

Oakley Sound Systems

Human Voice Module

PCB Issue 1

Builder's Guide

V1.0.5

Tony Allgood
Oakley Sound Systems
CARLISLE
United Kingdom

Introduction

This is the Builder's Guide for the issue 1 Human Voice Module (HVM) from Oakley Sound. This document contains a basic introduction to the circuit board and a full parts list for the components needed to populate the board or boards.

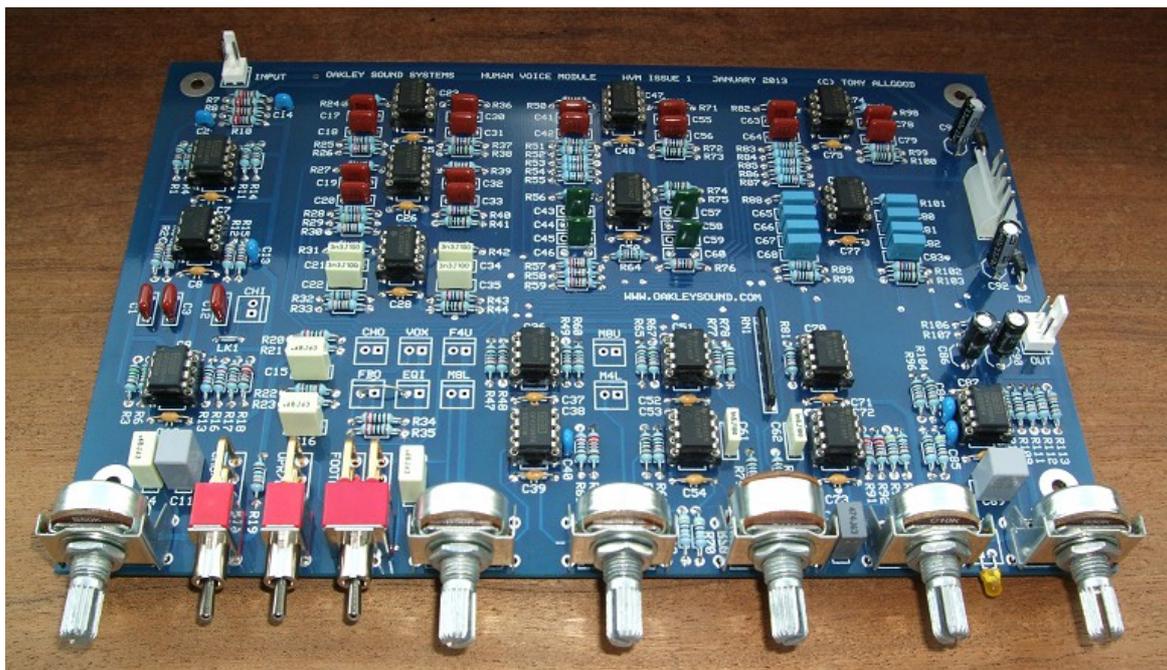
For the User Manual, which contains an overview of the unit and goes into some detail regarding the operation of the module, please visit the main project webpage at:

<http://www.oakleysound.com/hvm.htm>

For general information regarding where to get parts and suggested part numbers please see our useful Parts Guide at the project webpage or <http://www.oakleysound.com/parts.pdf>.

For general information on how to build our modules, including circuit board population, mounting front panel components and making up board interconnects please see our generic Construction Guide at the project webpage or <http://www.oakleysound.com/construct.pdf>.

This is an early version of the documentation. If you do find any errors, even silly little ones, please do let me know either directly by e-mail or via the forum.



