Handbook of Frequency Stability Analysis

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Dedication

This handbook is dedicated to the memory of Dr. James A. Barnes (1933-2002), a pioneer in the statistics of frequency standards.

James A. Barnes was born in 1933 in Denver, Colorado. He received a Bachelors degree in engineering physics from the University of Colorado, a Masters degree from Stanford University, and in 1966 a Ph.D. in physics from the University of Colorado. Jim also received an MBA from the University of Denver.

After graduating from Stanford, Jim joined the National Bureau of Standards, now the National Institute of Standards and Technology (NIST). Jim was the first Chief of the Time and Frequency Division when it was created in 1967 and set the direction for this division in his 15 years of leadership. During his tenure at NIST Jim made many significant contributions to the development of statistical tools for clocks and frequency standards. Also, three primary frequency standards (NBS 4, 5 and 6) were developed under his leadership. While division chief, closed-captioning was developed (which received an Emmy award) and the speed of light was measured. Jim received the NBS Silver Medal in 1965 and the Gold Medal in 1975. In 1992, Jim received the I.I. Rabi Award from the IEEE Frequency Control Symposium “for contributions and leadership in the development of the statistical theory, simulation and practical understanding of clock noise, and the application of this understanding to the characterization of precision oscillators and atomic clocks”. In 1995, he received the Distinguished PTTI Service Award. Jim was a Fellow of the IEEE. After retiring from NIST in 1982, Jim worked for Austron.

Jim Barnes died Sunday, January 13, 2002 in Boulder, Colorado after a long struggle with Parkinson’s disease. He was survived by a brother, three children, and two grandchildren.

Note: This biography is taken from his memoriam on the UFFC web site at: http://www.ieee-uffc.org/fcmain.asp?page=barnes.
Acknowledgments

The author acknowledges the contributions of many colleagues in the Time and Frequency community who have contributed the analytical tools that are so vital to this field. In particular, he wishes to recognize the seminal work of J.A. Barnes and D.W. Allan in establishing the fundamentals at NBS, and D.A. Howe in carrying on that tradition today at NIST. Together with such people as M.A. Weiss and C.A. Greenhall, the techniques of frequency stability analysis have advanced greatly during the last 45 years, supporting the orders-of-magnitude progress made on frequency standards and time dissemination.

I especially thank David Howe and the other members of the NIST Time & Frequency Division for their support, encouragement, and review of this Handbook.
Preface

The author has had the great privilege of working in the time and frequency field over the span of his career. I have seen atomic frequency standards shrink from racks of equipment to chip scale, and be manufactured by the tens of thousands, while primary standards and the time dissemination networks that support them have improved by several orders of magnitude. During the same period, significant advances have been made in our ability to measure and analyze the performance of those devices. This Handbook summarizes the techniques of frequency stability analysis, bringing together material that I hope will be useful to the scientists and engineers working in this field.
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1 Introduction

This handbook describes practical techniques for frequency stability analysis. It covers the definitions of frequency stability, measuring systems and data formats, preprocessing steps, analysis tools and methods, post processing steps, and reporting suggestions. Examples are included for many of these techniques. Some of the examples use the Stable32 program [1], which is a commercially available tool for studying and performing frequency stability analyses. Two general references [2], [3] for this subject are given below.

This handbook can be used both as a tutorial and as a reference. If this is your first exposure to this field, you may find it helpful to scan the sections to gain some perspective regarding frequency stability analysis. I strongly recommend consulting the references as part of your study of this subject matter. The emphasis is on time domain stability analysis, where specialized statistical variances have been developed to characterize clock noise as a function of averaging time. Methods are presented to perform those calculations, identify noise types and determine confidence limits. It is often important to separate deterministic factors such as aging and environmental sensitivity from the stochastic noise processes. One must always be aware of the possibility of outliers and other measurement problems that can contaminate the data.

Suggested analysis procedures are recommended to gather data, preprocess it, analyze stability and report results. Throughout these analyses, it is worthwhile to remember R.W. Hamming’s axiom that “the purpose of computing is insight, not numbers”. The analyst should feel free to use his intuition and experiment with different methods that can provide a deeper understanding.

References for Introduction

2 Frequency Stability Analysis

The time domain stability analysis of a frequency source is concerned with characterizing the variables x(t) and y(t), the phase (expressed in units of time error) and the fractional frequency, respectively. It is accomplished with an array of phase and frequency data arrays, x_i and y_i, respectively, where the index i refers to data points equally spaced in time. The x_i values have units of time in seconds, and the y_i values are (dimensionless) fractional frequency, Δf/f. The x(t) time fluctuations are related to the phase fluctuations by φ(t) = x(t)·2πν_0, where ν_0 is the nominal carrier frequency in Hz. Both are commonly called "phase" to distinguish them from the independent time variable, t. The data sampling or measurement interval, τ_0, has units of seconds. The analysis interval or period, loosely called “averaging time”, τ, may be a multiple of τ_0 (τ = mτ_0, where m is the averaging factor).

The goal of a time domain stability analysis is a concise, yet complete, quantitative and standardized description of the phase and frequency of the source, including their nominal values, the fluctuations of those values, and their dependence on time and environmental conditions.

A frequency stability analysis is normally performed on a single device, not a population of such devices. The output of the device is generally assumed to exist indefinitely before and after the particular data set was measured, which is the (finite) population under analysis. A stability analysis may be concerned with both the stochastic (noise) and deterministic (systematic) properties of the device under test. It is also generally assumed that the stochastic characteristics of the device are constant (both stationary over time and ergodic over their population). The analysis may show that this is not true, in which case the data record may have to be partitioned to obtain meaningful results. It is often best to characterize and remove deterministic factors (e.g., frequency drift and temperature sensitivity) before analyzing the noise. Environmental effects are often best handled by eliminating them from the test conditions. It is also assumed that the frequency reference instability and instrumental effects are either negligible or removed from the data. A common problem for time domain frequency stability analysis is to produce results at the longest possible analysis interval in order to minimize test time and cost. Computation time is generally not as much of a factor.

2.1. Background

The field of modern frequency stability analysis began in the mid 1960’s with the emergence of improved analytical and measurement techniques. In particular, new statistics became available that were better suited for common clock noises than the classic N-sample variance, and better methods were developed for high resolution measurements (e.g., heterodyne period measurements with electronic counters, and low noise phase noise measurements with double-balanced diode mixers). A seminal conference on short-term stability in 1964 [1], and the introduction of the 2-sample (Allan) variance in 1966 [2] marked the beginning of this new era, which was summarized in a special issue of the Proceedings of the IEEE in 1966 [3]. This period also marked the introduction of commercial atomic frequency standards, increased emphasis on low phase noise, and the use of the LORAN radio navigation system for global precise time and frequency transfer. The subsequent advances in the performance of frequency sources depended largely on the improved ability to measure and analyze their...
stability. These advances also mean that the field of frequency stability analysis has become more complex. It is the goal of this handbook to help the analyst deal with this complexity.

An example of the progress that has been made in frequency stability analysis from the original Allan variance in 1966 through Théo1 in 2003 is shown in the plots below. The error bars show the improvement in statistical confidence for the same data set, while the extension to longer averaging time provides better long-term clock characterization without the time and expense of a longer data record.
This handbook includes detailed information about these (and other) stability measures.

**References for Frequency Stability Analysis**

3 Definitions and Terminology

The field of frequency stability analysis, like most others, has its own specialized definitions and terminology. The basis of a time domain stability analysis is an array of equally spaced phase (really time error) or fractional frequency deviation data arrays, $x_i$ and $y_i$, respectively, where the index $i$ refers to data points in time. These data are equivalent, and conversions between them are possible. The $x$ values have units of time in seconds, and the $y$ values are (dimensionless) fractional frequency, $\Delta f/f$. The $x(t)$ time fluctuations are related to the phase fluctuations by $\phi(t) = x(t) \cdot 2\pi f_0$, where $f_0$ is the carrier frequency in hertz. Both are commonly called "phase" to distinguish them from the independent time variable, $t$. The data sampling or measurement interval, $\tau_0$, has units of seconds. The analysis or averaging time, $\tau$, may be a multiple of $\tau_0 \quad (\tau = m \tau_0 \quad \text{where} \quad m \quad \text{is} \quad \text{the} \quad \text{averaging factor})$. Phase noise is fundamental to a frequency stability analysis, and the type and magnitude of the noise, along with other factors such as aging and environmental sensitivity, determine the stability of the frequency source.

3.1 Noise Model

A frequency source has a sine wave output signal given by

$$V(t) = [V_0 + \varepsilon(t)]\sin[2\pi f_0 t + \phi(t)],$$

where $V_0 =$ nominal peak output voltage
$\varepsilon(t) =$ amplitude deviation
$f_0 =$ nominal frequency
$\phi(t) =$ phase deviation.

For the analysis of frequency stability, we are concerned primarily with the $\phi(t)$ term. The instantaneous frequency is the derivative of the total phase:

$$\nu(t) = f_0 + \frac{1}{2\pi} \frac{d\phi}{dt}.$$

For precision oscillators, we define the fractional frequency as

$$y(t) = \frac{\Delta f}{f} = \frac{\nu(t) - f_0}{f_0} = \frac{1}{2\pi f_0} \frac{d\phi}{dt} = \frac{dx}{dt},$$

where

$$x(t) = \phi(t) / 2\pi f_0.$$
3.2. Power Law Noise

It has been found that the instability of most frequency sources can be modeled by a combination of power-law noises having a spectral density of their fractional frequency fluctuations of the form $S_y(f) \propto f^\alpha$, where $f$ is the Fourier or sideband frequency in hertz, and $\alpha$ is the power law exponent.

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>$\alpha$</th>
</tr>
</thead>
<tbody>
<tr>
<td>White PM</td>
<td>2</td>
</tr>
<tr>
<td>Flicker PM</td>
<td>1</td>
</tr>
<tr>
<td>White FM</td>
<td>0</td>
</tr>
<tr>
<td>Flicker FM</td>
<td>-1</td>
</tr>
<tr>
<td>Random Walk FM</td>
<td>-2</td>
</tr>
<tr>
<td>Flicker Walk FM</td>
<td>-3</td>
</tr>
<tr>
<td>Random Run FM</td>
<td>-4</td>
</tr>
</tbody>
</table>

Examples of the four most common of these noises are shown in the table below:

3.3. Stability Measures

The standard measures for frequency stability in the time and frequency domains are the overlapped Allan deviation, $\sigma_y(\tau)$, and the SSB phase noise, $\xi(f)$, as described in more detail later in this handbook.

3.4. Differenced and Integrated Noise

Taking the differences between adjacent data points plays an important role in frequency stability analysis for performing phase to frequency data conversion, calculating Allan (and related) variances, and doing noise identification using the lag 1 autocorrelation method [2]. Phase data $x(t)$ may be converted to fractional frequency data $y(t)$ by taking the first
differences $x_{i+1} - x_i$ of the phase data and dividing by the sampling interval $\tau$. The Allan variance is based on the first differences $y_{i+1} - y_i$ of the fractional frequency data or, equivalently, the second differences $y_{i+2} - 2y_{i+1} + y_i$ of the phase data. Similarly, the Hadamard variance is based on third differences $x_{i+3} - 3x_{i+2} + 3x_{i+1} - x_i$ of the phase data.

Taking the first differences of a data set has the effect of making it less divergent. In terms of its spectral density, the $\alpha$ value is increased by 2. For example, flicker FM data ($\alpha = -1$) is changed into flicker PM data ($\alpha = +1$). That is the reason that the Hadamard variance is able to handle more divergent noise types ($\alpha \geq -4$) than the Allan variance ($\alpha \geq -2$) can. It is also the basis of the lag 1 autocorrelation noise identification method whereby first differences are taken until $\alpha$ becomes $\geq 0.5$. The plots below show random run noise differenced first to random walk noise and again to white noise.

![Random Run Noise](image1.png)

![Random Walk Noise](image2.png)

![White Noise](image3.png)

The more divergent noise types are sometimes referred to by their color. White noise has a flat spectral density (by analogy to white light). Flicker noise has an $f^{-1}$ spectral density, and is called pink or red (more energy toward lower frequencies). Continuing the analogy, $f^2$ (random walk) noise is called brown, and $f^3$ (flicker walk) noise is called black, although that terminology is seldom used in the field of frequency stability analysis.

Integration is the inverse operation of differencing. Numerically integrating frequency data converts it into phase data (with an arbitrary initial value). Such integration subtracts 2 from the original $\alpha$ value. For example, the random run data in the top left plot above was generated by simulating random walk FM data and converting it to phase data by numerical integration.
3.5. Glossary

See the Glossary chapter at the end of this handbook for brief definitions of many of the important terms used in the field of frequency stability analysis.

References for Definitions and Terminology


4 Standards

Standards have been adopted for the measurement and characterization of frequency stability, as shown in the references below [1]-[5]. These standards define terminology, measurement methods, means for characterization and specification, etc. In particular, IEEE-Std-1139 contains definitions, recommendations, and examples for the characterization of frequency stability.

References for Standards

2. MIL-PRF-55310, Oscillators, Crystal, General Specification For.
5 Time Domain Stability

The stability of a frequency source in the time domain is based on the statistics of its phase or frequency fluctuations as a function of time, a form of time series analysis [1]. This analysis generally uses some type of variance, a 2nd moment measure of the fluctuations. For many divergent noise types commonly associated with frequency sources, the standard variance, which is based on the variations around the average value, is not convergent, and other variances have been developed that provide a better characterization of such devices. A key aspect of such a characterization is the dependence of the variance on the averaging time used to make the measurement, which dependence shows the properties of the noise.

5.1 Sigma-Tau Plots

The most common way to express the time domain stability of a frequency source is by means of a sigma-tau plot that shows some measure of frequency stability versus the time over which the frequency is averaged. Log sigma versus log tau plots show the dependence of stability on averaging time, and show both the stability value and the type of noise. The power law noises have particular slopes, μ, as shown on the following log s vs. log τ plots, and α and μ are related as shown in the table below:

<table>
<thead>
<tr>
<th>Noise</th>
<th>α</th>
<th>µ</th>
</tr>
</thead>
<tbody>
<tr>
<td>W</td>
<td>2</td>
<td>-2</td>
</tr>
<tr>
<td>F</td>
<td>1</td>
<td>~ 2</td>
</tr>
<tr>
<td>W</td>
<td>0</td>
<td>-1</td>
</tr>
<tr>
<td>F</td>
<td>-1</td>
<td>0</td>
</tr>
<tr>
<td>RW</td>
<td>-2</td>
<td>1</td>
</tr>
</tbody>
</table>

The log σ versus log τ slopes are the same for the two PM noise types, but are different on a Mod sigma plot, which is often used to distinguish between them.
5.2. Variances

Variances are used to characterize the fluctuations of a frequency source [2, 3]. These are second-moment measures of scatter, much as the standard variance is used to quantify the variations in, say, the length of rods around a nominal value. The variations from the mean are squared, summed, and divided by one less than the number of measurements; this number is called the “degrees of freedom”.

Several statistical variances are available to the frequency stability analyst, and this section provides an overview of them, with more details to follow. The Allan variance is the most common time domain measure of frequency stability, and there are several versions of it that provide better statistical confidence, can distinguish between white and flicker phase noise,
and can describe time stability. The Hadamard variance can better handle frequency drift and more divergence noise types, and several versions of it are also available. The newer Total and Thêo1 variances can provide better confidence at longer averaging factors.

There are two categories of stability variances: unmodified variances, which use \( d \)th differences of phase samples, and modified variances, which use \( d \)th differences of averaged phase samples. The Allan variances correspond to \( d = 2 \), and the Hadamard variances to \( d = 3 \). The corresponding variances are defined as a scaling factor times the expected value of the differences squared. One obtains unbiased estimates of this variance from available phase data by computing time averages of the differences squared. The usual choices for the increment between estimates (the time step) are the sample period \( \tau_0 \) and the analysis period \( \tau \), a multiple of \( \tau_0 \). These give respectively the overlapped estimator and non-overlapped estimators of the stability.

<table>
<thead>
<tr>
<th>Variance Type</th>
<th>Characteristics</th>
</tr>
</thead>
<tbody>
<tr>
<td>Standard</td>
<td>Non-convergent for some clock noises – don’t use</td>
</tr>
<tr>
<td>Allan</td>
<td>Classic – use only if required – relatively poor confidence</td>
</tr>
<tr>
<td>Overlapping Allan</td>
<td>General Purpose - most widely used – first choice</td>
</tr>
<tr>
<td>Modified Allan</td>
<td>Used to distinguish W and F PM</td>
</tr>
<tr>
<td>Time</td>
<td>Based on modified Allan variance</td>
</tr>
<tr>
<td>Hadamard</td>
<td>Rejects frequency drift, and handles divergent noise</td>
</tr>
<tr>
<td>Overlapping Hadamard</td>
<td>Better confidence than normal Hadamard</td>
</tr>
<tr>
<td>Total</td>
<td>Better confidence at long averages for Allan</td>
</tr>
<tr>
<td>Modified Total</td>
<td>Better confidence at long averages for modified Allan</td>
</tr>
<tr>
<td>Time Total</td>
<td>Better confidence at long averages for time</td>
</tr>
<tr>
<td>Hadamard Total</td>
<td>Better confidence at long averages for Hadamard</td>
</tr>
<tr>
<td>Thêo1</td>
<td>Provides information over nearly full record length</td>
</tr>
<tr>
<td>ThêoH</td>
<td>Hybrid of Allan and Thêo1 variances</td>
</tr>
</tbody>
</table>

- All are second moment measures of dispersion – scatter or instability of frequency from central value.
- All are usually expressed as deviations.
- All are normalized to standard variance for white FM noise.
- All except standard variance converge for common clock noises.
- Modified types have additional phase averaging that can distinguish W and F PM noises.
- Time variances based on modified types.
- Hadamard types also converge for FW and RR FM noise.
- Overlapping types provide better confidence than classic Allan variance.
- Total types provide better confidence than corresponding overlapping types.
- Thêo1 (Theoretical Variance #1) provides stability data out to 75% of record length.
- Some are quite computationally intensive, especially if results are wanted at all (or many) analysis intervals (averaging times), \( \tau \).

The modified Allan deviation can be used to distinguish between white and flicker PM noise. For example, the W and F PM noise slopes are both \( \approx -1.0 \) on the ADEV plots below, but they can be distinguished as \(-1.5\) and \(-1.0\), respectively, on the MDEV plots.
The Hadamard deviation may be used to reject linear frequency drift when a stability analysis is performed. For example, the simulated frequency data for a rubidium frequency standard in the left plot below shows significant drift. Allan deviation plots for these data are shown in the right hand plots for the original and drift-removed data. Notice that, without drift removal, the Allan deviation plot has a $+\tau$ dependence at long $\tau$, a sign of linear frequency drift. However, the Hadamard deviation for the original data is nearly the same as the Allan deviation after drift removal, but it has lower confidence for a given $\tau$. 
5.2.1. Standard Variance

The classic N-sample or standard variance is defined as

\[ s^2 = \frac{1}{N-1} \sum_{i=1}^{N} (y_i - \bar{y})^2, \]

where the \( y_i \) are the \( N \) fractional frequency values, and \( \bar{y} = \frac{1}{N} \sum_{i=1}^{N} y_i \) is the average frequency. The standard variance is usually expressed as its square root, the standard deviation, \( s \). It is not recommended as a measure of frequency stability because it is non-convergent for some types of noise commonly found in frequency sources, as shown in the figure below.

References for Variances

The standard deviation (upper curve) increases with the number of samples of flicker FM noise used to determine it, while the Allan deviation (lower curve and discussed below) is essentially constant.

The problem with the standard variance stems from its use of the deviations from the average, which is not stationary for the more divergence noise types. That problem can be solved by instead using the first differences of the fractional frequency values (the second differences of the phase), as described for the Allan variance below.

In the context of frequency stability analysis, the standard variance is used primarily in the calculation of the B1 ratio for noise recognition.

Reference for Standard Variance


5.2.2. Allan Variance

The Allan variance is the most common time domain measure of frequency stability. Similar to the standard variance, it is a measure of the fractional frequency fluctuations, but has the advantage of being convergent for most types of clock noise. There are several versions of the Allan variance that provide better statistical confidence, can distinguish between white and flicker phase noise, and can describe time stability.

