## ANAN-8000DLE Test Report

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Figure 1: Apache Labs ANAN-8000DLE SDR Transceiver.



*Introduction:* This test report presents results of an RF lab test suite performed on an Apache Labs ANAN-8000DLE 200W direct-sampling/DUC SDR transceiver loaned by Apache Labs.

The Orion Mk II receiver/exciter board in this DUT uses the Crystek CVHD-950 TCXO. No noise-floor degradation with dither enabled was observed in the ANAN-8000DLE.

Software versions: Iss. 1: PowerSDR OpenHPSDR mRX PS v3.4.2

Firmware versions: Iss. 1: Orion MkII v1.6 (Protocol 1)

Performance Tests conducted in my home RF lab, July 5-31, 2017.

## A. Receiver 1 (RX1) Tests

*Note:* Frequency and level calibration (10.000 MHz, -70 dBm) performed at start.

1: MDS (Minimum Discernible Signal) is a measure of ultimate receiver sensitivity. In this test, MDS is defined as the RF input power which yields a 3 dB increase in the receiver noise floor, as measured on the S-meter.

*Test Conditions:* ATT as shown, NR off, NB off, ANF off, AGC Fixd, AGC Gain = 120, RX1 Meter: Sig, Avg., Dither off, Random off.

	3.6 N	ЛНz	14.1	MHz	28.1	MHz	50.1 MHz		
ATT dB	SSB 2.4kHz	CW 500Hz							
0	-126	-133	-126	-133	-126	-134	-135	-142	
-20	-106	-113	-105	-113	-106	-113	-120	-127	

 Table 1: MDS<sup>1</sup> in dBm (RX1)

Notes: 1. Dither and/or Random do not affect MDS.

2. Bypassing Alex preselector does not affect MDS

3. ADC clip level: -5 dBm

2: Reciprocal Mixing Noise occurs in a direct-sampling SDR receiver when phase noise generated within the ADC mixes with strong signals close in frequency to the wanted signal, producing unwanted noise products at the IF and degrading the receiver sensitivity. Reciprocal mixing noise in a direct-sampler is an indicator of the ADC clock's spectral purity.

In this test, a Wenzel 5 MHz OCXO with low phase noise is connected via a 7 dB pad and a 0-110 dB step attenuator to the DUT (ANT). The noise floor is read on the DUT Smeter in CW mode (500 Hz) with ANT terminated in 50 $\Omega$ . The input power P<sub>i</sub> increased to raise detected noise by 3 dB. Reciprocal mixing dynamic range (RMDR) = P<sub>i</sub> – MDS.

*Note:* The residual phase noise of the OCXO is the limiting factor in measurement accuracy.

*Test Conditions:* 5.000 MHz, 250 Hz CW, ATT 0 dB, NR off, ANF off, NB off, negative offset. AGC Fxd, AGC Gain 120. Dither off, Random off. BH-4 receive filter window, sample rate 192K, buffer size 1024, filter size 4096. RMDR *in dB* = input power ( $P_i$ ) – MDS (*both in dBm*). Here, MDS = -136 dBm (B = 250 Hz). RMDR = Pi - MDS. Phase noise = -(RMDR + 10 log B).

Offset kHz	P <sub>i</sub> dBm	RMDR dB	PN dBc/Hz
0.5	-27	108	-132
1	-22	113	-137
2	-19	116	-140
3	-17	118	-142
5	-15	120	-144
10	-13	122	-146
20	-10	125	-149
30	-8	127	-151
50	-6	129	-153

 Table 2: RMDR & Phase Noise.

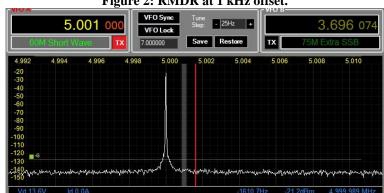


Figure 2: RMDR at 1 kHz offset.

**3:** Channel filter shape factor (-6/-60 dB). This is the ratio of the -60 dB bandwidth to the -6 dB bandwidth, which is a figure of merit for the filter's adjacent-channel's rejection. The lower the shape factor, the "tighter" the filter.

In this test, an in-channel RF test signal from the Wenzel OCXO is applied at -50 dBm. The bandwidths at -6 and -60 dB relative to the input power are determined by tuning the receiver across the signal and observing the S-meter.

*Test Conditions:* 5000 kHz nominal, SSB/CW modes, ATT = 0 dB, AGC med, NR off, NB off, ANF off, Dither off, Random off. BH-4 filter window. Audio tab: Sample rate 192K, buffer size 1024, filter type Linear Phase. *Note:* Switching to BH-7 window or Low Latency does not affect results.

Table 5: Channel Filter Shape Factors.								
	DSP Filter	Size 2048	DSP Filter	Size 16348				
Filter	Shape Factor	6 dB BW kHz	Shape Factor	6 dB BW kHz				
2.4 kHz SSB	1.06	2.40	1.01	2.40				
500 Hz CW	1.29	0.50	1.04	0.50				
250 Hz CW	1.58	0.25	1.08	0.25				
5 kHz AM	1.03	5.0						

Table 3:	Channel	Filter	Shape	Factor

*3a: Ultimate channel filter attenuation.* This test is conducted with the Wenzel OCXO as in *3.* above. A test signal is applied at a power level of -26 dBm, and the receiver is detuned until the S-meter drops no further. The final S-meter reading and the frequency offset are recorded. *Note:* The channel filters actually have deeper stopbands than measured.

*Test Conditions:* Test signal 5000 kHz at -26 dBm, SSB/CW modes, ATT = 0 dB, AGC med, NR off, NB off, ANF off. BH-4, sample rate 192K, filter size 2048. Switching to BH-7 window does not affect results.

<i>Test Results:</i> 2.4 kHz SSB: S-meter minimum = -117 dBm at 1.65 kHz offset.
Ultimate attenuation = $-26 - (-117) = 91 \text{ dB}$ .
Bandwidth for ultimate attenuation = $2 * 1.5 \approx 3.3$ kHz.

500 Hz CW: S-meter minimum = -123 dBm at 550 Hz offset. Ultimate attenuation = -26 - (-123) = **97 dB**. Bandwidth for ultimate attenuation =  $2 * 0.55 \approx 1.1$  kHz.

4: NR noise reduction, measured as SINAD. This test is intended to measure noise reduction on SSB signals close to the noise level.

A distortion test set or SINAD meter is connected to the DUT audio output. The test signal is offset 1 kHz from the receive frequency to produce a test tone, and RF input power is adjusted for a 6 dB SINAD reading (-122 dBm). NR is then turned on, and SINAD read at various NR settings.

*Test Conditions:* 14.100 MHz, 2.4 kHz USB, BH-4 RX filter, sampling rate 192K, buffer size 1024, filter size 2048, filter type Linear Phase, AGC Med, ATT = 0 dB, NB off, ANF off, NR/ANF Pre-AGC (in DSP Options), Dither off, Random off. Initial NR settings (defaults): Taps 64, Delay 16, Gain 100, Leak 100. (Varying Delay does not significantly affect SINAD readings.)

Table 4: NR SINAD.							
Taps	Delay	SINAD dB					
NR off	16	6					
64	16	13					
128	32	18					
256	16	24					
512	16	30					
1024	16	35					

This shows an SINAD improvement of 29 dB max. with NR at maximum for an SSB signal roughly 4 dB above the noise floor. This is an approximate measurement, as the amount of noise reduction is dependent on the original signal-to-noise ratio.

In on-air listening, NR was very effective in reducing band noise (as long as the desired signal was audible), and did not distort received audio.

5: Auto-Notch Filter (ANF) stopband attenuation. In this test, an RF signal is applied at a level  $\approx$  70 dB above MDS. The test signal is offset 1 kHz from the receive frequency to produce a test tone. ANF is activated and the test signal level is adjusted to raise the audio output 3 dB above noise floor. The stopband attenuation is equal to the difference between test signal power and MDS.

*Test Conditions:* 14.100 MHz, 2.4 kHz USB, sampling rate 192K, BH-4 RX filter, buffer size 1024, AGC med, ATT = 0 dB, NB off, ANF off, Dither off, Random off. Initial ANF settings (defaults): Taps 64, Delay 16, Gain 100, Leak 100. AGC Med, Gain 120 (max).

*Test Results:* Measured MDS = -126 dBm per Test 1.  $P_i = -56$  dBm. *NR/ANF Post-AGC:* Stopband attenuation =  $P_i - MDS = -46 - (-126) = 70$  dB. Tone completely suppressed.

*NR/ANF Pre-AGC:* Tone partly suppressed. Audible artifacts in RX audio output (raspy hum and low-level tone) as depicted in Figure 3.



Figure 3: Spectrum and waveform of RX audio with ANF (Pre-AGC). DS0-X 2012A, MY5546213D: Tue Jul 11 12:13:30 2017

6: Multi-Notch Filter (MNF) Test. Since v3.3.6, PowerSDR OpenHPSDR incorporates a new Multi-Notch Filter (MNF) tool, which allows the user to specify up to 1024 notches (bandstop filters). These notches are specified by absolute RF frequency and width; they will be invoked as needed when they overlap the detection passband. This feature is useful for suppressing interference which consistently appears on specific frequencies.

