



***Society of Cable  
Telecommunications  
Engineers***

---

**ENGINEERING COMMITTEE  
Interface Practices Subcommittee**

---

**AMERICAN NATIONAL STANDARD**

**ANSI/SCTE 96 2008**

**Cable Telecommunications Testing Guidelines**

## NOTICE

The Society of Cable Telecommunications Engineers (SCTE) Standards are intended to serve the public interest by providing specifications, test methods and procedures that promote uniformity of product, interchangeability and ultimately the long term reliability of broadband communications facilities. These documents shall not in any way preclude any member or non-member of SCTE from manufacturing or selling products not conforming to such documents, nor shall the existence of such standards preclude their voluntary use by those other than SCTE members, whether used domestically or internationally.

SCTE assumes no obligations or liability whatsoever to any party who may adopt the Standards. Such adopting party assumes all risks associated with adoption of these Standards, and accepts full responsibility for any damage and/or claims arising from the adoption of such Standards.

Attention is called to the possibility that implementation of this standard may require the use of subject matter covered by patent rights. By publication of this standard, no position is taken with respect to the existence or validity of any patent rights in connection therewith. SCTE shall not be responsible for identifying patents for which a license may be required or for conducting inquiries into the legal validity or scope of those patents that are brought to its attention.

Patent holders who believe that they hold patents which are essential to the implementation of this standard have been requested to provide information about those patents and any related licensing terms and conditions. Any such declarations made before or after publication of this document are available on the SCTE web site at <http://www.scte.org>.

All Rights Reserved

© Society of Cable Telecommunications Engineers, Inc. 2008  
140 Philips Road  
Exton, PA 19341

## TABLE OF CONTENTS

1.0	SCOPE .....	1
2.0	DEFINITIONS AND ACRONYMS .....	1
3.0	TEST PLAN CONSIDERATIONS .....	5
4.0	TEST EQUIPMENT .....	10
5.0	THE TOTAL TEST SYSTEM .....	31
6.0	MEASUREMENT ACCURACY .....	38
7.0	MEASUREMENT RECORDS .....	42
8.0	REFERENCE INFORMATION .....	43

### *LIST OF FIGURES*

FIGURE 1 – RF SPECTRUM ANALYZER BLOCK DIAGRAM	13
FIGURE 2 – RF RECEIVER BLOCK DIAGRAM	13
FIGURE 3 - ENVELOPE DETECTOR.	17
FIGURE 4 – SAMPLE DETECTION	20
FIGURE 5 - VIDEO FILTERING	20
FIGURE 6 - REFERENCE BROADBAND COMMUNICATIONS TEST SYSTEM	31
FIGURE 7 - REFERENCE BROADBAND COMMUNICATIONS TEST SYSTEM	33
FIGURE 8 - POWER (10*LOG) ADDITION	44
FIGURE 9 - VOLTAGE OR CURRENT (20*LOG) ADDITION	46
FIGURE 10 - GRAPH OF NNN/BNN CORRECTION	47

*LIST OF TABLES*

TABLE 1 - FIXED CHANNEL FILTER REQUIREMENTS	25
TABLE 2 – CTB MEASUREMENT ACCURACY ANALYSIS	41
TABLE 3 - POWER (10*LOG) ADDITION	44
TABLE 4 - VOLTAGE OR CURRENT (20*LOG) ADDITION	45

## 1.0 SCOPE

The test procedures that reference this document are intended to allow a competent technician or engineer to perform the tasks of determining, to a reasonable degree of certainty, the level of performance for the various parameters detailed. The procedures are general in nature and with sufficient forethought and preparation, can be adapted to individual devices, cascades or complete systems. The primary focus for these procedures is for bench or laboratory testing, but the principles discussed are equally applicable to field testing. When the suggestions made in this document conflict with the detailed steps of a specific procedure, the specific test procedure will take precedence.

In order to maintain the simplicity and reduce the overall size of the individual procedures, most theoretical and practical discussions regarding test equipment, methodology and variations in techniques, as well as information which is generic or repetitive in nature is discussed in this document. This will also allow alterations and/or updates to be handled more easily by reducing the total number of documents (or sections) which will be affected. Specific information or data required for a single test, or a limited number of tests, will be found in those procedures as needed.

Measurements can normally be separated into two types, absolute and relative. Absolute measurements are used for determining such items as signal levels, modulation deviation, etc. Relative measurements are made with respect to a reference level or parameter and some examples are distortion, frequency flatness, depth of modulation, etc. Absolute measurements are typically more difficult to make within the same tolerance limits as relative measurements since more measurement tolerances within the test equipment and test configuration must be considered. Relative measurements are often quite accurate since many of these tolerances are cancelled in the final calculations, especially when measurement conditions are carefully maintained. Relative measurements are often used as the basis for comparison between similar products and are valid when the measurement conditions are identical.

## 2.0 DEFINITIONS AND ACRONYMS

### **Carrier Level or Carrier Power**

Often used as a synonym for “signal level” or “channel power”. When the carrier level is being modulated with information, the term “channel power” is more appropriate.

### **Channel Power or Channel Level**

See definitions below for various types of signals. The level is usually presented in decibels with respect to one millivolt RMS in a 75  $\Omega$  system (dBmV).

#### **Continuous Wave (CW) Carrier Level**

The RMS voltage of the sinusoidal signal.

### **Analog Video Channel Level**

The RMS voltage of the sinusoidal signal during the video sync pulse.

### **Digital Channel Level**

The RMS voltage of a sinusoidal signal that would produce the same heating in a 75  $\Omega$  resistor as does the actual signal.

### **Intermittent Digital Channel Level**

For a signal that occupies one assigned time slot in a time division multiple access (TDMA) sequence of time slots, the level reported shall be the equivalent level as if the signal being measured (any one of the multiple signals included in the total sequence) was on continuously.

### **dB**

Decibels. Logarithmic expression of the ratio between two values

$$\text{For two powers: } x \text{ dB} = 10 * \log\left(\frac{P_2}{P_1}\right) \quad (1)$$

$$\text{For two voltages: } x \text{ dB} = 20 * \log\left(\frac{V_2}{V_1}\right) \quad (2)$$

### **dBc**

Decibels relative to carrier power. Signals greater than the carrier will have a positive result; signals less than the carrier will have a negative result.

$$x \text{ dBc} = \text{Disturbance Power (dB)} - \text{Carrier Power (dB)} = 10 * \log\left(\frac{P_{\text{disturbance}}}{P_{\text{carrier}}}\right) \quad (3)$$

### **-dBc (negative dBc)**

To avoid using negative numbers, the ratio between a disturbance (smaller than the carrier) and the carrier is often specified as a positive number with the units -dBc.

### **dBm**

Decibels relative to one milliwatt. 0 dBm equals 1 mW.

$$y \text{ dBm} = 10 * \log\left(\frac{x \text{ mW}}{1 \text{ mW}}\right) \quad (4)$$

where

x = the power in mW

y = the power in dBm

### **dBmV**

Decibels relative to one millivolt RMS. 0 dBmV equals 1 mV.

$$y \text{ dBmV} = 20 * \log\left(\frac{x \text{ mV}}{1 \text{ mV}}\right) \quad (5)$$

where

x = the voltage in mV

y = the voltage in dBmV

### **dBμV**

Decibels relative to one microvolt RMS. 0 dBμV equals 1 μV.

$$y \text{ dBμB} = 20 * \log\left(\frac{x \text{ μV}}{1 \text{ μV}}\right) \quad (6)$$

where

x = the voltage in μV

y = the voltage in dBμV

### **Unit Conversions**

#### **dBμV to dBmV**

To convert from dBμV to dBmV, subtract 60 from the value:

$$y \text{ dBmV} = x \text{ dBμV} - 60 \quad (7)$$

#### **dBmV to dBμV**

To convert from dBmV to dBμV, add 60 to the value:

$$y \text{ dBμV} = x \text{ dBmV} + 60 \quad (8)$$

#### **dBmV to dBm**

To convert from dBm to dBmV, the following equation can be used:

$$y \text{ dBmV} = 20 * \log\left(1000 * \sqrt{\left[10^{\left(\frac{x \text{ dBm}}{10}\right)} * Z\right]}\right) \quad (9)$$

where

x = the power in dBm

y = the voltage in dBmV

Z = the impedance in ohms

To convert from dBmV to dBm, the following equation can be used:

$$y \text{ dBm} = 10 * \log \left( 1000 * \left[ \frac{10^{\left( \frac{x \text{ dBmV}}{20} \right)}}{1000} \right]^2 * \left[ \frac{1}{Z} \right] \right) \quad (10)$$

where

x = the voltage in dBmV  
y = the power in dBm  
Z = the impedance in ohms

In a 75 Ω system:

$$0 \text{ dBm} = 48.75 \text{ dBmV}$$

In a 50 Ω system:

$$0 \text{ dBm} = 46.99 \text{ dBmV}$$

### **dBmV measurements of 75 Ω systems with 50 Ω equipment**

The difference when changing between 50 and 75 Ω systems is 48.75 - 46.99 = 1.76 dB. If measurements of a 75 Ω system are made in dBmV on 50 Ω test equipment, the results will be 1.76 dB too low. To obtain the correct dBmV value for the 75 Ω system, the loss of the impedance matching system (generally a transformer or Minimum Loss Pad (MLP)) and the 1.76 dB correction factor must be added to the result. When a 5.7 dB MLP is used, this total correction factor is 7.46 dB.

### **Flatness**

The maximum peak to valley excursion of the transmission response over the specified bandwidth.

### **Match**

See Return Loss

### **Peak, Peak Level, Peak Carrier Level, Peak of the Carrier, or Peak Signal Level**

”Peak” has two common uses:

1. The peak level of a signal or carrier is defined as the maximum voltage of that signal or carrier. Generally, a specific measurement period is defined, during which the maximum voltage is recorded. In some cases, only voltages that last for at least a pre-defined duration are recorded, with large voltage excursions of a shorter duration being ignored or limited by a narrow measurement bandwidth.
2. The more common usage of “peak” refers to the highest amplitude of a displayed signal on a spectrum analyzer. It is often necessary to position the spectrum analyzer marker on such a maximum value (generally with the “peak search”

feature) to get a proper reading. In this case, the quantity to be measured is the maximum value of the RMS voltage and is not the peak voltage or peak power of a signal.

Unless specifically stated, “peak” will refer to the second definition. The quantity to be measured is the maximum value of the RMS voltage and is ***not*** the peak voltage or peak power of a signal.

### **Return Loss**

Ratio between the level of a signal impinging on a port and the level of the signal reflected back from that same port.

### **Signal Level**

Signal Level has two common uses:

1. Signal Level can be used to define the amplitude of a baseband signal. When used in this way, the term “signal” is reserved for references to baseband information, while carrier level or channel power is used for references to modulated RF carriers.
2. Signal Level can be used to define the level of modulated RF carriers.

To avoid confusion between these two definitions, the term “Signal Level” should be avoided. “Baseband Signal Level” should be used for baseband signals and “Channel Power” should be used for modulated RF carriers.

### **Slope**

A measure of the monotonic frequency response of the network from low to high frequency. Slope is positive, or upward going, if the gain increases as the response is swept from low frequency to high frequency.

### **Tilt**

The variation in level across the operating range of the network. Positive tilt is defined to occur if the signals at lower frequencies are lower in amplitude than those at higher frequencies.

## **3.0 TEST PLAN CONSIDERATIONS**

Many factors must be considered in order to assure confidence that performance tests will provide valid and useable results. Care must be taken to insure that test results are not biased by pre-conceived notions of what is expected. The following are several factors to consider when establishing a test plan. These factors should be established for both the forward and return paths and both spectral components should be present simultaneously if actual operating conditions are to be analyzed.

### 3.1 Test Signal Source

The signal loading required for device or system evaluation must be determined as part of the test plan. Additionally, the same loading (or one that is at least substantially identical) must be used in all tests which are to be used for comparative purposes. Tests should be performed with a mixture of analog and digital signals that represent the anticipated operating conditions of the device. Though it is certainly possible, and in some cases useful, to test with a full loading of analog channels, there are meaningful arguments to be made for having a mixture of analog and digital channels so that one may optimize the set-up of a device for its intended use.

Some of the decisions to be made regarding the composition of the test spectrum (for both the forward and return path tests) are:

1. The number and frequency of the analog signals (channels)
  - a. Will the traditional FM band be included (ie., EIA channels 95-97, etc.)?
  - b. Will out-of-band data signals be included?
  - c. The worst-case conditions for some parameters will differ with different signal loading.
2. The number, frequency and type of digital signals (channels)
  - a. What assigned bandwidth, what data rates, what modulation protocol, what level relative to analog signal loading, etc.?
  - b. How much guardband will be provided between channels (if any)?

One example of a test spectrum plan is presented below as an illustration of the process and necessary items.

#### **Forward Signal Path; 54 to 1002 MHz**

1. Full 498 MHz CW carrier loading in the forward path, with channels 95-97 (82 total channels per EIA 542 channelization plan)
2. 200 MHz of digital signal loading using 6 MHz wide 64 QAM or 256 QAM channels. As an alternative, reduced amplitude CW carriers at the frequencies of data channels may be used, but this is not the preferred method.
3. The digital signals will be tested at -6 dBc relative to the analog portion of the spectrum.