The original non-overlapped Allan, or 2-sample variance, AVAR, is the standard time domain measure of frequency stability [1, 2]. It is defined as

The original Allan variance has been largely superseded by its overlapping version.
\[
\sigma_y^2(\tau) = \frac{1}{2(M-1)} \sum_{i=1}^{M-1} [y_{i+1} - y_i]^2,
\]

where \(y_i\) is the \(i\)th of \(M\) fractional frequency values averaged over the measurement (sampling) interval, \(\tau\). Note that these \(y\) symbols are sometimes shown with a bar over them to denote the averaging.

In terms of phase data, the Allan variance may be calculated as

\[
\sigma_y^2(\tau) = \frac{1}{2(N-2)\tau^2} \sum_{i=1}^{N-2} [x_{i+2} - 2x_{i+1} + x_i]^2,
\]

where \(x_i\) is the \(i\)th of the \(N = M+1\) phase values spaced by the measurement interval \(\tau\).

The result is usually expressed as the square root, \(\sigma_y(\tau)\), the Allan deviation, \(\text{ADEV}\). The Allan variance is the same as the ordinary variance for white FM noise, but has the advantage, for more divergent noise types such as flicker noise, of converging to a value that is independent on the number of samples. The confidence interval of an Allan deviation estimate is also dependent on the noise type, but is often estimated as \(\pm \sigma_y(\tau) / \sqrt{N}\).

### 5.2.3. Overlapping Samples

Some stability calculations can utilize (fully) overlapping samples, whereby the calculation is performed by utilizing all possible combinations of the data set, as shown in the diagram and formulae below. The use of overlapping samples improves the confidence of the resulting stability estimate, but at the expense of greater computational time. The overlapping samples are not completely independent, but do increase the effective number of degrees of freedom. The choice of overlapping samples applies to the Allan and Hadamard variances. Other variances (e.g., total) always use them.

Overlapping samples don’t apply at the basic measurement interval, which should be as short as practical to support a large number of overlaps at longer averaging times.
The following plots show the significant reduction in variability, hence increased statistical confidence, obtained by using overlapping samples in the calculation of the Hadamard deviation:

Non-Overlapping Samples

Overlapping Samples

\[ H \sigma_y^2(\tau) = \frac{1}{6(M-2)} \sum_{i=1}^{M-2} (y_{i+2} - 2y_{i+1} + y_i)^2 \]

\[ H \sigma_y^2(\tau) = \frac{1}{6m^2(M-3m+1)} \sum_{i=1}^{M-3m+1} \sum_{i+m-1}^{i+m} (y_{i+2m} - 2y_{i+m} + y_i)^2 \]
5.2.4. Overlapping Allan Variance

The fully overlapping Allan variance, or AVAR, is a form of the normal Allan variance, $\sigma_y^2(\tau)$, that makes maximum use of a data set by forming all possible overlapping samples at each averaging time $\tau$. It can be estimated from a set of $M$ frequency measurements for averaging time $\tau = m\tau_0$, where $m$ is the averaging factor and $\tau_0$ is the basic measurement interval, by the expression

$$\sigma_y^2(\tau) = \frac{1}{2m^2(M - 2m + 1)} \sum_{j=1}^{M-2m+1} \left( \sum_{i=j}^{j+m-1} [y_{i+m} - y_i] \right)^2.$$

This formula is seldom used for large data sets because of the computationally intensive inner summation. In terms of phase data, the overlapping Allan variance can be estimated from a set of $N = M+1$ time measurements as

$$\sigma_y^2(\tau) = \frac{1}{2(N - 2m)\tau^2} \sum_{i=1}^{N-2m} \left[ x_{i+2m} - 2x_{i+m} + x_i \right]^2.$$

Fractional frequency data, $y_i$, can be first integrated to use this faster formula. The result is usually expressed as the square root, $\sigma_y(\tau)$, the Allan deviation, ADEV. The confidence interval of an overlapping Allan deviation estimate is better than that of a normal Allan variance estimation because, even though the additional overlapping differences are not all statistically independent, they nevertheless increase the number of degrees of freedom and thus improve the confidence in the estimation. Analytical methods are available for calculating the number of degrees of freedom for an estimation of overlapping Allan variance, and using that to establish single- or double-sided confidence intervals for the estimate with a certain confidence factor, based on Chi-squared statistics.

Sample variances are distributed according to the expression

$$\chi^2 = \frac{df \cdot s^2}{\sigma^2},$$

where $\chi^2$ is the Chi-square, $s^2$ is the sample variance, $\sigma^2$ is the true variance, and $df$ is the number of degrees of freedom (not necessarily an integer). For a particular statistic, $df$ is determined by the number of data points and the noise type.
References for Allan Variance


5.2.5. Modified Allan Variance

The modified Allan variance, $\text{Mod} \sigma_y^2(\tau)$, MVAR, is another common time domain measure of frequency stability [1]. It is estimated from a set of $M$ frequency measurements for averaging time $\tau = m\tau_0$, where $m$ is the averaging factor and $\tau_0$ is the basic measurement interval, by the expression

$$\text{Mod} \sigma_y^2(\tau) = \frac{1}{2m^4(M - 3m + 2)} \sum_{j=1}^{M-3m+2} \left( \sum_{i=j}^{j+m-1} \left( \sum_{k=i}^{i+m-1} [y_{k+m} - y_k] \right) \right)^2.$$

In terms of phase data, the modified Allan variance is estimated from a set of $N = M+1$ time measurements as

$$\text{Mod} \sigma_y^2(\tau) = \frac{1}{2m^2\tau^2(N - 3m + 1)} \sum_{j=1}^{N-3m+1} \left( \sum_{i=j}^{j+m-1} \left[ x_{i+2m} - 2x_{i+m} + x_i \right] \right)^2.$$

The result is usually expressed as the square root, $\text{Mod} \sigma_y(\tau)$, the modified Allan deviation. The modified Allan variance is the same as the normal Allan variance for $m = 1$. It includes an additional phase averaging operation, and has the advantage of being able to distinguish between white and flicker PM noise. The confidence interval of a modified Allan deviation determination is also dependent on the noise type, but is often estimated as $\pm \sigma_y(\tau)/\sqrt{N}$. Use the modified Allan deviation to distinguish between white and flicker PM noise.
References for Modified Allan Variance


5.2.6. Time Variance

The time Allan variance, TVAR, with square root TDEV, is a measure of time stability based on the modified Allan variance [1]. It is defined as

\[ \sigma_x^2(\tau) = (\tau^2/3) \cdot \text{Mod} \sigma_y^2(\tau). \]

In simple terms, TDEV is MDEV whose slope on a log-log plot is transposed by +1 and normalized by \( \sqrt{3} \). The time Allan variance is equal to the standard variance of the time deviations for white PM noise. It is particularly useful for measuring the stability of a time distribution network.

It can be convenient to include TDEV information on a MDEV plot by adding lines of constant TDEV, as shown in the following figure:
5.2.7. Time Error Prediction

The time error of a clock driven by a frequency source is a relatively simple function of the initial time offset, the frequency offset, and the subsequent frequency drift, plus the effect of noise, as shown in the following expression:

\[
\Delta T = T_o + (\Delta f/f) \cdot t + \frac{1}{2} D \cdot t^2 + \sigma_x(t),
\]

where \(\Delta T\) is the total time error, \(T_o\) is the initial synchronization error, \(\Delta f/f\) is the sum of the initial and average environmentally induced frequency offsets, \(D\) is the frequency drift (aging rate), and \(\sigma_x(t)\) is the (rms) noise-induced time deviation. For consistency, units of dimensionless fractional frequency and seconds should be used throughout.

Because of the many factors, conditions, and assumptions involved, and their variability, clock error prediction is seldom easy or exact, and it is usually necessary to generate a timing error budget.

- **Initial Synchronization**
  The effect of an initial time (synchronization) error, \(T_o\), is a constant time offset due to the time reference, the finite measurement resolution, and measurement noise. The measurement resolution and noise depends on the averaging time.

- **Initial Syntonization**
  The effect of an initial frequency (syntonization) error, \(\Delta f\), is a linear time error. Without occasional resyntonization (frequency recalibration), frequency aging can cause this to be the biggest contributor toward clock error for many frequency sources (e.g., quartz crystal oscillators and rubidium gas cell standards). Therefore, it can be important to have a means for periodic clock syntonization (e.g., GPS or cesium beam standard). In that case, the syntonization error is subject to uncertainty due to the frequency reference, the measurement and tuning resolution, and noise considerations. The measurement noise can be estimated by the square root of the sum of the Allan variances of the clock and reference over the measurement interval. The initial syntonization should be performed, to the greatest extent possible, under the same environmental conditions (e.g., temperature) as expected during subsequent operation.

References for Time Variance

Environmental Sensitivity

After initial syntonization, environmental sensitivity is likely to be the largest contributor to time error. Environmental frequency sensitivity obviously depends on the properties of the device and its operating conditions. When performing a frequency stability analysis, it is important to separate the deterministic environmental sensitivities from the stochastic noise. This requires a good understanding of both the device and its environment.

Reference for Time Error Prediction


5.2.8. Hadamard Variance

The Hadamard variance is a three-sample variance similar to the two-sample Allan variance that is commonly applied for the analysis of frequency stability data that has highly divergent noise (\( \alpha < -2 \)) or linear frequency drift. There are normal, overlapping, modified, and total versions of the Hadamard variance, with the overlapping version providing better estimates for this statistic, and the Hadamard total variance offering improved confidence at large averaging factors.

The Hadamard [1] variance is based on the Hadamard transform [2], which was adapted by Baugh as the basis of a time-domain measure of frequency stability [3]. As a spectral estimator, the Hadamard transform has higher resolution than the Allan variance, since the equivalent noise bandwidth of the Hadamard and Allan spectral windows are \( 1.2337N^{-1}\tau^{-1} \) and \( 0.476\tau^{-1} \) respectively [4]. For the purposes of time-domain frequency stability characterization, the most important advantage of the Hadamard variance is its insensitivity to linear frequency drift, making it particularly useful for the analysis of rubidium atomic clocks [6, 7]. It has also been used as one of the components of a time-domain multivariance analysis [5], and is related to the third structure function of phase noise [8].

Because the Hadamard variance examines the second difference of the fractional frequencies (the third difference of the phase variations), it converges for the Flicker Walk FM (\( \alpha = -3 \)) and Random Run FM (\( \alpha = -4 \)) power-law noise types. It is also unaffected by linear frequency drift.

For frequency data, the Hadamard variance is defined as:

\[
H\sigma_y^2(\tau) = \frac{1}{6(M-2)} \sum_{i=1}^{M-2} \left[ y_{i+2} - 2y_{i+1} + y_i \right]^2,
\]

where \( y_i \) is the ith of \( M \) fractional frequency values at averaging time \( \tau \).
For phase data, the Hadamard variance is defined as:

\[ H\sigma^2_y(\tau) = \frac{1}{6\tau^2(N-3)} \sum_{i=1}^{N-3} \left[ x_{i+3} - 3x_{i+2} + 3x_{i+1} - x_i \right]^2, \]

where \( x_i \) is the \( i \)th of \( N = M+1 \) phase values at averaging time \( \tau \).

Like the Allan variance, the Hadamard variance is usually expressed as its square-root, the Hadamard deviation, HDEV or \( H\sigma_y(\tau) \).

### 5.2.9. Overlapping Hadamard Variance

In the same way that the overlapping Allan variance makes maximum use of a data set by forming all possible fully overlapping 2-sample pairs at each averaging time \( \tau \), the overlapping Hadamard variance uses all 3-sample combinations [9]. It can be estimated from a set of \( M \) frequency measurements for averaging time \( \tau = m\tau_0 \) where \( m \) is the averaging factor and \( \tau_0 \) is the basic measurement interval, by the expression:

\[ H\sigma^2_y(\tau) = \frac{1}{6m^2(M-3m+1)} \sum_{j=1}^{M-3m+1} \left\{ \sum_{i=j}^{j+m-1} \left[ y_{i+2m} - 2y_{i+m} + y_i \right] \right\}^2, \]

where \( y_i \) is the \( i \)th of \( M \) fractional frequency values at each measurement time.

In terms of phase data, the overlapping Hadamard variance can be estimated from a set of \( N = M+1 \) time measurements as:

\[ H\sigma^2_y(\tau) = \frac{1}{6(N-3m)\tau^2} \sum_{j=1}^{N-3m} \left[ x_{i+3m} - 3x_{i+2m} + 3x_{i+m} - x_i \right]^2, \]

where \( x_i \) is the \( i \)th of \( N = M+1 \) phase values at each measurement time.

Computation of the overlapping Hadamard variance is more efficient for phase data, where the averaging is accomplished by simply choosing the appropriate interval. For frequency data, an inner averaging loop over \( m \) frequency values is necessary. The result is usually expressed as the square root, \( H\sigma_y(\tau) \), the Hadamard deviation, HDEV. The expected value of the overlapping statistic is the same as the normal one described above, but the confidence interval of the estimation is better. Even though not all the additional overlapping differences are statistically independent, they nevertheless increase the number of degrees of freedom and thus improve the confidence in the estimation. Analytical methods are available for
calculating the number of degrees of freedom for an overlapping Allan variance estimation, and that same theory can be used to establish reasonable single- or double-sided confidence intervals for an overlapping Hadamard variance estimate with a certain confidence factor, based on Chi-squared statistics.

Sample variances are distributed according to the expression:

\[
\chi^2(p, \text{df}) = (\text{df} \cdot s^2) / \sigma^2,
\]

where \(\chi^2\) is the Chi-square value for probability \(p\) and degrees of freedom \(\text{df}\), \(s^2\) is the sample variance, \(\sigma^2\) is the true variance, and \(\text{df}\) is the number of degrees of freedom (not necessarily an integer). The \(\text{df}\) is determined by the number of data points and the noise type. Given the \(\text{df}\), the confidence limits around the measured sample variance are given by

\[
\sigma^2_{\text{min}} = (s^2 \cdot \text{df}) / \chi^2(p, \text{df}), \quad \text{and} \quad \sigma^2_{\text{max}} = (s^2 \cdot \text{df}) / \chi^2(1-p, \text{df}).
\]

### 5.2.10. Modified Hadamard Variance

By similarity to the modified Allan variance, a modified version of the Hadamard variance can be defined [15] that employs averaging of the phase data over the \(m\) adjacent samples that define the analysis \(\tau = m \cdot \tau_0\). In terms of phase data, the three-sample modified Hadamard variance is defined as

\[
\text{Mod} \sigma^2_H(\tau) = \frac{\sum_{j=1}^{N-4m+1} \left( \sum_{i=j}^{j+m-1} \left[ x_i - 3x_{i+m} + 3x_{i+2m} - x_{i+3m} \right] \right)^2}{6m^2 \tau^2 \left[ N - 4m + 1 \right]},
\]

where \(N\) is the number of phase data points \(x_i\) at the sampling interval \(\tau_0\), and \(m\) is the averaging factor, which can extend from 1 to \(\lfloor N/4 \rfloor\). This is an unbiased estimator of the modified Hadamard variance, \(\text{MHVAR}\). Expressions for the equivalent number of \(\chi^2\) degrees of freedom (edf) required to set \(\text{MHVAR}\) confidence limits are available in [2].

Clock noise (and other noise processes) can be described in terms of power spectral density, which can be modeled as a power law function \(S \propto f^\alpha\), where \(f\) is Fourier frequency and \(\alpha\) is the power law exponent. When a variance such as \(\text{MHVAR}\) is plotted on log-log axes versus averaging time, the various power law noises correspond to particular slopes \(\mu\). \(\text{MHVAR}\) was developed in Reference [15] for determining the power law noise type of Internet traffic statistics, where it was found to be slightly better for that purpose than the modified Allan variance, \(\text{MAVAR}\), when there were a sufficient number of data points. \(\text{MHVAR}\) could also be useful for frequency stability analysis, perhaps in cases where it was necessary to distinguish between short-term white and flicker PM noise in the presence of more divergent (\(\alpha = -3\) and \(-4\)) flicker walk and random run FM noises. The \(\text{Mod} \sigma^2_H(\tau)\) log-log slope \(\mu\) is related to the power law noise exponent by \(\mu = -3 - \alpha\).

The modified Hadamard variance concept can be generalized to subsume \(\text{AVAR}\), \(\text{HVAR}\), \(\text{MAVAR}\), \(\text{MHVAR}\), and \(\text{MHVARs}\) using higher-order differences:
where \( d \) = phase differencing order; \( d = 2 \) corresponds to MAVAR, \( d = 3 \) to MHVAR; higher-order differencing is not commonly used in the field of frequency stability analysis. The unmodified, nonoverlapped AVAR and HVAR variances are given by setting \( m = 1 \). The allowable power law exponent for convergence of the variance is equal to \( \alpha > 1 - 2d \), so the second difference Allan variances can be used for \( \alpha > -3 \) and the third difference Hadamard variances for \( \alpha > -5 \).

Confidence intervals for the modified Hadamard variance can be determined by use of the edf values of Reference [16].
5.2.11. Total Variance

The total variance, TOTVAR, is a relatively new statistic for the analysis of frequency stability. It is similar to the two-sample or Allan variance, and has the same expected value, but offers improved confidence at long averaging times [1-5]. The work on total variance began with the realization that the Allan variance can "collapse" at long averaging factors because of symmetry in the data. An early idea was to shift the data by 1/4 of the record length and average the two resulting Allan variances. The next step was to wrap the data in a circular fashion and calculate the average of all the Allan variances at

The total variance offers improved confidence at large averaging factor by extending the data set by reflection at both ends.

References for Hadamard Variance

9. This expression for the overlapping Hadamard variance was developed by the author at the suggestion of G. Dieter and S.T. Hutseell.
every basic measurement interval, $\tau_\circ$. This technique is very effective in improving the confidence at long averaging factors but requires end matching of the data. A further improvement of the total variance concept was to extend the data by reflection, first at one end of the record and then at both ends. This latest technique, called TOTVAR, gives a very significant confidence advantage at long averaging times, exactly decomposes the classical standard variance [6], and is an important new general statistical tool. TOTVAR is defined for phase data as

$$\text{Tot var}(\tau) = \frac{1}{2\tau^2(N-2)} \sum_{i=2}^{N-1} \left[ x^*_{i-m} - 2x^*_i + x^*_{i+m} \right]^2,$$

where $\tau = m \tau_\circ$ and the $N$ phase values $x$ measured at $\tau = \tau_\circ$ are extended by reflection about both endpoints to form a virtual sequence $x^*$ from $i = 3\cdot N$ to $i = 2\cdot N - 2$ of length $3\cdot N - 4$. The original data are in the center of $x^*$ with $i = 1$ to $N$ and $x^* = x$. The reflected portions added at each end extend from $j = 1$ to $N - 2$ where $x^*_{1-j} = 2x_1 - x_{1+j}$ and $x^*_{N+j} = 2x_N - x_{N-j}$.

Totvar can also be defined for frequency data as

$$\text{Tot var}(\tau) = \frac{1}{2(M-1)} \sum_{j=1}^{M-1} \left[ y^*_{i+j+1} - y^*_{i+j} \right]^2,$$

where the $M = N-1$ fractional frequency values, $y$, measured at $\tau = \tau_\circ$ (N phase values) are extended by reflection at both ends to form a virtual array $y^*$. The original data are in the center, where $y^*_{i} = y_i$ for $i = 1$ to $M$, and the extended data for $j = 1$ to $M-1$ are equal to $y^*_{1-j} = y_j$ and $y^*_{M+1-j} = y_{M+1-j}$.

The result is usually expressed as the square root, $\sigma_{\text{tot}}(\tau)$, the total deviation, TOTDEV. When calculated by use of the doubly reflected method described above, the expected value of TOTVAR is the same as AVAR for white and flicker PM or white FM noise. Bias corrections of the form $1/[1-a(\tau/T)]$, where $T$ is the record length, need to be applied for flicker and random walk FM noise, where $a=0.481$ and 0.750, respectively.

The number of equivalent $\chi^2$ degrees of freedom for TOTVAR can be estimated for white FM, flicker FM and random walk FM noise by the expression $b(T/\tau)-c$, where $b=1.500$, 1.168 and 0.927, and $c=0$, 0.222 and 0.358, respectively. For white and flicker PM noise, the edf for a total deviation estimate is the same as that for the overlapping ADEV with the number of $\chi^2$ degrees of freedom increased by 2.
References for Total Variance


5.2.12. Modified Total Variance

The modified total variance, MTOT, is another new statistic for the analysis of frequency stability. It is similar to the modified Allan variance, MVAR, and has the same expected value, but offers improved confidence at long averaging times. It uses the same phase averaging technique as MVAR to distinguish between white and flicker PM noise processes.