To avoid phase distortion, the notches are implemented with linear phase. They introduce no additional processing delay, nor do they consume any additional CPU cycles once the receiver is on frequency and the notches are set up. This is all accomplished by simply "cutting" the notches into the existing bandpass, filters rather than adding additional filters. (*Description courtesy Doug Wigley W5WC*.)

To test MNF, it is convenient to generate an FM test signal whose modulating frequency  $f_m$  and peak deviation  $\Delta f$  are chosen to yield a specific number of sideband pairs. The receiver is then set to USB and tuned to the carrier frequency of the test signal. The first notch is set at the carrier frequency  $f_c$ , and subsequent notches are set at  $f_c + nf_m$  where n is the order of the sideband pair. The number of configured notches equals the number of sideband pairs. With the receiver in USB, only the upper sidebands will appear in the detection channel, as illustrated in Figures 5 & 6.

When MNF is inactive, a loud composite tone will be heard in the headphones and the Smeter will indicate the average power of the FM carrier and sidebands. With MNF on, Smeter reading and the audio output should fall to the receiver noise floor.

*Test Conditions:* 14.100 MHz, 2.4 kHz USB, ATT 0 dB, NR off, ANF off, NB off, AGC Med, AGC Gain 120. Dither off, Random off. BH-4 receive filter window, sample rate 192K, buffer size 1024; filter type Linear Phase. S-meter reads Sig Avg. DUT and signal generator clocked from 10 MHz GPS-derived lab standard. Frequency calibration performed on DUT prior to starting test. MNF set up for 21 notches at 100 Hz intervals from 14100.0 (#0) to 14102.0 (#20). See Figure 28.

Test signal:  $f_c = 14100$  kHz.  $f_m = 100$  Hz.  $\Delta f = 1.2$ kHz. These settings yield 20 usable sideband pairs at 100 Hz intervals. The upper sidebands fall within the 2.4 kHz SSB bandwidth.

Options	CW	AGC/ALC	AM/SAM	FM	Audio	EER	NR/ANE	MNF
Options	CVV	AGC/ALC	AM/ SAM	FIM	AUGIO	EEN	IND/AINE	
Multi N	lotch Filt	ter						
Notch	#	C	Center Frequ	uency (	Mhz)	Widt	h (Hz)	
20	*	1 6	4.102000			0	-A	✓ Active
12000	lain.				. Internet	108	hat and	
			VFOA					
_								
	Add		Edit		Delete			ENTER Cancel
_							uation	

*Test Results:* The test results are shown in Figures 5 and 6 below.

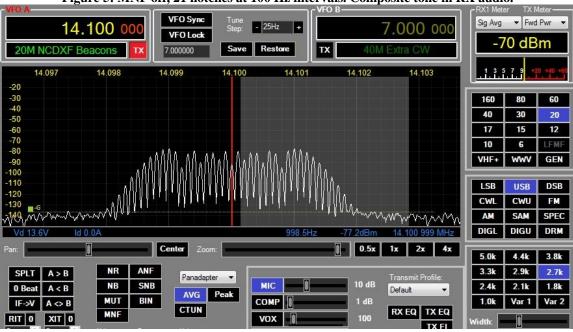
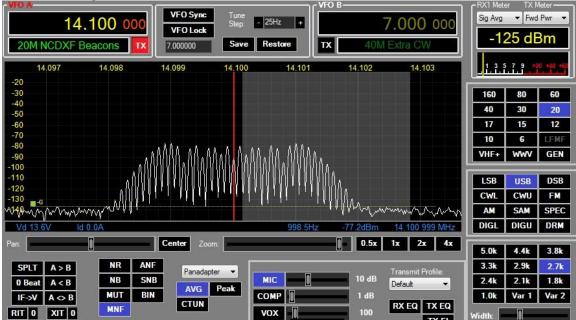


Figure 5: MNF off, 21 notches at 100 Hz intervals. Composite tone in RX audio.

Figure 6: MNF on, 21 notches at 100 Hz intervals. RX audio at noise floor.



7: AGC impulse response. The purpose of this test is to determine the Anan-8000DLE's AGC response in the presence of fast-rising impulsive RF events. Pulse trains with short rise times are applied to the receiver input.

*Test Conditions:* 3.6 MHz, 2.4 kHz LSB, sampling rate 192K, BH-4 RX filter, buffer size 1024, NR on, NB off/on as required, ANF off, ATT= 0 dB, AGC Fast, ANF off, MNF off, NR/ANF Pre- and Post-AGC (in DSP Options), Dither off, Random off. A pulse generator is connected to ANT1 via a step attenuator.

The pulse rise time (to 70% of peak amplitude) is 10 ns. Pulse duration **t** is varied from 20 to 95 ns. In all cases, pulse period  $\tau$  is 600 ms. The step attenuator is set at 36 dB. Pulse amplitude is  $16V_{pk}$  (e.m.f.)

With NR and noise blankers off, each pulse triggers the AGC, creating a "hole" in the background noise for the duration of the AGC hang time. There is no audible "tick" at the leading edge of the pulse for 20 ns  $\leq$  t  $\leq$  1.25 µs. This is an improvement over earlier versions of PowerSDR OpenHPSDR mRX, where ticks were always audible in this state.

The pulses create lines on the waterfall. With NR and NB off, the S-meter flicks up to S7 at  $\mathbf{t} = 20$  ns and S9 at  $\mathbf{t} = 95$  ns.

With NR/ANF Pre-AGC, activating NR suppresses background noise; the effect is to render "ticks" audible. With NR/ANF Post-AGC and t < 100 ns, NR reduces the ticks to quiet "holes" in the receiver audio.

With NR2 on, Pre-AGC, a crushing (crunch) sound is heard for each pulse. NR2 (Post-AGC) reduces the crunch to a quiet "hole" for each pulse..

NB or NB2 suppresses the audible effects of the pulses entirely, whether NR/ANF is Preor Post-AGC. The S-meter, spectrum scope and waterfall responses are suppressed.

*7a: SNB* (*Spectral Noise Blanker*). With NR off, SNB changes quiet "holes" to "thumps". SNB + NR muffles the "thumps". SNB + NR2 produces "clanking" sounds (Pre-AGC) or "thumps" (Post-AGC). Unlike NB, SNB does not suppress scope, waterfall or S-meter reaction to the pulses.

A simulation test was performed by playing the SM5BSZ *agctest-96.wav* file into the receiver (*Ref. 5*). The simulation exercised AGC response to impulses, but not NB. Only NR + Post-AGC suppressed the audible response almost completely, as well as reducing the noise level in the audio output. NR + Pre-AGC was far less effective.

8: S-meter tracking: This is a quick check of S-meter signal level tracking.

*Test Conditions:* 500 Hz CW, ATT = 0 dB, sampling rate 192K, BH-4 RX filter, buffer size 1024, AGC Med., ANF off, Dither off, Random off. RX1 Meter: Sig Avg. *Level calibration* (14.100 MHz, -70 dBm) is performed before starting the test. Next, starting at -120 dBm, the test signal power is increased and the level corresponding to each S-meter reading is noted.

Table 5: S-Meter Tracking. -120 -110 -100 -90 -40 -30 -20 P<sub>i</sub> dBm -80 -73 -10 -60 -50 -5 -119 -109 -99 -79 -72 -49 -39 -29 -19 -9 Rdg.dBm -89 -59 CLIP S3 >S9 S9+13 S9+21 S9+41 S9+52 S9+61 S-meter >S1 <S5 >S6 S8 S9+33

Table 6: S-Meter/ATT Tracking								
P <sub>i</sub> dBm		-73						
ATT dB	0	0 10 20 30 31						
Rdg. dBm	-72	-72	-72	-72	-72			
S-meter	>S9	>S9	>S9	>S9	>S9			

	Table 6:	S-Meter/ATT	Tracking
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9: Two-Tone IMD<sub>3</sub> (IFSS, Interference-Free Signal Strength) tested in CW mode (500 Hz), ATT = 0 dB, AGC Med. Test frequencies:  $f_1 = 14100 \text{ kHz}$ ,  $f_2 = 14102 \text{ kHz}$ . IMD<sub>3</sub> products: 14098/14104 kHz. IMD<sub>3</sub> product level was measured as absolute power in a 500 Hz detection bandwidth at various test-signal power levels and Dither/Random combinations, with 0 dB ATT selected. The ITU-R P-372.1 band noise levels for typical urban and rural environments are shown as datum lines. The S-meter is set at Sig Avg.

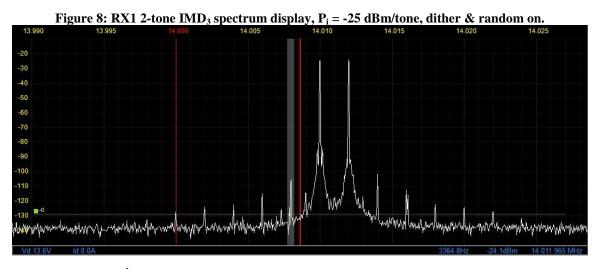
	IFSS (Interference-Free Signal Strength) Apache ANAN-8000DLE, RX1, ATT = 0	
-95 -	SW: PSDR OpenHPSDRmRX v3.4.2, FW V1.6 f1 = 14010 kHz, f2 = 14012 kHz CW, B = 500 kHz 11.07.2017 VA7OJ/AB4OJ	
-100 -		
-105 -		
-110 -		No Dith,
-115 -		Dith + R
-120 -		Rural No
-125 -	A// M/	
-130		
-135 -	70 -65 -60 -55 -50 -45 -40 -35 -30 -2	15 -20 -15
-7	70 -65 -60 -55 -50 -45 -40 -35 -30 -2 Input Power dBm/tone	-20 -15

Figure 7: RX1 IFSS (2-tone IMD<sub>3</sub>) vs. test signal level.