#### **Return Signal Path; 5 to 42 MHz**

1. 35 MHz of digital signal loading using 1.5 MHz wide QPSK channels

- |   |
|---|
| <p>2. Actual comparative testing, for noise and distortion performance, may be conducted using 6 CW carriers within the pass band of 5 to 42 MHz.</p> |
|---|

Many different combinations of CW carriers, AM modulated signals, noise and actual digital signals are possible. As an alternative to discrete digitally modulated carriers, highly filtered broadband noise may be used. In this case, the spacing between channels must be defined in order to provide notches for measuring distortion products. When a digital signal is specified, the modulation type and occupied bandwidth must be included.

Other signal loading/spectrum schemes may be devised to ensure comprehensive testing of equipment capabilities. Full analog channel loading is useful to demonstrate the worst-case performance for composite distortions, but the addition of digital signals has added other impairments such as Carrier-to-Intermodulation Noise (CIN).

### 3.1.1 Test Signal Frequencies

Signal carrier frequencies used in broadband telecommunications systems in the US generally follow EIA Standard 542. This standard is based upon the traditional broadcast television frequency plan, which now incorporates the prescribed aeronautical offsets of  $\pm 12.5$  kHz (or multiples thereof) for analog channels within specific portions of the transmitted spectrum. Although this frequency allocation plan represents “real life” conditions, it is not generally used for purposes of product qualification testing. The major reason for not using aeronautical offsets is economics and convenience. It would be quite expensive and troublesome to retune all of the existing fixed frequency signal generators in use today.

In addition to this, the original frequency plan with visual carriers placed 1.25 MHz above the lower channel band edge actually yields a more severe “worst case” distortion contribution than the newer requirements. With the aeronautical frequency offsets, the beats are bound within a larger spectral distribution pattern, given equal frequency stability requirements. Therefore, aeronautical offsets are not recommended for product qualification testing since this could alter the absolute levels of some distortion components. If aeronautical offsets are used, the distortion measurement procedure must be modified to ensure that the entire distortion beat spectrum is included within the passband of the measurement device.

Another consideration when determining frequency assignments for testing is where to operate high frequency data channels. It might be desirable to set these carriers with a +1.25 MHz offset relative to the standard analog channel visual carrier frequency assignments in order to cause CTB distortions to accumulate between the data channels rather than within them. If such a plan is anticipated the test signal generator should be configured to test accordingly.

Distortion products caused by analog channel loading are fairly predictable and it should be possible to determine which channels will be most affected by a particular distortion parameter. Different distortions are likely to have their “worst case” effect on different channels. The “worst case” channel may change depending on the analog signal loading, the signal distribution (how many channels are used, channel bandwidth and frequency allocation) and the degree and type of tilt implemented.

Distortions caused by digital carriers should also be predictable based upon their modulation characteristics and spectral positions. Some low bandwidth data signals may actually act more like analog carriers. Other data signals may manifest a third order distortion which closely resembles noise (carrier-to-intermodulation noise, CIN) and is additive to thermal noise. Some of the distortion products caused by digital carriers will be transient in nature since the signals themselves are transient.

### 3.1.2 Operating Test Source Levels

After the composition of the test spectrum is determined, the operational conditions under which the device or system is to be loaded must be established. The fundamental levels and tilts of the signals, which are encountered in “real life”, vary greatly. A house amp, for instance, encounters a wide range of input signal levels and tilt conditions based upon where it is located along the distribution line. The exact test conditions used must be determined with the understanding that every variable in test conditions requires an additional round of testing in order to acquire a complete picture of a device’s performance.

As implied in the preceding paragraph, it is essential to determine the carrier levels used for testing. These levels must be established for the device’s input and output to ensure that it is operating within its specified range. In some cases, as in line amplifiers and headend products, intermediate levels must also be established and maintained. Different sections of broadband communications plant will require different levels and ratios between forward and return signal levels. The “in-home” environment generally demands that return path signals be much higher in level than forward path signals. Equipment used in the in-home environment will normally not encounter as many simultaneous signals in the return path as the coaxial distribution plant. In addition to establishing the levels, the operational tilt at which a unit under test will operate must be determined. The tilt is often different for input, output and intermediate signal locations and the character of the tilt may be the inverse of the cable loss or linear.

In a complex spectrum, which contains both analog and digital signals, the digital signals must be set to a proper level relative to the adjacent analog carriers. This task is complicated when the digital signals vary in modulation scheme and bandwidth. To determine the “worst case” performance for a

particular parameter, an alternative spectrum loading configuration may be required. If it is not apparent which spectrum load will provide a “worst case” scenario then a complete range of anticipated level and tilt conditions must be used to determine the potential for performance variations as a device is actually applied in the field.

### 3.2 Device Under Test (DUT) Configuration

Tests can be conducted for an individual device, a cascade of similar devices or a combination of different devices to represent much or all sections of a complete system from headend to in-home terminal.

If anything more than an individual unit is to be tested then the amount and nature of losses to be incorporated between the various components must be considered. Different plant sections also have different characteristics. The trunk portions of the plant generally have more cable loss and less passive loss than the distribution and the characteristics of those losses will not react the same to temperature variations.

Sufficient test points must be included to allow for absolute and relative measurements at every potential input and output, for both forward and return signal paths.

### 3.3 Environmental Requirements

#### 3.3.1 Temperature Testing

1. It is recommended that all equipment should be tested over the full specified temperature range. For equipment used in outdoor plant, this should include soaking the instrument at an appropriate cold temperature (manufacturer’s minimum operating temperature) with the power off and verifying that the DUT will start and operate normally.
2. Measurements should be taken at both extremes and at ambient as a minimum. Other temperatures may prove to be useful for analytical tasks (i.e., product development duties may suggest that measurements be made at 20° intervals in Celsius).
3. A "Temperature vs. Time" chart is recommended to track temperature variations as well as their timing and duration. Sufficient temperature probes should be used to track ambient temperature as well as the “core” temperature of any cable which is used. Placement of the cable's temperature probe(s) is especially critical for accurate tracking.
4. The stability of the cable's temperature, rather than ambient, should provide the trigger for each round of testing.

#### 3.3.2 Additional Environmental Tests

The following is a list of additional environmental tests, which may be done if specified by the manufacturer. Detail on how to perform these environmental tests can be found in several reference specifications, including MIL spec PRF-28800.

1. Humidity – This test is typically done by verifying the performance of the DUT after soaking the DUT at an elevated temperature and humidity for an extended period of time.
2. Salt Spray Corrosion - Tests for performance in concentrations of salt spray or other corrosive elements are difficult and the user should refer to MIL specifications.
3. Water Resistance - Resistance to water may be measured using several different methods from drip to blowing rain to total submersion. Most test equipment intended for outdoor use is designed to withstand a blowing rain and should be tested using a controlled volume of water directed at the instrument from vertical  $\pm 90^\circ$ .
4. EMI and Susceptibility - These tests measure the level of radiated emissions from the instrument and the instrument's susceptibility to measurement errors caused by environments with high RF signal levels. The user should refer to IEC 801 specifications for the appropriate test procedures to be used.

### 3.3.3 Test Sample Considerations

1. Typically testing one DUT will verify if the device is within the manufacturer's published specifications. If the measurements results are very close to the published specifications, a larger sample may be required.
2. Devices tested should be from a random sampling of production released product, not early prototypes or lab samples.

## 4.0 TEST EQUIPMENT

### 4.1 RF Power Meters

The power meter is the fundamental test instrument for measuring signal levels. It provides the highest degree of measurement accuracy and is most suitable as a standard against which other signal measurement devices can be calibrated. A power meter provides no frequency selectivity and therefore the user must isolate the carrier to be measured. Analog television signals, with their visual, aural and color carriers, cannot be measured properly because the power meter will measure the total power of all three carriers.

The power meter may be used as a standard reference for comparison with other signal level measurement devices. The difference in level measurement between the

power meter and the frequency selective device may be used as a correction factor for future measurements. By making comparison measurements between the two devices across the entire frequency band of interest, a table of correction factors may be created. The two most common power sensing devices in use today are thermocouple sensors and diode detectors. Each method of measuring average power has some advantages and the preferred method is dependent upon the particular measurement situation.

#### 1. Thermocouple Sensors

Thermocouple sensors typically have better return loss and higher maximum input level. They are normally limited on the low-end to about +25 dBmV, but can easily measure to > +60 dBmV. Thermocouple sensors also measure true RMS power, even when measuring complex signals.

#### 2. Diode Detector Sensors

Diode detector sensors have the advantage of being able to measure lower level signals, typically to < 0 dBmV. The maximum input level to a diode sensor is typically about +25 dBmV, although higher-level sensors have recently become available. The response time of diode detector sensors is much quicker, so automated tests will operate faster. Older diode sensors may measure incorrectly with complex signals. It is also important to operate in the square law region of the sensor for best accuracy.

With the continuous improvement in power meter technology, the differences between measurement technologies are becoming smaller. For a thorough discussion of power meter fundamentals refer to Reference [6].

### 4.2 Signal Level Meters

A standard broadband communications signal level meter (SLM) is actually a frequency selective voltmeter. In its simplest form it combines an envelope amplitude detector circuit with a tuning circuit, a switchable attenuator and a display device. Modern high quality SLMs with digital readouts are actually very good devices for setting amplitude modulated (AM) signals. Though not as accurate as the power meter they have historically been more accurate than the more complex spectrum analyzers, since they are designed for a very specific task.

### 4.3 Signal Analyzers

#### 4.3.1 Spectrum Analyzers

A third type of signal amplitude measurement device is the spectrum analyzer. The spectrum analyzer is the most versatile device for measuring relative signal differentials, especially for those with a frequency offset from the reference carrier or signal. It provides a powerful visual display which allows for simultaneous viewing of a multitude of signals, both intentional and

unintentional. Spectrum analyzers have been developed to a very high degree of usefulness, incorporating some very powerful features.

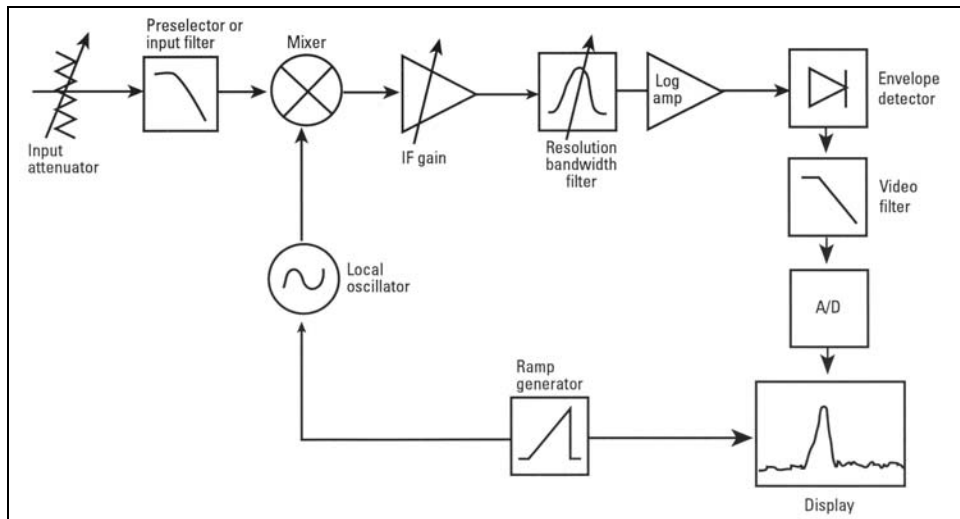
The spectrum analyzer is more flexible than the SLM or the power meter but historically has not been as accurate for measuring levels. However, the latest generation of digital spectrum analyzers have amplitude specifications similar to the best signal level meters.

Among the more useful features one should look for in a spectrum analyzer are:

1. Normalization or “zeroing” of certain functions in order to allow for accurate differential measurements.
2. Markers for absolute amplitude and frequency measurements and measurements relative to a user selectable reference.
3. Comprehensive self-test and self-calibration check.
4. Complete “on-screen display” of significant data points including equipment settings, levels, frequencies, etc. Additionally, the displayed information should be included in any printout or data storage schemes.
5. Standard data communications port for file transfer to printer, plotter or data storage device.

Some of the features, which are available for spectrum analyzers, may not be ideally suited for lab or bench test, although they are certainly appropriate for field use. A preamplifier is often required for better sensitivity. If an internal preamplifier is used, it is important to characterize it adequately in order to understand its impact on the measurement being made. If this characterization is not possible, an external amplifier is preferred.

