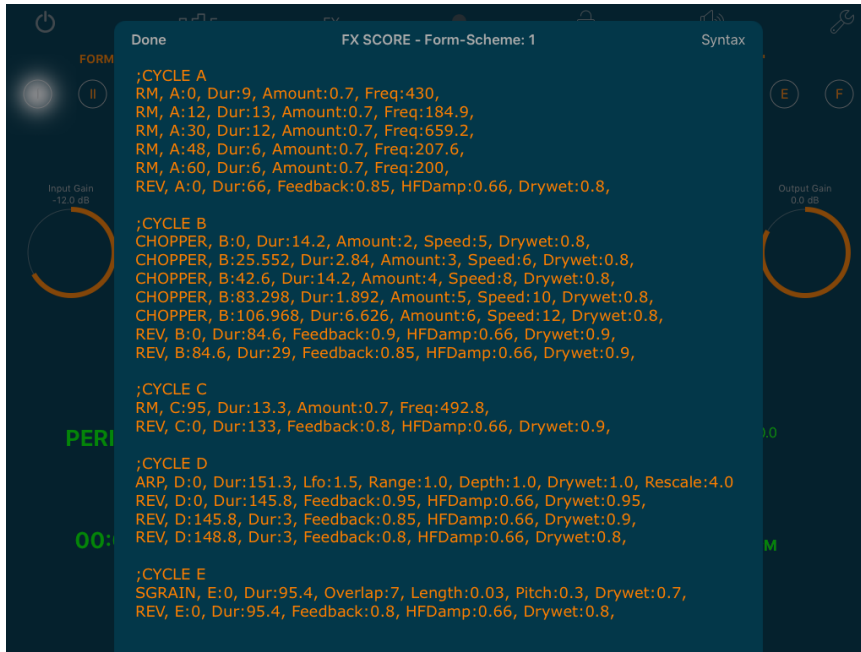
FX (*Fox extended*)

In **SOLO [Nr.19]** is given the opportunity to apply what the author calls *timbres* and *electronic effects*, and interface that go under the name of FX.

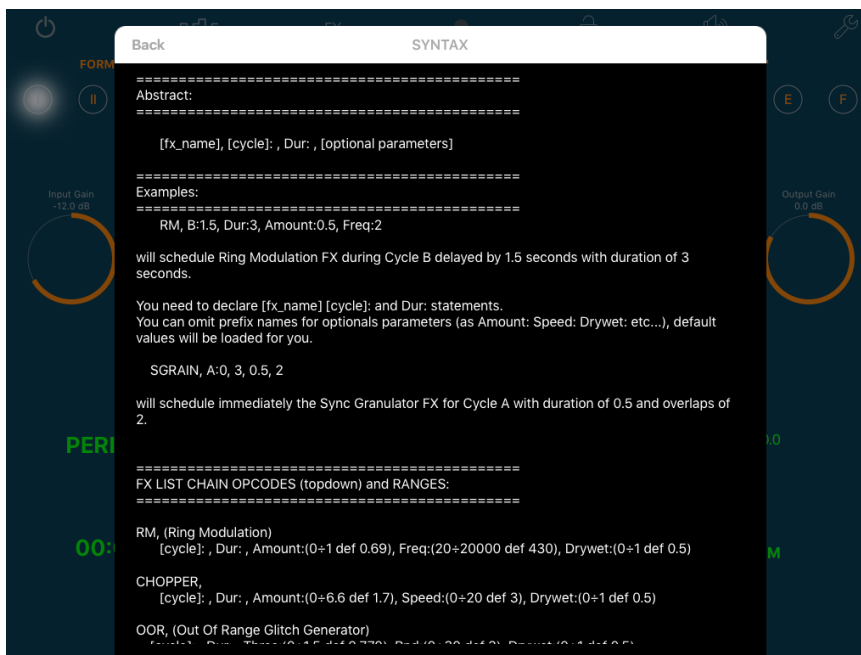
With a *tap* on the button:



opens a *toogle* that represents the editor for the *script* of the effects.



Another *tap* on *Syntax* to open an *abstract* syntax to be followed by the introduction of the effects of the FX *script*:



Obviously the introduction of the *effects* has been previously determined during the development phase of its six pages, depending on the choice of the *timbres* (I, II, III) and their permanence (*action time and duration*).