*Notes on 2-tone IMD<sub>3</sub> test:* This is a new data presentation format in which the amplitude relationship of the actual IMD<sub>3</sub> products to typical band-noise levels is shown, rather than the more traditional DR<sub>3</sub> (3<sup>rd</sup>-order IMD dynamic range) or SFDR (spuriousfree dynamic range). The reason for this is that for an ADC, SFDR referred to input power rises with increasing input level, reaching a well-defined peak ("sweet spot") and then falling off. In a conventional receiver, SFDR falls with increasing input power.

If the  $IMD_3$  products fall below the band-noise level at the operating site, they will generally not interfere with desired signals.

The SFDR behavior of an ADC invalidates the traditional DR<sub>3</sub> test for a direct-sampling SDR receiver. Our goal here is to find an approach to SFDR testing which holds equally for SDR and legacy receiver architecture. See *Reference 4*.



10: Two-Tone  $2^{nd}$ -Order Dynamic Range (DR<sub>2</sub>). The purpose of this test is to determine the range of signals far removed from an amateur band which the receiver can tolerate while essentially generating no spurious responses within the amateur band.

In this test, two widely-separated signals of equal amplitude  $P_i$  are injected into the receiver input. If the signal frequencies are  $f_1$  and  $f_2$ , the 2<sup>nd</sup>-order intermodulation product appears at  $(f_1 + f_2)$ . The test signals are chosen such that  $(f_1 + f_2)$  falls within an amateur band.

The two test signals are combined in a passive hybrid combiner and applied to the receiver input via a step attenuator. The receiver is tuned to the IMD product  $(f_1 + f_2)$  which appears as a 600 Hz tone in the speaker. The per-signal input power level P<sub>i</sub> is adjusted to raise the noise floor by 3 dB, i.e. IMD product at MDS. The P<sub>i</sub> value is then recorded. DR<sub>2</sub> = P<sub>i</sub> - MDS.

*Test Conditions:*  $f_1 = 6.1 \text{ MHz}$ ,  $f_2 = 8.1 \text{ MHz}$ , IMD2 product at 14.2 MHz. 500 Hz CW, AGC slow, ATT = 0 dB, NR off, NB off, CW neutral, ANF off, Alex preselector in or out as required. DR<sub>2</sub> in dB. Measured MDS = -133 dBm (dither & random on). RX S-meter: Sig Avg.

Table 7. KAI DK2. 11. 0.1 WIIIZ. 12. 0.1 WIIIZ. IND product. 14.2 WIIIZ.									
	Preselector	0	ut	in					
Dither & Random	MDS dBm	P <sub>i</sub> dBm	DR <sub>2</sub> dBm	P <sub>i</sub> dBm	DR <sub>2</sub> dBm				
off	-133	-74	59	-39	94				
on	-135	-72	61	-26	107				

Table 7: RX1 DR<sub>2</sub>. f<sub>1</sub>: 6.1 MHz. f<sub>2</sub>: 8.1 MHz. IMD product: 14.2 MHz

It will be observed that when the 20m Alex preselector is switched in, it suppresses the  $f_1$  and  $f_2$  signals. This virtually eliminates the 2nd-order IMD product. Any residual IMD is further reduced by dither and random, which are much more effective when the interfering signals are lower down the IFSS curve.

11: Noise Power Ratio (NPR): An NPR test is performed, using the test methodology described in detail in *Ref.* 2. The noise-loading source used for this test is a noise generator fitted with bandstop (BSF) and band-limiting filters (BLF) for the test frequencies utilized.

The noise loading  $P_{TOT}$  is applied to ANT1 and increased until ADC clipping just commences, and then backed off until no clipping is observed for at least 10 seconds. NPR is then read off the spectrum scope by observation. (NPR is the ratio of noise power in a channel outside the notch to noise power at the bottom of the notch.)

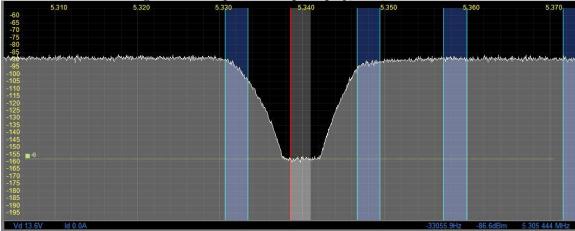
*Test Conditions:* Receiver tuned to bandstop filter center freq.  $f_0 \pm 1.5$  kHz, 2.4 kHz SSB, ATT = 0/20 dB, NR off, NB off, Notch off, ANF off, AGC Med, Alex on. Display Averaging: Log Recursive (default). Test results are presented in **Table 8**.

	Table 8: RX1 NPR Test Results <sup>2</sup>								
	BLF kHz		Alex	Ρτοτ	dBm	NPR	dB <sup>1</sup>	Theor. NPR <sup>3</sup>	
			ATT	0 dB	20 dB	0 dB	20 dB		
1940	602044	29.2	0	-18	-4	71	65	80	
1940	002044	29.2	1	-9	+3	78	68	00	
3886	604100	32.3	0	-17	+1	73	70	76.9	
3000	004100	32.3	1	-15	-4	71	60	70.9	
5340	605600	33.6	0	-17	-1	70	68	75.6	
5540	005600	005000	33.0	1	-15	-4	71	60	75.0
7600	3168160	35.1	0	-17	+3	70	67	74.1	
7000	5100100	33.1	1	-11	0	70	63	74.1	
11700	013000	37.6	0	-16	+4	64	65	71.6	
11700	013000	57.0	1	-13	+5	65	60	71.0	

## Notes on NPR test:

- 1. NPR readings were stable over time.
- 2. Enabling Dither and/or Random did not affect NPR results.
- 3. Theoretical NPR was calculated for the LTC2208-16 ADC using the method outlined in *Ref. 3*. The theoretical NPR value assumes that  $B_{RF}$  is not limited by any filtering in the DUT ahead of the ADC, and that the net gain between the antenna port and the ADC is 0 dB.

Figure 4: NPR spectrum scope display, RX1, 5340 kHz.



## B. Receiver 2 (RX2) Tests

*12: MDS (Minimum Discernible Signal)* is a measure of ultimate receiver sensitivity. In this test, MDS is defined as the RF input power which yields a 3 dB increase in the receiver noise floor, as measured at the audio output.

*Test Conditions:* ATT as shown, NR off, NB off, ANF off, AGC Fxd, threshold AGC gain 120, Dither off, Random off.

	3.6 MHz		14.1 MHz		14.1 MHz 28.1 MHz		50.1	MHz
ATT dB	SSB 2.4kHz	CW 500Hz	SSB 2.4kHz	CW 500Hz	SSB 2.4kHz	CW 500Hz	SSB 2.4kHz	CW 500Hz
0	-126	-133	-126	-132	-126	-132	-137	-144
20	-105	-112	-105	-112	-106	-112	-121	-129
Nation 4. Dither and/on Dan dame do not offer tMDC								

 Table 9: MDS<sup>1</sup> in dBm (RX2).

Notes: 1. Dither and/or Random do not affect MDS.

13: RX1/RX2 Crosstalk. In this test, RX1 and RX2 are set to the same frequency, mode and bandwidth. A test signal of amplitude  $P_i$  is applied to the RX1 input, and the RX2 input is terminated in 50 $\Omega$ . The value of Pi required to increase the RX2 audio output by 3 dB is recorded. The test is then repeated with the RX1 input terminated in 50 $\Omega$  and the test signal applied to the RX2 input. The receiver to which the test signal applied is muted.

*Test Conditions:* RX1 and RX2 set to 500 Hz CW and tuned to  $f_0$ . ATT = 0 dB, NR off, NB off, AGC: RX1: Med, Gain = 120. RX2: Fixd, Gain = 120. Repeat: AGC: RX1: Fixd, Gain = 120. RX2: Med, Gain = 120.

Table 10: KX1/KX2 Crosstalk.						
f₀ MHz	Test signal to	Pi dBm for +3 dB output	Output from			
50.1	RX1	-93	RX2			
50.1	RX2	-87	RX1			
1.0	RX1	-71	RX2			
1.9	RX2	-84	RX1			

Table 10: RX1/RX2 Crosstalk.

14: Noise Power Ratio (NPR): An NPR test is performed, using the test methodology described in detail in *Ref. 1*. The noise-loading source used for this test is a noise generator fitted with bandstop (BSF) and band-limiting filters (BLF) for the test frequencies utilized.

The noise loading  $P_{TOT}$  is applied to RX2 and increased until ADC clipping just commences, and then backed off until no clipping is observed for at least 10 seconds. NPR is then read off the spectrum scope by observation. (NPR is the ratio of noise power in a channel outside the notch to noise power at the bottom of the notch.)