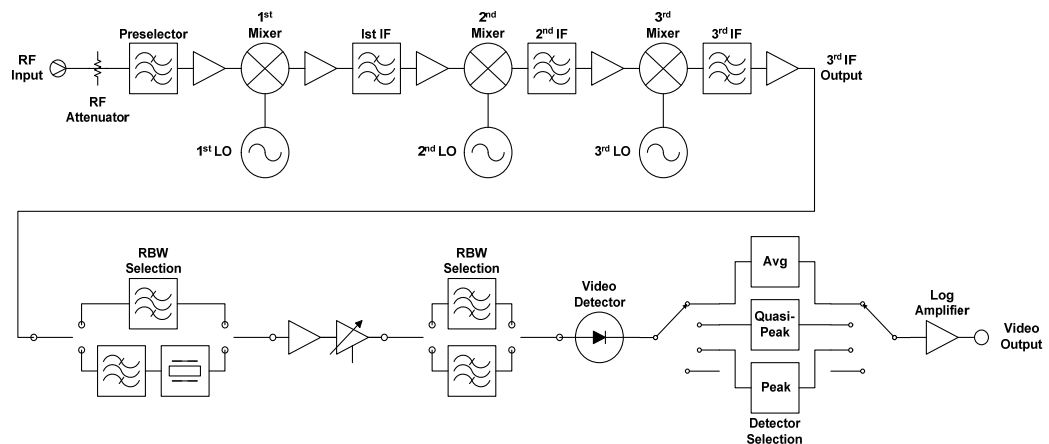
Another area of caution when using an analyzer (or signal level meter) is the accuracy of the internal attenuator. Historically attenuators are generally only capable of 10 dB steps and it is difficult to determine the true attenuation of each step. This becomes critical when attempting to make relative measurements. Newer analyzers do an excellent job of compensating accurately for the internal attenuator. A typical spectrum analyzer block diagram is shown in Figure 1.



**Figure 1 – RF Spectrum Analyzer Block Diagram**

#### 4.3.2 RF Receivers

An alternative solution to the RF spectrum analyzer is the RF Receiver. The RF Receiver typically provides more flexibility in the measurement configuration and allows higher dynamic range measurements, than typical Spectrum Analyzers. The RF receiver provides the frequency conversion to IF and demodulation of the signal along with additional measurement functionality. A typical RF Receiver block diagram is shown in Figure 2.



**Figure 2 – RF Receiver Block Diagram**

The following section discusses differences between the RF Spectrum Analyzer and RF Receiver, including advantages and disadvantages.

**RF Input Performance** - Both block diagrams (Figures 1 and 2) indicate RF pre-selection before the 1<sup>st</sup> mixer which limits the power level present at the 1<sup>st</sup> mixer and minimizes distortion products generated within the receiver's front

end. Pre-selection filtering is standard in most RF receivers, but has traditionally not been available in any but the most expensive spectrum analyzers. Many newer spectrum analyzers offer internal pre-selectors, but when using older analyzers, external pre-selection filters are required for best dynamic range.

RF Receivers also typically include a low noise preamplifier between the pre-selector and 1st mixer which establishes a lower noise figure for the RF front end and provides the best sensitivity to low level signals. If using an external preamplifier, it is important to place it after the external pre-selection filter and verify that the dynamic range of the preamplifier is sufficient to handle the signal levels being measured without contributing additional distortion.

Since RF receivers are specifically designed for receiving and demodulating narrowband communications signals, they are typically designed with an emphasis on minimizing the phase noise of the 1st local oscillator which is used to up convert the incoming signal to the first IF frequency. Spectrum Analyzers have traditionally been designed with an emphasis on sweep speed and may not have the same low phase noise performance RF Receiver. This is also changing over time, and spectrum analyzers with excellent phase noise performance are becoming much more affordable.

Detectors - The spectrum analyzer block diagram in Figure 1 shows an envelope detector for demodulation of the signal's AM component followed by a video peak or sampling detector. This demodulation / detector scheme is not ideal for noise-like signals, although correction factors can be used to compensate.

RF receivers typically provide a wider selection of detector choices for different measurement requirements. This selection of detectors will normally include a Square Law or RMS detector. The Square Law or RMS detector is also now available on the more expensive spectrum analyzers. The Square Law or RMS detector provides the actual "root-mean-square" value of the signal as well as for the noise. More detail is provided concerning detectors in Section 4.4.

### 4.3.3 Baseband Analyzers

The Baseband Analyzer's main use is to resolve signals components which are largely composed of frequencies in the low Hz to tens of MHz range, into their respective amplitudes. The spectral amplitudes are then plotted on the instrument's front panel display, in much the same format as the RF Spectrum Analyzer display (i.e. as a 'spectral density') using units of voltage, instead of units of power. Much of the RF Spectrum Analyzer display functionality is also built into the display of the Baseband Analyzer. However, the internal hardware of the Baseband Analyzer is closer to an oscilloscope than an RF Spectrum Analyzer.

The Baseband Analyzer consists mainly of three important functional blocks: the input ‘instrumentation amplifier’, the A/D (analog-to-digital) converter and the Signal Processor. The input instrumentation amplifier can be thought of as a high performance ‘Operational Amplifier’ design. It provides a stable input impedance, wide bandwidth, low noise, fast rise time and low distortion, for signals which can contain large voltage/amplitude swings. Most importantly, it maintains this key performance to DC voltages. These signals are then converted to discrete samples of the waveform by the A/D Converter, which are then converted to a spectral density plot, using the algorithm of a DFT (Discrete Fourier Transform). Besides providing the DFT, the Signal Processor also allows other computations to be carried out on (multiple) waveforms stored in memory.

High performance Baseband Analyzers contain enough resolution in the A/D Converter and performance of the amplifiers to reach dynamic range levels of 90 dBc, or better, from waveforms that have amplitudes in the volts range. Therefore, the Baseband Analyzer instrument is often used to measure the detected distortion in an RF signal, after demodulation.

#### 4.3.4 Measuring Digital Signals

##### 4.3.4.1 Digital Power Measurements

Measuring the power of a digitally modulated carrier is not as straightforward as the procedure for measuring analog video signals. By definition, the power of a digitally modulated carrier is the power as measured by a power meter which uses a thermocouple as a transducer. This is the true RMS value of the sinusoidal signal which would produce the same heating in a 75  $\Omega$  resistor as does the actual signal.

The correct way to measure the power in a noise-like digitally modulated carrier using a spectrum analyzer or signal level meter is to:

1. sample the average power values at equally spaced frequency points across the bandwidth of the carrier
2. integrate the linear value of these samples
3. correct the result for envelope detection error and noise equivalent bandwidth of the measurement instrument
4. convert the result to a logarithmic value for display

Because this measurement is made across a specified bandwidth, the bandwidth of the channel measured becomes an integral part of the measurement. The measurement result is typically specified in one of two ways. It may be expressed as:

xx dBmV (Occupied Channel BW)

Example: +23 dBmV (2 MHz)

or

xx dBmV (1 Hz)

Example: -40 dBmV (1 Hz)

Both of these results represent the same amount of power in a 2 MHz channel. Quite often in the first example, the “2 MHz” is dropped and the result is just +23 dBmV. In this case, the operator needs to know the bandwidth of the channel measured. In the second example, the total power of the channel is not known without knowing the channel bandwidth.

Some measurement devices assume a flat noise characteristic across the bandwidth of the channel and make a single measurement in the channel and calculate the total power of the channel using the known channel bandwidth. This approach is only accurate when the channel is indeed flat or has the same shape characteristic as was used to calibrate the instrument’s correction factor.

#### 4.3.4.2 Digital Impairments

Bit Error Rate (BER) is defined as the ratio of the bits in error in a data stream to the transmitted bits in a given time period. BER is accepted as the ultimate measure of success for the transport and distribution of digital signals.

BER measurements are dependent on the modulation type and type of Forward Error Correction (FEC) used. The performance of a digital communication channel in an RF network with FEC experiences a cliff effect which causes the BER to behave well in the presence of lower levels of impairment and degrade rapidly as the impairment is increased. Therefore, reliance on post FEC BER as the only indication of a communication channel’s performance provides very little information concerning operating margin. In normal operation, a post FEC BER display will generally indicate a zero BER. Indications of a non-zero BER following FEC will normally be accompanied by a low Modulation Error Ratio (MER) reading and should be cause for concern.

BER provides only a quantitative indication of a communications channel’s performance. It does not provide any information regarding the cause of the errors. Therefore, other parameters such as Modulation Error Ration (MER), Error Vector Magnitude (EVM), adaptive equalizer tap

analysis and constellation analysis should be used in conjunction with BER measurements. For those needing BER measurements, either at the transmission level or after forward error-correction, the techniques required are well documented in ANSI/SCTE 121 2006.

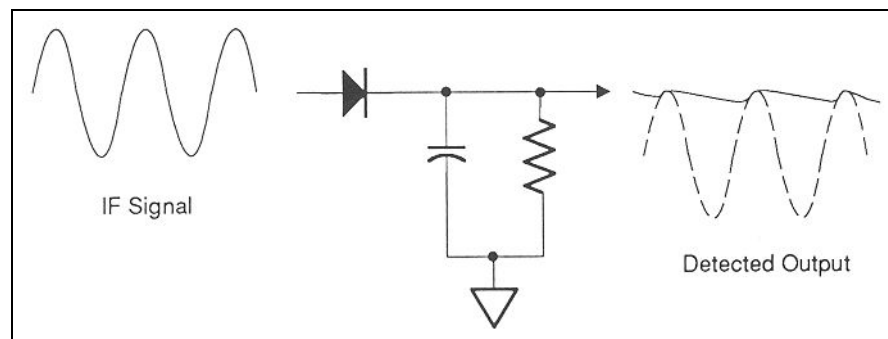
#### 4.4 Detectors

##### 4.4.1 IF Detectors

##### 4.4.1.1 Envelope or Linear Detector

Spectrum analyzers and signal level meters typically convert the IF signal to video with an envelope or linear detector. In its simplest form, an envelope detector is a diode followed by a parallel RC combination (see Figure 3). The detector is nonlinear as far as the RF carrier is concerned but linear as far as the modulation is concerned. The detector is often analyzed as a mixer with the carrier as the local oscillator, but may also be analyzed as a half wave or full wave rectifier. The purpose of the low pass filter is to separate the modulation from the RF or IF carrier.

The output of the IF chain is applied to the detector and the detector time constants are such that the voltage across the capacitor equals the peak value of the IF signal at all times. That is, the detector can follow the fastest possible changes in the envelope of the IF signal but not the instantaneous value of the IF sine wave itself (typically 10.7 MHz or 21.4 MHz).



**Figure 3 - Envelope Detector.**

For most measurements, a narrow enough IF resolution bandwidth is chosen to resolve the individual spectral components of the input signal. When measuring analog video signals, the IF resolution bandwidth needs to be sufficiently wide to pass enough of the horizontal sync spectral components for detection of the peak sync tip. The envelope detector follows the changing amplitude values of the peaks of the signal from the IF chain but not the instantaneous values and gives the analyzer or signal level meter its voltmeter characteristics.

When used with random noise, an envelope detector creates a reading which is lower than the true RMS value of the average noise. This difference is 1.05 dB. Thus, if we measure noise with a spectrum analyzer using voltage-envelope detection (the “linear” scale) and averaging, an additional 1.05 dB needs to be added to the result to compensate for averaging voltage instead of averaging voltage squared.

If the logarithmic display mode is being used, the log shaping used in spectrum analyzers amplifies noise peaks less than the rest of the noise signal. Because of this, the reported signal level is smaller than its true RMS value. The total correction for the log display mode combined with the detector characteristics is 2.51 dB, and should be used any time random noise is being measured in the log display mode. A thorough mathematical analysis of these correction factors is contained in Reference [14].

#### 4.4.1.2 RMS Detector

RMS detectors display the root of the mean of the square of the signal and are the only commonly used detector that can measure true power or the power in a non-sinusoidal signal. All of the previous detectors display power by assuming a sine wave input and calibrating the display. This is satisfactory until more than one random signal appears at the detector. RMS detectors read the true power by measuring the RMS voltage of the signal.

#### 4.4.1.3 Square Law Detector

Square Law detectors display the mean of the square of the signal and also measure the true power of the signal. The output of the square law detector is a linear function of the input power, a fact that is sometimes useful. Very early detectors were square law and linear detectors are square law at low signal levels. Wide range square law detectors became practical with the availability of analog squaring circuits. The Square Law detector does not reproduce the modulation waveform without distortion.

The basic difference between the square law detector and the linear detector can be expressed mathematically as follows.

$$\text{Linear Detector} \quad V_{OUT} = kV_{RF} \quad (11)$$

$$\text{Square Law Detector} \quad V_{OUT} = kP_{RF} = k(V_{RF})^2 \quad (12)$$

where

$$V_{OUT} = \text{low frequency output voltage}$$

$k$  = detector constant

$V_{RF}$  = RF voltage

$P_{RF}$  = RF power

#### 4.4.2 Video Detectors

##### 4.4.2.1 Positive and Negative Peak Detector

Positive and negative peak detectors take the output of the peak or linear detector and capture and display either the positive peak or the negative peak of the noise level. This detector is used for measuring the maximum or minimum level of the signal over a period of time.

##### 4.4.2.2 Sample Detector

Sample detectors are used in analyzers that have internal digital processing. In order to limit the size of the data being processed, the input from the linear detector is sampled and processed. Usually this is invisible because there are hundreds of sampled points across the screen.