FX library:

RM (*Ring Modulation*)

CHOPPER

OOOR (*Out Of Range Glitch Generator*)

ARP (*Spectral Arpeggiator*)

SGRAIN (*Sync Granulator*)

REV (*Schroeder Reverberation*)

The last *effect* is also active on the direct signal is not processed by the algorithm.

Here are the details of the *Syntax*⁴:

=====
Abstract:
=====

[fx_name], [cycle]: , Dur: , [optional parameters]

=====
Examples:
=====

RM, B:1.5, Dur:3, Amount:0.5, Freq:2

will schedule Ring Modulation FX during Cycle B delayed by 1.5 seconds with duration of 3 seconds.

You need to declare [fx_name] [cycle]: and Dur: statements.

You can omit prefix names for optional parameters (as Amount: Speed: Drywet: etc...), default values will be loaded for you.

SGRAIN, A:0, 3, 0.5, 2

will schedule immediately the Sync Granulator FX for Cycle A with duration of 0.5 and overlaps of 2.

⁴ Within the Syntax page you can: copy (Copy) text (for example to transcribe templates code within the script) and find the explanation of the terms with Define.

=====

FX LIST CHAIN OPCODES (topdown) and RANGES:

=====

RM, (Ring Modulation)

[cycle]: , Dur: , Amount:(0÷1 def 0.69), Freq:(20÷20000 def 430), Drywet:(0÷1 def 0.5)

CHOPPER,

[cycle]: , Dur: , Amount:(0÷6.6 def 1.7), Speed:(0÷20 def 3), Drywet:(0÷1 def 0.5)

OOR, (Out Of Range Glitch Generator)

[cycle]: , Dur: , Thres:(0÷1.5 def 0.770), Rnd:(0÷30 def 3), Drywet:(0÷1 def 0.5)

ARP, (Spectral Arpeggiator n.b. This effect does not work with Buffer Size <= 128, see Settings)

[cycle]: , Dur: , Lfo:(0.125÷10 def 2), Range:(0.001÷0.999 def 0.2), Depth:(0.7÷1 def 0.9), Drywet:(0÷1 def 0.5), Rescale:(0÷10 def 1.0)

SGRAIN, (Sync Granulator)

[cycle]: , Dur: , Overlap:(1÷10 def 2), Length:(0.005÷0.1 def 0.02), Pitch:(0÷1 def 0.2), Drywet:(0÷1 def 0.5)

REV, (Schroeder Reverberation)

[cycle]: , Dur: , Feedback:(0÷0.998 def 0.6), HFDamp:(0÷1 def 0.35), Drywet:(0÷1 def 0.5)

Once you have edited the FX through the script, running can observe their activations in the string (example):

→ RM → REVERB (Active!)

TIPS & TRICKS

- You should disable the screen rotation, otherwise the accelerometer doesn't work properly.
- Double tap on the **Knobs** to show the options. In the '**Edit**' field you can enter the value from the keyboard. Tap '**Random**' to generate a random value, '**Default**' to reset to the factory settings and configure the knobs dragging behavior. Step fine resolution sets how many points of movement result in a one degree

rotation in the knob's position. Only used in the horizontal/vertical slider modes. You can set Knobs behavior from Settings.

- Long Press Tap on the **Toggles** to open Control Manager Assignations.
- **SOLO [Nr.19]** can play audio in the background and/or mix it's own audio with other iOS apps. **SOLO [Nr.19]** also supports Apple IAA (i.e. Inter-App Audio for iOS 7 or higher) and [Audiobus](#), you can send and receive audio through applications.



Captures Output stereo signal (post Effects) and creates an audio file *wav* format 44.1 kHz 16 bit in the Documents Folder. Files created by **SOLO [Nr.19]**, are composed by id-name followed by a number, for example:

Output Rec 1.wav, Output Rec 7.wav etc...

Every new file will be numbered with the highest number on the list.



Sets the iDevice hardware volume.



Tools

Defaults Reset all default parameters, a warning message will be displayed.

By *default* when you first start, **SOLO [Nr.19]** is set for the implementation of *FORM-SCHEME (Version)*, starting from the first *CYCLE (A)*, first *PERIOD*.