*Test Conditions:* Receiver tuned to bandstop filter center freq.  $f_0 \pm 1.5$  kHz, 2.4 kHz SSB, ATT = 0 dB, NR off, NB off, Notch off, ANF off, AGC slow, Alex on. Test results are presented in **Table 11**.

BSF kHz BLF kHz		BWR dB	Alex	Ρτοτ	dBm	NPR	dB <sup>1</sup>	Theor. NPR <sup>3</sup>
			ATT	0 dB	20 dB	0 dB	20 dB	
1940	602044	29.2	0	-18	-1	71	68	80
1940	002044	29.2	1	-8	-1	79	66	00
3886	604100	32.3	0	-17	-2	70	68	76.9
3000		32.3	1	-14	-4	70	60	70.9
5240	60 5600	22.6	0	-18	+2	70	67	75.6
5340	605600	33.6	1	-14	-8	70	60	75.6
7600	3168160	400 05 4		-17	-1	69	67	74.1
7600	5100100	35.1	1	-11	+1	70	61	74.1
11700	11700 013000		0	-17	+3	66	63	71.6
11700	013000	37.6	1	-11	+1	66	60	/ 1.0

			2
Table 11:	RX2 NPR	Test	<b>Results<sup>4</sup></b>

## Notes on NPR test:

- 1. NPR readings were stable over time.
- 2. Enabling Dither and/or Random did not affect NPR results.
- 3. Theoretical NPR was calculated for the LTC2208-16 ADC using the method outlined in *Ref. 3*. The theoretical NPR value assumes that  $B_{RF}$  is not limited by any filtering in the DUT ahead of the ADC, and that the net gain between the antenna port and the ADC is 0 dB.

15: Two-Tone IMD<sub>3</sub> (IFSS, Interference-Free Signal Strength): Refer to Test 9 (above) for Test Description and Notes.

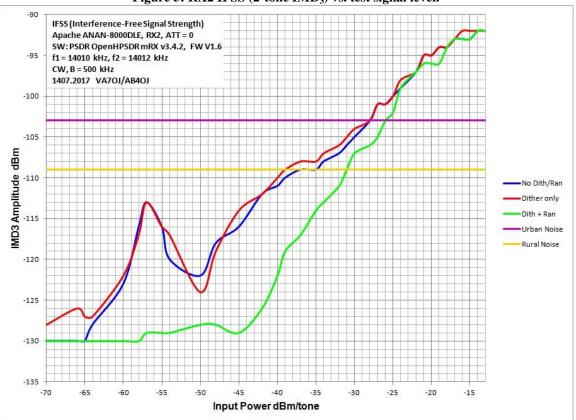


Figure 5: RX2 IFSS (2-tone IMD<sub>3</sub>) vs. test signal level.

*16. Two-Tone* 2<sup>*nd</sup></sup><i>-Order Dynamic Range (DR*<sub>2</sub>). Refer to Test 10 (above) for Test Description.</sup>

*Test Conditions:*  $f_1 = 6.1$  MHz,  $f_2 = 8.1$  MHz, IMD2 product at 14.2 MHz. 500 Hz CW, AGC slow, ATT = 0 dB, NR off, NB off, CW neutral, ANF off. DR<sub>2</sub> in dB. Measured MDS = -132 dBm (dither & random on).

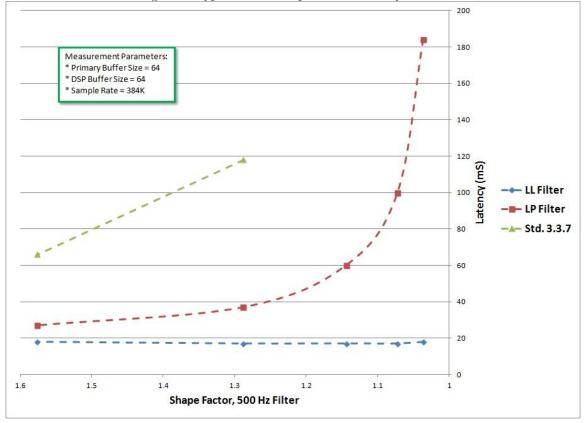
	Preselector	out <sup>1</sup>		i	n	
Dither & Random	MDS dBm	P <sub>i</sub> dBm	DR₂ dBm	P <sub>i</sub> dBm	DR <sub>2</sub> dBm	
off	-132	-31	101	-38	94	
on	-132	-30	102	-28	104	

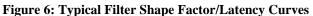
Table 12: RX2 DR<sub>2</sub>. f<sub>1</sub>: 6.1 MHz. f<sub>2</sub>: 8.1 MHz.

Notes: 1. The modest decrease in DR2 wth preselector out may be an IMD "sweet spot".

## 17: *Receiver Latency Test, January 16, 2016.* (Applies to ANAN series, PSDR OpenHPSDR mRX V3.3.9 and subsequent builds with LL/LP Filter Options)

Receiver latency was measured by applying a pulse train with fast rise-time to the RF input and measuring the time interval between the applied pulse (at the RF input) and the received pulse (at the audio output) with a 2-channel oscilloscope.





**Note:** LL: V3.3.9.0 (released June 2016) and all subsequent builds incorporate the LL (Low Latency) and LP (Linear Phase) filter options.

## C. Transmitter Tests

*Maximum temperatures:* Internal, 47°C. Case, 33°C.

• BH-4 TX Filter Window selected for all transmitter tests.

18: CW Power Output. In this test, the RF power output into a  $50\Omega$  load is measured at 3.6, 14.1, 28.1, 29.6 and 50.1 MHz in RTTY mode, at a primary DC supply voltage of +13.8V. A clamp-on ammeter is used to measure DC input current. A thermocouple-type power sensor and meter are connected to the ANT1 socket via a 50 dB power attenuator. PA Gain is adjusted on all bands before starting the test.

*Test Conditions:* 3.6, 14.1, 28.1, 29.6 and 50.1 MHz, 150W nominal. Set Tune Pwr to 100%, or check Use Drive Power and set Drive to 100%.. Adjust PA Gain and Wattmeter settings as required for nominal  $P_0$  and correct Fwd Pwr readings.

Table 15. C W Tower Output.					
Freq. MHz	Input Current A	Fwd Pwr W	Meas. Po W		
3.6	26.8	200	201		
7.1	31.5	200	200		
14.1	31.1	200	200		
21.1	30.9	200	200		
28.1	35.1	200	200		
50.1 <sup>1</sup>	35.6	150	150		

Table 13: CW Power Output.

Notes: 1. Po de-rated to 150W due to PSU current foldback.

18: SSB Peak Envelope Power (PEP). Here, an oscilloscope is terminated in  $50\Omega$  and connected to TX OUT via a 60 dB power attenuator. The scope vertical cursors are adjusted for 200W CW.

*Test Conditions:* 14.1 MHz, USB mode, dynamic mic connected, Drive 100%, Mic Gain 27 dB (COMP off), 33 dB (COMP on), compression 6 dB, Transmit Filter 200-3100 (default), supply voltage +13.8V. Leveler settings (default): Max. gain 5 dB, Decay 0.5s.

Speak loudly into the microphone for full-scale ALC reading. Figures 7 & 8 show the envelope for 200W PEP, without and with compression respectively. Figures 9 & 10 illustrate the effect of CESSB (controlled-envelope SSB) with 10 dB compression. *Note:* With MON on, mic/monitor latency  $\approx$  200 ms.

19: SSB ALC overshoot: A test was conducted in which white noise was applied from the internal noise generator, and the RF envelope observed on an oscilloscope terminated in  $50\Omega$  and connected to the DUT RF output via a 60 dB power attenuator.

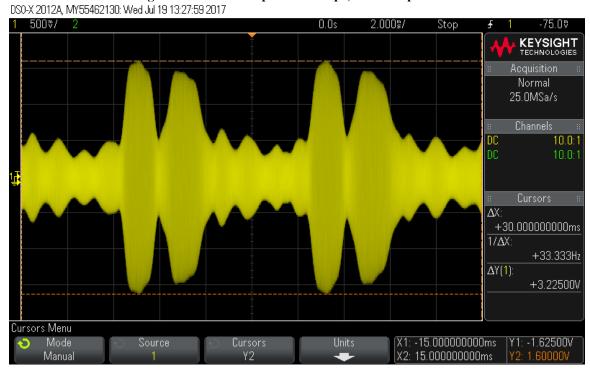
*Test Conditions:* 14100 kHz USB, compression 20dB. Test signal: white noise. Transmit Filter 200-3100 (default). Supply voltage +13.8V. Transmit Equalizer +15 dB (all 3 ranges), Preamp max. Test/Noise level +13 dB.

Test Results: No sign of ALC overshoot at 200W PEP. See Figure 11.



**Figure 7: 200W PEP speech envelope, no compression.** DS0-X 2012A, MY5546213D: Wed Jul 19 14:06:36 2017

Figure 8: 200W PEP speech envelope, 6 dB compression.