##### 4.4.2.3 Rectifying Detector

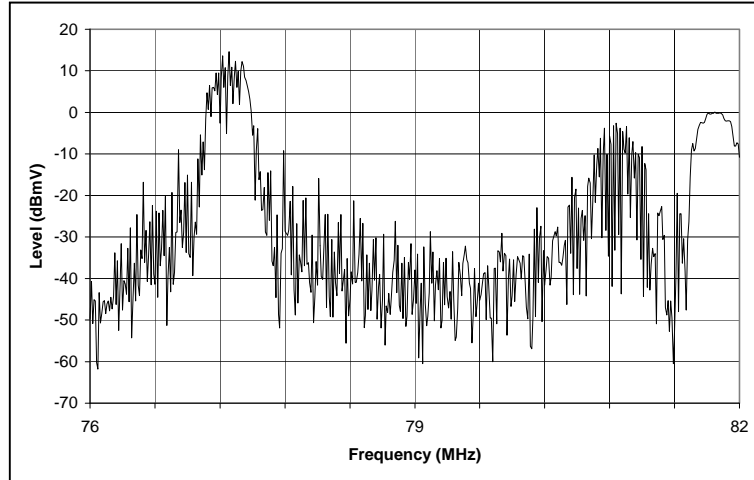
The rectifying detector is an analysis of the detector as a rectifier. The Linear detector is usually analyzed in terms of fast peak charge and slow discharge. The rectifying detector and the linear detector are equivalent.

##### 4.4.2.4 Rectifying Averaging Detector

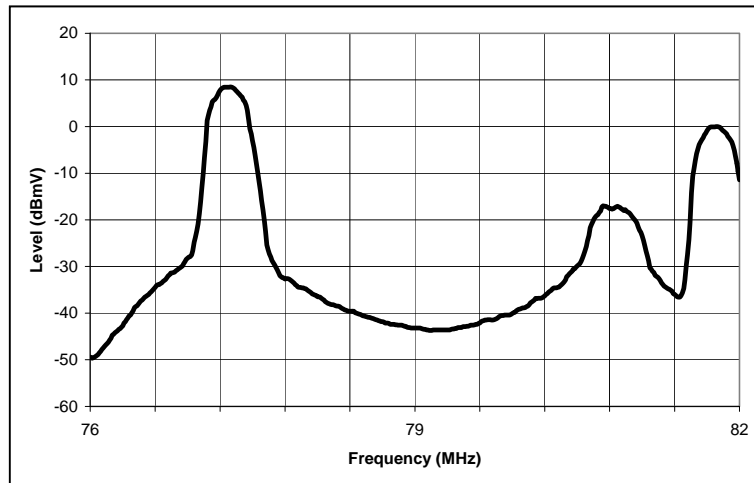
The use of the term "Rectifying Averaging Detector" became necessary when measuring noise because it is necessary to accurately know the bandwidth of the output of the linear detector. For all practical purposes, "Rectifying Averaging Detectors" are equivalent to linear detectors.

#### 4.4.3 Video Filtering

Spectrum analyzers display signals plus their own internal noise. To reduce the effect of noise on the displayed signal amplitude, the video can be smoothed or averaged, as shown in Figure 5. Most analyzers include a variable video filter for this purpose. The video filter is a low-pass filter that follows the detector and determines the bandwidth of the video circuits that drive the A/D converter. As the cutoff frequency of the video filter is reduced to the point at which it becomes equal to or less than the bandwidth of the selected IF resolution bandwidth filter, the video system can no longer follow the more rapid variations of the envelope of the signal passing through the IF chain. The result is a smoothing of the displayed signal.



**Figure 4 – Sample Detection**



**Figure 5 - Video Filtering**

The effect is most noticeable when measuring noise, particularly when a wide resolution bandwidth is used. As the video bandwidth is reduced, the peak-to-peak variations of the noise are reduced. The degree of noise reduction is a function of the ratio of video to resolution bandwidth. At ratios of 0.01 or less, the smoothing is very good.

#### 4.4.4 Video Averaging

Today's digital analyzers provide video averaging as an alternative approach for display smoothing. Video averaging is done over two or more sweeps on a point-by-point basis. At each new display point, the new data value is averaged with the previously averaged data. Video averaging retains the point-to-point accuracy of the analyzer at the same time it smooths the display. It

accomplishes this at the expense of update rate since it takes several sweeps to gradually converge to an average.

If the signal being measured is noise or a very low level signal near the noise, the effects of video filtering and video averaging are very similar. The significant difference between the two smoothing approaches is that video filtering is a real time measurement and its affect is seen on each point as the sweep progresses. Video averaging requires multiple sweeps and the averaging at each point takes place over the time period required to complete the multiple sweeps.

Using video filtering, a signal with a spectrum that changes with time will yield a different average on each sweep. But if video averaging is used, the final result will be much closer to the true average.

#### 4.5 Signal Sources

Signal generation devices for both forward and reverse signaling paths, whether analog, digital or both, form the foundation upon which all comparative measurement data is taken.

It is recommended that a “flat” spectrum be established at the various sources and that any desired spectrum tilt be imposed through an external device such as a tilt equalizer or equivalent (“True Tilt Networks,” for instance). This will simplify the alignment of the source and provide a more repeatable measurement. Tilt can be either positive or negative. The tilt used may vary as to the absolute amount, direction (positive or negative) and type (“linear” or “cable”). Some devices are aligned for linear tilt or no tilt at their output, while others are aligned for flat input prior to testing. Regardless of the alignment used, the input and output amplitudes must be clearly specified and documented for all tests.

Signals now come in two basic types — analog and digital (actually, all signals are analog in the RF domain). For testing purposes the actual difference between the two types relates more to the presence or absence of modulation. Continuous Wave (CW) signals are normally used to represent standard NTSC modulated signals, for all tests except cross modulation. It has been demonstrated that this type of signal will allow a close approximation of standard television signals for analysis of actual system performance under “worst case” conditions. Digital signals, on the other hand, are much more varied in nature, ranging from low-bandwidth (100 kHz) FSK control signals to relatively high-bandwidth signals, such as the 6 MHz 64 and 256 QAM channels currently being used within broadband communications plants. Digitally modulated signals do not have a stationary “carrier” which can be measured accurately with the customary “peak detection” instruments currently in use.

##### 4.5.1 Analog Signal Generation

The total test system requires a multi-carrier analog signal generator to simulate the channel loading found in real cable telecommunications systems. Typically, continuous wave (CW) carriers are used for laboratory testing, in order to produce repeatable results. For cross modulation measurements (XMOD), these CW carriers must be modulated by a well-defined signal.

To meet these requirements, the multi-carrier analog signal generator must have the following basic features:

- The generator must be capable of creating CW signals at all the required frequencies, according to the desired frequency plan.
- The output power of each of the CW signals must be individually adjustable, at a level sufficient to produce the desired testing conditions.
- The CW signals must be non-coherent, i.e. each must have their own free running reference signal and the stability of the reference must be sufficient to keep the beats within the measurement resolution bandwidth .
- The CW signal at each measurement frequency must have the ability to be turned off. The power of a signal when it is in the 'OFF' state must be at least 10 dB below the power of impairment being measured.
- Every CW signal must have the ability to be modulated with a 50 % duty cycle, downward only, square wave 100% modulation.
- The modulation signal must be coherent, i.e. the same modulation signal must be applied to all channels.
- The frequency of the modulation signal must be 15.750 kHz, or equal to the desired horizontal synchronization pulse frequency.
- The modulation at each measurement frequency must have the ability to be turned 'OFF'.

#### 4.5.1.1 Spectral Purity

One of the most important performance characteristics of a multi-carrier analog signal generator is spectral purity. The spectral purity of a signal describes how closely the actual signal matches the desired signal. Because no real signal generator is perfect, there will always be some difference between the actual signal and the desired signal. Consider the 55.25 MHz carrier in the output of a multi-carrier signal generator. Mathematically, the only spectral component in the frequency range from 52.25 to 58.25 MHz is a pure cosine at exactly 55.25 MHz. In reality, this frequency range will include a large signal at a frequency very close to 55.25 MHz, some low level interference signals at specific frequencies,

and noise. The interference signals may be unwanted distortion, spurious signals, or modulation occurring within the signal generator.

There are several possible sources of noise (e.g. quantization noise, thermal noise, impulse noise), but the noise of most multi-carrier analog signal generators may be treated as having two contributions, phase noise and broadband noise. Every signal that is generated includes some amount of phase noise. For an introductory discussion of phase noise, refer to Reference [10]. Every real device also has broadband noise associated with it. The amount of broadband noise depends primarily on the temperature and the noise figure of the device. For an introductory discussion of broadband noise, refer to Reference [9]. In order to measure typical devices, which have low distortion and noise levels, the spectral purity of the multi-carrier analog signal generator must be very good. For most applications, the distortion and noise levels of the multi-carrier analog signal generator must be at least 10 dB below the levels to be measured. If the internal CSO, CTB, or XMOD of the signal source(s) is produced in a way similar to the CSO, CTB, or XMOD of the Device Under Test (DUT), the internal CSO, CTB, or XMOD products must be at least 20 dB below the levels to be measured.

#### 4.5.1.2 Stability

Another critical characteristic of the multi-carrier analog signal generator is stability. To obtain repeatable measurements, the output power and frequency of each CW signal must be stable.

The output power stability of each signal directly affects the measurement repeatability. The performance of the device under test often varies depending on the signal power, so each 0.1 dB change in input power can produce a 0.1 or 0.2 dB change in the measurement. The measurements are often measured relative to the carrier power, so if the carrier power changes, the measured result will change by an equal amount. Because of these dependencies, it is imperative that the power of each signal remains constant, not only over time, but also when the surrounding signals are turned 'On' and 'Off', and when the signal itself is turned 'Off' and then back 'On'.

The frequency stability of the signals also affects the accuracy and repeatability of the measurements, because the frequency distribution of distortion products is dependent on the frequencies of the carrier signals. If the actual carrier frequencies drift, the frequencies of the individual distortion products will also drift. Some of the individual distortion frequencies may drift out of the measurement bandwidth, producing inaccurate measurements. The individual distortion frequencies may also

drift such that they beat together, producing low frequency variations in the measurements.

#### 4.5.2 Digital Signal Generation

The total test system may require the presence of digital signals in addition to the analog signals. The digital signals may be generated by a band of 16-QAM, 64-QAM, 256-QAM or QPSK modulators. The data stream used for these modulators is typically a pseudo-random sequence.

For some tests, it may be sufficient to use a highly filtered broadband noise source instead of QAM or QPSK signals. If filtered noise is used, care must be taken to assure that the peaks of the noise are not compressed. In general, no amplifiers should be used after the filters. The peak factor (sometimes called crest factor) of the filtered noise should be at least 13 dB. For more information about QAM and QPSK signals, refer to Reference [4].

#### 4.6 RF Post Amplifiers

An RF amplifier may be required to raise signal levels at the receiver to a more useable amplitude. One (or more) of these devices will typically be included in the test system. These amplifiers are normally used after a bandpass filter which limits the bandwidth of signals and total power which the amplifier must handle. It is recommended that NO amplifiers be placed between the signal sources and the DUT.

1. Gain  $\cong$  18 to 25 dB. Too much gain can be just as much a problem as too little.
2. NF  $\leq$  10 dB. The noise figure required will be determined by the nature of the measurement being made.
3. Input / Output Return Loss  $>$  14 dB
4. Frequency flatness within the 6 MHz measurement bandwidth  $<$  0.2 dB peak to valley
5. 1 dB gain compression  $\geq$  10 dB above the highest signal measured

#### 4.7 Filters

Filters are an essential part of the test system. These devices come in various configurations:

1. Bandpass filters allow a selected portion of the spectrum to pass through while reflecting all other portions.
2. Bandstop filters are the opposite of bandpass filters. They stop a selected portion of the spectrum by reflecting the energy, and allow all the rest to go through.

3. Highpass filters allow everything above a specified frequency to pass through while reflecting everything below.
4. Lowpass filters are the opposite of the highpass type, allowing everything below a specified frequency to pass while reflecting everything above it.

Various combinations of the types listed above may also be used.

No filter actually acts as a “brick wall”. Each will have some amount of crossover spectrum where its frequency response changes on a monotonic slope. Some bandpass filters may allow harmonics of the passband frequencies to pass through as well, relatively unaffected. This can cause problems in certain measurements and should be avoided whenever possible. In general, filter performance will degrade as the operating frequency increases. Although this may be compensated for by using more complex filters.

Additionally, filters can be either fixed or tunable. Fixed filters are generally preferred for testing, as their characteristics are more stable. Tunable filters provide flexibility, since they allow testing across a range of frequencies. In either case the filter should be carefully centered around the carrier under test. Table 1 provides the minimum frequency response and bandwidth characteristics for an acceptable filter:

	<b>Loss</b>	<b>Bandwidth</b>
<b>Flatness</b>	$\leq \pm 0.15$ dB	Carrier Frequency $\pm 1.5$ MHz
	$\leq \pm 0.5$ dB	Carrier Frequency $\pm 2.5$ MHz
<b>Rejection</b>	$\geq 25$ dB	Carrier Frequency $\pm 6.0$ MHz
	$\geq 50$ dB	$\geq$ Carrier Frequency $\pm 12.0$ MHz

**Table 1 - Fixed Channel Filter Requirements**

If the bandwidth of a filter is too narrow it may attenuate a beat product which has a significant frequency offset relative to the carrier, such as CSO. This problem is especially prevalent in the measurement of lower frequency channels as the “Q” of the filters for these frequencies is often very good and the bandwidth is less than desired. Misalignment of tunable filters is also an issue and can cause problems with any measurement, including that of noise.