Also by *default* the number of *Beats for PERIOD* and the metronome speed (*BPM*) are so set on the six *FORM-SCHEME (Versions I, II, III, IV, V and VI)*:

FORM SCHEME - VERSION I (also valid for the DEMO 1 version)

	<i>beats</i>	<i>BPM</i>
CYCLE A:	6	60
CYCLE B:	15	63.4
CYCLE C:	10	31.6
CYCLE D:	10	23.7
CYCLE E:	9	50.9
CYCLE F:	12	90

FORM SCHEME - VERSION II (also valid for the DEMO 2 version)

	<i>beats</i>	<i>BPM</i>
CYCLE A:	12	60

CYCLE B:	24	60
CYCLE C:	6	60
CYCLE D:	9	63.5
CYCLE E:	34	59.6
CYCLE F:	18	63.2

FORM SCHEME - VERSION III

	<i>beats</i>	<i>BPM</i>
CYCLE A:	30	59.2
CYCLE B:	9	60
CYCLE C:	20	59.1
CYCLE D:	13	57.8
CYCLE E:	6	60
CYCLE F:	45	59.2

FORM SCHEME - VERSION IV

	<i>beats</i>	<i>BPM</i>
CYCLE A:	40	52.6
CYCLE B:	6	60
CYCLE C:	12	53.3
CYCLE D:	18	53.2
CYCLE E:	8	53.3
CYCLE F:	28	5.3

FORM SCHEME - VERSION V

	<i>beats</i>	<i>BPM</i>
CYCLE A:	22	57.9
CYCLE B:	44	57.9
CYCLE C:	11	57.9
CYCLE D:	8	60
CYCLE E:	32	60
CYCLE F:	16	60

FORM SCHEME - VERSION VI

	<i>beats</i>	<i>BPM</i>
CYCLE A:	14	59.2
CYCLE B:	19	60
CYCLE C:	45	59.2
CYCLE D:	34	59.6
CYCLE E:	25	59.3
CYCLE F:	10	56.6

Samples

File manager can read/write in the Documents folder, you can also share files through AudioCopy, Document Picker (iCloud, Dropbox etc...) and AudioShare. File manager can also read the Built-in files, but you cannot rename or delete them. Touch on a file in the list to show the actions popover. Thus is possible **Load** file in the sampler, Open In... contains all system actions over the file, according to file type. For example you can send files as an e-mail, open file with a listed app etc...



You can rename files in Documents Folder, if you omit the file extension File Manager will fill the original audio file extension for you.

n.b. File manager shows only audio files supported, the other files in Documents folder, will not be seen.



You can pre-listen to the samples before you load them.



Access to the iPod library, this allows the application to read: user's songs, audio books, and audio podcasts. The file can be converted in .wav (PCM 16 bit 44.1 kHz) or .caf formats and stored in the local app Documents Folder. If the file exist at path, progressive number will be attached to the file name.



Starts audio capture from the microphone immediately and save new stereo audio file in wav format 44.1 kHz, 16 bit in the Documents Folder. Files created are composed by id-name followed by a number, for example:

Input Rec 4.wav, Input Rec 65.wav etc...

Every new file will be numbered with the highest number on the list.

You can load directly from the Audio Grab when stop the recording, if you have selected one of the two samplers.

n.b. Depending on the device input audio, you will see one or two V-Meters for the Left/Right channels, it means you will record a mono or stereo file in wav format 16 bit 44.1 kHz. Grab Audio always records signal from iPad microphone, even if your are connected in Audiobus or Inter-App Audio.



Paste Audio file from AudioCopy 2.0 clipboard.



You can import files from AudioCopy, Music Library, Document Picker (iCloud, Dropbox etc...) and AudioShare.



You can edit the File manager contents for deletion.
n.b. This is a destructive action.

Use iTunes to share files between your computer and your device

- 1) Open iTunes on your Mac or PC.
- 2) Connect your iPhone, iPad, or iPod touch to your computer using the USB cable that came with your device.
- 3) Click your device in iTunes.
- 4) In the left sidebar, click File Sharing.
- 5) Select an app to see which files are available for sharing in that app on your device.

MIDI

You can enable/disable the incoming MIDI signal from all the physical, virtual and network ports available. The MIDI Clock is managed by SYNC, see the SYNC section for more details.



If the Core MIDI Network is connected a wifi symbol will appear in the Network port cell.