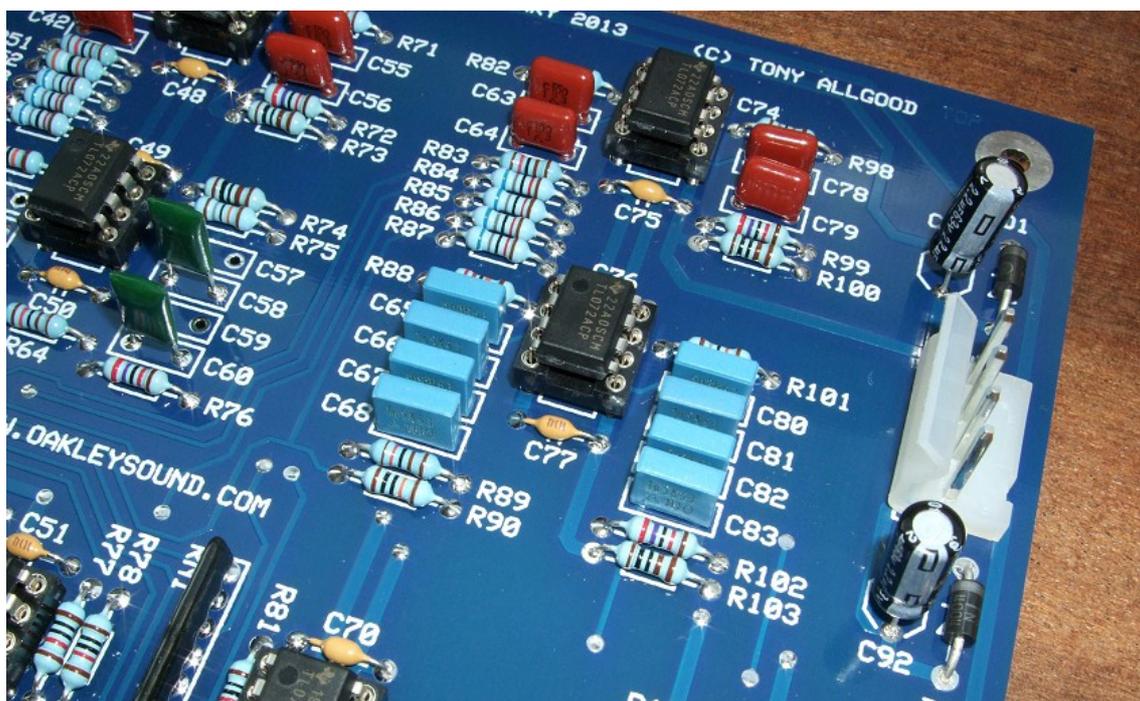
A completed issue 1 HVM circuit board awaiting its case.

The Oakley Sound Human Voice Module

The HVM does not have any built in power supply. It is expected that the circuit board be powered from a regulated split supply of +/-12V to +/-17V. Power is admitted onto the board via a four way 0.156" (2.96mm) header of MTA or KK type.

The module is designed to be powered by our own Rack Power Supply Unit (RPSU) or any other split rail compatible supply such as the Oakley PSU. The benefits of using the RPSU is its small size and ability to take a variety of different AC power sources. Because of the inherent dangers of using a component level mains transformer in your project I recommend that you use an external low voltage AC source such as that obtained from a wallwart or line lump. The RPSU can then be fed from a low voltage alternating current supply so all the high voltages are kept inside the wallwart or line lump. The User Manual for the RPSU will go into more detail of the various options for powering your unit. A single RPSU should be capable of driving both the HVM and SE330 should both projects be built into the same case.

The board size is 198mm (width) x 131mm (depth). Power consumption of the HVM, with no audio passing through it, is +/-80mA at +/-15V.



The power header on the HVM board. This is a 0.156" (3.96mm) Molex KK. This is similar to the MTA156 type connector that we normally use on our MOTM compatible modules but the two are not interchangeable. However, the KK is cheaper and only requires a cheap crimper to make the leads.

Parts List

For general information regarding where to get parts and suggested part numbers please see our useful Parts Guide at the project webpage or <http://www.oakleysound.com/parts.pdf>.

The components are grouped into values, the order of the component names is of no particular consequence.