A calculation of MTOT begins with an array of N phase data points (time deviates, $x_i$) with sampling period $\tau_0$ that are to be analyzed at averaging time $\tau=m\tau_0$. MTOT is computed from a set of N-3m+1 subsequences of 3m points. First, a linear trend (frequency offset) is removed from the subsequence by averaging the first and last halves of the subsequence and dividing by half the interval. Then the offset-removed subsequence is extended at both ends by uninverted, even reflection. Next the modified Allan variance is computed for these 9m points. Finally, these steps are repeated for each of the N-3m+1 subsequences, calculating MTOT as their overall average. These steps, similar to those for MTOT, but acting on fractional frequency data, are shown in the diagram below:
### Phase Data $x_i$, $i = 1$ to $N$

#### N-3m+1 Subsequences:

- $x_i$ for $i = n$ to $n + 3m - 1$

#### Linear Trend Removed:

- $\bar{x}_i = x_i - c_i$, $i$, $c_i$ = freq offset

#### Extended Subsequence:

- $x_{n-l}$ to $x_{n+l-1}$

#### Uninverted, Even Reflection:

- $x_{n+3m-l}$ to $x_{n+3m+l-1}$

#### 9 m-Point Averages:

#### 6m 2nd Differences:

#### Calculate Mod $\sigma^2_\tau$ for Subsequence:

- $\text{mod } \sigma^2_\tau = \frac{1}{2} \sum_{i=1}^{N-3m+1} \left( z_i^0(m) \right)$, where $z_i^0(m) = x_i(m) - 2x_{n+3m}(m) + x_{n+6m}(m)$

Computationally, the MTOT process requires three nested loops:

- An outer summation over the N-3m+1 subsequences. The 3m-point subsequence is formed, its linear trend is removed, and it is extended at both ends by uninverted, even reflection to 9m points.
- An inner summation over the 6m unique groups of m-point averages from which all possible fully overlapping second differences are used to calculate MVAR.
- A loop within the inner summation to sum the phase averages for three sets of m points.

The final step is to scale the result according to the sampling period, $\tau_0$, averaging factor, m, and number of points, N. Overall, this can be expressed as:

$$\text{Mod Tot var}(\tau) = \frac{1}{2(m \tau_0)^2(N - 3m + 1)} \sum_{n=1}^{N-3m+1} \left\{ \frac{1}{6m} \sum_{i=n-3m}^{n+3m-1} \left[ z_i^0(m) \right]^2 \right\}$$

where the $z_i^0(m)$ terms are the phase averages from the triply-extended subsequence, and the prefix $0$ denotes that the linear trend has been removed. At the largest possible averaging factor, $m = N/3$, the outer summation consists of only one term, but the inner summation has 6m terms, thus providing a sizable number of estimates for the variance.

### Reference for Modified Total Variance

5.2.13. **Time Total Variance**

The time total variance, TTOT, is a similar measure of time stability, based on the modified total variance. It is defined as

\[ \sigma_x^2(\tau) = (\tau^3/3) \cdot \text{Mod} \sigma_{\text{total}}^2(\tau). \]

5.2.14. **Hadamard Total Variance**

The Hadamard total variance, HTOT, is a total version of the Hadamard variance. As such, it rejects linear frequency drift while offering improved confidence at large averaging factors.

An HTOT calculation begins with an array of N fractional frequency data points, \( y_i \) with sampling period \( \tau_0 \) that are to be analyzed at averaging time \( \tau = m\tau_0 \). HTOT is computed from a set of N-3m+1 subsequences of 3m points. First, a linear trend (frequency drift) is removed from the subsequence by averaging the first and last halves of the subsequence and dividing by half the interval. Then the drift-removed subsequence is extended at both ends by uninverted, even reflection. Next the Hadamard variance is computed for these 9m points. Finally, these steps are repeated for each of the N-3m+1 subsequences, calculating HTOT as their overall average. These steps are shown in the diagram below:
N-3m+1 Subsequences: \[ y_i \text{ for } i = n \text{ to } n+3m-1 \]

Linear Freq Drift Removed: \[ y_i^0 = y_i - c_i \cdot i, \quad c_i = \text{freq drift} \]

Extended Subsequence: \[ y_{n+3m-l}^0 = y_{n+l-1}^0 \text{ for } 1 \leq l \leq 3m \]

Uninverted, Even Reflection: \[ y_{n+3m+l-1}^0 = y_{n+3m-l}^0 \]

9 m-Point Averages: 

6m 2nd Differences:

Calculate Hadamard second differences: \[ z_n(m) = y_n(m) - 2y_{n+m}(m) + y_{n+2m}(m) \]

Then Find HTOT as Average of Subestimates

Computationally, the HTOT process requires three nested loops:

- An outer summation over the N-3m+1 subsequences. The 3m-point subsequence is formed, its linear trend is removed, and it is extended at both ends by uninverted, even reflection to 9m points.
- An inner summation over the 6m unique groups of m-point averages from which all possible fully overlapping second differences are used to calculate HVAR.
- A loop within the inner summation to sum the frequency averages for three sets of m points.

The final step is to scale the result according to the sampling period, \( \tau_0 \), averaging factor, m, and number of points, N. Overall, this can be expressed as:

\[
TotalH\sigma_y^2(m, \tau_0, N) = \frac{1}{6(N-3m+1)} \sum_{n=1}^{N-3m+1} \left( \frac{1}{6m} \sum_{i=n-3m}^{n+3m-1} (H_i(m))^2 \right),
\]

where the \( H_i(m) \) terms are the \( z_n(m) \) Hadamard second differences from the triply extended, drift-removed subsequences. At the largest possible averaging factor, \( m = N/3 \), the outer summation consists of only one term, but the inner summation has 6m terms, thus providing a sizable number of estimates for the variance. The Hadamard total variance is a biased estimator of the Hadamard variance, so a bias correction is required that is dependent on the power law noise type and number of samples.
The following plots show the improvement in the consistency of the overlapping Hadamard deviation results compared with the normal Hadamard deviation, and the extended averaging factor range provided by the Hadamard total deviation [10].
A comparison of the overlapping and total Hadamard deviations shows the tighter error bars of the latter, allowing an additional point to be shown at the longest averaging factor.

The Hadamard variance may also be used to perform a frequency domain (spectral) analysis because it has a transfer function that is a close approximation to a narrow rectangle of spectral width $1/(2\cdot N\cdot \tau_0)$, where $N$ is the number of samples, and $\tau_0$ is the measurement time [3]. This leads to a simple expression for the spectral density of the fractional frequency fluctuations $S_y(f) = 0.73 \cdot \tau_0 \cdot H_{\mathcal{F} \mathcal{T}}^2(\tau) / N$, where $f = 1/(2\cdot \tau_0)$, which can be particularly useful at low Fourier frequencies.
The Picinbono variance is a similar three-sample statistic. It is identical to the Hadamard variance except for a factor of 2/3 [4]. Sigma-z is another statistic that is similar to the Hadamard variance that has been applied to the study of pulsars [5].

It is necessary to identify the dominant power law noise type as the first step in determining the estimated number of chi-squared degrees of freedom for the Hadamard statistics so their confidence limits can be properly set [6]. Because the Hadamard variances can handle the divergent flicker walk FM and random run FM power law noises, techniques for those noise types must be included. Noise identification is particularly important for applying the bias correction to the Hadamard total variance.

**References for Hadamard Total Variance**


**5.2.15. Thêo1**

The Thêo1 statistic is a two-sample variance similar to the Allan variance that provides improved confidence, and the ability to obtain a result for a maximum averaging time equal to 75 % of the record length.

Thêo1 [1] is defined as follows:

\[
\text{Thêo1}(m, \tau_0, N) = \frac{1}{(N - m)(m \tau_0)^2 \cdot 0.75} \sum_{i=1}^{N-m} \sum_{\delta=0}^{m/2-1} \left( \frac{1}{m/2-\delta} \right) \left\{ (x_i - x_{i+\delta+m/2} + (x_{i+m} - x_{i+\delta+m/2}) \right\}^2 ,
\]

where \( m \) = averaging factor, \( \tau_0 \) = measurement interval, and \( N \) = number of phase data points, for \( m \) even, and \( 10 \leq m \leq N - 1 \). It consists of \( N - m \) outer sums over the number of phase data points \(-1\), and \( m/2 \) inner sums. Thêo1 is the rms of frequency differences averaged over an averaging time \( \tau = 0.75 (m-1) \tau_0 \).
A schematic for a Théo1 calculation is shown in the figure below. This example is for eleven phase samples \( (N = 11) \) at the largest possible averaging factor \( (m = 10) \).

![Schematic for Théo1 calculation](image)

The single outer summation \( (i = 1 \text{ to } 1) \) at the largest possible averaging factor consists of \( m/2 = 5 \) terms, each with two phase differences. These terms are scaled by their spans \( m/2 - \delta = 5 \) thru 1 so that they all have equal weighting. A total of 10 terms contribute to the Théo1 statistic at this largest-possible averaging factor. The averaging time, \( \tau \), associated with a Théo1 value is \( \tau = 0.75 \cdot m \cdot \tau_0 \), where \( \tau_0 \) is the measurement interval. Théo1 has the same expected value as the Allan variance for white FM noise, but provides many more samples that provide improved confidence and the ability to obtain a result for a maximum \( \tau \) equal to \( 3/4 \) of the record length, \( T \). Théo1 is a biased estimator of the Allan variance, \( \text{Avar} \), for all noise types except white FM noise, and it therefore requires the application of a bias correction. Reference [2] contains the preferred expression for determining the Théo1 bias as a function of noise type and averaging factor:

\[
\text{Théo1 Bias} = \frac{\text{Avar}}{\text{Théo1}} = a + \frac{b}{m^c},
\]

where \( m \) is the averaging factor and the constants \( a, b \) and \( c \) are given in the table below. Note that the effective tau for a Théo1 estimation is \( \tau = 0.75 \cdot m \cdot \tau_0 \), where \( \tau_0 \) is the measurement interval.

<table>
<thead>
<tr>
<th>Noise</th>
<th>Alpha</th>
<th>a</th>
<th>b</th>
<th>c</th>
</tr>
</thead>
<tbody>
<tr>
<td>RW FM</td>
<td>-2</td>
<td>2.70</td>
<td>-1.53</td>
<td>0.85</td>
</tr>
<tr>
<td>F FM</td>
<td>-1</td>
<td>1.87</td>
<td>-1.05</td>
<td>0.79</td>
</tr>
<tr>
<td>W FM</td>
<td>0</td>
<td>1.00</td>
<td>0.00</td>
<td>0.00</td>
</tr>
<tr>
<td>F PM</td>
<td>1</td>
<td>0.14</td>
<td>0.82</td>
<td>0.30</td>
</tr>
<tr>
<td>W PM</td>
<td>2</td>
<td>0.09</td>
<td>0.74</td>
<td>0.40</td>
</tr>
</tbody>
</table>

Empirical formulae have been developed [1] for the number of equivalent \( \chi^2 \) degrees of freedom for the Théo1 statistic, as shown in the following table:
5.2.16. NewThêo1, ThêoBR and ThêoH

Because Thêo1 has the same bias-corrected expected value as the Allan variance, and because it covers a different (larger) range of averaging factors, 10 to N-2 versus 1 to (N-1)/2, it is useful to combine Thêo1 and AVAR results into a composite stability analysis. Several forms of this composite statistic have evolved, from simply plotting the two, to matching the Thêo1 points on-average to the AVAR points at those averaging times where both are available, to the New Thêo1 and ThêoH statistics described below.

The NewThêo1 algorithm of Reference [2] provides a method of automatic bias correction for a Thêo1 estimation based on the average ratio of the Allan and Thêo1 variances over a range of averaging factors:

\[
\text{NewThêo1}(m, \tau_0, N) = \left[ \frac{1}{n+1} \sum_{i=0}^{n} \text{Avar}(m = 9 + 3i, \tau_0, N) \right] \text{Thêo1}(m = 12 + 4i, \tau_0, N),
\]

where \( n = \left\lfloor \frac{N}{30} - 3 \right\rfloor \) and \( \left\lfloor \cdot \right\rfloor \) denotes the floor function.

NewThêo1 was used in Reference [2] to form a composite AVAR/ NewThêo1 result called LONG, which has been superseded by ThêoH (see below).

ThêoBR [3] is an improved bias-removed version of Thêo1 given by

<table>
<thead>
<tr>
<th>Thêo1 EDF Formulae</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Noise</strong></td>
</tr>
<tr>
<td>RW FM</td>
</tr>
<tr>
<td>F FM</td>
</tr>
<tr>
<td>W FM</td>
</tr>
<tr>
<td>F PM</td>
</tr>
<tr>
<td>W PM</td>
</tr>
</tbody>
</table>

where \( r = 0.75m \), and with the condition \( \tau_0 \leq T/10 \).
TheoBR\( (m, \tau_0, N) \) = \( \left[ \frac{1}{n+1} \sum_{i=0}^{n} \text{Avar}(m = 9 + 3i, \tau_0, N) \right] \text{Thêo l}(m, \tau_0, N) \),

where \( n = \left\lfloor \frac{N}{6} - 3 \right\rfloor \) and \( \left\lfloor \cdot \right\rfloor \) denotes the floor function.

TheoBR is computationally intensive for large data sets, but it can determine an unbiased estimate of the Allan variance over the widest possible range of averaging times without explicit knowledge of the noise type.

TheoH is a hybrid statistic that combines TheoBR and AVAR:

\[
\text{TheoH}(m, \tau_0, N) = \begin{cases} 
\text{Avar}(m, \tau_0, N) & \text{for } 1 \leq m < \frac{k}{\tau_0}, \\
\text{TheoBR}(m, \tau_0, N) & \text{for } \frac{k}{0.75\tau_0} \leq m \leq N - 1, \text{ even}
\end{cases}
\]

where \( k \) is the largest available \( \tau \leq 20\% \text{ T} \).

It is the best statistic available for estimating the stability of a frequency source at large averaging factors. An example of a TheoH plot is shown in the figure below:
ThêoH is a composite of AVAR and bias-corrected ThêoBR analysis points at a number of averaging times sufficiently large to form a quasi-continuous curve. The data are a set of 1001 simulated phase values measured at 15-minute intervals taken over a period of about 10 days. The AVAR results are able to characterize the stability to an averaging time of about two days, while Thêo1 is able to extend the analysis out to nearly a week, thus providing significantly more information from the same data set. The analysis requires less than five seconds on a 1 GHz Pentium processor.

References for Thêo1, NewThêo1, ThêoBR and ThêoH


5.2.17. MTIE

The maximum time interval error, MTIE, is a measure of the maximum time error of a clock over a particular time interval. This statistic is very commonly used in the telecommunications industry. It is calculated by moving an n-point \((n = \tau/\tau_o)\) window through the phase (time error) data and finding the difference between the maximum and minimum values (range) at each window position. MTIE is the overall maximum of this time interval error over the entire data set:

\[
MTIE(\tau) = \max_{1 \leq k \leq N-n} \left\{ \max_{k \leq s \leq k+n} (x_s) - \min_{k \leq s \leq k+n} (x_s) \right\}
\]

where \(n = 1,2,\ldots, N-1\) and \(N = \) number of phase data points.

MTIE is a measure of clock error commonly used in the telecommunications industry.

MTIE is a measure of the peak time deviation of a clock and is therefore very sensitive to a single extreme value, transient or outlier. The time required for an MTIE calculation increases geometrically with the averaging factor, \(n\), and can become very long for large data sets (although faster algorithms are available – see Reference 4 below).

The relationship between MTIE and Allan variance statistics is not completely defined, but has been the subject of recent theoretical work [1, 2]. Because of the peak nature of the MTIE statistic, it is necessary to express it in terms of a probability level, \(\beta\), that a certain value is not exceeded.
For the case of white FM noise (important for passive atomic clocks such as the most common rubidium and cesium frequency standards), MTIE can be approximated by the relationship

$$MTIE(\tau, \beta) = k_\beta \sqrt{(h_0 \cdot \tau)} = k_\beta \sqrt{2 \cdot \sigma_y(\tau) \cdot \tau},$$

where $k_\beta$ is a constant determined by the probability level, $\beta$, as given in the table below, and $h_0$ is the white FM power-law noise coefficient.

<table>
<thead>
<tr>
<th>$\beta$, %</th>
<th>$k_\beta$</th>
</tr>
</thead>
<tbody>
<tr>
<td>95</td>
<td>1.77</td>
</tr>
<tr>
<td>90</td>
<td>1.59</td>
</tr>
<tr>
<td>80</td>
<td>1.39</td>
</tr>
</tbody>
</table>

The maximum time interval error (MTIE) and rms time interval error (TIE rms) are clock stability measures commonly used in the telecom industry [3, 5]. MTIE is determined by the extreme time deviations within a sliding window of span $\tau$, and is not as easily related to such clock noise processes as TDEV [1]. MTIE is computationally intensive for large data sets [7].

### References for MTIE


### 5.2.18. TIE rms

The rms time interval error, TIE rms, is another clock statistic commonly used by the telecommunications industry. TIE rms is defined by the expression

$$TIE_{rms} = \sqrt{\frac{1}{N-n} \sum_{i=1}^{N-n} (x_{i+n} - x_i)^2},$$
where n = 1, 2, ..., N-1 and N = # phase data points.

For no frequency offset, TIE rms is approximately equal to the standard deviation of the fractional frequency fluctuations multiplied by the averaging time. It is therefore similar in behavior to TDEV, although the latter properly identifies divergent noise types.

**Reference for TIE rms**


### 5.2.19. Integrated Phase Jitter and Residual FM

Integrated phase jitter and residual FM are other ways of expressing the net phase or frequency jitter by integrating it over a certain bandwidth. These can be calculated from the amplitudes of the various power law terms.

The power law model for phase noise spectral density (see section 6.1) can be written as

\[ S_\phi(f) = K \cdot f^x, \]

where \( S_\phi \) is the spectral density of the phase fluctuations in rad\(^2\)/Hz, \( f \) is the modulation frequency, \( K \) is amplitude in rad\(^2\), and \( x \) is the power law exponent. It can be represented as a straight line segment on a plot of \( S_\phi(f) \) in dB relative to 1 rad\(^2\)/Hz versus log \( f \) in hertz. Given two points on the plot \((f_1, dB_1)\) and \((f_2, dB_2)\), the values of \( x \) and \( K \) may be determined by

\[
x = \frac{dB_1 - dB_2}{10 \cdot (\log f_1 - \log f_2)},
\]

and

\[
K = 10^{\left(\frac{dB}{10} - x \log f_1\right)}.
\]

The integrated phase jitter can then be found over this frequency interval by

\[
\Delta \phi^2 = \int_{f_1}^{f_2} S_\phi(f) \cdot df = \int_{f_1}^{f_2} K \cdot f^x \cdot df
\]

\[
\Delta \phi^2 = \frac{K}{x+1} (f_2^{x+1} - f_1^{x+1}) \text{ for } x \neq -1
\]

\[
\Delta \phi^2 = K \cdot (\log f_2 - \log f_1) \text{ for } x = -1
\]

It is usually expressed as \( \Delta \phi \) in rms radians.
Similarly, the spectral density of the frequency fluctuations in Hz²/Hz is given by

\[ S_v(f) = \nu_0^2 \cdot S_y(f) = f^2 \cdot S_{\Delta \phi}(f) = K \cdot f^{x+2}, \]

where \( \nu_0 \) is the carrier frequency in hertz, and \( S_y(f) \) is the spectral density of the fractional frequency fluctuations (see section 6.1).

The integrated frequency jitter or residual FM is therefore

\[ \Delta f^2 = \int_{f_1}^{f_2} S_v(f) \cdot df = \int_{f_1}^{f_2} K \cdot f^{x+2} \cdot df \]

\[ \Delta f^2 = \frac{K}{x+3} (f_2^{x+3} - f_1^{x+3}) \text{ for } x \neq 3 \]

\[ \Delta f^2 = K \cdot (\log f_2 - \log f_1) \text{ for } x = -3 \]

It is usually expressed as \( \Delta f \) in rms hertz.

The value of \( S_{\Delta \phi}(f) \) in dB can be found from the more commonly used \( \xi(f) \) measure of SSB phase noise to carrier power ratio in dBC/Hz by adding 3 dB. The total integrated phase noise is obtained by summing the \( \Delta \phi^2 \) contributions from the straight-line approximations for each power law noise type. The ratio of total phase noise to signal power in the given integration bandwidth is equal to 10 log \( \Delta \phi^2 \).