Channel sets the input channel for the selected port, the *SOLO* Pad will receive the Note On/Off messages in OMNI (from all 16 channels) or, one of the 16 channels.

n.b. *SOLO* does not use MIDI Note On/Off messages.

n.b. *SOLO* does not utilise the incoming Midi Clock.

Enable nRPN

NRPN stands for "Non-Registered Parameter Number" and is part of the MIDI specification for control of electronic musical instruments. NRPNs allow for manufacturer-specific or instrument-specific MIDI controllers that are not part of the basic MIDI standard.

Unlike other MIDI controllers (such as velocity, modulation, volume, etc.), NRPNs require more than one piece of controller data to be sent. First, controller 99 - NRPN Most Significant Byte (MSB) - followed by 98 - NRPN Least Significant Byte (LSB) sent as a pair specify the parameter to be changed. Controller 6 then

sets the value of the parameter in question. Controller 38 may optionally then be sent as a fine adjustment to the value set by controller 6.

This fine adjustment is part of the conventional MIDI controller specification, where any of the first 32 controls can be optionally paired with a control offset 32 higher. This is the rare 14-bit Continuous Controller feature of the MIDI specification, and NRPNs simply take advantage of that existing option in the same way to offer 16,384 possible values instead of only 128.

(<http://en.wikipedia.org/wiki/NRPN>)

When you enable nRPN, you should not employ CC 99 and 98 as normal CC in order to avoid undesired overlap. An nRPN message must be sent with this order: CC 99 98 6 38 where the first couple is the CC number and the second couple is the data. You can MIDI learn as for the normal CC, and nRPN works for all knobs and parameters. Although an nRPN CC number can be greater than 127, you will not be able to learn CC beyond 127.

Control Manager



MIDI



Accelerometer X



Accelerometer Y



LFO

After selecting **MIDI**, **SOLO [Nr.19]** enters into "learn mode" i.e. is waiting for a send message, the type of Control Change (CC are only supported yet). Now it's possible to send messages with MIDI hardware, Virtual Midi and Network Midi. Flashing screen means the association between the parameter and the MIDI message has been successfully and displays the corresponding control Change and Channel. In alternative you can enter direct CC and Midi Channel number through number keyboard. You will also receive an additional setting: 'Widget Range' that allows you to resize the range of action on the parameter. This is because you can use all 128 steps MIDI to control precisely particular part of the parameters.

Accelerometer X and **Y** are used to control the parameters through the horizontal or vertical inclination of the device.

LFO, Low Frequency Modulation is a table look-up oscillator with seven built-in shapes. The frequency of LFO is expressed in Hz but you can get the period time by dividing $1/\text{freq}$.

Settings

Sampling Rate, for full-range audio, the recommended sampling rate is 44.1 kHz. Using a lower rate **SOLO [Nr.19]** will reduce the number of samples that has to calculate, this will facilitate your Device's burden, but it will also reduce the frequency range. If your device is struggling at 44.1 kHz, you should try a lower rate and/or increase Buffer Size.

N.B. DEMO require Sampling Rate to 44.1 kHz, See Settings for change this value.

Buffer Size set the Audio Device latency. When you run **SOLO [Nr.19]** from **Audiobus**, or **Inter-App** this value will be taken from **Audiobus** or **Inter-App Host**. Warning! When an app runs first and keeps running in the background, the Sampling Rate and Buffer Size will impose for the whole iOS while running.

N.B. SOLO [Nr.19] require a minimum buffer of 512, on iPad two convolution reverb require 1024.

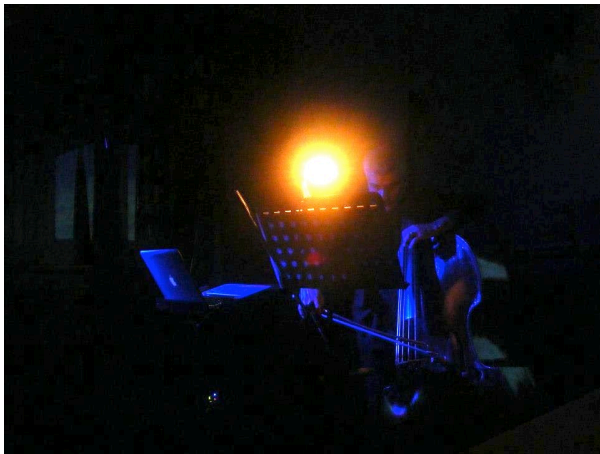
Save/Restore the Last App Setup, when you exit from the application, save the status (all **SOLO [Nr.19]** and Fx Parameters) and reload upon next opening of the app.