A quick note on European part descriptions. R is shorthand for ohm. K is shorthand for kilo-ohm. R is shorthand for ohm. So 22R is 22 ohm, 1K5 is 1,500 ohms or 1.5 kilohms. For capacitors: 1uF = one microfarad = 1000nF = one thousand nanofarad.

To prevent loss of the small '.' as the decimal point, a convention of inserting the unit in its place is used. eg. 4R7 is a 4.7 ohm, 4K7 is a 4700 ohm resistor, 6n8 is a 6.8 nF capacitor.

Resistors

All resistors 5% or better 0.25W types. Those items marked with 1% need to be 1% metal film (or better) 0.25W metal film resistors. For the sake of clarity it is recommended to make all resistors 1% metal film types.

47R	R70, R69
110R	R94
220R	R23, R20
220R 1%	R107, R106, R34, R35
1K	R74, R103, R66, R73, R108, R38, R100, R44, R41
1K2	R28, R25, R83, R51, R89, R64, R32
2K2	R16, R61
3K3	R113
4K7	R93, R105
10K	R77, R79, R110, R60, R78, R67, R49, R80, R17, R48, R65, R109, R68, R104, R96, R95, R81, R47
10K 1%	R1, R4, R11, R14
15K	R3, R6
22K	R91, R92, R18, R62
22K 1%	R7, R8, R9, R10
27K	R37, R99, R40, R72, R76, R102, R43
33K	R12, R5, R2, R15
68K	R55, R87, R84, R86, R54, R53, R85
100K	R90, R13, R63, R19, R21, R22, R46, R97
100K 1%	R111, R112
180K	R71, R101, R39, R75, R98, R42, R36
220K	R57, R27, R58, R50, R26, R59, R88, R31, R45, R24, R56, R82
330K	R30, R33
390K	R52
470K	R29

10K x 8 resistor pack. 9 pin single in line (SIL) package RN1

Capacitors

100nF axial multilayer ceramic	C25, C27, C47, C49, C76, C23, C53, C39, C6, C7, C9, C73, C28, C54, C38, C5, C8, C87, C10, C72, C88, C70, C52, C48, C37, C71, C51, C36, C50, C75, C74, C24, C77, C26
33pF 2.5mm C0G ceramic	C40, C84, C85
100pF 2.5mm C0G ceramic	C2, C14
470pF 2.5mm C0G ceramic	C13
1n5 polyester	C61, C62
3n3 polyester	C35, C22, C34, C21
3n9 polyester**	C58, C44, C60, C46
8n2 polyester**	C81, C83, C66, C68
12nF polyester	C63, C64, C78, C79
15nF polyester	C3
18nF polyester	C55, C56, C42, C41
47nF polyester	C33, C19, C20, C32
56nF polyester	C31, C30, C18, C17, C12
100nF polyester	C1
680nF polyester	C16, C15, C69, C4
1uF polyester	C89, C29, C11
10uF, 35V electrolytic	C91, C92
10uF, 16V non polarised* elect	C86, C90

* Non polarised capacitors can sometimes be called bipolar. Unlike ordinary polarised electrolytic capacitors they can be put into the board any way around.

** Both 8n2 and 3n9 may be difficult to find from your normal parts supplier. There are places on Ebay that sell both values, usually the old green cased 'mylar' capacitors, and these seem easy enough to get. However, provision on the board has been made to substitute two smaller value capacitors instead which when combined make up a close approximation of the required value.

1n5 polyester	C68, C66, C83, C81
6n8 polyester	C67, C65, C82, C80
560pF 5mm ceramic or polystyrene	C46, C44, C60, C58
3n3 polyester	C45, C43, C59, C57

Discrete Semiconductors

1N4004 diode	D1, D2
Green or yellow 3mm LED	ON

Integrated Circuits

OPA2134PA	U1, U3, U8, U12, U14, U17
TL072	U2, U4, U5, U6, U7, U9, U10, U11, U13, U15, U16

IC sockets are recommended. You need seventeen 8-pin DIL sockets.

Pots

All pots 16mm Alpha or Alps types.