References for Integrated Phase Noise and Residual FM


5.2.20. Dynamic Stability

A dynamic stability analysis uses a sequence of sliding time windows to perform a dynamic Allan (DAVAR) or Hadamard (DHVAR) analysis, thereby showing changes (nonstationarity) in clock behavior versus time. It is able to detect variations in clock stability (noise bursts, changes in noise level or type, etc.) that would be difficult to see in an ordinary overall stability analysis. The results of a dynamic stability analysis are presented as a 3D surface plot of log sigma versus log tau or averaging factor as a function of time or window number.

An example of a DAVAR plot is shown below. This example is similar to the one of Figure 2 in Reference [2], showing a source with white PM noise that changes by a factor of 2 at the middle of the record.
5.3. Confidence Intervals

It is wise to include error bars (confidence intervals) on a stability plot to indicate the degree of statistical confidence in the numerical results. The confidence limits of a variance estimate depend on the variance type, the number of data points and averaging factor, the statistical confidence factor desired, and the type of noise. This section describes the use of $\chi^2$ statistics for setting the confidence intervals and error bars of a stability analysis.

It is generally insufficient to simply calculate a stability statistic such as the Allan deviation, thereby finding an estimate of its expected value. That determination should be accompanied by an indication of the confidence in its value as expressed by the upper and (possibly) lower limits of the statistic with a certain confidence factor. For example, if the estimated value of the Allan deviation is $1.0 \times 10^{-11}$, depending on the noise type and size of the data set, one could state with 95% confidence that the actual value does not exceed (say) $1.2 \times 10^{-11}$. It is always a good idea to include such a confidence limit in reporting a statistical
result, which can be shown as an upper numeric limit, upper and lower numeric bounds, or (equivalently) error bars on a plot. Even though those confidence limits or error bars are themselves inexact, they should be included to indicate the validity of the reported result. If you are unfamiliar with the basics of confidence limits, it is recommended that an introductory statistics book be consulted for an introduction to this subject. For frequency stability analysis, the emphasis is on various variances, whose confidence limits (variances of variances) are treated with chi-squared ($\chi^2$) statistics. Strictly speaking, $\chi^2$ statistics apply to the classical standard variance, but they have been found applicable to all of the other variances (Allan, Hadamard, total, Théo1, etc.) used for frequency stability analysis. A good introduction to confidence limits and error bars for the Allan variance may be found in Reference [1]. The basic idea is to (1) choose an single or double-sided confidence limits (upper or upper and lower bounds), (2) choose an appropriate confidence factor (e.g. 95 %), (3) determine the number of equivalent $\chi^2$ degrees of freedom (edf), (4) use the inverse $\chi^2$ distribution to find the normalized confidence limit(s), and (5) multiply those by the nominal deviation value to find the error bar(s).

5.3.1. Simple Confidence Intervals

The simplest confidence interval approximation, with no consideration of the noise type, sets the $\pm 1 \sigma (68 \%)$ error bars at $\pm \sigma(\tau) \sqrt{N}$, where $N$ is the number of frequency data points used to calculate the Allan deviation.

A more accurate determination of this confidence interval can be made by considering the noise type, which can be estimated by the B1 bias function (the ratio of the standard variance to the Allan variance). That noise type is then be used to determine a multiplicative factor, $K_n$, to apply to the confidence interval:

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>$K_n$</th>
</tr>
</thead>
<tbody>
<tr>
<td>Random Walk FM</td>
<td>0.75</td>
</tr>
<tr>
<td>Flicker FM</td>
<td>0.77</td>
</tr>
<tr>
<td>White FM</td>
<td>0.87</td>
</tr>
<tr>
<td>Flicker PM</td>
<td>0.99</td>
</tr>
<tr>
<td>White PM</td>
<td>0.99</td>
</tr>
</tbody>
</table>

5.3.2. Chi-Squared Confidence Intervals

Chi-squared statistics can be applied to calculate single and double-sided confidence intervals at any desired confidence factor. These calculations are based on a determination of the number of degrees of freedom for the estimated noise type. Most stability plots show $\pm 1 \sigma$ error bars for its overlapping Allan deviation plot.

The error bars for the modified Allan and time variances are also determined by Chi-squared statistics, using the number of MVAR degrees of freedom for the particular noise type, averaging factor, and number of data points. During the Run function, noise type estimates are made at each averaging factor (except the last, where the noise type of the previous averaging factor is used).

Sample variances are distributed according to the expression
\[ \chi^2 = \frac{edf \cdot s^2}{\sigma^2}, \]

where \( \chi^2 \) is the Chi-square, \( s^2 \) is the sample variance, \( \sigma^2 \) is the true variance, and \( edf \) is the equivalent number of degrees of freedom (not necessarily an integer). The \( edf \) is determined by the number of analysis points and the noise type. Procedures exist for establishing single- or double-sided confidence intervals with a selectable confidence factor, based on \( \chi^2 \) statistics, for many of its variance functions. The general procedure is to choose a single- or double-limited confidence factor, \( p \), calculate the corresponding \( \chi^2 \) value, determine the \( edf \) from the variance type, noise type and number of analysis points, and thereby set the statistical limit(s) on the variance. For double-sided limits,

\[
\sigma^2_{\text{min}} = s^2 \cdot \frac{edf}{\chi^2(p,edf)} \quad \text{and} \quad \sigma^2_{\text{max}} = s^2 \cdot \frac{edf}{\chi^2(1-p,edf)}.
\]

### 5.4. Degrees of Freedom

The equivalent number of \( \chi^2 \) degrees of freedom (edf) associated with a statistical variance (or deviation) estimate depends on the variance type, the number of data points, and the type of noise involved. In general, the progression from the original two-sample (Allan) variance to the overlapping, total, and Théo1 variances has provided larger edfs and better confidence. The noise type matters because it determines the extent that the points are correlated. Highly correlated data have a smaller edf than does the same number of points of uncorrelated (white) noise. An edf determination therefore involves (1) choosing the appropriate algorithm for the particular variance type, (2) determining the dominant power law noise type of the data, and (3) using the number of data points to calculate the corresponding edf.

#### 5.4.1. AVAR, MVAR, TVAR and HVAR EDF

The equivalent number of \( \chi^2 \) degrees of freedom (edf) for the Allan variance (AVAR), the modified Allan variance (MVAR) and the related time variance (TVAR), and the Hadamard variance (HVAR) is found by a combined algorithm developed by C.A. Greenhall, based on its generalized autocovariance function [2].
Overlapping ADEV EDF for W FM Noise

Overlapping ADEV EDF for N=100
This method for estimating the edf for the Allan, modified Allan and Hadamard variances supersedes the following somewhat simpler empirical approximations (which may still be used).

The equivalent \# of $\chi^2$ degrees of freedom (edf) for the fully overlapping Allan variance (AVAR) can be estimated by the following approximation formulae for each power law noise type:

<table>
<thead>
<tr>
<th>Power Law Noise Type</th>
<th>AVAR edf, where $N = #$ phase data points, $m =$ averaging factor $= \tau / \tau_0$</th>
</tr>
</thead>
<tbody>
<tr>
<td>W PM</td>
<td>$\frac{(N + 1)(N - 2m)}{2(N - m)}$</td>
</tr>
<tr>
<td>F PM</td>
<td>$\exp \left[ \ln \left( \frac{(N - 1)}{2m} \right) \ln \left( \frac{(2m + 1)(N - 1)}{4} \right) \right]^{1/2}$</td>
</tr>
<tr>
<td>W FM</td>
<td>$\left[ \frac{3(N - 1)}{2m} - \frac{2(N - 2)}{N} \right] \frac{4m^2}{4m^2 + 5}$</td>
</tr>
<tr>
<td>F FM</td>
<td>$\frac{2(N - 2)^2}{2.3N - 4.9}$ For $m = 1$</td>
</tr>
</tbody>
</table>
The edf for the modified Allan variance (MVAR) can be estimated by the same expression as the overlapping Hadamard variance (see below) with the arguments changed as follows (valid for \(-2 \leq \alpha \leq 2\)): MVAR and TVAR edf for N, m and \(\alpha = MVAR\) edf for N+1, m and \(\alpha - 2\).

The edf for the fully overlapping Hadamard variance (HVAR) can be found by an earlier algorithm also developed by C.A. Greenhall based on its generalized autocovariance function. The HVAR edf is found either as a summation (for small m cases with a small number of terms) or from a limiting form for large m, where \(1/edf = (1/p)(a0 - a1/p)\), with the coefficients as follows:

<table>
<thead>
<tr>
<th>Power Law Noise Type</th>
<th>HVAR edf coefficients</th>
</tr>
</thead>
<tbody>
<tr>
<td>FM</td>
<td>a0 (7/9)</td>
</tr>
<tr>
<td>FM</td>
<td>a1 (1/2)</td>
</tr>
</tbody>
</table>

\[ \frac{5N^2}{4m(N+3m)} \quad \text{For } m > 1 \]

\[ \left( \frac{N-2}{m} \right)^{- \frac{(N-1)^2}{2} - \frac{3m(N-1) + 4m^2}{(N-3)^2}} \]

## 5.4.2. TOTVAR and TTOT EDF

The edf for the total variance (TOTVAR) and the related total time variance (TTOT) is given by the formula \(b(T/\tau) - c\), where \(T\) is the length of the data record, \(\tau\) is the averaging time, and \(b\) & \(c\) are coefficients that depend on the noise type, as shown in the following table:

<table>
<thead>
<tr>
<th>Power Law Noise Type</th>
<th>TOTVAR edf Coefficients</th>
</tr>
</thead>
<tbody>
<tr>
<td>FM</td>
<td>b 1.50  (0)</td>
</tr>
<tr>
<td>FM</td>
<td>c 0.22</td>
</tr>
<tr>
<td>Random Walk FM</td>
<td>b 0.93  (0.36)</td>
</tr>
</tbody>
</table>

## 5.4.3. MTOT EDF

The edf for the modified total variance (MTOT) is given by the same formula \(b(T/\tau) - c\), where \(T\) is the length of the data record, \(\tau\) is the averaging time, and \(b\) and \(c\) are coefficients that depend on the noise type as shown in the following table:


<table>
<thead>
<tr>
<th>Power Law Noise Type</th>
<th>MTOT edf Coefficients</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>b</td>
</tr>
<tr>
<td>White PM</td>
<td>1.90</td>
</tr>
<tr>
<td>Flicker PM</td>
<td>1.20</td>
</tr>
<tr>
<td>White FM</td>
<td>1.10</td>
</tr>
<tr>
<td>Flicker FM</td>
<td>0.85</td>
</tr>
<tr>
<td>Random Walk FM</td>
<td>0.75</td>
</tr>
</tbody>
</table>

### 5.4.4. Thêo1 EDF

The equivalent number of $\chi^2$ degrees of freedom (edf) for the Thêo1 variance is determined by the following approximation formulae for each power low noise type.

<table>
<thead>
<tr>
<th>Power Law Noise Type</th>
<th>Thêo1 edf, where $N = #$ phase data points, $\tau=0.75m$, $m =$ averaging factor $= \tau/\tau_0$</th>
</tr>
</thead>
<tbody>
<tr>
<td>White PM</td>
<td>$edf = \left( \frac{0.86(N_x + 1)(N_x - \frac{4}{3}\tau)}{N_x - \tau} \right) \left( \frac{\tau}{\tau + 1.14} \right)$</td>
</tr>
<tr>
<td>Flicker PM</td>
<td>$edf = \left( \frac{4.798N_x^2 - 6.374N_x\tau + 12.387\tau}{(\tau + 36.6)^{1/2}(N_x - \tau)} \right) \left( \frac{\tau}{\tau + 0.3} \right)$</td>
</tr>
<tr>
<td>White FM</td>
<td>$edf = \left[ \frac{4.1N_x + 0.8}{\tau} - \frac{3.1N_x + 6.5}{N_x} \right] \left( \frac{\tau^{3/2}}{\tau^{3/2} + 5.2} \right)$</td>
</tr>
<tr>
<td>Flicker FM</td>
<td>$edf = \left( \frac{2N_x^2 - 1.3N_x\tau - 3.5\tau}{N_x\tau} \right) \left( \frac{\tau^3}{\tau^3 + 2.3} \right)$</td>
</tr>
<tr>
<td>Random Walk FM</td>
<td>$edf = \left( \frac{4.4N_x - 2}{2.9\tau} \right) \left( \frac{(4.4N_x - 1)^2 - 8.6\tau(4.4N_x - 1) + 11.4\tau^2}{(4.4N_x - 3)^2} \right)$</td>
</tr>
</tbody>
</table>
References for Confidence Intervals


5.5. Noise Identification

Identification of the dominant power law noise type is often necessary for setting confidence intervals and making bias corrections during a frequency stability analysis. The most effective means for power noise identification are based on the $B_1$ and $R(n)$ functions and the lag 1 autocorrelation.

5.5.1. Power Law Noise Identification

It is often necessary to identify the dominant power law noise process (WPM, FPM, WFM, FFM, RWFM, FWFM or RRFM) of the spectral density of the fractional frequency fluctuations, $S_x(f) = \text{const} f^{-\alpha}$ ($\alpha = 2$ to $-4$), to perform a frequency stability analysis. For example, knowledge of the noise type is necessary to determine the equivalent number of chi-squared degrees of freedom (edf) for setting confidence intervals and error bars, and it is essential to know the dominant noise type to correct for bias in the newer Total and Théol
variances. While the noise type may be known a priori or estimated manually, it is desirable to have an analytic method for power law noise identification that can be used automatically as part of a stability analysis algorithm.

There is little literature on the subject of power-law noise identification. The most common method for power law noise identification is simply to observe the slope of a log-log plot of the Allan or modified Allan deviation versus averaging time, either manually or by fitting a line to it. This obviously requires at least two stability points. During a stability calculation, it is desirable (or necessary) to automatically identify the power law noise type at each point, particularly if bias corrections and/or error bars must be applied.

5.5.2. Noise Identification Using $B_1$ and $R(n)$

A noise identification algorithm that has been found effective in actual practice, and that works for a single $\tau$ point over the full range of $-4 \leq \alpha \leq 2$ is based on the Barnes $B_1$ function, which is the ratio of the N-sample (standard) variance to the two-sample (Allan) variance, and the $R(n)$ function [3], which is the ratio of the modified Allan to the normal Allan variances. The $B_1$ function has as arguments the number of frequency data points, $N$, the dead time ratio, $r$ (which is set to 1), and the power law t-domain exponent, $\mu$. The $B_1$ dependence on $\mu$ is used to determine the power law noise type for $-2 \leq \mu \leq 2$ (W and F PM to FW FM). For a $B_1$ corresponding to $\mu = -2$, the $\alpha = 1$ or 2 (F PM or W PM noise) ambiguity can be resolved with the $R(n)$ ratio using the modified Allan variance. For the Hadamard variance, for which RR FM noise can apply, ($m = 3, \alpha = -4$), the $B_1$ ratio can be applied to frequency (rather than phase) data, and adding 2 to the resulting $\mu$.

The overall noise $B_1/R(n)$ noise identification process is therefore:

1. Calculate the standard and Allan variances for the applicable $\tau$ averaging factor.
2. Calculate $B_1(N, r=1, \mu) = \frac{N(1 - N^\alpha)}{2(N - 1)(1 - 2^\alpha)}$.
3. Determine the expected $B_1$ ratios for $\alpha = -3$ through 1 or 2.
4. Set boundaries between them and find the best power law noise match.
5. Resolve an $\alpha = 1$ or 2 ambiguity with the modified Allan variance and $R(n)$.
6. Resolve an $\alpha = -3$ or $-4$ ambiguity by applying $B_1$ to frequency data.

The boundaries between the noise types are generally set as the geometric means of their expected values. This method cannot distinguish between W and F PM at unity averaging factor.

5.5.3. The Autocorrelation Function

The autocorrelation function (ACF) is a fundamental way to describe a time series by multiplying it by a delayed version of itself, thereby showing the degree by which its value at one time is similar to its value at a certain later time. More specifically, the autocorrelation at lag $k$ is defined as
\[ \rho_k = \frac{E[(z_t - \mu)(z_{t+k} - \mu)]}{\sigma_z^2} \]

where \( z_t \) is the time series, \( \mu \) is its mean value, \( \sigma_z^2 \) is its variance, and \( E \) denotes the expected value. The autocorrelation is usually estimated by the expression

\[ r_k = \frac{1}{N} \sum_{t=1}^{N-k} (z_t - \bar{z})(z_{t+k} - \bar{z}) \]

\[ \frac{1}{N} \sum_{t=1}^{N} (z_t - \bar{z})^2 \]

where \( \bar{z} \) is the mean value of the time series and \( N \) is the number of data points [4].

### 5.5.4. The Lag 1 Autocorrelation

The lag 1 autocorrelation is simply the value of \( r_1 \) as given by the expression above. For frequency data, the lag 1 autocorrelation is able to easily identify white and flicker PM noise, and white (uncorrelated) FM noise, for which the expected values are \(-1/2\), \(-1/3\) and zero, respectively. The more divergent noises have positive \( r_1 \) values that depend on the number of samples, and tend to be larger (approaching 1). For those more divergent noises, the data are differenced until they become stationary, and the same criteria as for WPM, FPM and WFM are then used, corrected for the differencing. The results can be rounded to determine the dominant noise type or used directly to estimate the noise mixture.

### 5.5.5. Noise Identification Using \( r_1 \)

An effective method for identifying power law noises using the lag 1 autocorrelation [5] is based on the properties of discrete-time fractionally integrated noises having spectral densities of the form \((2 \sin \pi f)^{-2\delta}\). For \( \delta < \frac{1}{2} \), the process is stationary and has a lag 1 autocorrelation equal to \( \rho_1 = \delta / (1-\delta) \) [6], and the noise type can therefore be estimated from \( \delta = r_1 / (1+r_1) \). For frequency data, white PM noise has \( \rho_1 = -1/2 \), flicker PM noise has \( \rho_1 = -1/3 \), and white FM noise has \( \rho_1 = 0 \). For the more divergent noises, first differences of the data are taken until a stationary process is obtained as determined by the criterion \( \delta < 0.25 \). The noise identification method therefore uses \( p = \text{round}(2\delta) - 2d \), where round \((2\delta)\) is \( 2\delta \) rounded to the nearest integer and \( d \) is the number of times that the data is differenced to bring \( \delta \) down to \( < 0.25 \). If \( z \) is a \( \tau \)-average of frequency data \( y(t) \), then \( \alpha = p \); if \( z \) is a \( \tau \)-sample of phase data \( x(t) \), then \( \alpha = p + 2 \), where \( \alpha \) is the usual power law exponent \( f^{-\alpha} \), thereby determining the noise type at that averaging time. The properties of this power law noise identification method are summarized in the table below. It has excellent discrimination for all common power law noises for both phase and frequency data, including difficult cases with mixed noises.
### Lag 1 Autocorrelation for Various Power Law Noises

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>α</th>
<th>Phase Data*</th>
<th>d=0 ACF of Phase Data</th>
</tr>
</thead>
<tbody>
<tr>
<td>W PM</td>
<td>2</td>
<td><img src="image1" alt="Graph" /></td>
<td><img src="image2" alt="Graph" /></td>
</tr>
<tr>
<td>F PM</td>
<td>1</td>
<td><img src="image3" alt="Graph" /></td>
<td><img src="image4" alt="Graph" /></td>
</tr>
<tr>
<td>W FM</td>
<td>0</td>
<td><img src="image5" alt="Graph" /></td>
<td><img src="image6" alt="Graph" /></td>
</tr>
<tr>
<td>F FM</td>
<td>-1</td>
<td><img src="image7" alt="Graph" /></td>
<td><img src="image8" alt="Graph" /></td>
</tr>
<tr>
<td>RW FM</td>
<td>-2</td>
<td><img src="image9" alt="Graph" /></td>
<td><img src="image10" alt="Graph" /></td>
</tr>
</tbody>
</table>

* The differencing operation changes the appearance of the phase data to that shown 2 rows higher.