A bandpass filter acts as an open circuit to all frequencies that are outside of its passband. This means that it will cause out of band reflections that may affect the amplitude of other signals in the test system through phase addition and cancellation and can affect the operation of the device under test. This effect can be minimized through the use of a fixed attenuator between the filter and the system. The loss of the fixed attenuator must, of course, be accounted for to ensure that signal levels are adequate for the measurement task.

## 4.8 Coaxial Test Devices

### 4.8.1 RF Attenuators

Attenuators are most useful for improving impedance matching between various devices in the test system and for adjusting signal levels to more desirable amplitudes. These devices come in two basic styles, fixed and switchable. As with everything else within the test environment, caution must be exercised when using attenuators. Some fixed attenuators have such a wide specification tolerance that they are almost useless. The performance of switchable attenuators needs to be verified regularly depending on time and frequency of use. Like other passive devices used within the test system these should be carefully analyzed for accuracy and frequency response characteristics. Records should be made of their initial performance and regular verification tests performed to identify any degradation.

General recommendations for attenuators include the following:

1. 75 $\Omega$  switchable attenuator(s):
  - a. dB steps, 0 - 70 dB range,  $\pm 0.1$  dB tolerance.
  - b. dB steps, 0 - 10 dB range,  $\pm 0.05$  dB tolerance.
  - c. Return loss > 20 dB across the frequency range of interest
2. 75 $\Omega$  fixed attenuator(s)
  - a. 3, 6, 10 and 20 dB values,  $\pm 0.05$  dB tolerance.
  - b. Return loss > 20 dB across the frequency range of interest

### 4.8.2 Cables, Connectors and Adapters

Cables, connectors and adapters are used to interconnect the various components of test apparatus and can introduce undesirable measurement errors and uncertainties if not recognized, compensated for, properly selected, or installed. All of these components will introduce additional reflections and losses in a test system that can significantly alter the frequency response. Even small reflections can lead to significant power holes or “suck outs” if they happen to occur at multiples of  $\frac{1}{2}$  wavelength at any in-band frequency.

#### 4.8.2.1 Cables

The proper cable type depends on the application. For fixed applications a “semi-rigid” or “hardline” cable, or a cable with a solid outer conductor is usually preferred for reasons of cost, performance, and shielding. If a

flexible cable is required, then a braided outer conductor is usually chosen. For flexible cables, the user needs to be aware that the cable's performance can change with bending and the user may need to be concerned with factors such as amplitude stability, phase stability, and shielding effectiveness during flexing.

The cable loss must be accounted for in the test system design, or calibrated out. If long runs of cable are required, the user may need to be concerned about the cable's "structural return loss", which is the result of minor periodicities within the cable all adding in phase at some frequency to produce a power hole or "suck out".

#### 4.8.2.2 Connectors and Adapters

High quality test connectors should be used in any test system and all connectors must be properly installed and torqued. The most common high quality test connector for use in 75  $\Omega$  systems is the "precision" 75  $\Omega$  Type N connector. Caution must be used when using Type N connectors. The 50  $\Omega$  and 75  $\Omega$  types are not intermateable. Inserting the larger-diameter pin of a 50  $\Omega$  male connector into a 75  $\Omega$  female will result in severe damage to the female contact. Conversely, the smaller pin of a 75  $\Omega$  male connector will have intermittent or no contact if used with a 50  $\Omega$  female.

The basic figure of merit for a connector is its "return loss" and the return loss of all connectors, cables, and adapters in a system must be better than that which is to be measured, preferably by at least 10 dB. A loose or damaged connector can seriously degrade the performance of any test system. Test connectors are very delicate and must be inspected and cleaned regularly. The most common failure modes are bent or broken fingers on the female contact, worn or damaged mating interface surfaces, loose coupling nuts, or dirt or contamination in the interface.

#### 4.8.3 Impedance Matching Devices

It may be necessary to employ one or more impedance matching devices within a test system. The broadband communications industry has standardized on a 75 $\Omega$  impedance. Much of the test equipment is geared toward the more common 50 $\Omega$  world of radio communications. In order to use 50 $\Omega$  test equipment one must use an adequate impedance matching device to prevent high levels of signal reflections, which would invalidate any and all measurements. Devices currently available for impedance matching include:

1. Minimum Loss Pad

A minimum loss pad (MLP) is a broadband resistive matching device which provides the correct impedance on both ports over a specified

frequency range. The advantage of the MLP is that it is non-reactive and has the broadest bandwidth of available matching devices. The disadvantage of the MLP is that it has the highest insertion loss. MLPs from  $50\Omega$  to  $75\Omega$  are available with good return loss to 3 GHz. The insertion loss of an MLP when matching  $50\Omega$  to  $75\Omega$  is calculated to be 5.71 dB.

## 2. Matching Transformer

A matching transformer is a broadband matching device which provides the correct impedance on both ports over a specified frequency range. The advantage of the matching transformer is that it has the lowest insertion loss of available matching devices. The disadvantage of the matching transformer is that it is reactive and typically has a lower useable bandwidth. Matching transformers from  $50\Omega$  to  $75\Omega$  are available with good return loss to  $> 1$  GHz. The insertion loss of a matching transformer is typically less than 1 dB.

## 3. Series Resistor

The series resistor is the simplest matching device used to increase the input impedance of a  $50\Omega$  measurement device to  $75\Omega$ . A series resistor is a unidirectional device providing a well matched impedance on the  $75\Omega$  side, but poor match to the  $50\Omega$  side. The advantage of the series resistor is that it is non-reactive and has a broad bandwidth. The disadvantage is the lack of match on the  $50\Omega$  side. The insertion loss of a  $25\Omega$  series resistor used to match from  $50\Omega$  to  $75\Omega$  is calculated to be 1.76 dB.

## 4.9 General Purpose Test Equipment

### 4.9.1 Digital Multimeters (DMM)

Digital Multimeters, also referred to as DVMs, DMMs, or voltmeters, are used throughout these procedures to make measurements on both AC and DC supply voltages. Most modern DMMs have the resolution required for the measurements being made with 4 ½ digits being the lowend, but the user should verify that the accuracy is acceptable for their specific requirements. Typical accuracy for a modern DMM is  $< \pm 0.1\%$  for VDC and  $< \pm 1.0\%$  for VAC. When using the DMM for measuring the amplitude of the ferro-resonant quasi-square wave AC power signal, it is critical that a DMM is used with a true-RMS reading vs. the lower cost peak reading measurement.

### 4.9.2 Oscilloscopes

Two general types of oscilloscopes are currently available. Analog oscilloscopes display continuously variable voltages and have traditionally been

used for video measurements due to better screen resolution. Analog scopes are capable of displaying signals within the frequency range of the CRT display. Very high frequency signals become too dim to see and very low frequency signals become a slow moving dot across the screen. The CRT in analog scopes has a characteristic called intensity grading that makes the trace brighter where the signal features occur most often. This feature makes it easy to distinguish signal details by observing intensity levels. The fastest analog scopes can display frequencies up to about 1 GHz.

Digital storage oscilloscopes (DSOs) use an A/D converter to convert the analog voltage to bits and store the signal information as a series of samples. Digital scopes are capable of displaying any frequency within the sample range of the A/D converter with stability and brightness. For repetitive signals, the upper frequency limit of the digital scope is a function of the front end analog components of the scope. For single shot events, the bandwidth may be limited by the sample rate of the A/D converter. Digital scopes continue to evolve and improve significantly in capability and are becoming much more affordable. Some key oscilloscope characteristics are discussed below.

**Bandwidth** - An oscilloscope's bandwidth specifies the frequency at which a sinusoidal signal is measured at 70.7% of its true amplitude, or the -3 dB point. Without adequate bandwidth, the high frequency details of the signal will be lost, and the displayed signal will be distorted. To guarantee adequate bandwidth, the oscilloscope's bandwidth should be 5 times greater than the highest frequency to be measured.

**Sample Rate** - Sample rate specifies how often the digital oscilloscope's A/D converter samples the analog signal. In order to accurately reconstruct a signal and avoid aliasing, Nyquist theory says that the signal must be sampled at least twice as fast as its highest frequency component. Accurate reconstruction of the signal is dependent upon sample rate and the interpolation method used to fill between samples. Some oscilloscopes allow both  $\sin(x)/x$  interpolation for sinusoidal signals and linear interpolation for square wave and pulse signals. Using  $\sin(x)/x$  interpolation, the sample rate should be at least 2.5 times greater than the highest frequency being measured. Using linear interpolation, the sample rate should be at least 10 times greater than the highest frequency signal component.

**Vertical Sensitivity** – Vertical sensitivity specifies the ability of an oscilloscope to amplify a weak signal and is usually measured in mV per division. Modern oscilloscopes are typically capable of 1 mV/div.

**Acquisition Modes** - Trace averaging can be used to provide better measurement accuracy on low level signals and is a requirement for many of these measurements. When using a digital sampling oscilloscope to measure noise and transient type signals, some form of Peak Detect sampling is required.

### 4.9.3 AC / DC Power Supplies

Three types of power supplies are used throughout these procedures.

DC power supplies are used for powering preamps or peripheral equipment which is part of the test set-up. The critical requirements for these supplies are:

- sufficient current capacity to prevent the supply from going out of regulation during the test, typically 2 times greater than the current required by the test system
- low enough noise performance to prevent the noise of the power supply from affecting the measurement results

Line voltage AC power supplies are used for powering the test equipment and peripheral equipment which is used as part of the test set-up. Grounding of these supplies should be done carefully to prevent ground loops, especially when measuring low level hum signals and low level distortion products.

Ferro-resonant quasi-square wave (trapezoidal signal with  $< 100$  V/ms slew rate) AC power supplies are used for powering distribution equipment under test. The rated current for these supplies should be at least 30% greater than the desired test current.

### 4.10 Test Equipment Calibration

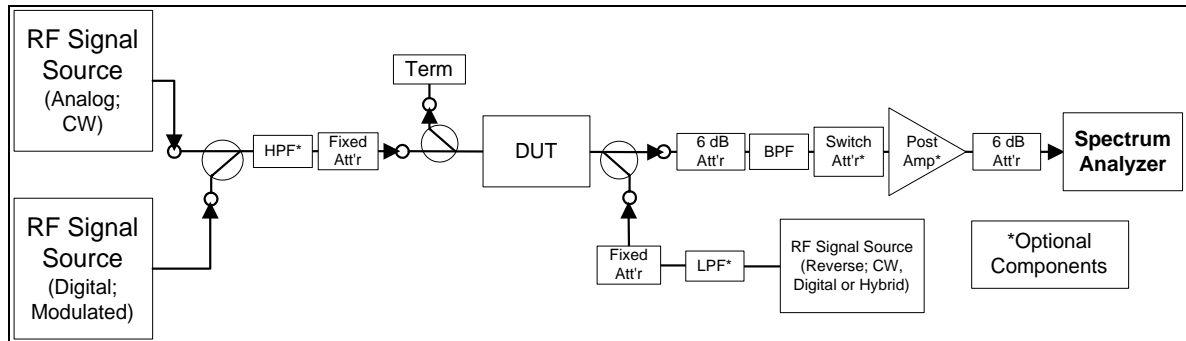
No test can be valid if taken with measurement devices that have not been properly calibrated. The user should follow the manufacturer's recommendations for appropriate factory calibration requirements, warm-up time and field calibration before performing any tests. There are at least two levels of calibration required:

1. Calibration traceable to the original factory specifications
2. Verification which ensures that a piece of equipment is operating properly at the time the measurements are made.

The factory calibration is normally done at regular intervals, as specified by the equipment manufacturer. The operational calibration is done at least daily or even several times a day, depending upon the stability of the test equipment. This frequency of verification will verify that the test unit is performing properly when the testing is initiated. Many modern test instruments have a "self-calibration" feature, which will perform operational checks during boot-up or upon command by the operator. Older equipment, lacking this feature, should be checked against acceptable frequency and amplitude standards. Note that reference standards also require verification and calibration.

## 5.0 THE TOTAL TEST SYSTEM

A simplified block diagram of a total test system is shown in Figure 6. This diagram will serve to demonstrate the configuration of the various components and interconnection of a basic test system. This represents only one way by which valid testing may be achieved. Certainly there are alternative ways of accomplishing the same end. This is presented here simply as a means of detailing some of the major issues involved in establishing a proper test system.