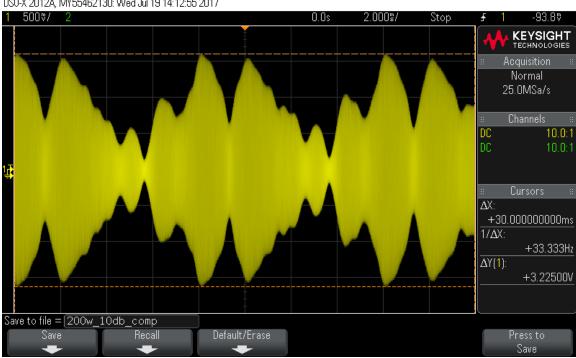
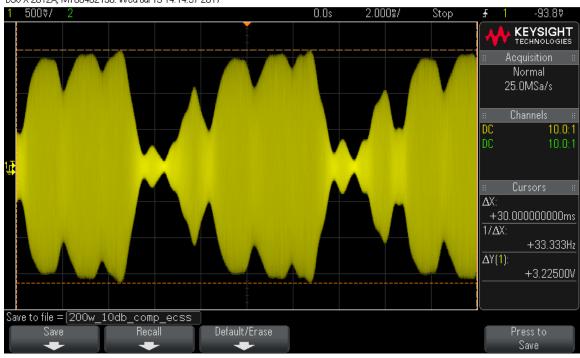
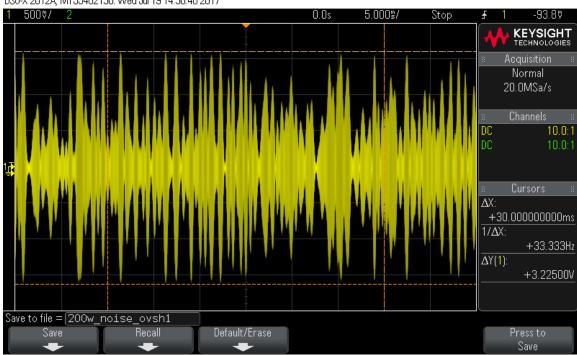


Figure 9: 200W speech envelope, 10dB compression, CESSB off. DS0-X 2012A, MY55462130: Wed Jul 19 14:12:55 2017

Figure 10: 200W speech envelope, 10dB compression, CESSB on. DS0-X 2012A, MY5546213D: Wed Jul 19 14:14:37 2017



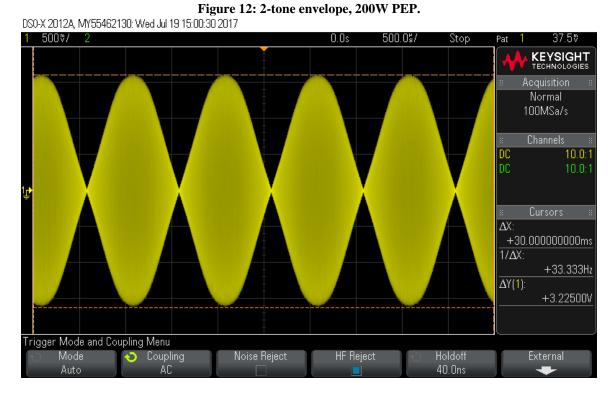


**Figure 11: 200W white noise test** (±3 vert. div. = 200W PEP). DS0-X 2012A, MY55462130: Wed Jul 19 14:38:40 2017

20: ALC Compression Check. In this test, a 2-tone test signal is applied to the USB port from the internal 2-tone generator. An oscilloscope is connected to the DUT RF output via a 60 dB power attenuator. RF Power is initially adjusted for 200W CW output.

*Test Conditions:* 14100 kHz USB, compression off. Test signal: 2-tone. Transmit Filter 200-3100 (default). Test tones: 700 and 1700 Hz, at equal amplitudes. Supply voltage +13.8V.

Test Result: No flat-topping of the 2-tone envelope was observed (see Figure 12.)



**20a:** Subjective TX audio test: In this test, a headset is plugged into the microphone and headphone jacks and a transmitted SSB signal is monitored with MON active.

## Test Procedure:

- a. Set COMP to 6 dB.
- b. Adjust Mic Gain for no ALC COMP on TX Meter with CESSB off.
- c. Set TX EQ off. .
- d. Transmit alternately with COMP off and on. Observe that COMP gives monitored TX audio more audible "punch" and penetrating power.

Test Results: With Mic Gain = 16 dB, CESSB off, COMP on yields audible improvement in audio "punch",

**21:** *Transmitter 2-tone IMD Test.* In this test, a 2-tone test signal is applied from the internal tone generator. A spectrum analyzer is connected to the DUT RF output via a 50 dB power attenuator. RF Power is initially adjusted for rated CW output on each band in turn. (Due to lab power supply limitations, it was found necessary to de-rate to 100W on 50 MHz.)

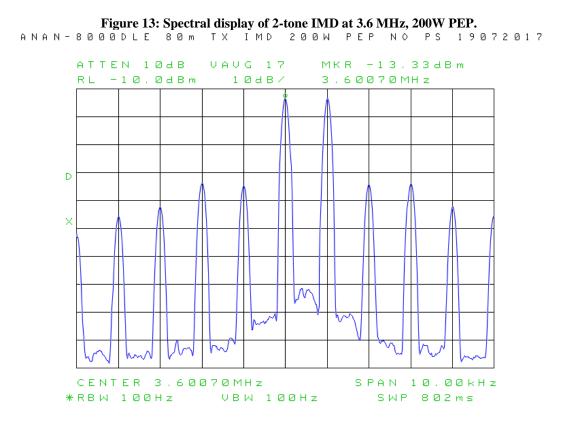
*Test Conditions:* 3.6, 14.1 and 50.1 MHz USB, compression off. Test signal: 2-tone. Transmit Filter 200-3100 (default). Test tones: 700 and 1700 Hz, at equal amplitudes. Supply voltage +13.8V. The -10 dBm reference level RL equates to 200W CW output (= 0 dBc).

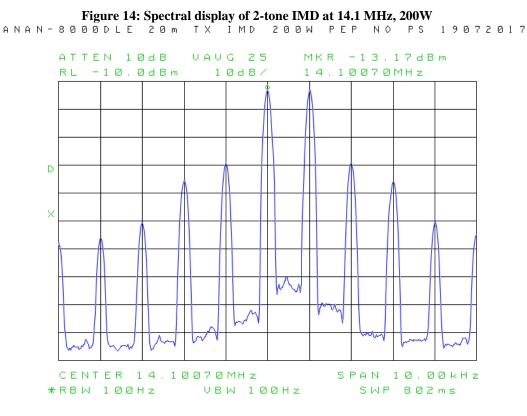
Adjust test tone levels for 200W PEP (each tone at -6 dBc). Figures 13 through 15 show the two test tones and the associated IMD products for each test case.

Table 14: 2-tone TX IMD.					
2-tone TX IMD Products at Rated P <sub>o</sub>					
IMD Products Rel. Level dBc (0 dBc = 1 tone)					
Freq. MHz	3.6 14.1 28.1 50.1 <sup>1</sup>				
IMD <sub>3</sub> (3 <sup>rd</sup> -order)	-33 -27 -25 -31				
IMD <sub>5</sub> (5 <sup>th</sup> -order)	-32 -34 -30 -41				
IMD <sub>7</sub> (7 <sup>th</sup> -order)	der) -30 -48 -37 -49				
IMD <sub>9</sub> (9 <sup>th</sup> -order) -44 -54 -40 -49					
Add 6 dB for IMD referred to 2-tone PEP					
Notes: 1. 6n	n IMD m	easured a	t 100W P	EP.	

**20a.** Noise IMD Test. This test is similar to Test 20, except that a white-noise baseband is applied from the internal noise generator. See Figure 17. Note that the IMD skirts are steeper at the lower power level.

*Test Conditions:* 14.1 MHz USB, 10 dB compression, CESSB on.. TX EQ on (adjust as required for 200W PEP.) Transmit Mode: Noise. Level (dB): +13. Measured at -43 dBm/Hz noise power  $\approx$  175W PEP and 6 dB lower.

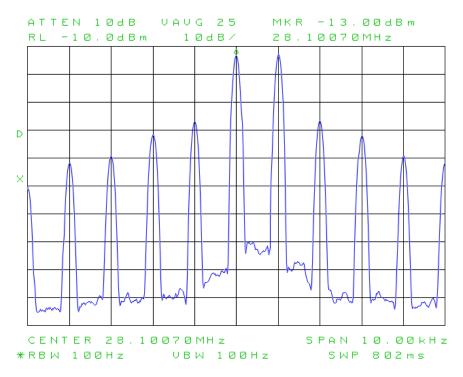






#### Figure 15: Spectral display of 2-tone IMD at 28.1 MHz, 200W PEP.

ANAN-8000DLE 10m TX IMD 200W PEP NO PS 19072017



21

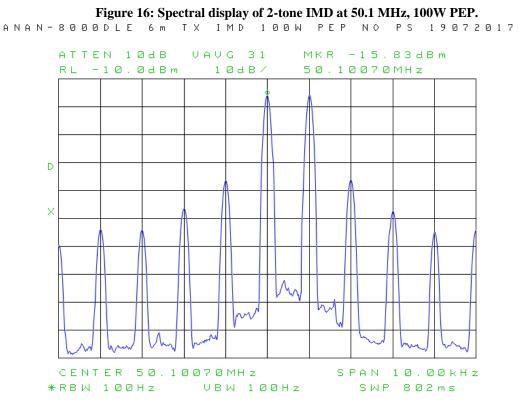
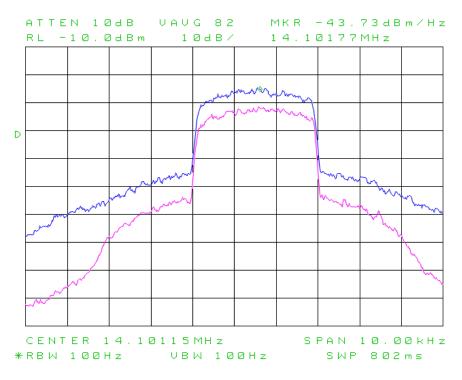


Figure 17: Noise modulation, showing IMD skirts.