10K log dual gang	FREQUENCY
10K reverse log	Q
47K linear	LEVEL, OUTPUT, GAIN, WET/DRY

Six Alpha pot brackets.

Six knobs to suit.

Switches

SPDT toggle PCB mounted	CHO/VOX, UPR/LWR
DPDT toggle PCB mounted	FOOTAGE

Interconnects

4 way 3.96mm Molex KK or MTA header	PSU
2 way 0.1" Molex KK or MTA header	INPUT
3 way 0.1" Molex KK or MTA header	OUT

Other interconnects may be required if you intend to use the HVM differently to the suggested design.

Wire Links

These are to be fitted if you are building the suggested design. Use a small piece of uninsulated wire like a resistor wire clipping.

Link out LK1. This connects the pre-amp output to the chorus filter network.

Link pin 2 of FBO to pin 2 of EQ1. Pin 2 is the round pin on each header. This connects the output of the wet/dry balance circuit to the input of the parametric equaliser.

Modifications

The HVM is configured to work well with line level signals like those coming from a mixing desk's auxiliary outputs. It can also be plugged into most synthesiser outputs, including modulators, with no issues. However, you may wish to use the HVM on an output that is weaker than normal. Like, for example, an older machine like the Logan String Melody. In this case it is a good idea to increase the input sensitivity.

The easiest way of doing this is to change the value of R6. This sets the gain of the post input level amplifier. Doubling the value of R6, to say 33K, will double the sensitivity of the input.

If you find the volume level too low when used in the Vox mode then you can increase the value of R81. Currently it's at 10K. Increasing it to 22K will double the gain of the filter network summing amp, based around U13a, and produce an increase in Vox volume. However, I have chosen 10K for a reason; the Vox mode's filter networks do produce some very large increases in signal level at certain specific frequencies which may distort the signal. Feel free to experiment though.

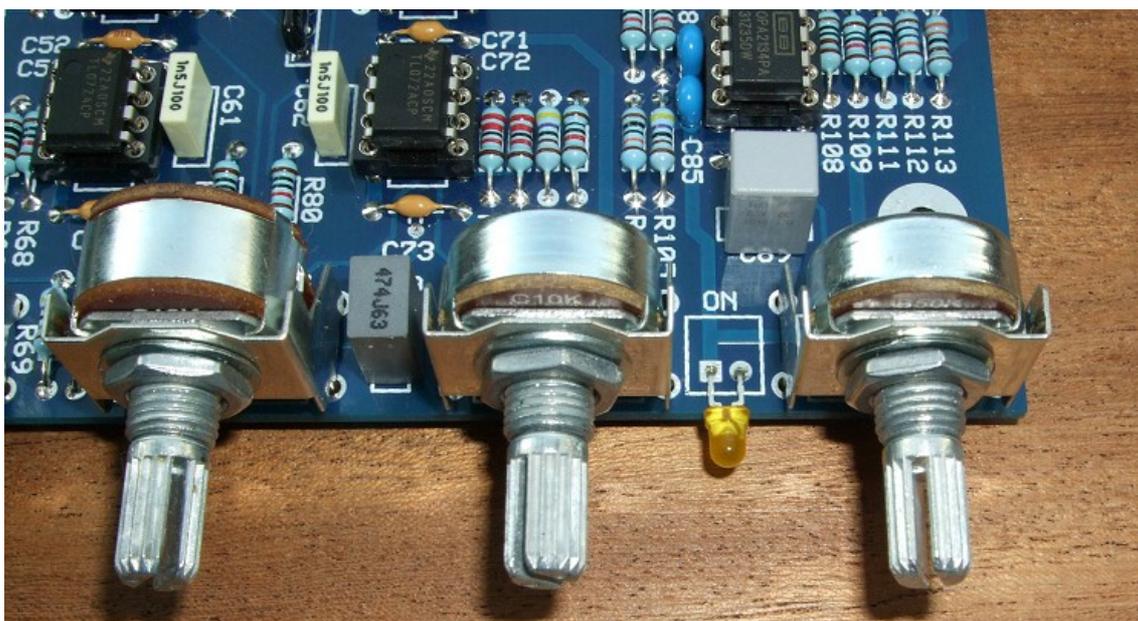
If you do make some interesting changes to the circuit do let the Oakley community know via our friendly online forum at Muffwiggles. I, for one, am always interested to hear what people do with their Oakley Sound projects.