**Figure 6 - Reference Broadband Communications Test System**

### 5.1 A Reference Broadband Communications Test System

Signal sources are shown for both the forward and the return signal paths. The forward path signal sources are combined together at the proper levels and then fed into the input of the test system through a highpass filter (HPF) and a fixed attenuator. The return signal source is treated in a similar fashion; from the source to a lowpass filter (LPF) to a fixed attenuator. The highpass and lowpass filters are optional and are intended to prevent the introduction of out-of-band spurious signals into the test system. The fixed attenuators are required to provide a broadband return impedance match to the directional couplers used at the Device Under Test (DUT) ports. A minimum value of 6 dB is recommended.

After filtering and attenuating, the signal sources are directed through directional couplers (DCs), which are used to provide forward and return signal injection ports and complementary test points. The values for the DCs should be selected to ensure proper signal level control while close attention is given to their bandwidth, frequency response and isolation characteristics.

The DUT may consist of a discrete component, a single device or a complete system. The output of the DUT is connected through a DC, which as mentioned above, provides an output test point for the forward path as well as an input port for the return signal spectrum. The output of the return signal source is connected to the tap port of this DC. If the device to be tested is intended for two-way applications it is critical that both signal paths be present simultaneously during all testing in order to

ensure that all possible distortion products are present, including those derived from loop isolation and common path deficiencies.

After the output DC, a fixed attenuator is used to ensure adequate match between the DUT and the rest of the test equipment. Again, a minimum of 6 dB is recommended. The output of the attenuator is passed through a bandpass filter, switchable attenuator and post amplifier. The bandpass filter minimizes the possibility of overdriving the input of the amplifier by restricting the number of channels which will be present at its input. The bandpass filter actually represents a set of filters, whether fixed or tunable, which will be used to test the designated channels. The switchable attenuator serves two purposes:

1. Ensure that the input level to the amplifier is maintained well above the amplifier's noise floor
2. Maintain a stable input level into the measurement/display device, in this case a spectrum analyzer, in order to minimize the need for constant resetting of the measurement device's controls.

The output of the amplifier is connected to another fixed attenuator and then to the signal display device. Again, a minimum of 6 dB attenuation is recommended. This test equipment configuration may be moved from the forward test point to the return test point as required.

It should be noted that quite a bit of RF loss is incorporated within the test chain. This is an unfortunate, but necessary, requirement. It is important to remember that the filters present a well-matched impedance only to those frequencies in the passband. All other frequencies will be reflected back into the source with very little return loss. The fixed attenuators are required to minimize these reflections. Directional couplers may also present mismatched conditions, although to a lesser degree. A final note; the unused port of the return test point DC is shown to be terminated. Good engineering test practices dictate that all unused ports should be terminated.

## 5.2 Test System Characterization

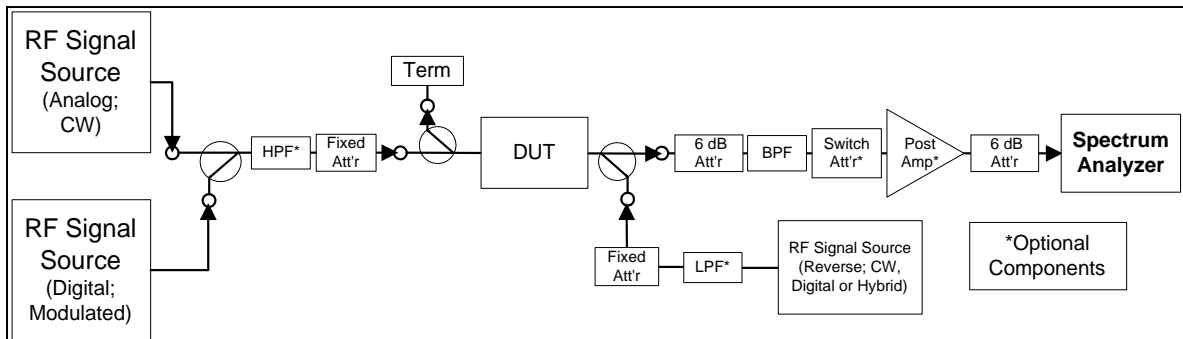
The components used in the test system must be completely characterized for their effects on the test signals. Factors such as gain, loss, frequency response, etc. must be known, and often factored out, in order to ensure accurate measurements DUT. Amplifiers used will contribute some amount of self-generated distortion and noise components that will interfere with the measurement of the parameters actually generated within the DUT. Filters will have some degree of loss and group delay and could have a major impact upon frequency response as well. Other measurement errors may be caused by a filter's bandwidth limitation.

Cables will add frequency dependent losses to any signal and the magnitude of the losses will vary over temperature. Fortunately the variations in losses are reasonably

predictable and can be calculated for the different frequencies. Unfortunately the losses and the effects of changes in temperature differ by manufacturer, size and, in some cases, age of the cable. Another problem may arise if the cable is damaged to any significant degree. A damaged cable can add reflections which result in standing waves. This will be evident by a consistent, repetitive ripple in the frequency response of the cable. Because of these potential problems every piece of cable, whether mainline or drop, should be subjected to a thorough sweep analysis over the maximum possible frequency range and over the entire range of anticipated temperatures.

The performance of every passive device, including cables, connectors, test fixtures, fixed and switchable attenuators, matching devices (pads and transformers), etc. should be tested in order to demonstrate its suitability for the intended purposes. These tests should include frequency response, absolute loss over the entire frequency range, and performance over the temperature range to which the device will be subject. Switchable devices such as A/B switches and switchable attenuators should be verified in every possible position to ensure against potential errors. A complete record of this verification effort, perhaps in an engineering log format, should be compiled for future reference. Additionally, a record should be kept of the swept responses to allow for later comparison if a significant change is suspected.

### 5.3 Characterization Data



**Figure 7 - Reference Broadband Communications Test System**

Every component within the Reference Broadband Communications Test System should be properly and adequately “characterized” in order to determine what, if any, affect it might have on the ultimate test results. Figure 7 will be used to outline those areas which should be analyzed for their potential influence upon the end-of-line measurements.

Knowledge of such parameters as loss (or gain) and frequency response within the test system is especially critical to ensuring proper and accurate data. Return loss (input and output) and distortion characteristics of the different components are also important. Additionally, other factors must be considered within the individual

scopes of specific tests (such as hum or group delay performance) and should be well known in order to demonstrate overall device or system capabilities.

Losses within the test system, from cables, connectors, interfaces, etc., will affect various signal levels and could, possibly, cause the equipment to be operated outside of its specified or desired range. On the other hand, the gain of any active device must be determined as accurately as possible in order to ensure proper application within device or system tests. Noise and distortion generation characteristics of these devices must be established to permit eventual discrimination between these effects and those which are actually introduced by the device or system under test. Frequency response measurements of the entire test system, and its discrete sections or components, will allow the operator to account for any differences in parametric performance within the entire passband of the desired spectrum. Return loss characterization will ensure that proper impedance matching is maintained within the test system.

The same test procedures used for evaluating products and systems should be used to qualify the test system itself, before any other evaluations are attempted.

#### 5.3.1 Using Characterization Data

The characterization data established for test system components should be used in the calculation of actual measurement results when possible. Signal losses through the test system should be accounted for in all absolute and relative level measurements.

The characterization data will be useful in three primary ways:

1. Permit the operator to set all levels with a greater degree of accuracy
2. Allow the operator to “normalize” the measurement results by subtracting the test equipment’s contributions
3. Provide a history to determine if any of the components have changed or degraded

The second point above will benefit from a small amount of expansion. It is perfectly legitimate to deduct the parametric performance of the test system (noise, distortions, delay, etc.) from that of the actual product and/or system, but only if those aspects of the test system are accurately determined and documented. Such parameters should be measured for the test system and then deducted from those measurements performed on the DUT.

#### 5.3.2 Signal Sources

One of the most common failures in system or device testing is inaccurate setting of the signal source. Fortunately, this is one of the easiest oversights to

correct. The first step in testing is to ensure that all source signals are correct, and then take multiple spectrum plots of the various signals by zooming in on particular portions of the spectrum. Special notice should be given to those areas where analog-to-digital signal level offsets occur. A “full spectrum” plot, for both forward and reverse signal paths, should be recorded.

### 5.3.3 Connectors, Cables and Adapters

Some components may degrade or “wear-out” over time — connectors, adapters and cables are especially prone to this problem. Extreme care must be exercised with these devices as such damage is not commonly apparent except under the closest of visual inspection. Also, degradation with these components tends to be catastrophic in nature, often with them failing at the most inopportune time. Such problems will not normally be discoverable through testing so it is advisable to follow manufacturers’ recommendations regarding handling and use of these components, especially if a typical (or maximum) number of applications is specified.

One sign that a problem may be developing with these components is the sudden appearance of signal interference (ingress) within the test spectrum. Shielding effectiveness testing, perhaps using a near-field probe, may also be used to guard against signal ingress and/or egress which should be evident as the various interfaces involved with these devices begin to degrade.

### 5.3.4 Filters and In-Line Fixed Attenuators

Filters come in two broad categories; bandpass or bandstop. Bandpass filters may cover a specific span of frequencies or be categorized as highpass or lowpass filters. Bandstop filters may also be called traps. All of these filters should be qualified for both loss and actual spectrum coverage. An additional parameter to be concerned about with filters is group delay. Generally, both loss and group delay should be measured versus frequency. Special attention should be given to these parameters at a filter’s band edge(s). Tunable bandpass filters have the added concern that they will have different losses and passband characteristics dependant upon the frequency to which they are tuned. As these are frequency selective devices, of one sort or another, the out-of-band return loss will be quite low, perhaps even 0 dB in some instances.

The use of in-line fixed attenuators is recommended for all frequency selective devices in order to ensure adequate impedance matching characteristics, especially for those portions of the spectrum which fall outside of the tuning range of the filters (or tuners, as on the spectrum analyzer). In-line fixed attenuators are also recommended to ensure good match at the input and output DC's.

### 5.3.5 Switchable Attenuators

These devices are quite useful for setting levels and in helping to perform relative signal level measurements. There are two areas of concern with these devices, however, where problems can arise in product/system evaluations — attenuation setting accuracy and long-term wear and tear, which will degrade performance over time. As with other devices within the test system these items should be checked with a network analyzer to ascertain and document the precision of the various switch positions. A record of the initial settings and performance will also provide a point to which current operation can be compared, as time passes.

### 5.3.6 RF Post Amplifiers

As with the previous components mentioned, a record of the swept frequency response which documents gain, operational bandwidth and frequency response, will provide a good measure of protection against misuse of RF post amplifiers. The noise performance and distortion characteristics of RF post amplifiers also require special attention.

The fundamental measure of an amplifier's noise performance is “noise figure.” Unfortunately, this metric is usually not uniform across an amplifier’s operational bandwidth, especially when multi-octave, broadband devices are under consideration. It is essential to have an accurate reading of the noise figure of all amplifiers, which will be used within the test system.

Another area of concern with an amplifier’s performance is its distortion characteristics. This relates to the amount of composite triple beat (CTB), composite second order (CSO) or cross modulation (XMOD) the amplifier generates. The generation of these distortions is dependent upon both the spectrum loading and the output signal level (and tilt) at which the amplifier is required to operate — the higher the output signal level the greater the distortion. It is generally assumed that active devices will have a “linear” range in which the change in distortion products will track with a change in output level. Unfortunately, this assumption of linearity may or may not be correct. The only way to know what the range of linearity is (and its limits) is to measure the distortion performance of the amplifier over the anticipated operational range.

If the amplifier is carrying digital signals, carrier-to-intermodulation noise distortion (CIN) will also be generated. Since this distortion is noise-like, it has to be characterized by measuring the carrier-to-composite noise (CCN) with digital signals present as well as carrier-to-thermal noise (CTN) without digital signals, and subtracting on a 10 log basis to obtain CIN. For additional information, refer to ANSI/SCTE 17 2007.

In the test system shown in Figure 7 the spectrum loading will be limited by the bandpass filter, which precedes the amplifier. The bandpass filter will also

limit, to a large degree, the tilt, which the amplifier sees. The switchable attenuator on the input of the amplifier should be used to limit the signal level range to a reasonable spread but it should be noted that one must be careful not to degrade the system's noise performance by putting in too much attenuation.

It is recommended that the distortion and noise performance measurements be performed upon the entire cascade of devices (in Figure 7) between the two 6 dB fixed attenuators. Testing should be conducted at several specific frequencies as well as at a number of different output signal levels (and at different temperatures, if applicable) so that meaningful comparisons can be made with the measurements made upon the DUT.

### 5.3.7 General Concerns

**Return Loss** — Every component within the test system should be qualified for input and output return loss (IRL/ORL, also called input and output “match”). Better match (reflected in higher RL numbers) will decrease the potential for degenerative interaction between devices, which are coupled together. This concern is the major reason for using fixed attenuators at most (or all) filter interfaces.

**Bandwidth and Frequency Response** — It is necessary to ensure that all components within the test system are capable of handling the full range of frequencies to be encountered within the total scope of the testing. The best way to do this is through sweep testing of each part, using a network analyzer or similar device. The swept response will also provide detailed information as to changes in response relative to frequency. Special care should be taken in documenting areas of transition, where changes are likely to occur at a more pronounced rate.