ANAN-8000DLE TX Noise IMD B: 175W R: 45W 20072017



21: AM sidebands and THD with single-tone modulation. As in Test 20 above, the spectrum analyzer is connected to the DUT RF output via a 20 dB 25W attenuator. On the GUI, RF Power is adjusted for 50W resting carrier. A 1 kHz test tone is applied from the internal tone generator. The spectrum analyzer records the carrier and sideband parameters.

*Test Conditions:* 14100 kHz AM, 50W carrier output, Transmit Mode: Tone, Level: -15 dB. Adjust test tone level for -7 dBc test tone level (90% modulation.) Figure 18 shows the carrier and sideband levels. Calculated THD  $\approx$  3%.

 Figure 18: AM Sidebands for 90% Modulation.

 ANAN-8000DLE AM 20m 50W CXR m=0.9 THD~3% 200717

 DISCRETE SIDEBAND SEARCH RESULTS

 CARRIER FREQ:
 14.10 MHz

 CARRIER POWER:
 -12.7 dBm

 OFFSET FREQ
 0FFSET + OFFSET

 .998 kHz
 -7.0

 .998 kHz
 -7.0

 .998 kHz
 -29.8

 .2996 kHz
 -35.2

 .4004 kHz
 -42.2

 .003 kHz
 -54.3

FOUND: 5 SETS OF SIDEBANDS

22: Transmitter harmonics & spectral purity. Once again, the spectrum analyzer is connected to the DUT RF output via a 60 dB power attenuator. RF Power is adjusted for 200W CW output on each band in turn. RL = -10 dBm equates to 200W. The spectrum analyzer's harmonics capture utility is started.

**Test Conditions:** 3.6, 14.1 and 50.1 MHz, TUNE mode, 200W to  $50\Omega$  load. Harmonic data is presented for all frequencies tested (Figures 19 through 21), and a spur sweep from 1 – 68 MHz in Figure 22. It will be seen that harmonics are well within specifications. Non-harmonic spurs are within the -60 dBc limit specified in FCC Part 97.307(e). In addition, a spur sweep in the range 14090-14190 kHz shows that the Orion MkII C326 capacitor upgrade has greatly reduced the transmitted spurs at ±48 kHz offset. See Figure 23.

#### Figure 18.

ANAN-8000DLE TX Harmonics 80m 200W CW 20072017

HARMONIC MEASUREMENT RESULTS FUNDAMENTAL SIGNAL: 3.600 MHz -7.7 dBm HARMONIC LEVEL dBc FREQUENCY 7.200 MHz 2 -54.3 10.80 MHz -38.8 3 -87.7 \* 14.40 MHz 4 -68.3 18.00 MHz 5 -95.2 21.60 MHz 6 7 -94.3 25.20 MHz -98.8 28.80 MHz 8 MEASURED LEVEL MAY BE \* NOISE OR LOST SIGNAL. TOTAL HARMONIC DISTORTION = 1.2 % (OF HARMONICS MEASURED)

**Figure 19.** ANAN-8000DLE TX Harmonics 20m 200W CW 20072017

HARMONIC MEASUREMENT RESULTS

FUNDAMENTAL SIGNAL: 14.10 MHz -7.0/dBm HARMONIC LEVEL dBc FREQUENCY 2 -71.8 \* 28.20 MHz З -91.5 \* 42.30 MHz 4 -95.0 56.40 MHz -94.3 5 70.50 MHz 84.60 MHz 6 -100.2 98.70 MHz 7 -97.3 -95.0 8 112.8 MHz \* MEASURED LEVEL MAY BE NOISE OR LOST SIGNAL. TOTAL HARMONIC DISTORTION = Ø % (OF HARMONICS MEASURED)

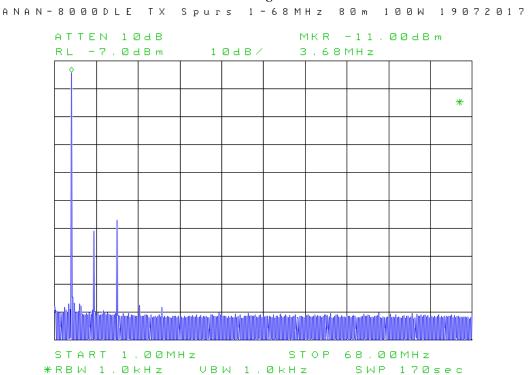
#### Figure 20.

HARMONIC MEASUREMENT RESULTS FUNDAMENTAL SIGNAL: 28.10 MHz -7.2 d.B.m. HARMONIC LEVEL dBc FREQUENCY -65.5 56.20 MHz 2 -67.2 84.30 MHz 3 -87.5 112.4 MHz 4 -57.5 5 140.5 MHz -80.7 168.6 MHz 6 -72.8 196.7 MHz 7 224.8 MHz -81.7 8 TOTAL HARMONIC DISTORTION = .2 % (OF HARMONICS MEASURED)

Figure 21. ANAN-8000DLE TX Harmonics 6m 100W CW 19072017

HARMONIC MEASUREMENT RESULTS FUNDAMENTAL SIGNAL: 50.10 MHz -9.8 dBm LEVEL dBc HARMONIC FREQUENCY 2 -84.5 \* 100.2 MHz 3 -52.3 150.3 MHz 4 -74.0 200.4 MHz -60.3 5 250.5 MHz -71.7 6 300.6 MHz 7 -65.0 350.7 MHz 8 -73.0 400.8 MHz MEASURED LEVEL MAY BE \* NOISE OR LOST SIGNAL. TOTAL HARMONIC DISTORTION = .3 ½ (OF HARMONICS MEASURED)

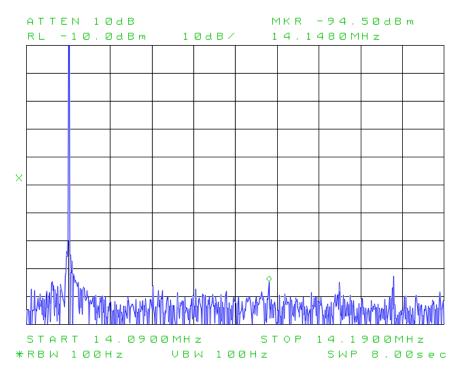
*Note:* At 50.1 MHz, the 3rd harmonic level exceeds the -60 dBc maximum specified in FCC Part 97.307(e) for the frequency range 30-225 MHz.



#### Figure 22.

Figure 23. Spurs to +90 kHz offset. Note spur at +48 kHz.

ANAN-8000DLE 20m 100W Spurs to +90kHz 200717



*23: Transmitted composite noise.* A Perseus SDR receiver is connected to the DUT RF output via a 60 dB power attenuator. Figure 23 is the resulting plot.

Test Conditions: 3.6, 14.1 and 28.1 MHz, TUNE mode, 100W output.

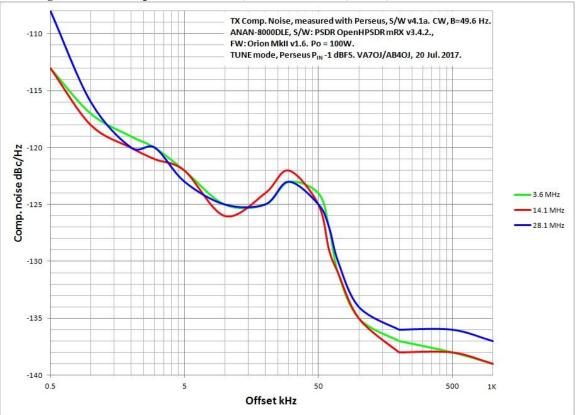


Figure 23: TX composite noise at 3.6, 14.1 & 28.1 MHz, 100W, measured on Perseus RX.

23: Spectral display of CW keying sidebands. The spectrum analyzer is connected to the DUT RF output via a 60 dB RF power attenuator. The -5 dBm reference level equates to 100W. A series of dits is transmitted at 60 wpm.

*Test Conditions:* 14.1 MHz CW, 100W output. Keying speed 60 wpm using internal keyer. CW key-down & key-up delays 2 ms. Spectrum analyzer RBW is 10 Hz, video-averaged; sweep time < 2 sec. Figure 24 shows the transmitter output  $\pm 2.5$  kHz from the carrier.