Mounting the Pots, LEDs and Switches

If you are using the recommended Alpha pots, then they can support the PCB with specially manufactured pot brackets. You may not normally need any further support for the board although I do tend to use the rear pair of mounting holes and attach these to threaded hex spacers which then attach to the bottom panel of the case. When constructing the board, fit the pot brackets to the pots by the nuts and washers supplied with the pots. Now fit them into the appropriate holes in the PCB. But only solder the three, or six, pins that connect to the pot. **Do not** solder the pot bracket at this stage. When you have soldered all the pots you can fit the board to your front panel. Position the PCB at right angles to the panel, the pot's own pins will hold it fairly rigid for now. Then you can solder each of the brackets. This will give you a very strong support and not stress the pot connections.

The Alpha pots are labelled with an A, B or C suffix. For example: 50KB or 10KC. Alpha and ALPS use the key; A = log, B = linear and C = reverse log. So a 47KB is a 50 kilohm linear pot.

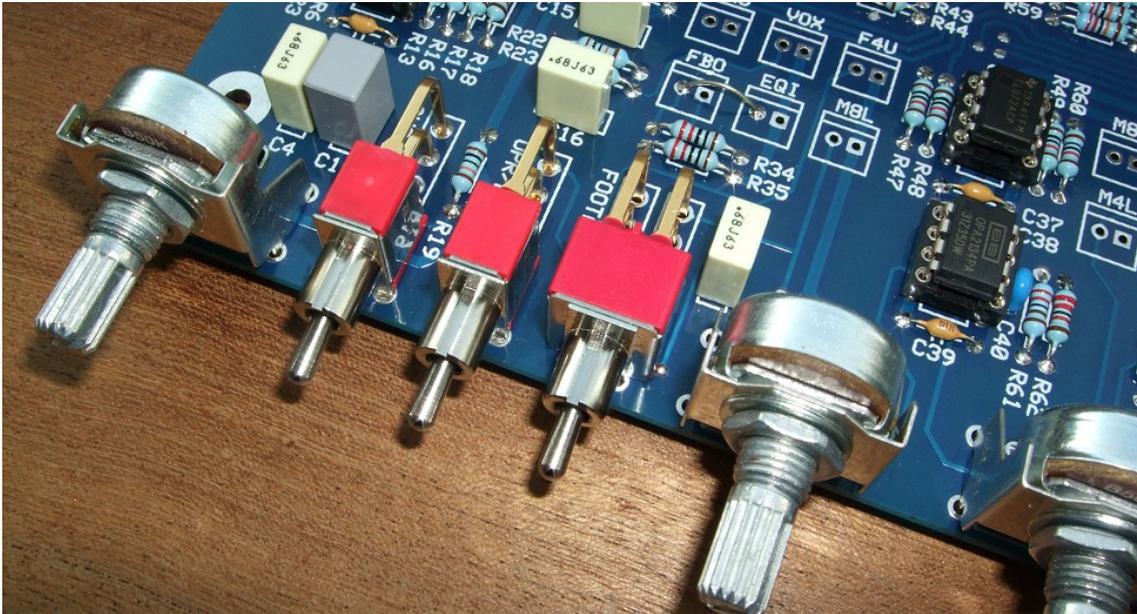
The single board mounted LED must be fitted carefully if you are using the directly mounted technique. Although this sounds fiddly, it's actually quite easy and it reduces wiring, interference and possible errors.



A close up of the LED. This should light up when the power is on. You can chose any colour you wish but I tend to like either green or yellow.