**Group Delay** — This parameter is growing in importance, especially relative to the carriage of broadband, digitally modulated signals. Each component should be checked across the entire operational band(s) and special attention should be given to band edges as the frequency sensitive components within the test system will exhibit greater amounts of delay versus frequency within the transition areas of the spectrums involved.

**Temperature** — Most (if not all) of the parameters under consideration within this document are sensitive to changes in temperature. If the product/system evaluation is designed to cover a specific temperature range (i.e., -40° to + 60° C) then those components, which are subject to this environmental specification, must be characterized across the entire range. This will probably entail measurements at the desired ambient temperature as well as the limits of the specified temperature range. It may also be helpful to have characterization data for a limited set of intermediate temperatures.

### 5.3.8 Records

1. Swept frequency response plots are particularly useful for documentation of signal transfer characteristics and provide a permanent record of the following:
  - a. Loss vs. frequency for all passive components, including filters, cables/connector assemblies, splitters and directional couplers
  - b. Gain vs. frequency for active devices
  - c. Input and output return loss (IRL, ORL) of each component within the test system
2. Tables containing the results from noise figure and distortion measurements, at the specific frequencies to be tested, are required. Plots taken directly off of the spectrum analyzer's screen (or other measurement device) are also helpful in recording this performance.
3. A set of nomographs or tables detailing the DUT's contribution to the total end-of-line noise may prove useful. These should be based upon the addition of two powers (or voltages), just like the noise-near-noise table found in Section 8.2. Since the noise of the test system is not coherent with the noise of the DUT, the (C/N) of the test system can be subtracted from combined performance of the test system and the DUT to reveal the actual performance of the DUT itself. This is not possible with the distortion performance since there is no way of knowing the relative coherence or phase of the distortion products.
4. Current calibration records for all applicable devices (meters, spectrum analyzers, etc.) are essential to demonstrating continued diligence to performing and maintaining accurate measurements.

## **6.0 MEASUREMENT ACCURACY**

The goal of these procedures is to ensure that the measurements taken are accurate and representative of "real life." Every test has the potential for errors and the source of these errors must be known and minimized, if possible. Measurement errors may be sorted into two separate classes, repeatable and random.

### **6.1 Repeatable Errors**

Repeatable errors include errors in the test equipment being used and related to the specifications of the signal sources, signal conditioning and signal measurement devices, as well as errors in the signal conditioning components of the test system. For a typical system, which includes a spectrum analyzer, amplifier, attenuator and/or filter, these sources of error include:

1. Amplitude measurement accuracy of the spectrum analyzer, including absolute and relative accuracy, and resolution

2. Frequency response (spectrum analyzer, amplifier, filter)
3. Attenuator accuracy (spectrum analyzer, fixed and switchable attenuators)
4. Log amplifier accuracy (spectrum analyzer)
5. Noise and distortion contribution (signal generator, spectrum analyzer, amplifier)
6. Signal level inaccuracies which contribute to noise and/or distortion components
7. Bandwidth limits
8. Impedance matching deficiencies

Many of the repeatable errors related to the test equipment can be minimized and/or eliminated by “normalization” and “characterization”. In addition, the consistent use of certain test techniques can further decrease the impact of these factors.

## 6.2 Random Errors

Random errors include the following:

1. The accuracy with which the operating levels of the DUT are set.
2. Errors in reading the signal level meter (or spectrum analyzer) when setting the carrier levels. Many errors of this type are greatly minimized by digital readouts on the SLM or the spectrum analyzer’s marker.
3. Errors which occur in reading both the carrier level and the composite distortion level due to the visual resolution of the spectrum analyzer display. Again, many errors of this type are greatly minimized by digital readouts on the SLM or the spectrum analyzer’s marker.
4. Errors produced by variations in the alignment of the non-coherent carrier frequencies.
5. Signal source level drift between the time levels are adjusted and measurements are completed.

## 6.3 Measurement Accuracy Analysis

A mathematical analysis of the combined potential errors, representing a calculation for the measurement of CTB, using a signal level meter (SLM) to set operational levels, is detailed below. Note that it is assumed for these calculations that the CTB level is not greater than 70 dBc below the carrier level, and that the CTB level is at least 10 dB above the spectrum analyzer’s observed noise level.

1. The root-sum-of-the-squares (RSS) of errors (in a linear format) is the square root of the sum of the squares of the individual random errors (also in a linear format). The  $\pm$ dB error is converted to linear format by the following formula:

$$\text{linear error (\%)} = \left( 10^{\frac{\text{error}[dB]}{10}} - 1 \right) * 100\% \quad (13)$$

For example,  $\pm 0.4$  dB is equivalent to either:

$$\left(10^{\frac{+0.4}{10}} - 1\right) * 100\% = (1.0965 - 1) = +9.65\% \quad (14)$$

or

$$\left(10^{\frac{-0.4}{10}} - 1\right) * 100\% = (+0.912 - 1) = -8.80\% \quad (15)$$

**NOTE:** The dB format always refers to a ratio of two numbers and the linear error quantities (+9.65% and -8.80%) are “geometrically” centered but not “arithmetically” centered.

2. For  $n=1 \dots N$  linear error terms the following formula applies:

For Positive Errors:-

$$\text{RSS linear sum (\%)} = \sqrt{\sum_{n=1}^N (\text{linear error})^2} * 100\% \quad (16)$$

For Negative Errors:

$$\text{RSS linear sum (\%)} = -\sqrt{\sum_{n=1}^N (\text{linear error})^2} * 100\% \quad (17)$$

**Example:** For a CTB measurement with discrete potential errors of  $\pm 2.0$  dB,  $\pm 1.0$  dB,  $\pm 0.4$  dB and three (3) potential errors of  $\pm 0.2$  dB each, the worst-case analysis reveals the following linear quantities:

Accuracy FACTOR	ACCURACY SPECIFICATION	LINEAR ERROR ( $\pm\%$ )
Operating level setting ( $\pm 1$ dB) affect on CTB	$\pm 2.0$ dB	+58.49/-36.90
Analyzer Incremental Log Accuracy (0 to -70 dB from reference level)	$\pm 1.0$ dB	+25.89/-20.57
Effect of random carrier frequency distribution	$\pm 0.4$ dB	+9.65/-8.80
Effect of 16 dB worst case output match	$\pm 0.2$ dB	+4.71/-4.50
Display Visual Resolution, reading carrier level	$\pm 0.2$ dB	+4.71/-4.50

Display Visual Resolution, reading distortion product level	±0.2 dB	+4.71/-4.50
<b>Total linear RSS (%) error</b>		+65.20 % & -44.08 %
<b>Total RSS (±dB) error</b>		+2.18 dB & -2.51 dB

**Table 2 – CTB Measurement Accuracy Analysis**

3. The total linear RSS error (%) is determined by the following formula:

For Positive Errors:

$$\begin{aligned} & \sqrt{(0.5849)^2 + (0.2589)^2 + (0.0965)^2 + (0.0471)^2 + (0.0471)^2 + (0.0471)^2} * 100\% \\ & = 0.6520 * 100\% = 65.20\% \end{aligned} \quad (18)$$

For Negative Errors:

$$\begin{aligned} & -\sqrt{(0.3690)^2 + (0.2057)^2 + (0.0880)^2 + (0.0450)^2 + (0.0450)^2 + (0.0450)^2} * 100\% \\ & = -0.4385 * 100\% = -43.85\% \end{aligned} \quad (19)$$

4. Use the following formula to convert this linear term back into a logarithmic format (dB):

$$\text{RSS (dB)} = 10 \log \left( 1 + \frac{\text{RSS linear sum (\%)}}{100\%} \right) \quad (20)$$

5. From the above example this formula provides a total RSS (dB) of

$$\begin{aligned} \text{RSS (dB)} &= 10 \log \left( 1 + \frac{\text{RSS linear sum (\%)}}{100\%} \right) = 10 \log \left( 1 + \frac{65.20\%}{100\%} \right) \\ &= +2.18 \text{ dB} \end{aligned} \quad (21)$$

$$\begin{aligned} \text{RSS (dB)} &= 10 \log \left( 1 + \frac{\text{RSS linear sum (\%)}}{100\%} \right) = 10 \log \left( 1 + \frac{-43.85\%}{100\%} \right) \\ &= -2.51 \text{ dB} \end{aligned} \quad (22)$$

6. All of the above is incorporated into the following formula:

For Positive Errors:

$$\text{RSS accuracy (dB)} = 10 \log \left( 1 + \sqrt{\sum_{n=1}^N \left( 10^{\frac{\text{accuracy}_n(\text{dB})}{10}} - 1 \right)^2} \right) \dots \quad (23)$$

For Negative Errors:

$$\text{RSS accuracy (dB)} = 10 \log \left( 1 - \sqrt{\sum_{n=1}^N \left( 10^{\frac{\text{accuracy}_n \text{ (dB)}}{10}} - 1 \right)^2} \right) \dots \quad (24)$$

where  $n = 1 \dots N$  individual terms and final probable accuracy is expressed in dB

These maximum errors assume that the equipment has been thoroughly warmed up and stabilized at the measurement temperature and that all recommended calibration procedures have been followed.

If the spectrum analyzer input attenuator must be changed, then the accuracy shown above could be further degraded by the incremental accuracy of the attenuator. If the final composite distortion levels are close to the visible noise floor, then the accuracy will be further degraded by the difficulty in discerning a distortion level near the noise floor of the instrument.

If a spectrum analyzer is used to set the carrier levels the accuracy could be limited by its absolute amplitude accuracy. Whatever unit is used to set the device's/system's levels, whether a spectrum analyzer or a signal level meter, it should have been carefully correlated with a power meter so that this error can be minimized.

**NOTE:** Each discrete set of equipment used in these measurements must be analyzed for potential errors. Information regarding each instrument's guaranteed accuracy, and the conditions for that guarantee, should be available from the test equipment manufacturers.

## 7.0 MEASUREMENT RECORDS

### 7.1 Test System Documentation Requirements

The following documentation should be developed and/or collected prior to the commencement of actual testing.

1. Complete test system diagram which details all components including gain and loss, for both forward and return paths.
2. Functional block diagram for the device or system to be tested.
3. Detailed test procedures, test equipment lists and setup diagrams for each unique test system configuration (cascade, bench, etc.). The lists and diagrams should include the following basic information:
  - a. Test equipment make, model and calibration history
  - b. Test equipment and system setup, including settings for each test performed
  - c. Operational levels and tilts

- d. Graphical and table characterization data for all devices used in the test system, including amplifiers, filters, splitters, directional couplers, switches, attenuators, etc. This data should be over the frequency range and temperature range to be used.
4. Other documentation which may be needed including independent lab verification of compliance with various governmental requirements such as FCC Part 15, UL, etc. It is often desirable to have the manufacturer state such qualitative standards as MTBF or MTTF.

## 7.2 Measurement Records

Measurement records typically include the actual levels measured and all information pertinent to the conditions under which the data was collected.

1. Date, time and temperature
2. Unit tested; manufacturer, model number, serial number.
3. Signal loading; bandwidth, operating levels, tilt, etc., for input and output.
4. Actual channels or frequencies where the parameter was measured
5. Test equipment identification (make, model number, serial number), calibration date, etc.
6. Measured performance, including:
  - a. Per channel under test, with actual carrier level and actual, relative distortion/noise level
  - b. Correction factors for analyzer settings, beat near noise/noise near noise correction, etc.
  - c. Corrected distortion/noise level
7. Accuracy analysis to demonstrate potential margin for measurement error
8. Test personnel data

## 8.0 REFERENCE INFORMATION

### 8.1 Power and Voltage Addition

#### 8.1.1 Power Addition

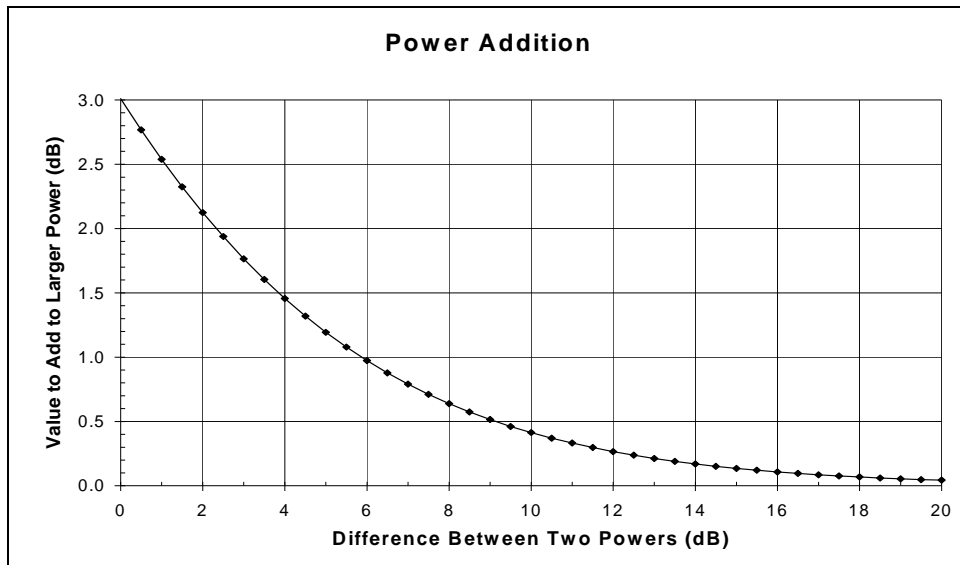
Addition or subtraction of two powers that are represented as decibels requires an equation or look up table. The equation for adding two signals on a power basis is:

$$\text{Sum (dB)} = 10 \log \left[ 10^{\left(\frac{P_1}{10}\right)} + 10^{\left(\frac{P_2}{10}\right)} \right] \quad (25)$$

To add two power levels, the following table can be used. Find the box corresponding to the difference between the two signal levels and add the amount in the table to the larger of the two individual levels.