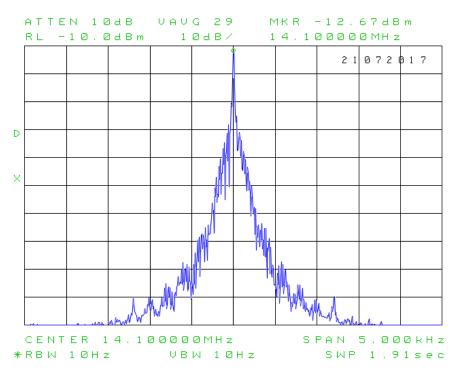


Figure 24: Keying sidebands at 60 wpm, Weight = 50%, 14.1 MHz, 100W. ANAN8000DLE CW spectrum 60wpm dits 2ms delays

24. CW keying envelope. The oscilloscope is terminated in  $50\Omega$  and connected to the DUT RF output via a 60 dB 2RF power attenuator. A series of dits is transmitted from the internal keyer at 60 wpm.

*Test Conditions:* 14.1MHz CW, 100W output. Keying speed = 60 wpm using internal keyer. CW key-down & key-up delays 2 ms\*. The keying envelope is shown in Figure 25.

\* Delays  $\geq$  25 ms will cause distortion of the keying envelope (uneven dit timing).

1 500♥/ 2	2.600\$	20.00g/ S	itop Л	1 2380 • KEYSIGHT
				KEYSIGHT TECHNOLOGIES Acquisition :: Normal 2.50MSa/s Channels :: 10.0:1 10.0:1
Display Menu Persistence Off	Clear Clea sistence Displa			Intensity 20%

Figure 25: Keying envelope at 60 wpm 130: Sat Jul 22 14:26:45 2017

## 25: PureSignal Adaptive-Predistortion Linearization Test.

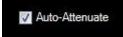
Test Setup: PS FEEDBACK-PS INPUT jumper in place on DUT rear panel (default).

## PowerSDR Configuration:

1. On Setup, General, Hardware Config tab, uncheck as follows:

Pure Signal	
Disable Pure Signal	

2. On PureSignal 2.0/Advanced form, check Auto Attenuate:



*Test Conditions:* 14.1 and 50.1 MHz USB, compression off. Test signal: 2-tone. Transmit Filter 200-3100 (default). Test tones: 700 and 1700 Hz, at equal amplitudes. Supply voltage +13.8V. The -7 dBm reference level RL on spectrum analyzer equates to 200W CW output on 14.1 MHz (0 dBc). PS-A off (initially), then on.

At 50.1 MHz, set reference level to -10 dBm = 100 W CW output (0 dBc).

## Test Procedure:

To start PureSignal, click on Linearity on the menu bar at the top of the main console. The PureSignal form (Figure 26) will open. If desired, click "Information" to open a PureSignal user guide.

1. On Setup, Tests tab, set up Two-Tone Test as follows:

Two Tone Test				
Applied: DSP Output				
Freq #1: 700 🚔				
Freq #2: 1700 🚔				
Level (dB): 0.000 🚔				
RF Power: 100 🚔				
Vise Drive Power				
Invert for LS Modes				
Start				

- 2. Click **Start** and adjust RF Power for 200W PEP (-6 dBc per tone) on spectrum analyzer. Click **Start** again to stop transmitting.
- 3. Now return to PureSignal form and click **Two-tone**. Non-linearized 2-tone spectrum will be displayed. Store or capture screen image on spectrum analyzer.
- 4. Next, turn PS-A on and verify green Correcting indicator.

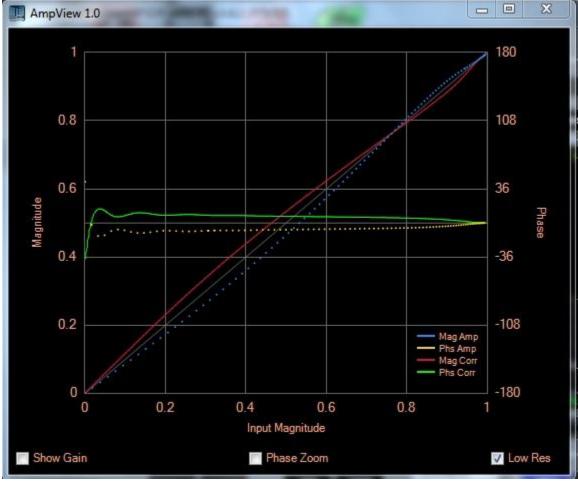
Correcting

- 5. The linearized 2-tone spectrum will be displayed. Store or capture screen image on spectrum analyzer. (See Figure 25).
- 6. Click AmpView to display phase/gain transfer curve screen. (See Figure 24).
- 7. Click **Two-Tone** again to stop 2-tone test. Do not click **OFF**.
- 8. Re-run AM sidebands/THD test (see Test 21) with PureSignal on. Record test data from spectrum analyzer. (See Figure 29). Calculate THD.

🛄 PureSignal 2.0				. 🗆 🗙
Two-Tone Single Cal An	npView Advanced	Save	Restore	OFF
MOX Wait (sec) 0.2 🚽 👽	Auto-Attenuate		V PIN	
CAL Wait (sec) 0.0 🚔	Relax Tolerance		V MAP	
AMP Delay (ns) 150 🌲			🔽 STBL	
			TINT (de	B) 0.5 🔻
Calibration Information				
bldr.rx <mark>0</mark> sln.chk	0 state	1		
bldr.cm 0 dg.cnt	0 feedbk	148		
bldr.cc 0 cor.cnt	1395 GetPk 0.4	4157060		
bldr.cs 0	SetPk 0.4	4072		
Bottom Panadapter Displays RX 2			📑 Alwa	ys On Top

Figure 25: PureSignal (Linearity) Advanced form.

Figure 26: AmpView screen for 200W 2-tone test, PureSignal on.



PureSignal Test Results:

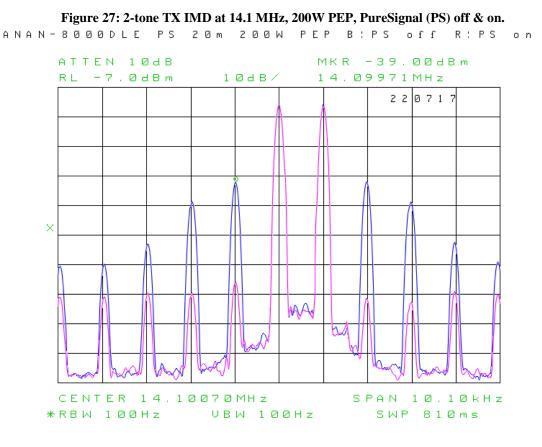
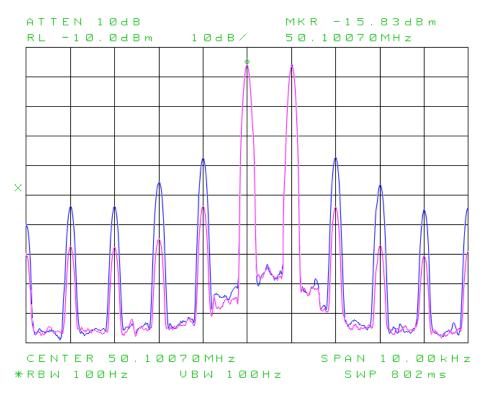


Figure 28: 2-tone TX IMD at 50.1 MHz, 100W PEP, PureSignal (PS) off & on.

ANAN-8000DLE 6m LNA off R:PS on B:PS off 220717



ANAN-8000DLE 20m AM m=0.9 50W CXR PS on 220717

DISCRETE SIDEBAND SEARCH RESULTS

CARRIER FREQ: 14.10 MHz CARRIER POWER: -12.2 dBm

OFFSET FREQ       - OFFSET       + OFFSET         dBc       dBc         .998					
1.997       kHz       -55.3       -56.0         2.996       kHz       -57.7       -57.0         3.995       kHz       -64.3       -62.8	OFFSET	FREQ		:T +	
1.997       kHz       -55.3       -56.0         2.996       kHz       -57.7       -57.0         3.995       kHz       -64.3       -62.8				·	
	1.997 2.996 3.995	k H z k H z k H z	-55.3 -57.7 -64.3	 	-56.0 -57.0 -62.8

FOUND: 5 SETS OF SIDEBANDS

#### 26. References:

- 1. Apache Labs website: <u>https://apache-labs.com/</u>
- 2. "Noise Power Ratio (NPR) Testing of HF Receivers" http://www.ab4oj.com/test/docs/npr\_test.pdf
- 3. "Theoretical maximum NPR of a 16-bit ADC" http://www.ab4oj.com/test/docs/16bit\_npr.pdf
- 4. "HF Receiver Testing: Issues & Advances" http://www.nsarc.ca/hf/rcvrtest.pdf
- 5. "A New Look at SDR Testing" http://www.ab4oj.com/sdr/sdrtest2.pdf
- 6. "Testing AGC in receivers" by SM5BSZ http://www.sm5bsz.com/lir/agctest/agctest.htm
- 7. Quick Start Guide for the OpenHPSDR mRX PS CFC Audio Tools http://tinyurl.com/cfcdoc

**27.** *Acknowledgements:* I would like to thank Abhi Arunoday of Apache Labs, Warren Pratt NROV and Doug Wigley W5WC for making this ANAN-800DLE available to me for testing and evaluation. Thanks are also due to Warren, Doug and Abhi for their invaluable guidance in configuring the radio and activating PureSignal.