### Lag 1 Autocorrelation for Various Power Law Noises and Differences

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>α</th>
<th>Lag 1 Autocorrelation, r₁†</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>d=0</td>
<td>d=1</td>
</tr>
<tr>
<td></td>
<td>x(t)</td>
<td>y(t)</td>
</tr>
<tr>
<td>W PM</td>
<td></td>
<td></td>
</tr>
<tr>
<td>F PM</td>
<td></td>
<td></td>
</tr>
<tr>
<td>W FM</td>
<td></td>
<td></td>
</tr>
<tr>
<td>F FM</td>
<td></td>
<td></td>
</tr>
<tr>
<td>RW FM</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

† Shaded values are those used for noise ID for the particular noise and data type.
5.5.6. Noise ID Algorithm

The basic lag 1 autocorrelation power law noise identification algorithm is quite simple. The inputs are a vector \( z_1, \ldots, z_N \) of phase or frequency data, the minimum order of differencing \( d_{\text{min}} \) (default = 0), and the maximum order of differencing \( d_{\text{max}} \). The output is \( p \), an estimate of the \( \alpha \) of the dominant power law noise type, and (optionally) the value of \( d \).

```
Done = False, d = 0
While Not Done
    \( \bar{z} = \frac{1}{N} \sum_{i=1}^{N} z_i \)
    \( r_1 = \frac{\sum_{i=1}^{N-1} (z_i - \bar{z})(z_{i+1} - \bar{z})}{\sum_{i=1}^{N} (z_i - \bar{z})^2} \)
    \( \delta = \frac{r_1}{1 + r_1} \)
    If \( d >= d_{\text{min}} \) And (\( \delta < 0.25 \) Or \( d >= d_{\text{max}} \))
        \( p = -2(\delta + d) \)
        Done = True
    Else
        \( z_1 = z_2 - z_1, \ldots, z_{N-1} = z_N - z_{N-1} \)
        \( N = N - 1 \)
        \( d = d + 1 \)
    End If
End While
Note: May round \( p \) to nearest integer
```

The input data should be for the particular averaging time, \( \tau \), of interest, and it may therefore be necessary to decimate the phase data or average the frequency by the appropriate averaging factor before applying the noise identification algorithm. The \( d_{\text{max}} \) parameter should be set to 2 or 3 for an Allan or Hadamard (2 or 3-sample) variance analysis, respectively. The alpha result is equal to \( p+2 \) or \( p \) for phase or frequency data, respectively, and may be rounded to an integer (although the fractional part is useful for estimated mixed noises). The algorithm is fast, requiring only the calculation of one autocorrelation value and first differences for several times. It is independent of any particular variance. The lag 1 autocorrelation method yields good results, consistently identifying pure power noise for \( \alpha = 2 \) to \(-4 \) for sample sizes of about 30 or more, and generally identifying the dominant type of mixed noises when it is at least 10% larger than the others. For a mixture of adjacent noises, the fractional result provides an indication of their ratio. It can handle all averaging factors.
Before analysis, the data should be preprocessed to remove outliers, discontinuities, and deterministic components. Acceptable results can be obtained from the lag 1 autocorrelation noise identification method for \( N \geq 32 \), where \( N \) is the number of data points. The algorithm tends to produce jumps in the estimated alpha for mixed noises when the differencing factor, \( d \), changes (although the alpha value when rounded to an integer is still consistent). This can be avoided by using the same \( d \) for the entire range of averaging times, at the expense of higher variability when a lower \( d \) would have been sufficient. The lag 1 autocorrelation method for power law noise identification is a fast and effective way to support the setting of confidence intervals and to apply bias corrections during a frequency stability analysis, as shown in the example below:

\[
\begin{align*}
\text{Overlapping Allan Deviation, } \sigma_a(\tau), & \quad \text{seconds} \\
\text{Linear Frequency Drift Removed} & \\
\text{Noise Type} & \\
\text{WPM} & \\
\text{FPM} & \\
\text{WFM} & \\
\text{FFM} & \\
\text{RWFM} & \\
\text{FWFM} \end{align*}
\]

Frequency Stability and Noise Analysis of Two Hydrogen Masers

References for Noise Identification

5.6. Bias Functions

Several bias functions are defined and used in the analysis of frequency stability, as defined below. In particular, $B_1$, the ratio of the standard variance to the Allan variance, and $R(n)$, the ratio of the modified Allan variance to the normal Allan variance, are used for the identification of power law noise types (see section 5.2.2), and the $B_2$ and $B_3$ bias functions are used to correct for dead time in a frequency stability measurement.

5.7. $B_1$ Bias Function

The $B_1$ bias function is the ratio of the N-sample (standard) variance to the 2-sample (Allan) variance with dead time ratio $r = T/\tau$, where $T =$ time between measurements, $\tau =$ averaging time, and $\mu =$ exponent of $\tau$ in Allan variance for a certain power law noise process:

$$B_1(N, r, \mu) = \frac{\sigma^2(N, T, \tau)}{\sigma^2(2, T, \tau)}.$$

The $B_1$ bias function is useful for performing power law noise identification by comparing its actual value to those expected for the various noise types (see section 5.2.2).

5.8. $B_2$ Bias Function

The $B_2$ bias function is the ratio of the 2-sample (Allan) variance with dead time ratio $r = T/\tau$ to the 2-sample (Allan) variance without dead time ($r = 1$):

$$B_2(r, \mu) = \frac{\sigma^2(2, T, \tau)}{\sigma^2(2, \tau, \tau)}.$$

5.9. $B_3$ Bias Function

The $B_3$ bias function is the ratio of the N-sample (standard) variance with dead time ratio $r = T/\tau$ at multiples $M = \tau/\tau_0$ of the basic averaging time $\tau_0$ to the N-sample variance with the same dead time ratio at averaging time $\tau$:

$$B_3(N, M, r, \mu) = \frac{\sigma^2(N, M, T, \tau)}{\sigma^2(N, T, \tau)}.$$

The product of the $B_2$ and $B_3$ bias functions is used for dead time correction, as discussed in section 5.7.

5.10. $R(n)$ Bias Function

The $R(n)$ function is the ratio of the modified Allan variance to the normal Allan variance for $n =$ number of phase data points. Note: $R(n)$ is also a function of $\alpha$, the exponent of the power law noise type:

$$R(n) = \text{Mod} \frac{\sigma^2_y(\tau)}{\sigma^2_y(\tau)}$$

The $R(n)$ bias function is useful for performing power law noise identification by comparing its actual value to those expected for the various noise types (see section 5.2.2).
5.11. TVAR Bias Function

The TOTVAR statistic is an unbiased estimator of the Allan variance for white and flicker PM noise, and for white FM noise. For flicker and random walk FM noise, TOTVAR is biased low as $\tau$ becomes significant compared with the record length. The ratio of the expected value of TOTVAR to AVAR is given by the expression

$$B(\text{TOTAL}) = 1 - a \cdot (\tau/T), \quad 0 < \tau \leq T/2,$$

where $a = 1/3 \cdot \ln 2 = 0.481$ for flicker FM noise, $a = 3/4 = 0.750$ for random walk FM noise, and $T$ is the record length. At the maximum allowable value of $\tau = T/2$, TOTVAR is biased low by about 24% for RW FM noise. This bias function should be used to correct all reported TOTVAR results.

5.12. MTOT Bias Function

The MTOT statistic is a biased estimator of the modified Allan variance. The MTOT bias factor (the ratio of the expected value of Mod Totvar to MVAR) depends on the noise type but is essentially independent of the averaging factor and number of data points, as shown in the following table:

<table>
<thead>
<tr>
<th>Noise</th>
<th>Bias Factor</th>
</tr>
</thead>
<tbody>
<tr>
<td>W PM</td>
<td>1.06</td>
</tr>
<tr>
<td>F PM</td>
<td>1.17</td>
</tr>
<tr>
<td>W FM</td>
<td>1.27</td>
</tr>
<tr>
<td>F FM</td>
<td>1.30</td>
</tr>
<tr>
<td>RW FM</td>
<td>1.31</td>
</tr>
</tbody>
</table>

This bias factor should be used to correct all reported MTOT results.

5.13. Thêo1 Bias

The Thêo1 statistic is a biased estimator of the Allan variance. The Thêo1 bias factor (the ratio of the expected value of Thêo1 to AVAR) depends on both noise type and averaging factor:

$$\text{Thêo1 Bias} = \frac{\text{AVAR}}{\text{Thêo1}} = a + b/m^c,$$

where $m$ is the averaging factor and the constants $a$, $b$ and $c$ are given in the table below. Note that the effective tau for a Thêo1 estimation is $t = 0.75 \cdot m \cdot t_0$, where $t_0$ is the measurement interval.

<table>
<thead>
<tr>
<th>Noise</th>
<th>Alpha</th>
<th>a</th>
<th>b</th>
<th>c</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>RW FM</td>
<td>-2</td>
<td>2.70</td>
<td>-1.53</td>
</tr>
<tr>
<td></td>
<td>F FM</td>
<td>-1</td>
<td>1.87</td>
<td>-1.05</td>
</tr>
<tr>
<td>W FM</td>
<td>0</td>
<td>1.00</td>
<td>0.00</td>
<td>0.00</td>
</tr>
<tr>
<td>F PM</td>
<td>1</td>
<td>0.14</td>
<td>0.82</td>
<td>0.30</td>
</tr>
<tr>
<td>W PM</td>
<td>2</td>
<td>0.09</td>
<td>0.74</td>
<td>0.40</td>
</tr>
</tbody>
</table>
5.14. Dead Time

Dead time can occur in frequency measurements because of instrumentation delay between successive measurements, or because of a deliberate wait between measurements. It can have a significant effect on the results of a stability analysis, especially for the case of large dead time (e.g., frequency data taken for 100 seconds, once per hour).

Dead time can occur in frequency measurements and can significantly affect a subsequent stability analysis. Methods are available to correct for dead time and thus obtain unbiased results.

References for Bias Functions

Dead time corrections can be applied by dividing the calculated Allan deviation by the square root of the product of the Barnes $B_2$ and $B_3$ bias ratios. These corrections are particularly important for non-white FM noise with a large dead time ratio. Restricting the dead time corrections to Allan deviations is a conservative approach based on the $B_2$ and $B_3$ definitions. Those bias functions depend critically on the power law noise type. Requiring manual noise selection avoids the problem of noise identification for biased data having the wrong sigma-tau slope. Dead time correction is problematic for data having multiple noise types. In addition to introducing bias, measurement dead time reduces the confidence in the results, lowers the maximum allowable averaging factor, and prevents proper conversion of frequency to phase. Moreover, no information is available about the behavior of the device under test during the dead time. It is recommended that these issues be avoided by making measurements with zero dead time.

Dead time that occurs at the end of a measurement can be adjusted for in an Allan deviation determination by using the Barnes $B_2$ bias function [1], the ratio of the two-sample variance with dead time ratio $r = T/\tau$ to the two-sample variance without dead time. Otherwise, without this correction, one can determine only the average frequency and its drift. When such data are used to form frequency averages at longer tau, it is necessary to also use the $B_3$ bias function [2], the ratio of the variance with distributed dead time to the variance with all the dead time at the end. Those bias corrections are made by use of the product of $B_2$ and $B_3$. The power law noise type must be known in order to calculate these bias functions. Simulated periodically sampled frequency data with distributed dead time for various power law noise processes shows good agreement with the $B_2$ and $B_3$ bias function corrections, as shown in the figures below.

**Frequency Stability Plots for Common Power Law Noises with Large Measurement Dead Time ($r = T/\tau = 36$)**

Simulated Data Sampled for $\tau = 100$ Seconds Once Per Hour for 10 Days
Nominal $1 \times 10^{-11}$ Stability at $\tau = 100$ Seconds Shown by Lines
Plots Show Stability of Simulated Data Sets for Continuous, Sampled and Dead Time-Corrected Data

![Frequency Stability Plots](image)

White PM ($\mu = -2$) $\sqrt{B_2} = 0.82$ at AF = 1
Flicker PM ($\mu = -2$) $\sqrt{B_2} = 0.82$ at AF = 1

White FM ($\mu = -1$) $\sqrt{B_2} = 1.00$ at AF = 1
These simulations show that the $B_2$ and $B_3$ bias corrections are able to support reasonably accurate results for sampled frequency stability data having a large dead time, when the power law noise type is known. The slope of the raw sampled stability plot does not readily identify the noise type, however, and mixed noise types would make correction difficult. The relatively small number of data points reduces the confidence of the results, and limits the allowable averaging factor. Moreover, the infrequent measurements provide no information about the behavior of the clock during the dead time, and prevent a proper conversion of frequency to phase. Sparsely sampled data are therefore not recommended for the purpose of stability analysis.
References for Dead Time


5.15. Unevenly Spaced Data

Unevenly spaced phase data can be handled if they have associated timetags by using the individual timetag spacing when converting them to frequency data. Then, if the tau differences are reasonably small, the data may be analyzed by use of the average timetag spacing as the analysis tau, in effect placing the frequency data on an average uniform grid. While completely random data spacing is not amenable to this process, tau variations of ±10% will yield reasonable results as long as the exact interval is used for phase to frequency conversion.

An example of unevenly spaced data is two-way satellite time & frequency transfer (TWSTFT) measurements made on Monday, Wednesday, and Friday of each week, where the data spacing is either one or two days.

The TWSTFT data are simulated as 256 points of white PM noise with an Allan deviation (ADEV) level of $\sigma_\tau(\tau) = 1 \times 10^{-11}$ at 1-day. A composite plot of the TWSTFT TDEV results is shown above. The corresponding TDEV is $5.77 \times 10^{-12}$ sec at $\tau=1$ day (TDEV = MDEV).
divided by $\sqrt{3}$), as shown in curve A. Note that these time stability plots include points at all possible tau values. The green line shows that the $-0.5$ slope of the TDEV plot for W PM noise. The TWSTFT data are sampled once on Monday, Wednesday and Friday of each week. These sampled data therefore have an average tau of $7/3 = 2.33$ days, and their TDEV is shown in curve B. If the missing points are replaced by linearly interpolated phase values, the TDEV becomes highly distorted, as shown in curve C. If the sampled phase data are converted to frequency data using their timetag differences to determine the individual measurement intervals, the average tau, $\tau_{avg}$, is close to 2.33 days (depending on the final fractional week that is included), and the resulting TDEV is shown in curve D. It is essentially identical to that for the sampled phase data shown in curve B. It is interesting that, although the converted frequency results are different depending on whether the average or individual (more correct) taus are used, the (integrated) TDEV results are not (at least for a white PM noise process).

None of the results is in good agreement with the nominal simulation. The result with the linearly interpolated phase points is particularly bad for $\tau<\tau_{avg}$, and is similar to that of Tavella and Leonardi, as shown in Figure 1 of Reference [1]. As they point out in that paper, because the true sampling interval is $\tau_{avg}$, it is not possible to estimate the noise at shorter times, especially for an uncorrelated white noise process. They further suggest that the higher level of the estimated noise is related to the ratio of the true and interpolated sampling times ($\approx 2.33$) and the $\sqrt{\tau}$ dependence of TDEV. By applying a correction factor of $\sqrt{2.33} \approx 1.5$, the longer-tau TDEV estimates are lowered to the correct level. These factors are smaller for other non-white PM and FM noise processes. The adjusted method of using frequency data converted from phase data by using individual tau values adjusted for the timetag spacing is recommended because it does not use interpolation, does not present results at unrealistically low tau, and uses the best frequency estimates.

Another situation is data that are taken in bursts. In that case, the best approach is probably to analyze the segments separately, perhaps averaging those results to obtain better statistical confidence. One could obtain reasonable results for the shorter averaging times, but cannot apply standard techniques to analyze the complete data set.

### References for Unevenly Spaced Data


### 5.16. Histograms

A histogram shows the amplitude distribution of the phase or frequency fluctuations, and can provide insight regarding them. One can expect a normal (Gaussian) distribution for a reasonably sized data set, and a different (e.g., bimodal) distribution can be a sign of a problem.

For a normal distribution, the standard deviation is approximately equal to the half-width at half-height ($\text{HWHA} = 1.177s$).
An example of a histogram for a set of white FM noise is shown below:
5.17. Frequency Offset

It is often necessary to estimate the frequency offset from either phase or frequency data. Frequency offset is usually calculated from phase data by either of three methods:

1. A least squares linear fit to the phase data (optimum for white PM noise):
   \[ x(t) = a + bt, \text{ where slope } = y(t) = b. \]

2. The average of the first differences of the phase data (optimum for white FM noise):
   \[ y(t) = \text{slope} = \frac{x(t+\tau) - x(t)}{\tau}. \]

3. The difference between the first and last points of the phase data:
   \[ y(t) = \text{slope} = \frac{x(\text{end}) - x(\text{start})}{M-1}, \text{ where } M = \# \text{ phase data points}. \]
   This method is used mainly to match the two endpoints.

5.18. Frequency Drift

Most frequency sources have frequency drift, and it is often necessary (and usually advisable) to characterize and remove this systematic drift before analyzing the stochastic noise of the source. The term drift refers to the systematic change in frequency due to all effects, while aging includes only those effects internal to the device. Frequency drift is generally analyzed by fitting the trend of the frequency record to an appropriate mathematical model (e.g., linear, log, etc.), often by the method of least squares. The model may be based on some physical basis or simply a convenient equation, using either phase or frequency data, and its suitability may be judged by the degree to which it produces white (i.e., uncorrelated) residuals.

Frequency drift is the systematic change in frequency due to all effects, while frequency aging is the change in frequency due to effects within the device. Thus, for a quartz crystal oscillator, aging refers to a change in the resonant frequency of its quartz crystal resonator, while drift would also include the effects of its external environment. Therefore, drift is the quantity that is usually measured, but it is generally done under constant environmental conditions to the greatest extent possible so as to approximate the aging of the device.

5.19. Drift Analysis Methods

Several drift methods are useful for phase or frequency data as described below. The best method depends on the quality of the fit, which can be judged by the randomness of the residuals.
### 5.20. Phase Drift Analysis

Three methods are commonly used to analyze frequency drift in phase data:

1. A least squares quadratic fit to the phase data:

   \[ x(t) = a + bt + ct^2, \text{ where } y(t) = x'(t) = b + 2ct, \text{ slope } = y'(t) = 2c. \]

   This continuous model can be expressed as
   \[
   x_n = a + \tau_0 bn + \tau_0^2 cn^2 \quad \text{for } n = 1, 2, 3, ..., N, \]
   and \( \tau_0 \) is the sampling interval for discrete data, where the \( a, b \) and \( c \) coefficients have units of sec, sec/sec and sec/sec², respectively, and the frequency drift slope and intercept are \( 2c \) and \( b \), respectively. The fit coefficients can be estimated by the following expressions [1]:

   \[
   \hat{a} = \left( \frac{\sum_{n=1}^{N} x_n + \tau_0 B \sum_{n=1}^{N} nx_n + \tau_0^2 C \sum_{n=1}^{N} n^2 x_n}{G} \right)
   \]

   \[
   \hat{b} = \left( \frac{B \sum_{n=1}^{N} x_n + \tau_0 D \sum_{n=1}^{N} nx_n + \tau_0^2 E \sum_{n=1}^{N} n^2 x_n}{G \tau_0} \right)
   \]

   \[
   \hat{c} = \left( \frac{C \sum_{n=1}^{N} x_n + \tau_0 E \sum_{n=1}^{N} nx_n + \tau_0^2 F \sum_{n=1}^{N} n^2 x_n}{G \tau_0^2} \right)
   \]

   where the \( A-F \) terms are as follows
\[ A = 3\left[3N(N + 1) + 2\right] \]
\[ B = -18(2N + 1) \]
\[ C = 30 \]
\[ D = 12(2N + 1)(8N + 11) / \left[(N + 1)(N + 2)\right] \]
\[ E = \frac{-180}{(N + 2)} \]
\[ F = \frac{180}{\left[(N + 1)(N + 2)\right]} \]
\[ G = N(N - 1)(N - 2) \]

A quadratic fit to the phase data is the optimum model for white PM noise.

2. The average of the second differences of the phase data:
\[ y(t) = \frac{[x(t+\tau) - x(t)]}{\tau}, \text{ slope } = \frac{[y(t+\tau) - y(t)]}{\tau} = \frac{[x(t+2\tau) - 2x(t+\tau) + x(t)]}{\tau^2}. \]

This method is optimum for random walk FM noise.

3. A three-point fit at the start, middle and end of the phase data:
\[ \text{slope } = 4\left[x(\text{end}) - 2x(\text{mid}) + x(\text{start})\right]/(M\tau)^2, \text{ where } M = \# \text{ data points}. \]

It is the equivalent of the bisection method for frequency data.