$\Delta$	0.0	0.1	0.2	0.3	0.4	0.5	0.6	0.7	0.8	0.9
0.0	3.01	2.96	2.91	2.86	2.81	2.77	2.72	2.67	2.63	2.58
1.0	2.54	2.50	2.45	2.41	2.37	2.32	2.28	2.24	2.20	2.16
2.0	2.12	2.09	2.05	2.01	1.97	1.94	1.90	1.87	1.83	1.80
3.0	1.76	1.73	1.70	1.67	1.63	1.60	1.57	1.54	1.51	1.48
4.0	1.46	1.43	1.40	1.37	1.35	1.32	1.29	1.27	1.24	1.22
5.0	1.19	1.17	1.15	1.12	1.10	1.08	1.06	1.04	1.01	0.99
6.0	0.97	0.95	0.93	0.91	0.90	0.88	0.86	0.84	0.82	0.81
7.0	0.79	0.77	0.76	0.74	0.73	0.71	0.70	0.68	0.67	0.65
8.0	0.64	0.63	0.61	0.60	0.59	0.57	0.56	0.55	0.54	0.53
9.0	0.51	0.50	0.49	0.48	0.47	0.46	0.45	0.44	0.43	0.42
10.0	0.41	0.40	0.40	0.39	0.38	0.37	0.36	0.35	0.35	0.34
11.0	0.33	0.32	0.32	0.31	0.30	0.30	0.29	0.28	0.28	0.27
12.0	0.27	0.26	0.25	0.25	0.24	0.24	0.23	0.23	0.22	0.22
13.0	0.21	0.21	0.20	0.20	0.19	0.19	0.19	0.18	0.18	0.17
14.0	0.17	0.17	0.16	0.16	0.15	0.15	0.15	0.14	0.14	0.14
15.0	0.14	0.13	0.13	0.13	0.12	0.12	0.12	0.12	0.11	0.11
16.0	0.11	0.11	0.10	0.10	0.10	0.10	0.09	0.09	0.09	0.09
17.0	0.09	0.08	0.08	0.08	0.08	0.08	0.07	0.07	0.07	0.07
18.0	0.07	0.07	0.07	0.06	0.06	0.06	0.06	0.06	0.06	0.06
19.0	0.05	0.05	0.05	0.05	0.05	0.05	0.05	0.05	0.05	0.04
20.0	0.04	0.04	0.04	0.04	0.04	0.04	0.04	0.04	0.04	0.04

**Table 3 - Power (10\*Log) Addition**



**Figure 8 - Power (10\*Log) Addition**

8.1.2 Voltage Addition

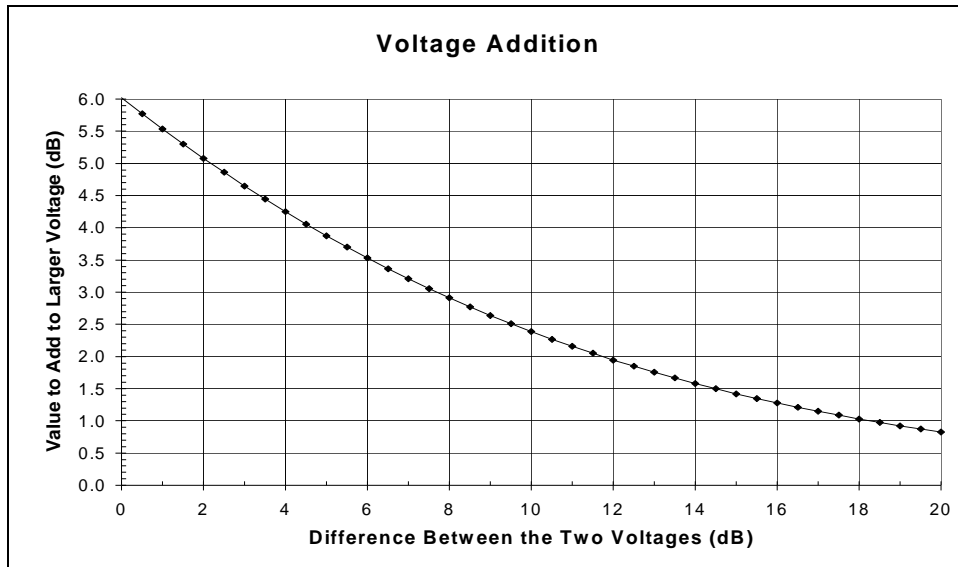
Addition or subtraction of two voltages that are represented as decibels requires an equation or look up table. The equation for adding two signals on a voltage basis is:

$$\text{Sum (dB)} = 20\log \left[ 10^{\left(\frac{V_1}{20}\right)} + 10^{\left(\frac{V_2}{20}\right)} \right] \quad (26)$$

To add two voltage levels, the following table can be used. Find the box corresponding to the difference between the two signal levels and add the amount in the table to the larger of the two individual levels.

$\Delta$	0.0	0.1	0.2	0.3	0.4	0.5	0.6	0.7	0.8	0.9
0.0	6.02	5.97	5.92	5.87	5.82	5.77	5.73	5.68	5.63	5.58
1.0	5.53	5.49	5.44	5.39	5.35	5.30	5.26	5.21	5.17	5.12
2.0	5.08	5.03	4.99	4.95	4.90	4.86	4.82	4.78	4.73	4.69
3.0	4.65	4.61	4.57	4.53	4.49	4.45	4.41	4.37	4.33	4.29
4.0	4.25	4.21	4.17	4.13	4.10	4.06	4.02	3.98	3.95	3.91
5.0	3.88	3.84	3.80	3.77	3.73	3.70	3.66	3.63	3.60	3.56
6.0	3.53	3.50	3.46	3.43	3.40	3.36	3.33	3.30	3.27	3.24
7.0	3.21	3.18	3.15	3.12	3.09	3.06	3.03	3.00	2.97	2.94
8.0	2.91	2.88	2.85	2.83	2.80	2.77	2.74	2.72	2.69	2.66
9.0	2.64	2.61	2.59	2.56	2.53	2.51	2.48	2.46	2.44	2.41
10.0	2.39	2.36	2.34	2.32	2.29	2.27	2.25	2.22	2.20	2.18
11.0	2.16	2.13	2.11	2.09	2.07	2.05	2.03	2.01	1.99	1.97
12.0	1.95	1.93	1.91	1.89	1.87	1.85	1.83	1.81	1.79	1.77
13.0	1.75	1.74	1.72	1.70	1.68	1.67	1.65	1.63	1.61	1.60
14.0	1.58	1.56	1.55	1.53	1.51	1.50	1.48	1.47	1.45	1.44
15.0	1.42	1.41	1.39	1.38	1.36	1.35	1.33	1.32	1.31	1.29
16.0	1.28	1.26	1.25	1.24	1.22	1.21	1.20	1.19	1.17	1.16
17.0	1.15	1.14	1.12	1.11	1.10	1.09	1.08	1.06	1.05	1.04
18.0	1.03	1.02	1.01	1.00	0.99	0.98	0.96	0.95	0.94	0.93
19.0	0.92	0.91	0.90	0.89	0.88	0.87	0.86	0.86	0.85	0.84
20.0	0.83	0.82	0.81	0.80	0.79	0.78	0.77	0.77	0.76	0.75

**Table 4 - Voltage or Current (20\*log) Addition**



**Figure 9 - Voltage or Current (20\*Log) Addition**

8.2 Noise-near-Noise/Beat-near-Noise (NnN/BnN) Correction

8.2.1 Calculating NnN/BnN Correction

$$\text{Correction (dB)} = \left| 10 \log \left( 1 - 10^{\frac{-\text{Noise Drop}}{10}} \right) \right| \text{dB} \quad (27)$$

8.2.2 Table of NnN/BnN Correction

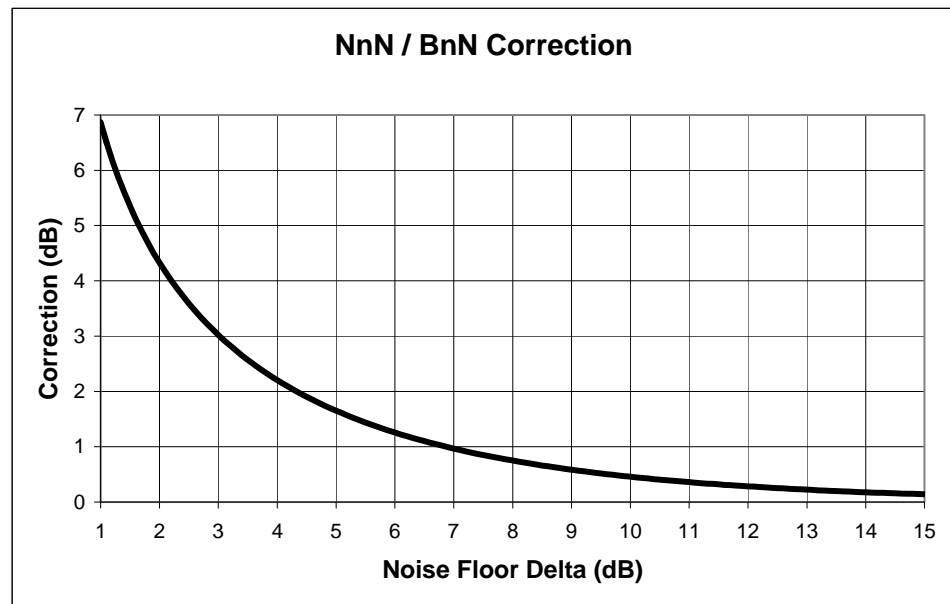
Noise Floor Delta	Correction
0.5	9.6
1	6.9
1.5	5.3
2.0	4.3
2.5	3.6
3.0	3.0
3.5	2.6
4.0	2.2
4.5	1.9
5.0	1.7

8.5	0.7
9.0	0.6
9.5	0.5
10.0	0.5

Noise Floor Delta	Correction
5.5	1.4
6.0	1.3
6.5	1.1
7.0	1.0
7.5	0.9
8.0	0.7

Noise Floor Delta	Correction
10.5	0.4
11.0	0.4
11.5	0.3
12.0	0.3
12.5	0.3
13.0	0.2
13.5	0.2
14.0	0.2
14.5	0.2
15.0	0.1

**Note:** Correction factors are shown down to a Noise Floor Delta of 0.5 dB and, of course, the formula will allow one to derive a value for any observed Drop, however any Drop < 2.0 dB is subject to significant potential errors. Therefore, it is recommended that, for Noise Floor Deltas < 2.0 dB, 4.3 dB be added to the observed distortion/noise level and the total value be expressed as “greater than” (>)  $x$  dB.



**Figure 10 - Graph of NnN/BnN Correction**

### 8.3 Informative References

The following documents may provide valuable information to the reader but are not required when complying with this standard.

1. ANSI/SCTE 17 2007: Test Procedure for Carrier to Noise (C/N, CCN, CIN, CTN).
2. Cable Television Proof of Performance; Jeffrey L. Thomas; Hewlett-Packard Professional Books; 1995.
3. Cable Television System Measurements Handbook; Hewlett-Packard Company; July, 1993.
4. Digital Communications, J.G. Proakis, McGraw-Hill, latest edition.
5. EIA/CEA 542 (April 2002): Cable Television Channel Identification Plan
6. Fundamentals of RF and Microwave Power Measurements, Application Note 64-1B, Agilent Technologies, April 2000

7. Introduction to Communications Engineering; Gagliardi, Robert M., John Wiley and Sons; Chapter 4.
8. MIL-PRF-28800 (6/24/1996): Test Equipment for Use with Electrical and Electronic Equipment
9. NCTA Recommended Practices For Measurement On Cable Television Systems, Third Edition; National Cable Television Association.
10. Phase Noise Theory and Measurements: A Short Review; Goldberg, Bar-Giora; Microwave Journal, Vol. 43, No. 1, January, 2000.
11. ANSI/SCTE 121 2006: Test Procedure for Forward Path (Downstream) Bit Error Rate
12. Spectrum Analysis . . . . Spectrum Analyzer Basics, Application Note 150, Agilent Technologies, November, 1989. (1<sup>st</sup> in a very useful series of application notes on spectrum analysis.)
13. Spectrum Analyzer Fundamentals, Tektronix App. Note 26W-7037-1, 1989.
14. Spectrum Analyzer Measurements and Noise, Application Note 1303, Agilent Technologies, April 1998.