Remove the front panel so that you just have the board again. Get the LED and find the cathode. Make sure the cathode of the LED will go into the square pad, pin 1, on the board. Carefully bend the LED's legs at a point 6mm away from the plastic body of the LED. The legs should be bent by 90 degrees so that the legs are pointing straight down. Check to see if they fit into the board. The bottom of the LED's body should fit just flush to the board edge. Fit the LED to the board but do not solder it in at this stage. Let its legs poke through, there's no need to cut them down yet. Now fit the front panel again to the board and tighten the pot

nuts. You should find that the board now fits snugly into position and the LED should be just poking out of its hole neatly, albeit loosely. Align the LED if it isn't quite straight and then solder it, trimming its leads nice and short afterwards.



The three switches. These are made by Multicom (part numbers 1MS1T2B4M7RE for the two SPDT and 1MD1T2B4M7RE for the single DPDT.)

With panel removed once again, you can now fit the switches. The PCB mountable switches should fit tightly into their respective holes on the board. Indeed, they'll probably work for short time without soldering them so don't forget to do so.

Make sure the switch body is flat against the board. Now refit the front panel and make sure the round switch barrel fits into its hole in the front panel. Now solder all the pins on each of the switches including the two securing pins to the front.

That completes the soldering of the front panel components.

Power Supply

Power is admitted onto the board using the usual Oakley power system. That is a single four way 0.156" (3.96mm) MTA or Molex KK header. For the rack projects I am recommending the Molex system over the MTA simply because the tool needed to make the interconnects is much cheaper. It is also possible to solder the crimp terminals used in the KK system. It is, of course, quite possible to solder your power leads straight onto the board. The board is protected against reverse polarity with a couple of diodes. Note that these diodes work by shorting out your power supply in the event of a reversal. Because of this they should not be relied on if the reversal continues for a long time as this will stress both the diodes and the power supply. The diodes are simply there to save the board's ICs from certain death in the event of the power being inadvertently reversed during testing. There is no overvoltage protection to be careful not to put anything over +/-17V into the board.

The pin out for the power connector, PSU:

Pin 1 +12V to +17V
Pin 2 Ground (0V)
Pin 3 Ground (0V)
Pin 4 -12V to -17V

Pin 1 is indicated by the little diagonal on the header's legend.

Both ground connections need not be connected to your power supply. However, you do most certainly need at least one of them – either pin 2 or pin 3 will do if you using only one connection.

A special compact low voltage supply has been made for this, and other Oakley rack projects, in the form of the Oakley RPSU. You don't have to use this as any decent split rail supply will do but it is an easy build and is flexible enough to suit most builders' needs. Please see the RPSU Builder's Guide and User Manual for more details.

Inputs and Outputs

If you are building the suggested design then you have only one input and one output. Both input and output are balanced and you should use TRS jack sockets. TRS, tip-ring-sleeve, sockets are ones with three poles and are also used for headphone outputs.

Even though both input and outputs are balanced they are still compatible with unbalanced signals. This means you can plug ordinary two pole mono jacks into the TRS sockets and the unit will work as expected.

Audio projects should really be built into metal cases to prevent interference and because of this we now face a bit of a conundrum. Ideally, balanced input and output sockets should be grounded to the case. If you have internal mains transformer then the case will be grounded via the power supply so this will ground the socket and any connecting cables. But this only works if your sockets have a metal frame and can be guaranteed to make a permanent electrical connection to the casing when they are tightened up. It also means that your PSU must be grounding the case directly which, unlike with the internal mains transformer, is not what I recommend when using an external wall wart.

Therefore I prefer to use plastic framed chassis mounted sockets for my rack projects. This allows the case to be grounded via the power supply and keep the sleeve terminals of all the sockets locally isolated from the case. Grounding the sockets must therefore be done via a short wire link back to either the PSU's main ground star point (which is not so easily done on the RPSU if both four way headers are being used) or the main HVM PCB.

