

BHI Noise Eliminating DSP LS

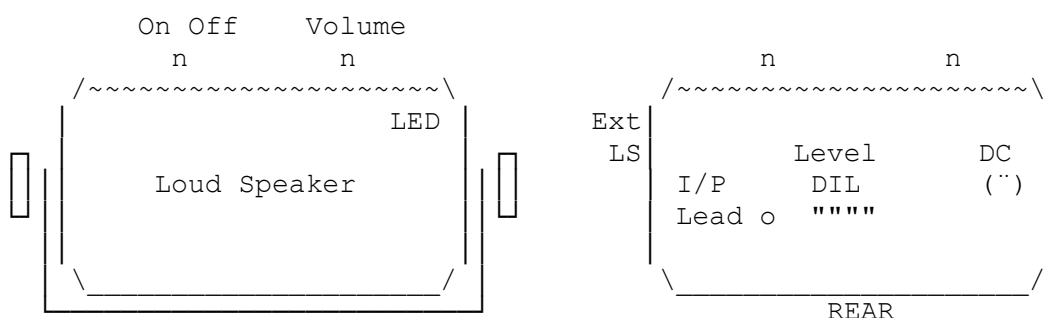
By G8MNY

(Updated Mar 14)

(8 Bit ASCII graphics use code page 437 or 850, Terminal Font)

A few years ago I bought several of BHI NES LS after reading the Dec 2002 Radcom Article by G4HCL, & hearing one actually demonstrated at a rally. The demo is what really sold it to me. I also went to a ham club where one of the 3 man company gave a talk on it's development & the trials & tribulations of getting the product out. It was interesting to learn that the dedicated IC used was designed for noise elimination on fitted car phone microphones & not on Rx noise.

The units have quite amazing performance, compared to other DSP types, don't be fooled by the little amplified speaker box & it's relatively high cost, it really is the state of the art surface mount 64 pin Digital Signal Processing noise reduction unit running at 16MHz. Other models & PCB rig upgrades not tested by me but I assume they are much the same & based on the same IC.



It has 8 levels of noise reduction set up on DIL switch on the rear, as well as an input volume control (does same as Rig Volume!) & a noise reduction on/off switch that also lights a red/green LED visible through the speaker grill. The unit is powered by the usual 2.3mm DC connector (+ in the middle) & it has a series idiot diode for protection as well as a 12V regulator. In the OFF mode the unit is just an amplified speaker, with the time delay, frequency response & input impedance seen below. In the ON mode there is software AGC as well as the effective noise & tone eliminating function.

Here are some performance figures I measured from my one...

AUTOMATIC NOISE & TONE REDUCTION

DIL	SSB	1kHz	
SETTING	NOISE *	TONE	
1 Min	9 dB	4 dB	Best no DSP artifacts setting
2	11 dB	5 dB	
3	12 dB	6 dB	
4	13 dB	8 dB	
5	16 dB	16 dB	
6	19 dB	21 dB	
7	24 dB	25 dB	
8 Max	33 dB	65 dB	Hi Artifacts, no squelch needed mode.

* Noise & Tones was measured on a flat averaging analogue meter.

At setting 8 MAX there is no need to use a SQUELCH on FM, in fact a chopping squelch is unhelpfully as the DSP retraining time for the noise change causes the noise to jumps on & fade off. The only problem using it at max under noisy conditions, is knowing when an over has occurred as you are left with almost silence, just the DSP remnants musical tones peaking at -33dB.

FREQUENCY RESPONSE (external Hi Z connections) 25Hz - 4.5KHz @ -3dB. Which makes it quite acceptable for non HiFi Broadcast Radio signals. A bigger external LS is needed to hear any bass!

INPUT IMPEDANCE is 10k Ohms. (20k on cheaper model)

TIME DELAY was 27ms, which I guess is the time to go from Analogue to Digital & back to Analogue.

In the off mode the audio still goes Analogue-Digital-Analogue but without any noise processing, you still get the frequency filtering due to the 12kHz sample rate, e.g. a 6kHz turnover frequency & filtering that gives -3dB @ 4.5kHz. The processor clocks at 16MHz. Higher clock rates (I tested up to 55MHz) give proportional higher frequency responses, but after about 32MHz (-3dB @9MHz) much of the improvement was lost in unwanted musical effects.

IN USE

I found that you soon get used to the musical artifacts (like digital phones) that the highest level of DSP gives you on very noisy signals, as these are much preferred to normal very noisy audio. At moderate eliminating levels, moderate noise reduction occurs & the recovered speech is more like normal. With extremely noisy weak signals (0dB S/N) the recovery process starts to fail & little usable audio comes though, at these levels straining with the normal signal MAY JUST be better.

The tone reduction feature may be of specific interest to HF/VHF operators who have computer birdies etc. on channel. This unit can automatically eliminate carriers by up to 65dB if needed in under a couple of seconds, allowing weaker Dx to be worked underneath, although the tone will slightly reappear on speech peaks or changing noise floor.