### 5.21. Frequency Drift Analysis

Four methods are commonly used to analyze frequency drift in frequency data:

1. A least squares linear regression to the frequency data:
\[ y(t) = a + bt, \text{ where } a = \text{ intercept, } b = \text{ slope } = y'(t). \]

Linear frequency drift can be estimated by a linear least squares fit to the frequency data, \( y(t) = a + bt \). That continuous model can be expressed as \( y_n = a + \tau_0bn \) for \( n = 1, 2, 3, ..., M \) where \( M \) is the number of frequency data points, \( \tau_0 \) is the sampling interval for the discrete data. The frequency drift intercept and slope are \( a \) and \( b \), and have units of sec and sec/sec, respectively. The fit coefficients can be estimated by the following expressions:
A linear fit to the frequency data is the optimum model for white FM noise.

2. The frequency averages over the first and last halves of the data:

\[ \text{slope} = 2 \left[ \frac{y(2\text{nd half}) - y(1\text{st half})}{(N\tau)} \right], \]

where \( N \) = number of points.

This bisection method is optimum for white and random walk FM noise.

3. A log model of the form (see MIL-O-55310B) that applies to frequency stabilization:

\[ y(t) = a \cdot \ln(bt + 1), \]

where slope = \( y'(t) = ab/(bt+1) \).

4. A diffusion (\( \sqrt{t} \)) model of the form

\[ y(t) = a + b(t+c)^{1/2}, \]

where slope = \( y'(t) = \frac{1}{2}b(t+c)^{-1/2} \).

References for Drift


5.22. All Tau

Stability calculations made at all possible tau values can provide an excellent indication of the variations in the results, and are a simple form of spectral analysis. In particular, cyclic variations are often the result of interference between the sampling rate and some periodic instability (such as environmental sensitivity). However, an all tau analysis is computationally intensive and can therefore be slow. For most purposes, however, it is not necessary to calculate values at every tau, but instead to do so at enough points to provide a nearly continuous curve on the display device (screen or paper). Such a “many tau” analysis can be orders of magnitude faster and yet provide the same information.
Environmental sensitivity should be treated separately from noise when performing a stability analysis. However, it can be very difficult to distinguish between those different mechanisms for phase or frequency variations. It is often possible to control the environmental conditions sufficiently well during a stability run so that environmental effects such as temperature and supply voltage are negligible. Determining how well those factors have to be controlled requires knowledge of the device’s environmental sensitivities. Those factors should be measured individually, if possible, over the largest reasonable
excursions to minimize the effect of noise. Environmental sensitivity can best be determined by considering the physical mechanisms that apply within the unit under test. Useful information about the environmental sensitivity of frequency sources can be found in the references below. Some environmental factors affect phase and frequency differently, which can cause confusion. For example, temperature affects the phase delay through a cable. Dynamically, however, a temperature ramp produces a rate of change of phase that mimics a frequency change in the source. Because environmental sensitivity is highly dependent on device and application, it does not receive detailed consideration in this handbook. More information will be found in the following references.

### References for Environmental Sensitivity


### 5.25. Parsimony

In any measurement or analysis, it is desirable to minimize the number of extraneous parameters. This is not just a matter of elegance; the additional parameters may be implicit or arbitrary and thereby cause confusion or error if they are ignored or misunderstood. For example, the Allan deviation has the important advantage that, because of its convergence characteristics for all common clock noises, its expected value is independent of the number of data points. Many of the techniques used in the analysis of frequency stability, however,
do require that certain parameters be chosen before they can be performed. For example, drift removal requires the choice of a model to which the data will be fit (perhaps using the criterion of white residuals). Outlier removal is an especially difficult case, where judgment often enters into the decision as to whether or not a datum is anomalous. A listing of some of these nonparsimonious parameters is given in the table below:

<table>
<thead>
<tr>
<th>Type</th>
<th>Process</th>
<th>Parameter</th>
<th>Criterion</th>
<th>Remarks</th>
</tr>
</thead>
<tbody>
<tr>
<td>Pre-processing</td>
<td>Outlier Removal</td>
<td>Number of sigmas</td>
<td>Apply best judgment</td>
<td>Use of MAD-based robust statistics is recommended.</td>
</tr>
<tr>
<td></td>
<td>Drift Removal</td>
<td>Remove or not</td>
<td>Is drift deterministic?</td>
<td>It is generally wise to remove deterministic drift before noise analysis.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Is its cause known?</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Model</td>
<td>White residuals</td>
<td>White residuals are sign that</td>
<td>Model may have physical basis (e.g., diffusion process)</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>model is appropriate.</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Convergence limit</td>
<td>For iterative fit.</td>
<td></td>
<td>Generally uncritical – can hide deeply in algorithm</td>
</tr>
<tr>
<td></td>
<td>Remove average frequency</td>
<td>Model</td>
<td>Noise type</td>
<td>Not necessarily simple arithmetic mean of frequency.</td>
</tr>
<tr>
<td></td>
<td>Phase to frequency conversion</td>
<td>Tau</td>
<td>Accuracy</td>
<td>Is measurement interval known exactly? Is it the same for each point?</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Tau</td>
<td>See above</td>
<td></td>
</tr>
<tr>
<td>Frequency to phase conversion</td>
<td>Initial phase</td>
<td>Set to zero.</td>
<td>Generally arbitrary. Doesn’t affect subsequent stability analysis.</td>
<td></td>
</tr>
<tr>
<td>Analysis</td>
<td>Drift Estimation</td>
<td>Model</td>
<td>Smallest residuals</td>
<td>Can be critical, especially for predictions. Known physics can help choose.</td>
</tr>
</tbody>
</table>

Table 1. Non-Parsimonious Parameters in Frequency Stability Analysis
<table>
<thead>
<tr>
<th>Section</th>
<th>Frequency Estimation</th>
<th>Gaps</th>
<th>Allan Deviation</th>
</tr>
</thead>
<tbody>
<tr>
<td>Convergence limit</td>
<td>Noise model</td>
<td>Skip, fill, omit, or exclude</td>
<td>Number of data points</td>
</tr>
<tr>
<td></td>
<td>Lowest residuals for known noise type</td>
<td>Number, distribution, noise type</td>
<td>Time available for measurement. Must remove outliers.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Dead time</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Property of measuring system</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Dead time</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Maximum AF</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Confidence degrades</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Noise ID method to support error bars</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Hadamard Deviation</td>
<td>All Allan deviation parameters</td>
<td>To use rather than Allan deviation or separate drift removal</td>
<td>To use rather than Allan deviation</td>
</tr>
<tr>
<td></td>
<td>See above</td>
<td>Easier handling of clock with drift or divergent noise</td>
<td>Better confidence at long tau.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Noise ID method to support bias removal</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Total Deviation Thêø1</td>
<td>All Allan deviation parameters</td>
<td>To use rather than Allan deviation</td>
<td>Noise ID method to support bias removal</td>
</tr>
<tr>
<td></td>
<td>See above</td>
<td>Better confidence at long tau.</td>
<td>Critical for these biased estimators</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Less commonly used</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Generally unambiguous</td>
</tr>
<tr>
<td>Dynamic Stability</td>
<td>Window and step size</td>
<td>Resolution, number of windows</td>
<td>Affects calc time</td>
</tr>
<tr>
<td></td>
<td>Variance type</td>
<td>Data properties: noise, drift</td>
<td>AVAR or HVAR</td>
</tr>
<tr>
<td></td>
<td>Viewpoint, mesh, color</td>
<td>Visibility</td>
<td>Personal preference</td>
</tr>
<tr>
<td>Spectral Analysis</td>
<td>Type – Parametric or non-parametric, Windowing – Bias reduction</td>
<td>Convention</td>
<td>Analysis tools available Knowledge of analyst</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Clarity</td>
<td>Uncritical</td>
</tr>
</tbody>
</table>
5.26. Transfer Functions

Variances can be related to the spectral density of the fractional frequency fluctuations by their transfer functions. For example, for the Hadamard variance, this relationship is

$$\sigma^2_H(\tau) = \int_0^{f_h} S_y(f) \cdot |H_H(f)|^2 \, df,$$

where $\sigma^2_H(\tau)$ is the three-sample, zero dead-time, binomially weighted Hadamard variance, $S_y(f)$ is the spectral density of the fractional frequency fluctuations, $H_H(f)$ is its transfer function, and $f_h$ is the upper cutoff frequency (determined by hardware factors). The transfer function is determined by the statistic’s time-domain sampling pattern.

The transfer functions for the most common variances used in frequency stability analysis are shown in the table below:

| Variance     | Magnitude Squared Transfer Function $|H(f)|^2$ |
|--------------|-----------------------------------|
| Allan        | $2 \left( \frac{\sin \pi \tau f}{\pi \tau f} \right)^2 \sin^2 \pi \tau f$ |
| Hadamard     | $2^4 \left( \frac{\sin \pi \tau f}{\pi \tau f} \right)^2 \sin^4 \pi \tau f$ |
For $\pi f \ll 1$, the transfer function of the Allan variance behaves as $(\pi f)^2$, indicating that it is convergent for power law processes $S_f^{\alpha}$ down to as low as $\alpha = -2$ (Random Walk FM), while the transfer function of the Hadamard variance behaves as $(\pi f)^4$, indicating that it is convergent for power law processes as low as $\alpha = -4$ (Random Run FM).

The squared magnitudes of these transfer functions are shown in the plots below:

These responses have their peaks where the frequency is one-half the sampling rate, and nulls where it is a multiple of the sampling rate (i.e., at $f = n/\tau$, where $n$ is an integer). As a
spectral estimator, the Hadamard variance has slightly higher resolution than the Allan variance, since the equivalent noise bandwidths of the Hadamard and Allan spectral windows are $0.411\tau^{-1}$ and $0.476\tau^{-1}$, respectively [5].

Similar transfer functions exist for the modified, total and Thêo1 variances.

### References for Transfer Functions

6 Frequency Domain Stability

Frequency stability can also be characterized in the frequency domain in terms of a power spectral density (PSD) that describes the intensity of the phase or frequency fluctuations as a function of Fourier frequency. Spectral stability measures are directly related to the underlying noise processes, and are particularly appropriate when the phase noise of the source is of interest.

6.1. Noise Spectra

The random phase and frequency fluctuations of a frequency source can be modeled by power law spectral densities of the form

\[ S_y(f) = h(\alpha)f^\alpha, \]

where:

- \( S_y(f) \) = one-sided power spectral density of \( y \), the fractional frequency fluctuations, 1/Hz
- \( f \) = Fourier or sideband frequency, Hz
- \( h(\alpha) \) = intensity coefficient
- \( \alpha \) = exponent of the power law noise process.

The most commonly encountered noise spectra are

- White (\( f^0 \))
- Flicker (\( f^{-1} \))
- Random Walk (\( f^{-2} \))
- Flicker Walk (\( f^{-3} \)).

Examples of these noise types are shown in the figure below.
Power law spectral models can be applied to both phase and frequency power spectral densities. Phase is the time integral of frequency, so the relationship between them varies as $1/f^2$:

$$S_\phi(f) = \frac{S_y(f)}{(2\pi f)^2},$$

where $S_\phi(f) = \text{PSD of the time fluctuations, sec}^2/\text{Hz}$.

Two other quantities are also commonly used to measure phase noise:

$$S_\phi(f) = \text{PSD of the phase fluctuations, rad}^2/\text{Hz and its logarithmic equivalent } \ell(f), \text{dBC/Hz.}$$

The relationship between these is

$$S_\phi(f) = (2\pi
_0)^2 \cdot S_x(f) = \left(\frac{\nu_0}{f}\right)^2 \cdot S_y(f)$$

and

$$\ell(f) = 10 \cdot \log\left[\frac{1}{2} \cdot S_\phi(f)\right],$$

where $\nu_0$ is the carrier frequency, hertz.

The power law exponent of the phase noise power spectral densities is $\beta = \alpha - 2$. These frequency-domain power law exponents are also related to the slopes of the following time domain stability measures:

- Allan variance $\sigma^2_y(\tau)$ $\mu = -(\alpha + 1), \alpha < 2$
- Modified Allan variance $\text{Mod } \sigma^2_y(\tau)$ $\mu' = -(\alpha + 1), \alpha < 3$
- Time variance $\sigma^2_x(\tau)$ $\eta = -(\alpha - 1), \alpha < 3$

The spectral characteristics of the power law noise processes commonly used to describe the performance of frequency sources are shown in the following table:

### Spectral Characteristics of Power Law Noise Processes

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>$\alpha$</th>
<th>$\beta$</th>
<th>$\mu$</th>
<th>$\mu'$</th>
<th>$\eta$</th>
</tr>
</thead>
<tbody>
<tr>
<td>White PM</td>
<td>2</td>
<td>0</td>
<td>-2</td>
<td>-3</td>
<td>-1</td>
</tr>
<tr>
<td>Flicker PM</td>
<td>1</td>
<td>-1</td>
<td>-2</td>
<td>-2</td>
<td>0</td>
</tr>
<tr>
<td>White FM</td>
<td>0</td>
<td>-2</td>
<td>-1</td>
<td>-1</td>
<td>1</td>
</tr>
<tr>
<td>Flicker FM</td>
<td>-1</td>
<td>-3</td>
<td>0</td>
<td>0</td>
<td>2</td>
</tr>
<tr>
<td>Random Walk FM</td>
<td>-2</td>
<td>-4</td>
<td>1</td>
<td>1</td>
<td>3</td>
</tr>
</tbody>
</table>

### 6.2. Power Spectral Densities

Four types of power spectral density are commonly used to describe the stability of a frequency source:
PSD of Frequency Fluctuations $S_y(f)$

The power spectral density of the fractional frequency fluctuations $y(t)$ in units of $1/\text{Hz}$ is given by $S_y(f) = h(\alpha) \cdot f^\alpha$, where $f = \text{sideband frequency, Hz}$.

PSD of Phase Fluctuations $S_\phi(f)$

The power spectral density of the phase fluctuations in units of $\text{rad}^2/\text{Hz}$ is given by $S_\phi(f) = (2\pi\nu_0)^2 \cdot S_y(f)$, where $\nu_0 = \text{carrier frequency, Hz}$.

PSD of Time Fluctuations $S_x(f)$

The power spectral density of the time fluctuations $x(t)$ in units of $\text{sec}^2/\text{Hz}$ is given by $S_x(f) = h(\beta) \cdot f^\beta = S_y(f)/(2\pi f)^2$, where: $\beta = \alpha - 2$. The time fluctuations are related to the phase fluctuations by $x(t) = \phi(t)/(2\pi\nu_0)$. Both are commonly called "phase" to distinguish them from the independent time variable, $t$.

SSB Phase Noise $\mathcal{E}(f)$

The SSB phase noise in units of $\text{dBc/Hz}$ is given by $\mathcal{E}(f) = 10 \cdot \log\left[\frac{1}{2} \cdot S_\phi(f)\right]$. This is the most common function used to specify phase noise.

6.3. Phase Noise Plot

The following diagram shows the slope of the SSB phase noise, $\mathcal{E}(f)$, $\text{dBc/Hz}$ versus log $f$, Fourier frequency, Hz for various power law noise processes.

![SSB Phase Noise Diagram](image)

Figure 6.3 SSB Phase Noise Plot
6.4. Spectral Analysis

Spectral analysis is the process of characterizing the properties of a signal in the frequency domain, either as a power spectral density for noise, or as the amplitude and phase at discrete frequencies. Spectral analysis can thus be applied to both noise and discrete components for frequency stability analysis. For the former, spectral analysis complements statistical analysis in the time domain. For the latter, spectral analysis can aid in the identification of periodic components such as interference and environmental sensitivity. Time domain data can be used to perform spectral analysis via the Fast Fourier Transform (FFT), and there is much technical literature on that subject [2, 3]. While, in principle, time and frequency domain analyses provide equivalent information, in practice, one or the other is usually preferred, either because of measurement and/or analysis convenience, or because the results are more applicable to a particular application. Spectral analysis is most often used to characterize the short-term (< 1 second) fluctuations of a frequency source as a plot of phase noise power spectral density versus sideband frequency, while a time domain analysis is most often used to provide information about the statistics of its instability over longer intervals (> 1 second). Modern instrumentation is tending to merge these approaches by digitizing the signal waveform at high sampling rates, thereby allowing FFT analysis at relatively high Fourier frequencies. Nevertheless, there are many pitfalls and subtleties associated with spectral analysis that must be considered if meaningful results are to be obtained.

6.5. PSD Windowing

Data windowing is the process of applying a weighting function that falls off smoothly at the beginning and end to avoid spectral leakage in an FFT analysis. Without windowing, bias will be introduced that can severely restrict the dynamic range of the PSD result. The most common windowing types are Hanning, Hamming and Multitaper. The classic Hanning and Hamming windows can be applied more than one time.

6.6. PSD Averaging

Without filtering or averaging, the variances of the PSD results are always equal to their values regardless of the size of the time domain data set. More data provides finer frequency resolution, not lower noise (while the data sampling time determines the highest Fourier frequency). Without averaging, for white noise, each spectral result has only two degrees of freedom. Some sort of filtering or averaging is usually necessary to provide less noise in the PSD results. This can be accomplished by dividing the data into sections, performing an FFT analysis on each section separately, and then averaging those values to obtain the final PSD result. The averaging factor improves the PSD standard deviation by the square root of the averaging factor. The tradeoff in this averaging process is that each section of the data is shorter, yielding a result with coarser frequency resolution that does not extend to as low a Fourier frequency.

6.7. Multitaper PSD Analysis

The multitaper PSD analysis method offers a better compromise between bias, variance and spectral resolution. Averaging is accomplished by applying a set of orthogonal windowing (tapering) functions called discrete prolate spheroidal sequences (DPSS) or Slepian functions to the entire data array. An example of seven of these functions for order J=4 is shown below.
The 1st function resembles a classic window function, while the others sample other portions of the data. The higher windows have larger amplitude at the ends that compensates for the denser sampling at the center. These multiple tapering functions are defined by two parameters, the order of the function, J, which affects the resolution bandwidth, and the number of windows, which affects the variance. A higher J permits the use of more windows without introducing bias, which provides more averaging (lower variance) at the expense of lower spectral resolution, as shown in the following table:

<table>
<thead>
<tr>
<th>Order J</th>
<th># Windows</th>
</tr>
</thead>
<tbody>
<tr>
<td>2.0</td>
<td>1-3</td>
</tr>
<tr>
<td>2.5</td>
<td>1-4</td>
</tr>
<tr>
<td>3.0</td>
<td>1-5</td>
</tr>
<tr>
<td>3.5</td>
<td>1-6</td>
</tr>
<tr>
<td>4.0</td>
<td>1-7</td>
</tr>
<tr>
<td>4.5</td>
<td>1-8</td>
</tr>
<tr>
<td>5.0</td>
<td>1-9</td>
</tr>
</tbody>
</table>

The resolution BW is given by $2J/Nt$, where N is the number of data points sampled at time interval t. An adaptive algorithm can be used to weight the contributions of the individual tapers for lowest bias. The multitaper PSD has a flat-topped response for discrete spectral components that is nevertheless narrower than an averaged periodogram with the same variance. It is therefore particularly useful for examining discrete components along with noise.
6.8. PSD Notes

A carrier frequency parameter applies to the $S_d(f)$ and $S(f)$ PSD types. The number of Fourier frequency points is always the power of 2 greater than or equal to one-half of the number of time domain data points, $n$. The spacing between Fourier frequency points is $1/nt$, and the highest Fourier frequency is $1/2t$. If averaging is done, the value of $n$ is reduced by the averaging factor. The PSD fit is a least-squares power law line through octave-band PSD averages [6].

For characterizing frequency stability, a spectral analysis is used primarily for the analysis of noise (not discrete components), and should include the quantitative display of power law noise in common PSD units, perhaps with fits to integer power law noise processes. Amplitude corrections need to be made for the noise response of the windowing functions. The amplitude of discrete components should be increased by the log of the BW (Fourier frequency spacing in hertz), which is a negative number for typical sub-hertz bandwidths.