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In this digital reconstruction of the original historical technical equipment, I feel compelled to remember the great Maestro **Karlheinz Stockhausen**; to thank the **family Stockhausen**, and the **Universal Edition Wien** for having welcomed my project.

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BIO



Enrico Francioni has achieved degree in double-bass and electro-acoustic music at Music Conservatory "G.Rossini" in Pesaro (Italy).

He interpreted and recorded (WP) the Suite I for double-bass of F.Grillo.

His works have been selected and executed at: Oeuvre-Ouverte (Bourges-France), Festival cinque giornate (Milano-Italy), FrammentAzioni and Taukay Web Radio (Udine-Italy), XVII C.I.M. (Venezia-Italy), VoxNovus

60x60 and UnTwelveMix, Out Of Range (Frontone-Italy), EMUfest (Roma-Italy), ICMC (NYC), BASS2010 (Berlin-Germany), First International Csound Conference (Hannover-Germany), Bellagio Festival (Italy), Sguardi Sonori and DI_stanze (Foligno-Italy), Estudio de Musica Electroacustica (Montevideo-Uruguay), International Csound Conference (Boston-USA), Italian Composers Forum, Acusmatiq 9.0 (Ancona-Italy), Dance Immersion 2014 (Cagli-Italy), Suoni inauditi 2015 (Livorno-Italy), ICSC 2015 (St. Petersburg, Russia), Festival di musica contemporanea 2015 (Cagliari-Italy).

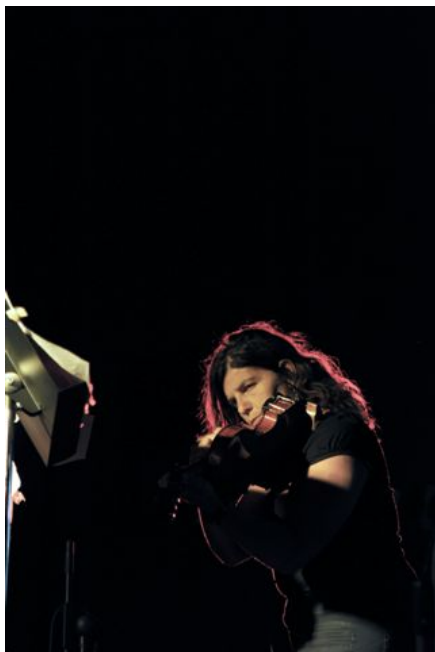
He is author of instrumental chamber music, theater, electro-acoustic, educational and essays (Csound Journal, Edipan, EurArte, Mondo Sonoro and Wicky).

As composer and performer, he was rewarded in international competitions.

He has recorded for Dynamic, Agorà, Orfeo, RSI, RAI, ROF, Agenda.

He was double-bass teacher at Music Conservatory "G.Rossini" in Pesaro (Italy) and he's involved in several educational music activities.

BIO



Simonetta Sargenti was born in Milan. She completed a M.A. Degree in Violin and in Composition at the Conservatory “G. Verdi” in Milan (Italy) and studied Electroacoustic Music at the Sezione Musica Contemporanea, Scuola Civica in Milan and Centro di Sonologia Computazionale in Padua (Italy). As a professional performer, her interests mainly lie in the application of technology to the musical domain, with a special focus on the 20th-century repertoire. Her compositions involve several instrumentations, including live electronics and magnetic tape, and have been performed in several European countries. She recently contributed to audio-visual installations at the Villa Simonetta in Milan. She is active as a musicologist and researcher in electroacoustic music and music theory and analysis. She

held masterclasses on XX century’s music and on her own compositions at the University of Technology in Kaunas (Lithuania), Universidade Catholica Portuguesa in Porto (Portugal), Conservatorio Superior de Musica in Malaga among other. She teaches History of Music at the Conservatory ‘G. Cantelli’ of Novara and History and Analysis of Electroacoustic Music at the Conservatory ‘G. Rossini’ of Pesaro.