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e-mail: farson@shaw.ca http://www.ab4oj.com/ Iss. 2, July 29, 2017.

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## Appendix 1: CFC Audio Tools Test

The purpose of this test is to verify the effect of CFC tailoring on transmitted voice audio. A dynamic microphone is connected to the ANAN-8000DLE, a custom CFC profile is created and the transmitted RF envelope waveforms of an SSB signal with CFC alternately disabled and enabled are displayed and captured on a digital oscilloscope connected to ANT1 via a 60 dB RF power attenuator.

### 27. CFC TX Audio Tools Test.

Test Conditions: 14100 kHz, USB, 200W PEP, default TX filter (200-3100 Hz). Inrad M629 dynamic MIC connected to MIC jack. MIC GAIN, COMP, Equalizer (Figure 30), Leveler and CFC (Figure 31) set up per instructions in Ref. 7: http://tinyurl.com/cfcdoc The TX audio profile thus created is saved as "VA7OJ Test" under the Transmit tab.

3-Band E	Settings	10-Band Equ	alizer		10
Receive Equ			361201		
Enabled					
	·		and a second		Reset
Preamp Low			Mid	High	
- 15	idB			1 -	15dB
			-		
6 0	ыв		<u>ь</u> :		0dB
-12	dB –				-12dB
					-12dB
Fransmit Equ	ıalizer				-12dB
	ualizer d	3 📥 25	0 1000 1	4000 🗂 160	
Fransmit Equ Transmit Equ Enables FREQ	ualizer d				-12dB
Fransmit Equ I Enablex FREQ Preamp	d 32 ÷				00
Fransmit Equ I Enablex FREQ Preamp	ualizer d				
Fransmit Equ I Enables FREQ Preamp ↓ 15	aalizer d 32 ♀ idB ☐				00 € - 15dB ⊃
Fransmit Equ I Enables FREQ Preamp ↓ 15	d 32 ÷				00

34

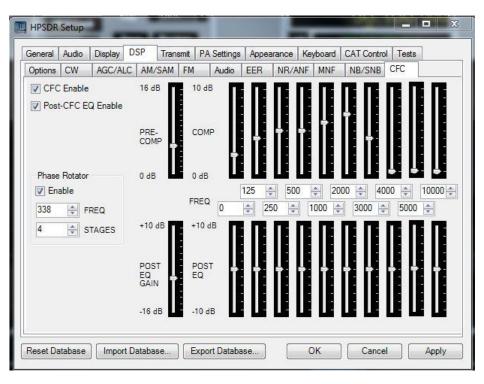
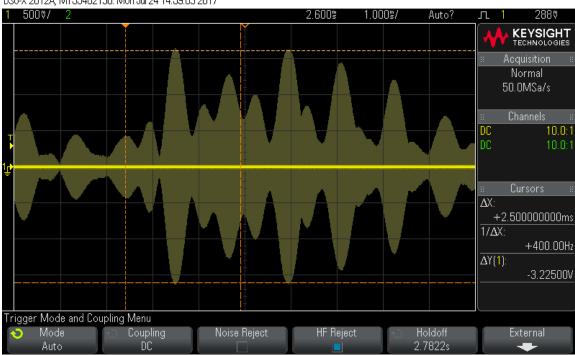


Figure 31: CFC tab settings.

*Test Results:* Figures 32 and 33 illustrate the RF envelope waveform with CFC off and the VA7OJ Test profile enabled, respectively.



**Figure 32: Transmitted RF envelope, USB, 200WPEP, CFC off.** DS0-X 2012A, MY55462130: Mon Jul 24 14:59:03 2017



Figure 33: Transmitted RF envelope, USB, 200WPEP, CFC on, VA7OJ Test profile. DS0-X 2012A, MY55462130: Mon Jul 24 15:09:46 2017

*Subjective listening test:* By listening via the Monitor facility, it was determined that the CFC-processed audio had much more pre-emphasis, greatly increased "punch" and a large improvement in articulation as compared to the audio with CFC off.

## Appendix 2: Aliasing Suppression Test, Dec. 23, 2015.

The purpose of this test is to measure the suppression of aliasing artifacts generated in Nyquist Zone 1 when a swept signal covering a frequency offset range in Nyquist Zone 2 is applied to the RX1 input.

The Orion MkII Nyquist frequency ( $f_s/2$ ) is 61.44 MHz. Thus, Nyquist Zone 1 is 0 – 61.44 MHz and Zone 2 is 61.44 – 122.88 MHz.

19. Aliasing suppression for input sweep range 61.44 MHz  $\leq f_0 \leq$  71.44 MHz. An RF sweep generator is connected to the RX1 input via a 20 dB attenuator, and set up as follows:

- Start Freq: 61.44MHz ( $f_s/2$ )
- Stop Freq: 71.44MHz ( $f_{s}/2 + 10$  MHz)
- Sweep Time: 50 ms
- Output level: -60 dBm
- Attenuator: 10 dB

*Test conditions:* ANAN-8000DLE, running under PowerSDR OpenHPSDR v3.3.6, configured as follows:

- Attenuator: 0 dB
- Preselector: None fitted
- AGC Gain: 120
- Dither & Random: OFF
- Zoom: Minimum
- AVG: ON, Avg. Time 240 ms
- Center freq. 61.170 MHz
- Stop freq. 61.440 MHz
- Span: ± 270 kHz
- FFT: Mid (bin width 5.859 Hz)
- Window: Blackman-Harris
- Weighted Log

The test result is shown in Figure 6. An aliasing artifact is observed at 61.386 MHz. See Figure 27.



Figure 27: Aliasing artifacts vs. frequency.

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# Appendix 3: Investigation of ALEX Preselector PIM, August 2-5, 2017

A comparison of the RX1 IFSS charts for the ANAN-8000DLE and the stand-alone Rev. 3 Orion MkII board revealed a significant degradation of front-end  $IMD_3$  performance with the ALEX preselector in the signal path. (Refer to Test 9 and Figure 7 on p. 8 above.)

As a result, an additional IFSS test was conducted under the same conditions as for Test 9, but with ALEX alternately on and off.  $IMD_3$  amplitudes are significantly higher with ALEX on, the worst case being 16 dB degradation at -30 dBm/tone input. This leads to a conclusion that PIM (passive IMD) is occurring in inductor cores in the ALEX filters.

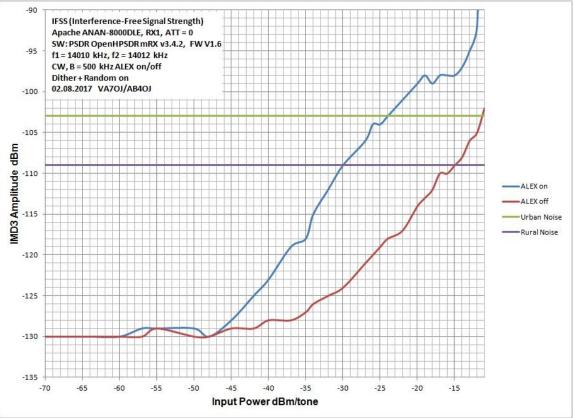


Figure 38: RX1 IFSS (2-tone IMD<sub>3</sub>) vs. test signal level, with ALEX on and off.

To quantify the PIM, two-port NPR tests were conducted on the BPF1 section of the ALEX board. (It is assumed that the BPF1 and BPF2 sections are identical.) Tests were performed with the Wandel & Goltermann RS-50 and RS-25 noise generators and RE-25 noise receiver. The generators were fitted with band-limiting (BLF) and bandstop (BSF) filters as listed in Table 15, and the receiver with modules corresponding to the bandstop filters. The generator in use was connected to the ALEX input via a low-PIM 75/50 $\Omega$  transformer, and the ALEX output was connected to the receiver via a MCL BMP-5075 50/75 $\Omega$  matching pad. *Test Conditions:* NPR tests run at 0, -10 and +5 dBm noise loading. Intrinsic (test system) NPR measured initially at each level by jumpering the  $50\Omega$  transformer output and matching-pad input.

Noise Loading dBm		0		-10		+5				
BSF kHz	BLF kHz	Intr. NPR	ALEX NPR	Intr. NPR	ALEX NPR	Intr. NPR	ALEX NPR			
1940	60-2044	77.7	70.6	79.9	79.9	77.0	67.9			
2438	60-2600	76.6	52.9	76.6	70.5	78.4	48.8			
3886	60-4100	79.0	53.3	77.6	64.1	77.7	48.3			
4650	60-5600	78.0	55.8	76.8	66.7	73.4*	51			
5340	60-5600	77.7	56.1	76.3	67.1	74.7	50.9			
7600	316-8160	73.3	63.5	77.7	76.9	67.0*	55.9			
11700	0-13000	74.2	62.4	73.7	68.3	73.3	57.4			

Table 15: Two-port NPR of ALEX Preselector

\*Intrinsic NPR degraded due to RE-25 REU module overload at +5 dBm.

*Test Results:* Severe NPR degradation was observed, increasing sharply with noise loading, This will severely compromise the receivers' IMD performance under strong-signal conditions with ALEX in the signal path.

This issue can be rectified by replacing the existing inductors in the ALEX filters with parts using cores of sufficient cross-section to minimize saturation at high input levels. Choice of core material will also be an important factor in this remediation process.

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