<table>
<thead>
<tr>
<th>References for Spectral Analysis</th>
</tr>
</thead>
</table>
7 Domain Conversions

The stability of a frequency source can be specified and measured in either the time or the frequency domain. Examples of these stability measures are the Allan variance, \( \sigma_y^2(\tau) \), in the time domain, and the spectral density of the fractional frequency fluctuations, \( S_y(f) \), in the frequency domain. Conversions between these domains may be made by numerical integration of their fundamental relationship, or by an approximation method based on a power law spectral model for the noise processes involved. The latter method can be applied only when the dominant noise process decreases toward higher sideband frequencies. Otherwise, the more fundamental method based on numerical integration must be used. The general conversion from time to frequency domain is not unique because white and flicker phase noise have the same Allan variance dependence on \( \tau \). When performing any of these conversions, it is necessary to choose a reasonable range for \( \sigma \) and \( \tau \) in the domain being converted to. The main lobe of the \( \sigma_y(\tau) \) and Mod \( \sigma_y(\tau) \) responses occur at the Fourier frequency \( f = 1/2\tau \).

Time domain frequency stability is related to the spectral density of the fractional frequency fluctuations by the relationship

\[
\sigma_y^2(M, T, \tau) = \int_0^T S_y(f) \cdot |H(f)|^2 \cdot df,
\]

where \( |H(f)|^2 \) is the transfer function of the time domain sampling function.

The transfer function of the Allan (two-sample) time domain stability is given by

\[
|H(f)|^2 = 2 \left[ \sin^4(\pi\tau f) / (\pi\tau f)^2 \right],
\]

and therefore the Allan variance can be found from the frequency domain by the expression

\[
\sigma_y^2(\tau) = 2 \int_0^{f_0} S_y(f) \frac{\sin^4(\pi\tau f)}{(\pi\tau f)^2} df.
\]

The equivalent expression for the modified Allan variance is

\[
\text{Mod} \sigma_y^2(\tau) = \frac{2}{N^4 \pi^2 \tau_0^2} \int_0^{f_0} \frac{S_y(f) \sin^6(\pi\tau f)}{f^2 \sin^2(\pi\tau_0 f)} df.
\]

The stability of a frequency source can be specified and measured in either the time or the frequency domain. One domain is often preferred to specify the stability because it is most closely related to the particular application. Similarly, stability measurements may be easier to perform in one domain than the other. Conversions are possible between these generally equivalent measures of frequency stability.
7.1. Power Law Domain Conversions

Domain conversions may be made for power law noise models by using the following equation and conversion formulae:

$$\sigma^2_y(\tau) = h_{-2} \left(\frac{(2\pi)^2}{6}\tau + h_{-2} \log e \tau + h_0 \frac{1}{2\tau} + h_1 \frac{1.038 + 3\log e \cdot 2\pi f_h \tau}{2\pi^2 \tau^2} + h_2 \frac{3f_h}{(2\pi)^3 \tau^3}\right)$$,

where the $h_i$ terms define the level of the various power law noises.

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>$\sigma^2_y(\tau)$</th>
<th>$S_y(f)$</th>
</tr>
</thead>
<tbody>
<tr>
<td>RW FM</td>
<td>$A \cdot f^2 \cdot S_y(f) \cdot \tau^{-1}$</td>
<td>$A^{-1} \cdot \tau^{-1} \cdot \sigma^2_y(\tau) \cdot f^{-2}$</td>
</tr>
<tr>
<td>F FM</td>
<td>$B \cdot f^0 \cdot S_y(f) \cdot \tau^0$</td>
<td>$B^{-1} \cdot \tau^0 \cdot \sigma^2_y(\tau) \cdot f^{-1}$</td>
</tr>
<tr>
<td>W FM</td>
<td>$C \cdot f^0 \cdot S_y(f) \cdot \tau^{-1}$</td>
<td>$C^{-1} \cdot \tau^{-1} \cdot \sigma^2_y(\tau) \cdot f^{-1}$</td>
</tr>
<tr>
<td>F PM</td>
<td>$D \cdot f^{-2} \cdot S_y(f) \cdot \tau^{-2}$</td>
<td>$D^{-1} \cdot \tau^{-2} \cdot \sigma^2_y(\tau) \cdot f^{-1}$</td>
</tr>
<tr>
<td>W PM</td>
<td>$E \cdot f^{-2} \cdot S_y(f) \cdot \tau^{-2}$</td>
<td>$E^{-1} \cdot \tau^{-2} \cdot \sigma^2_y(\tau) \cdot f^{-2}$</td>
</tr>
</tbody>
</table>

where

- $A = 4\pi^3/6$
- $B = 2\cdot\ln2$
- $C = 1/2$
- $D = 1.038 + 3\cdot\ln(2\pi f_h \tau_0)/4\pi^2$
- $E = 3f_h / 4\pi^2$

and $f_h$ is the upper cutoff frequency of the measuring system in hertz, and $\tau_0$ is the basic measurement time. This factor applies only to white and flicker PM noise. The above conversion formulae apply to the ThêoH hybrid statistic as well as to the Allan variance.

7.2. Example of Domain Conversions

This section shows an example of time and frequency domain conversions. First, a set of simulated power law noise data is generated, and the time domain properties of this noise are analyzed by use of the overlapping Allan deviation. Next, the same data are analyzed in the frequency domain with an $f(f)$ PSD. Then, a power law domain conversion is done, and those results are compared with those of the spectral analysis. Finally, the other power spectral density types are examined.

For this example, we generate 4097 phase data points of simulated white FM noise with a 1-second Allan deviation value $\sigma^2_y(1) = 1\times10^{-11}$ and a sampling interval $\tau = 1$ msec. The number of points is chosen as an even power of 2 for the frequency data. The generated set of simulated white FM noise is shown as frequency data in Figure 7.2a, and their overlapping Allan deviation is shown in Figure 7.2b. The $\sigma^2_y(1)$ white FM noise fit parameter is 1.08e-11, close to the desired value.
The power spectrum for the phase data is calculated by use of a 10 MHz carrier frequency and a $\xi(f)$ power spectral density type, the SSB phase noise to signal ratio in a 1 Hz BW as a function of sideband frequency, $f$, as shown in Figure 7.2c. The fit parameters show an $\xi(1)$ value of -79.2 dBc/Hz and a slope of -19.6 dB/decade, in close agreement with the expected values of -80 dBc/Hz and -20 dB/decade.

The expected PSD values that correspond to the time domain noise parameters used to generate the simulated power-law noise can be determined by the power law domain conversion formulae of section 7.1, as shown in the table below.
The other types of PSD that are commonly used for the analysis of frequency domain stability analysis are $S_\phi(f)$, the spectral density of the phase fluctuations in rad^2/Hz; $S_x(f)$, the spectral density of the time fluctuations in sec^2/Hz; and $S_y(f)$, the spectral density of the fractional frequency fluctuations in units of 1/Hz. The expected value of all these quantities for the simulated white FM noise parameters with $\sigma_y(1) = 1.00\times10^{-11}$, $\tau = 1.00\times10^{-3}$, and $f_o = 10$ MHz are shown in the following table.

<table>
<thead>
<tr>
<th>Type</th>
<th>dB/dec</th>
<th>PSD</th>
<th>Type</th>
<th>Mu</th>
<th>Sigma</th>
</tr>
</thead>
<tbody>
<tr>
<td>RWFM</td>
<td>-40</td>
<td>None</td>
<td>RWFM</td>
<td>+1</td>
<td>0.000000e+00</td>
</tr>
<tr>
<td>FFM</td>
<td>-30</td>
<td>None</td>
<td>FFM</td>
<td>0</td>
<td>0.000000e+00</td>
</tr>
<tr>
<td>WFM</td>
<td>-20</td>
<td>-80.0</td>
<td>WFM</td>
<td>-1</td>
<td>1.000000e-11</td>
</tr>
<tr>
<td>FPM</td>
<td>-10</td>
<td>None</td>
<td>FPM</td>
<td>-2</td>
<td>0.000000e+00</td>
</tr>
<tr>
<td>WPM</td>
<td>0</td>
<td>None</td>
<td>WPM</td>
<td>-2</td>
<td>0.000000e+00</td>
</tr>
<tr>
<td>All</td>
<td>-80.0</td>
<td>All</td>
<td>All</td>
<td>1</td>
<td>1.000000e-11</td>
</tr>
</tbody>
</table>

References for Domain Conversion

These spectral densities are shown in the plots below:

![Figure 7.2c](image1.png)

**Figure 7.2c** $\gamma(f)$ Power Spectral Density

![Figure 7.2d](image2.png)

**Figure 7.2d** $S_{\phi}(f)$ Power Spectral Density
Figure 7.2e  $S_x(f)$ Power Spectral Density

Figure 7.2f  $S_y(f)$ Power Spectral Density
8 Noise Simulation

It is valuable to have a means of generating simulated power law clock noise having the desired noise type (white phase, flicker phase, white frequency, flicker frequency and random walk frequency noise), Allan deviation, frequency offset, frequency drift, and perhaps a sinusoidal component. This can serve as both a simulation tool and as a way to validate stability analysis software, particularly for checking numerical precision, noise recognition, and modeling. A good method for power-law noise generation is described in Reference 8. The noise type and time series of a set of simulated phase data are shown below:

<table>
<thead>
<tr>
<th>Noise Type</th>
<th>Phase Data Plot</th>
</tr>
</thead>
<tbody>
<tr>
<td>Random Walk FM</td>
<td><img src="random_walk_flicker.png" alt="Random Walk FM" /></td>
</tr>
<tr>
<td>$\alpha = -2$</td>
<td></td>
</tr>
<tr>
<td>Random Run Noise</td>
<td><img src="random_run_noise.png" alt="Random Run Noise" /></td>
</tr>
<tr>
<td>Flicker FM</td>
<td><img src="flicker_freq.png" alt="Flicker FM" /></td>
</tr>
<tr>
<td>$\alpha = -1$</td>
<td></td>
</tr>
<tr>
<td>Flicker Walk Noise</td>
<td><img src="flicker_walk.png" alt="Flicker Walk Noise" /></td>
</tr>
<tr>
<td>White FM</td>
<td><img src="white_freq.png" alt="White FM" /></td>
</tr>
<tr>
<td>$\alpha = 0$</td>
<td></td>
</tr>
<tr>
<td>Random Walk Noise</td>
<td><img src="random_walk_white.png" alt="Random Walk Noise" /></td>
</tr>
<tr>
<td>Flicker PM</td>
<td><img src="flicker_phase.png" alt="Flicker PM" /></td>
</tr>
<tr>
<td>$\alpha = 1$</td>
<td></td>
</tr>
<tr>
<td>Flicker Noise</td>
<td><img src="flicker_phase.png" alt="Flicker Noise" /></td>
</tr>
</tbody>
</table>
8.1. White Noise Generation

White noise generation is straightforward. One popular technique is to first generate two independent uniformly distributed random sequences [1], and combine them using the Box-Muller transform [2], [3] to produce a white spectrum with Gaussian deviates. Another method is to generate 12 independent random sequences uniformly distributed between 0 and 1, add them, and subtract 6 [4]. This will via the central limit theorem produce a Gaussian distribution having zero mean and unit variance. White noise can be numerically integrated and differentiated to transform it by \(1/f^2\) and \(f^2\), respectively, to produce simulated noise having any even power law exponent.

8.2. Flicker Noise Generation

Flicker noise is more difficult to generate because it cannot be described exactly by a rational transfer function, and much effort has been devoted to generating it [5]-[9]. The most common methods involve linear filtering by RC ladder networks [5], or by FFT transformation [7]-[9]. The FFT method can produce noise having any integer power law exponent from \(\alpha = -2\) (RW FM) to \(\alpha = +2\) (W PM) [7], [8].

8.3. Flicker Walk and Random Run Noise Generation

The more divergent flicker walk FM (\(\alpha = -3\)) and random run FM (\(\alpha = -4\)) power law noise types may be generated by using the \(1/f^2\) spectral property of a frequency to phase conversion. For example, to generate RR FM noise, first generate a set of RW FM phase data. Then treat this RW FM phase data as frequency data, and convert it to a new set of RR FM phase data.

8.4. Frequency Offset, Drift, and Sinusoidal Components

Beside the generation of the desired power law noise, it is desirable to include selectable amounts of frequency offset, frequency drift, and a sinusoidal component in the simulated clock data.
References for Noise Simulation


**9 Measuring Systems**

Frequency measuring systems are instruments that accept two or more inputs, one of which may be considered to be the reference, and compare their relative phase or frequencies. These systems can take many forms, from the direct use of a frequency counter to elaborate low-noise, high-resolution multichannel clock measuring systems with associated archival databases. They can be custom built or bought from several organizations specializing in such systems. The most important attribute of a frequency measuring system is its resolution, which, for high performance devices, requires 1 picosecond/second ($10^{-12}$) or better resolution, and more elaborate hardware than a counter. The resolution of a digital frequency, period or time interval counter is determined mainly by its speed (clock rate) and the performance of its analog interpolator (if any). That resolution generally improves linearly with the averaging time of the measurement, and it can be enhanced by preceding it with a mixer that improves the resolution by the heterodyne factor, the ratio of the RF input to the IF beat frequencies. Noise is another important consideration for a high-performance measuring system whose useful resolution may be limited by its noise floor, the scatter in the data when the two inputs are driven coherently by the same source. The performance of the measuring system also depends on the stability of its reference source. A low noise ovenized quartz crystal oscillator may be the best choice for a reference in the short term (1-100 seconds), while a active hydrogen maser generally provides excellent stability at averaging times out to several days, and cesium beam tube devices at longer averaging times.

Three methods are commonly used for making precise time and frequency measurements, as described below:

**9.1. Time Interval Counter Method**

The time interval counter method divides the two sources being compared down to a much lower frequency (typically 1 pulse/second) and measures their time difference with a high resolution time interval counter:

![Figure 9.1 Block Diagram of a Time Interval Counter Measuring System](image)

This measurement method is made practical by modern high-resolution interpolating time interval counters that offer 10 digit/second resolution. The resolution is not affected by the division ratio, which sets the minimum measurement time, and determines how long data can...
be taken before a phase spillover occurs (which can be hard to remove from a data set). A source having a frequency offset of $1 \times 10^{-6}$ can, for example, be measured for only about 5.8 days before a 1 pps phase spillover occurs after being initially set at the center. Drift in the trigger point of the counter can be a limitation to this measurement method.

9.2. Heterodyne Method

The heterodyne method mixes (subtracts) the two sources being compared, and measures the period of the resulting audio-frequency beat note. The measurement resolution is increased by the heterodyne factor (the ratio of the carrier to the beat frequency).

![Figure 9.2 Block Diagram of a Heterodyne Measuring System](image)

This heterodyne technique is a classic way to obtain high resolution with an ordinary period counter. It is based on the principle that phase information is preserved in a mixing process. For example, mixing a 10 MHz source against a 9.999 MHz Hz offset reference will produce a 100 Hz beat signal whose period variations are enhanced by a factor of $10$ MHz/100 Hz = $10^5$. Thus a period counter with 100 nanosecond resolution (10 MHz clock) can resolve clock phase changes of 1 picosecond. A disadvantage of this approach is that a stable offset reference is required at exactly the right frequency. Even worse, it can measure only frequency, requires a priori knowledge of the sense of the frequency difference, and often has dead time between measurements.

9.3. Dual Mixer Time Difference Method

The third method, in effect, combines the best features of the first two, using a time interval counter to measure the relative phase of the beat signals from a pair of mixers driven from a common offset reference:
This dual mixer time difference (DMTD) setup is arguably the most precise way of measuring an ensemble of clocks all having the same nominal frequency. When expanded to multiple channels by adding additional buffer amplifiers and mixers, and time tagging the zero-crossings of the beat notes for each channel, this arrangement allows any two of the clocks to be intercompared. The offset reference need not be coherent, nor must it have particularly low noise or high accuracy, because its effect cancels out in the overall measurement process. For best cancellation, the zero-crossings should be coincident or interpolated to a common epoch. Additional counters can be used to count the whole beat note cycles to eliminate their ambiguity, or the zero-crossings can simply be time tagged. The measuring system resolution is determined by the time interval counter or time-tagging hardware, and the mixer heterodyne factor. For example, if two 5 MHz sources are mixed against a common 5 MHz - 10 Hz offset oscillator (providing a $5 \times 10^6/10 = 5 \times 10^5$ heterodyne factor), and the beat note is time tagged with a resolution of 100 nsec (10 MHz clock), the measuring overall system resolution is $10^{-7}/5 \times 10^5 = 0.2$ psec.

Multichannel DMTD clock measuring systems have been utilized by leading national and commercial metrology laboratories for a number of years [1-5]. An early commercial version is described in Reference [3], and a newer technique is described in Reference [8]. A direct digital synthesizer (DDS) can be used as the offset reference to allow measurements to be made at any nominal frequency within its range. Cross-correlation methods can be used to reduce the DDS noise. Instruments using those techniques are available that automatically make both time and frequency domain measurements.

9.4. Measurement Problems and Pitfalls

It can be difficult to distinguish between a bad unit under test and a bad measurement. When problems occur in time-domain frequency stability measurements, they usually cause results that are worse than expected. It is nearly impossible for a measurement problem to give better than correct results, and there is considerable justification in saying that the best results are the
correct ones. Two possible exceptions to this are (1) misinterpretation of the scale factor, and (2) inadvertent coherency (e.g., injection locking of one source to another due to inadequate isolation. Lack of stationarity (changes in the source itself), while not a measurement problem per se, must also be considered. In general, the more devices available and the more measurements being made, the easier it is to sort things out.

One common problem is hum that contaminates the measurements due to ground loops. Because the measurement interval is usually much longer than the period of the power line frequency, and not necessarily coherent with it, aliased "beats" occur in the frequency record. Inspection of the raw data can show this, and the best cure is often isolation transformers in the signal leads. In fact, this is a wise precaution to take in all cases.

All sorts of other mechanisms (electrical, electromagnetic, magnetic, thermal, barometric, vibrational, acoustic, etc.) exist that can interfere with time domain frequency measurements. Think about all the environmental sensitivities of the unit under test, and control as many of them as possible. Be alert to day-night and weekly cycles that indicate human interference. Stories abound about correlations between elevators moving and cars coming and going ("auto-correlations") that have affected clock measurements. Think about what can have changed in the overall test setup. Slow periodic fluctuations will show up more distinctly in an all tau (rather than an octave tau) stability plot.

In high-precision measurements, where picoseconds matter (e.g. \(1 \times 10^{-15} = 1\) psec/1000 seconds), it is important to consider the mechanical rigidity of the test setup (e.g. \(1\) psec = 0.3 mm). This includes the electrical length (phase stability) of the connecting cables. Teflon dielectric is an especially bad choice for temperature stability, while foamed polyethylene is much better. Even a few degrees of temperature variation will cause the phase of a high-stability source to "breathe" as it passes through 100 feet of coaxial cable.

Phase jumps are a problem that should never be ignored. Examination of the raw phase record is critical because a phase jump (frequency impulse) appears strangely in a frequency record as a white FM noise characteristic [10]. Some large phase jumps are related to the carrier period (e.g., a malfunctioning digital frequency divider).

It is difficult to maintain the integrity of a measuring system over a long period, but, as long as the operating conditions of the unit under test and the reference are undisturbed, gaps in the data record may be acceptable. An uninterruptible power system is indispensable to maintain the continuity of a long run.

### 9.5. Measuring System Summary

A comparison of the relative advantages and disadvantages of these methods is shown in the following table:

<table>
<thead>
<tr>
<th>Method</th>
<th>Advantages</th>
<th>Disadvantages</th>
</tr>
</thead>
<tbody>
<tr>
<td>Divider &amp; Time Interval Counter</td>
<td>Provides phase data</td>
<td>Modest resolution</td>
</tr>
<tr>
<td></td>
<td>Covers wide range of carrier frequencies</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Easily expandable at low cost</td>
<td>Not suitable for short tau</td>
</tr>
</tbody>
</table>
SECTION 9  MEASURING SYSTEMS

<table>
<thead>
<tr>
<th>Mixer and Period Counter</th>
<th>Resolution enhanced by heterodyne factor</th>
<th>No phase data</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Provides direct frequency data</td>
<td>No frequency sense</td>
</tr>
<tr>
<td></td>
<td>Usable for short tau</td>
<td>Requires offset reference</td>
</tr>
<tr>
<td></td>
<td>Expandable at reasonable cost</td>
<td>Single carrier frequency</td>
</tr>
</tbody>
</table>

| Dual Mixer Time Difference | High resolution, low noise              | Single carrier frequency |
|                           | Provides phase data                     |                           |
|                           | Offset reference noise & inaccuracy      | Relatively complex       |
|                           | cancels                                 |                            |

It is preferable to make continuous zero dead time phase measurements at regular intervals, and a system using a dual-mixer time interval measurement is recommended. An automated high-resolution multi-channel clock (phase) measuring system with a high-performance (e.g., hydrogen maser) reference is a major investment, but one that can pay off in better productivity. It is desirable that the measurement control, data storage, and analysis functions be separated to provide robustness and networked access to the data. A low-noise reference not only supports better measurement precision but also allows measurements to be made faster (with less averaging).