Input Socket

The two way header, named INPUT, should be wired so that pin 1 goes to the ring of the input socket and pin 2 goes to the tip. The sleeve, or ground, of the input socket connects to the sleeve of the output socket.

Output Socket

The three way header, named OUT, should be wired so that pin 1 connects to the tip of the output socket and pin 3 goes to the ring. The sleeve, or ground, of the output socket connects to pin 2 of the OUT header.

Other I/O Options

The HVM is designed to be very flexible. It has four main circuit parts and all are available with individual inputs and outputs should you wish to use them. I cannot go into every possible combination since I will leave this to the builder of the unit to decide. However, I will give a quick run down of what is available and a description of the slightly cryptic nomenclature of the headers on the board. It is advisable that if you are planning on some cunning interfacing that you are able to read the schematic well.

The four main circuit sections are pre-amplifier, chorus filter network, vox filter network and parametric equaliser.

1. Pre-amplifier.

Input via the balanced input of the header INPUT. Pin 2 is hot (tip) and pin 1 is cold (ring).

Output via pin 2 (middle pad) of the UPR/LWR switch. The output is also hardwired to the wet/dry mixer pot's dry end. Thus the pre-amplifier's balanced output is also available via the FBO (filter bank output) header. Pin 2 is hot (tip) and pin1 is cold (ring).

2. Chorus filter network

Input via the CHI (chorus input) header. Pin 2 is signal, pin1 is ground. But this can be, and is in the suggested design, connected directly to the pre-amp output via LK1. If you want to use CHI as a separate input you must remove LK1.

Output via the CHO (chorus output) header. Pin 2 is the signal, pin 1 is ground.

3. Vox filter network

Input via one of four inputs, F4U, M8U, M4L & M8L. Each of these inputs represent the connection that was made on the original VP-330. F4U being Female 4" Upper, M8L being Male 8" lower, etc. If you have an old string machine with a split keyboard and two available footages these inputs are the ones to use. It is these I used on my own RS-101 upgrade project.

The four inputs are accessed by their own headers. Again, pin 2 is the signal input and pin 1 is ground.

The output of the Vox filter network is available from the VOX header. Pin 2 being the signal and pin 1 being ground.

Alternatively, the outputs from both Vox and Chorus filter networks can be obtained from the wet/dry mixer output, header FBO. The CHO/VOX switch will then determine the choice of filter heard and the wet/dry pot will also allow for the pre-amplifier's output to be mixed in.

4. Parametric equaliser

Input is available via the EQI (equaliser input) header. Pin 2 is the signal, pin 1 is ground.

Output is available via the HVM's main balanced output, OUT. Pin 1 is hot (tip), pin 2 is ground (sleeve) and pin 3 is cold (ring).

Final Comments

If you have any problems with the module, an excellent source of support is the Oakley Sound Forum at Muffwiggler.com. Paul Darlow and I are on this group, as well as many other users and builders of Oakley modules.

I'd love to hear about what you have done with your module. Please do post pictures of your finished module at the Oakley Sound forum on Muffwigglers.

And if you can't get your project to work, then Oakley Sound Systems are able to offer a 'get you working' service. If you wish to take up this service please e-mail me, Tony Allgood, at my contact e-mail address found on the website. I can service either fully populated PCBs or whole modules. You will be charged for all postage costs, any parts used and my time at 25GBP per hour. Most faults can be found and fixed within one hour, and I normally return modules within a week. The minimum charge is 25GBP plus return postage costs.

If you have a comment about this user guide, or have found a mistake in it, then please do let me know. But please do not contact me or Paul Darlow directly with questions about sourcing components or general fault finding. Honestly, we would love to help but we do not have the time to help everyone individually by e-mail.

Last but not least, can I say a big thank you to all of you who helped and inspired me. Thanks especially to all those nice people on Muff's Forum and the Synth-diy and Analogue Heaven mailing lists.

Tony Allgood at Oakley Sound

Cumbria, UK

© March 2013 - updated August 2013

No part of this document may be copied by whatever means without my permission.