There is a software AF AGC both on processing & non processing modes so too loud a level will cause this to give the odd "click" on heavy AGC action.

ON CW as long as the wanted signal is stronger than others in the Rx passband normal CW will be heard totally noise free. Slow Morse having longer than 1 sec CW elements will be attenuated as the DSP action tries to remove it. Also strong signals may have clicky starts again as the DSP gets it wrong.

ON AM/SSB/FM there is little difference in the audio quality just much reduced background noise. With music, there is a tendency to eliminate all but the loudest instruments.

WITH TV SOUND

On sports commentary, the background crowd noise is well reduced, making the commentator's voice stand out rather than be drowned out, something deaf people may find very useful. For HIFI use I did try higher clocks up to 55MHz where it gave flat response to 15kHz, but the artifacts became quite noticeable. However with 32MHz clock a reasonable 9kHz treble compromise sounded OK to me.

HAM ATV SOUND

I have to listen to several channels at a time when using 23cms ATV repeater GB3HV :- 144.75 FM talkback locally, 5.5MHz subcarrier carrying Engineering sound 144.75 Rx @ GB3HV, as well as 6MHz subcarrier HiFi Sound. So using several of the cheaper NES boxes is a option for me.

THE PROBLEMS

My main gripe with the unit is the use of a tiny DIL switch to control the depth of noise reduction. Although the maker suggest most people leave this set at level (cheaper model). This should really have been a easy access BCD switch as off mode is useful in tuning in SSB. I have done this modification to the cheaper model, as well as increased the 100uF LS cap to 470uF & made the unit operate as a dumb LS when there is no power.

Other problems are slight RFI from the unit if used close to your Rx aerial, (e.g. when using Handhelds). And it has slight susceptability to alternator whine when /M on Tx, this is down to the series diode & 12V reg not doing anything as there is no voltage headroom!

Also be aware when running from same PSU as HF rig, as a few mV drop on rig earth lead, will appear in LS input lead & sound like Tx RFI in the LS., In that case try no PSU earth to LS, only LS earth from rig.

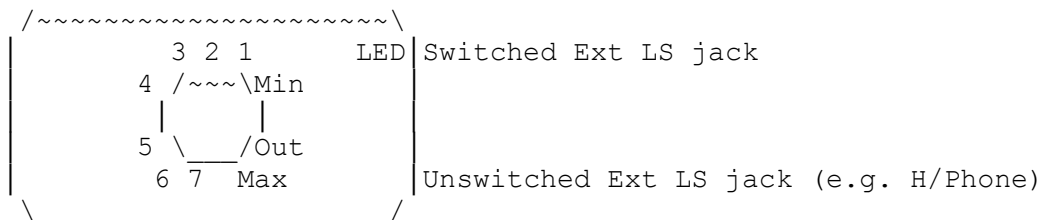
For more information on the unit see www.bhinstruments.co.uk or see the RADCOM article.

(P.S. I have no commercial interest in this product/co, this is just my) (personal experiences of these units, that I hope others may be interested in.)

MY MODIFICATIONS

Five modifications can all be done to the cheavit WITHOUT the on off processing switch, DIL processing level preset switch or volume control.

- A/ Add centre speaker 10 way noise processing level & in/out control switch.
- B/ Add dual Green ON & Red through mode dual LED.
- C/ Add no power LS only mode (uses a DIL relay).
- D/ Improve LF LS output, change 100u for 470u
- E/ Add 2 External LS sockets, one with switched LS.



Tools needed: 10mm Punch (10mm drill body & a metal hole)
 LED & 3.5mm Jack holes can be made with soldering iron/drill.
 Anti Static earthed soldering iron.
 Use Anti Static handling precautions.

- Parts Needed:
- BCD 10 position switch (RS cat 327-938)
 - Small shrouded skirt knob.
 - 15cm of 5 way coloured ribbon wire.
 - Dil 11V C/O reed Relay, & metal screen.
 - Dual colour LED (-ve common).
 - Small PNP transistor.
 - 2x 3.5mm surface mounting jack socket.
 - Small 470uF 10V cap.
 - Several colours of thin connecting wire.

function, green otherwise.

12/Wire up the LS live connections via the relay to switch to through on power off. Picking up external input on bottom of input attenuator P1 r.h.s. R.
(Note step 13 when wiring to the relay.)

Test that the relay puts LS signal though when the power is off & unit functions correctly when powered.

13/Assemble & checking nothing fouls. Test for LS magnet - relay interaction. If there is (e.g. not switching audio on power on/off) make a magnetic shield for the relay, e.g. cut & bend a steel lamination or tin can, to rap around the LS side of the relay. Also try a lamination strip shield taped to the LS magnet.

14/Make 2 wholes for an external LS 3.5mm jack sockets on the LHS by the upper & lower pillars leaving just enough room to clear the pillar but still clear the LS! Wire one up to up from the Amp & 2nd to cut of off the LS.

See my Tech Bul on "Rig DC Power & RF Hazards".

Y don't U send an interesting bul?