9.6. Data Format

A one-column vector is all that is required for a phase or frequency data array. Because the data points are equally spaced, no time tags are necessary. Nevertheless, the use of timetags is recommended (see section 9.4 below), particularly to identify anomalies or to compare several sources. Time tagging is generally required for archival storage of clock measurements, but a single vector of extracted gap-filled data is sufficient for analysis. The recommended unit for phase data is seconds, while frequency data should be in the form of dimensionless fractional frequency. Double-precision exponential ASCII numeric format is recommended for ease of reading into most analysis software, with comma or space-delimited fields and one data point per line. The inclusion of comments and headers can pose problems, but most software will reject lines that start with a # or some other non-numeric character.

9.7. Data Quantization

The phase or frequency data must be gathered with sufficient resolution to show the variations of interest, and it must be represented with sufficient precision to convey those variations after removal of fixed offsets (see section 10.1 below). Nevertheless, highly quantized data can still contain useful information, especially after they are combined into longer averaging times. An example of highly quantized frequency data is the random telegraph signal shown below. Although these data have a non-Gaussian amplitude distribution (their histogram consists of two spikes), the random occurrences of the two levels produce a white FM noise characteristic.
9.8. Timetags

Timetags are often associated with phase or frequency data, and can be usefully applied to the analysis of these data.

Timetags are highly desirable for frequency stability measurements, particularly for identifying the exact time of some anomaly. The preferred timetag is the Modified Julian Date (MJD) expressed as a decimal fraction and referenced to UTC. Based on the astronomical Julian Date, the number of days since noon on January 1, 4713 BC, the MJD is the Julian Date - 2,4000,000.5. It is widely used, purely numeric, can have any required resolution, is easily converted to other formats, is non-ambiguous over a two-century (1900-2099) range, and is free from seasonal (daylight saving time) discontinuities.

Analysis software can easily convert the MJD into other formats such as year, month, day, hour, minute and second. The MJD (including the fractional day) can be obtained from the C language `time()` function by dividing its return value by 86400 and adding 40587.

9.9. Archiving and Access

There is no standard way to archive and access clock data. For some purposes, it is sufficient to simply save the raw phase or frequency data to a file, identifying it only by the file name. At the other extreme, multichannel clock measuring systems may require an elaborate database to store a large collection of data, keep track of the clock identities and transactions, provide security and robust data integrity, and serve the archived data via a network. It may also be necessary to integrate the clock data with other information (e.g., temperature) from a data acquisition system.
References for Measuring Systems

6. Data Sheet, Model 5110A Time Interval Analyzer, Symmetricom, Inc. (formerly Timing Solutions Corp.), Boulder, CO 80301 USA.
11. Data Sheet, Model 5120A Phase Noise Test Set, Symmetricom, Inc. (formerly Timing Solutions Corp.), Boulder, CO 80301 USA.
### 10 Analysis Procedure

A frequency stability analysis can proceed along several paths, as the circumstances dictate. Nevertheless, the following example shows a typical analysis flow. Using simulated data for a high-stability rubidium frequency standard, the purpose of the analysis is to characterize the noise in the presence of an outlier, large frequency offset and significant drift.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
<th>Plot</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Open and examine a phase data file. The phase data is just a ramp with slope corresponding to frequency offset.</td>
<td><img src="image1.png" alt="Phase Data Plot" /></td>
</tr>
<tr>
<td></td>
<td>0.000000000000000e+00</td>
<td></td>
</tr>
<tr>
<td></td>
<td>8.999873025741449e-07</td>
<td></td>
</tr>
<tr>
<td></td>
<td>1.799890526185009e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>2.699869215098003e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>3.599873851209537e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>4.49987627663997e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>5.399836191440859e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>6.29983612216789e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>7.19983672345638e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>8.099785679257264e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>8.99974486024524e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>9.89973269842008e-06</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Etc.</td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Convert the phase data to frequency data and examine it. An obvious outlier exists that must be removed to continue the analysis.</td>
<td><img src="image2.png" alt="Frequency Data Plot" /></td>
</tr>
<tr>
<td></td>
<td>Visual inspection of data is an important preprocessing step!</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Analyst judgment may be needed for less obvious outliers.</td>
<td></td>
</tr>
</tbody>
</table>
3 In an actual analysis, one should try to determine the cause of the outlier. The frequency spike of $1 \times 10^{-9}$ corresponds to a phase step of 900 nsec over a single 900-second measurement interval, nine 10 MHz carrier cycles. Data taken at a higher rate would help to determine whether the anomaly happened instantaneously or over some finite period. Timetags can help to relate the outlier to other external events.

<table>
<thead>
<tr>
<th>Relative Phase Nanoseconds</th>
</tr>
</thead>
<tbody>
<tr>
<td>1053.0</td>
</tr>
<tr>
<td>Data Point</td>
</tr>
</tbody>
</table>

3 Remove the outlier. The noise and drift are now visible. A line shows a linear fit to the frequency data, which appears to be quite appropriate.

<table>
<thead>
<tr>
<th>Frequency, ppm</th>
</tr>
</thead>
<tbody>
<tr>
<td>100.020</td>
</tr>
<tr>
<td>Time, Days</td>
</tr>
</tbody>
</table>

4 Remove the frequency offset from the phase data. The resulting quadratic shape is due to the frequency drift. One can just begin to see phase fluctuations around the quadratic fit to the phase data.

<table>
<thead>
<tr>
<th>Phase, Nanoseconds</th>
</tr>
</thead>
<tbody>
<tr>
<td>500</td>
</tr>
<tr>
<td>Time, Days</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Time, Days</th>
</tr>
</thead>
<tbody>
<tr>
<td>0.0</td>
</tr>
<tr>
<td>0.0</td>
</tr>
</tbody>
</table>
5. Remove the frequency drift, leaving the phase residuals for noise analysis, which is now clearly visible.

Some experience is needed to interpret phase data like these. Remember that frequency corresponds to the slope of the phase, so the frequency is lowest near the end of the record, where the phase slope is the most negative.

6. Convert the phase residuals to frequency residuals. Alternatively, remove the frequency drift from the frequency data of Step #3. There are subtle differences in removing the linear frequency drift as a quadratic fit to the phase data compared with removing it as a linear fit to the frequency data (different noise models apply). Other drift models may be more appropriate. Analyst judgment is needed to make the best choices.

7. Perform a stability analysis using the overlapping Allan deviation. The results show white FM noise at short averaging times ($\tau^{-0.5}$ slope), and flicker FM noise at longer tau ($\tau^{0}$ slope), both at the simulated levels shown in the annotations of the first plot.

### 10.1. Data Precision

There are relatively few numerical precision issues relating to the analysis of frequency stability data. One exception, however, is phase data for a highly stable frequency source having a relatively large frequency offset. The raw phase data will be essentially a straight
line (representing the frequency offset), and the instability information is contained in the small deviations from the line. A large number of digits must be used unless the frequency offset is removed by subtracting a linear term from the raw phase data. Similar considerations apply to the quadratic phase term (linear frequency drift). Many frequency stability measures involve averages of first or second differences. Thus, while their numerical precision obviously depends upon the variable digits of the data set, there is little error propagation in forming the summary statistics.

### 10.2. Preprocessing

Preprocessing of the measurement data is often necessary before the actual analysis is performed, which may require data averaging, or removal of outliers, frequency offset and drift.

Phase data may be converted to frequency data, and vice versa. Phase and frequency data can be combined for a longer averaging time. Frequency offset may be removed from phase data by subtracting a line determined by the average of the first differences, or by a least squares linear fit. An offset may be removed from frequency data by normalizing it to have an average value of zero. Frequency drift may be removed from phase data by a least squares or 3-point quadratic fit, or by subtracting the average of the second differences. Frequency drift may be removed from frequency data by subtracting a least-squares linear fit, by subtracting a line determined by the first differences or by calculating the drift from the difference between the two halves of the data. The latter, called the bisection drift, is equivalent to the three-point fit for phase data. Other more specialized log and diffusion models may also be used. The latter are particularly useful to describe the stabilization of a frequency source. In general, the objective is to remove as much of the deterministic behavior as possible, obtaining random residuals for subsequent noise analysis.

### 10.3. Gaps, Jumps and Outliers

It is common to have gaps and outliers in a set of raw frequency stability data. Missing or erroneous data may occur due to power outages, equipment malfunctions and interference. For long-term tests, it may not be possible or practical to repeat the run, or otherwise avoid such bad data points. Usually the reason for the gap or outlier is known, and it is particularly important to explain all phase discontinuities. Plotting the data will often show the bad points, which may have to be removed before doing an analysis to obtain meaningful results.

Frequency outliers are found by comparing each data point with the median value of the data set plus or minus some multiple of the absolute median deviation. These median statistics are more robust because they are insensitive to the size of the outliers. Outliers can be replaced by gaps or filled with interpolated values.

Frequency jumps can also be a problem for stability analysis. Their occurrence indicates that the statistics are not stationary, and it may be necessary to divide the data into portions and analyze them separately.
Gaps and outliers can occur in clock data due to problems with the measuring system or the frequency source itself. Like death and taxes, gaps and outliers can be avoided but not eliminated.

### 10.4. Gap Handling

Gaps should be included to maintain the proper implied time interval between measurements, and a value of zero (0) is often used to denote a gap. For phase data, zero should be treated as valid data if it is the first or last point. For fractional frequency data, valid data having a value of zero can be replaced by some very small value (e.g. 1e-99). Many analysis functions can produce meaningful results for data having gaps by simply skip those points that involve a gap. For example, in the calculation of the Allan variance for frequency data, if either of the two points involved in the first difference is a gap, that Allan variance pair is skipped in the summation.

Gaps may be filled in phase or frequency data by replacing them with interpolated values, by first removing any leading and trailing gaps, and then using the two values immediately before and after any interior gaps to determine linearly interpolated values within the gap.

A zero value in fractional frequency data can also occur as the result of the conversion of two equal adjacent phase data points (perhaps because of limited measurement resolution), and the value should be adjusted to, say, 1e-99 to distinguish it from a gap.

### 10.5. Uneven Spacing

Unevenly spaced phase data can be handled if they have associated timetags by using the individual timetag spacing when converting it to frequency data. Then, if the tau differences are reasonably small, the data may be analyzed by using the average timetag spacing as the analysis tau, in effect placing the frequency data on an average uniform grid. While completely random data spacing is not amenable to this process, tau variations of ±10 % will yield reasonable results as long as the exact intervals are used for the phase to frequency conversion.

### 10.6. Analysis of Data with Gaps

Care must be taken when analyzing the stability of data with missing points and/or gaps. Missing points can be found by examining the timetags associated with the data, and gaps can then be inserted as placeholders to maintain equally spaced data. Similarly, outliers can be replaced with gaps for the same reason. These gaps can span multiple points. Some analysis processes can be performed with data having gaps by skipping over them, perhaps at some speed penalty, but other calculations cannot be. It is therefore often necessary to replace the gaps with interpolated values. Those points are not real data, however, and, if there are many of them, the results will be suspect. In these cases, judgment is needed to assure a credible result. It may be more prudent to simply analyze a gap-free portion of the data.
10.7. Phase-Frequency Conversions

Phase to frequency conversion is straightforward for data having gaps. Because two phase points are needed to determine each frequency point (as the difference between the phase values divided by their tau), a single phase gap will cause two frequency gaps, and a gap of \( N \) phase points causes \( N+1 \) frequency gaps.

Conversion from frequency to phase is more problematic because of the need to integrate the frequency data. The average frequency value is used to calculate the phase during the gap, which can cause a discontinuity in the phase record. Analysis of phase data resulting from the conversion of frequency data having a large gap is not recommended.

10.8. Drift Analysis

Drift analysis functions generally perform well for data having gaps, provided that missing data are represented by gaps to maintain a regular time sequence.

10.9. Variance Analysis

Variance analysis functions can include provisions for handling gaps. Some of these functions yield satisfactory results in all cases, while others have speed limitations, or provide unsatisfactory results for data having large gaps. The latter is most apparent at longer averaging times where the averaging factor is comparable to the size of the gap. The speed limitations are caused by more complex gap checking and frequency averaging algorithms, while the poor results are associated with the total variances for which conversion to phase data is required. In all cases, the results will depend on coding details included in addition to the basic variance algorithm. Filling gaps can often help for the total variances. Two general rules apply for the variance analysis of data having large gaps: (1) use unconverted phase data, and (2) check the results against the normal Allan deviation (which has the simplest, fastest gap handling ability).

10.10. Spectral Analysis

Gap filling in spectral analysis functions can affect the low frequency portion of the spectrum.

10.11. Outlier Recognition

The median absolute deviation (MAD) is recommended as its means of outlier recognition. The MAD is a robust statistic based on the median of the data. It is the median of the scaled absolute deviations of the data points from their median value, defined as

\[
\text{MAD} = \text{Median} \left\{ \frac{|y(i) - m|}{0.6745} \right\},
\]

where \( m = \text{Median} \{ y(i) \} \), and the factor 0.6745 makes the MAD equal to the standard deviation for normally distributed data. Each frequency data point, \( y(i) \), is compared with the median value of the data set, \( m \), plus or minus the desired multiple of the MAD.

While the definition of an outlier is somewhat a matter of judgment, it is important to find and remove such points in order to use the rest of the data, based on their deviation from the median of the data, using a deviation limit in terms of the median absolute deviation (a 5-sigma limit is common). This is a robust way to determine an outlier, which is then replaced.
by a gap. An automatic outlier removal algorithm can iteratively apply this method to remove all outliers, which should be an adjunct to, and not a substitute for, visual inspection of the data.

It is important to explain all outliers, thereby determining whether they are due to the measurement process or the device under test. An important first step is to correlate the bad point with any external events (e.g., power outages, equipment failures, etc.) that could account for the problem. Failures of the measurement system, frequency reference, or environmental control are often easier to identify if multiple devices are under test. Obviously, a common gap in all measurement channels points to a failure of the measurement system, while a common change in all measurement readings points to a reference problem. Auxiliary information such as monitor data can be a big help in determining the cause of outliers. A log of all measurement system events should be kept to facilitate outlier identification. Discrete phase jumps are a particular concern, and, if they are related to the RF carrier frequency, may indicate a missing cycle or a problem with a digital divider. A phase jump will correspond to a frequency spike with a magnitude equal to the phase change divided by the measurement interval. Such a frequency spike will produce a stability record that appears to have a (large magnitude) white FM noise characteristic, which can be a source of confusion.

References for Gaps, Jumps, and Outliers


10.12. Data Plotting

Data plotting is often the most important step in the analysis of frequency stability. Visual inspection can provide vital insight into the results, and is an important preprocessor before numerical analysis. A plot also shows much about the validity of a curve fit.

Phase data are plotted as line segments connecting the data points. This presentation properly conveys the integral nature of the phase data. Frequency data are plotted as horizontal lines between the frequency data points. This shows the averaging time associated with the frequency measurement, and mimics the analog record from a frequency counter. As the density of the data points increases, there is essentially no difference between the two plotting methods. Missing data points are shown as gaps without lines connecting the adjacent points.

10.13. Variance Selection

It is the user's responsibility to select an appropriate variance for the stability analysis. The overlapping Allan variance is recommended in most cases, especially where the frequency drift is small or has been removed. The Allan variance is the most widely used time-domain stability measure, and the overlapping form provides better confidence than the original
"normal" version. The total and Théo1 variance can be used for even better confidence at large averaging factors (but at the expense of longer computation time). The modified Allan variance is recommended to distinguish between white and flicker PM noise, and, again, a total form of it is available for better confidence at long τ. The time variance provides a good measure of the time dispersion of a clock due to noise, while MTIE measures the peak time deviations. TIE rms can also be used to assess clock performance, but TVAR is generally preferred. Finally, the overlapping Hadamard variance is recommended over its normal form for analyzing stability in the presence of divergent noise or frequency drift. In all cases, the results are reported in terms of the deviations.

The choice of τ interval depends mainly on whether interference mechanisms are suspected that cause the stability to vary periodically. Normally, octave or decade spacing is used (the former has even spacing on a log-log plot, while the latter provides τ multiples of ten). The all τ option can be useful as a form of spectral analysis to detect cyclic disturbances (such as temperature cycling).

### 10.14. Three-Cornered Hat

Any frequency stability measurement includes noise contributions from both the device under test and the reference. Ideally, the reference noise would be low enough that its contribution to the measurement is negligible. Or, if the noise of the reference is known, it can be removed by subtracting its variance. A special case is that of two identical units where half of the measured variance comes from each, and the measured deviation can be corrected for one unit by dividing it by $\sqrt{2}$. Otherwise, it may be useful to employ the so-called "three-cornered hat" method for determining the variance of an individual source. Given a set of three pairs of measurements for three independent frequency sources a, b and c whose variances add

\[ \sigma_{ab}^2 = \sigma_a^2 + \sigma_b^2 \]
\[ \sigma_{ac}^2 = \sigma_a^2 + \sigma_c^2 \]
\[ \sigma_{bc}^2 = \sigma_b^2 + \sigma_c^2 . \]

The individual variances may be determined by the expressions

\[ \sigma_a^2 = \frac{1}{2} \left[ \sigma_{ab}^2 + \sigma_{ac}^2 - \sigma_{bc}^2 \right] \]
\[ \sigma_b^2 = \frac{1}{2} \left[ \sigma_{ab}^2 + \sigma_{bc}^2 - \sigma_{ac}^2 \right] \]
\[ \sigma_c^2 = \frac{1}{2} \left[ \sigma_{ac}^2 + \sigma_{bc}^2 - \sigma_{ab}^2 \right] . \]
Although useful for determining the individual stabilities of units having similar performance, the method may fail by producing negative variances for units that have widely differing stabilities, if the units are correlated, or for which there are insufficient data. The three sets of stability data should be measured simultaneously. The three-cornered hat method should be used with discretion, and it is not a substitute for a low noise reference. It is best used for units having similar stability (e.g., to determine which unit is best). Negative variances are a sign that the method is failing (because it was based on insufficient measurement data, or because the units under test have disparate or correlated stability). This problem is most likely to arise at long tau.

The three-cornered hat function may be used to correct a stability measurement for the noise contribution of the reference, as shown in the following diagram:

```
The Unit Under Test (UUT), denoted as source A, is measured against the reference, denoted by B and C, by identical stability data files A-B and A-C. The reference is measured against itself by stability data file B-C, which contains the a priori reference stability values multiplied by $\sqrt{2}$.

An example of the use of the three-cornered hat function to correct stability data for reference noise is shown below. Simulated overlapping Allan deviation stability data for the unit under test versus the reference were created by generating and analyzing 512 points of frequency data with $\tau = 1$ second and $\sigma_y(1 \text{ sec}) = 1 \text{e-11}$. The resulting stability data are shown in the following table.

<table>
<thead>
<tr>
<th>Tau</th>
<th>#</th>
<th>Sigma</th>
<th>Min Sigma</th>
<th>Max Sigma</th>
</tr>
</thead>
<tbody>
<tr>
<td>1.000e+00</td>
<td>511</td>
<td>9.448e-12</td>
<td>9.108e-12</td>
<td>9.830e-12</td>
</tr>
<tr>
<td>2.000e+00</td>
<td>509</td>
<td>7.203e-12</td>
<td>6.923e-12</td>
<td>7.520e-12</td>
</tr>
<tr>
<td>4.000e+00</td>
<td>505</td>
<td>5.075e-12</td>
<td>4.826e-12</td>
<td>5.367e-12</td>
</tr>
<tr>
<td>8.000e+00</td>
<td>497</td>
<td>3.275e-12</td>
<td>3.058e-12</td>
<td>3.546e-12</td>
</tr>
<tr>
<td>1.600e+01</td>
<td>481</td>
<td>2.370e-12</td>
<td>2.157e-12</td>
<td>2.663e-12</td>
</tr>
<tr>
<td>3.200e+01</td>
<td>449</td>
<td>1.854e-12</td>
<td>1.720e-12</td>
<td>2.025e-12</td>
</tr>
<tr>
<td>6.400e+01</td>
<td>385</td>
<td>1.269e-12</td>
<td>1.147e-12</td>
<td>1.441e-12</td>
</tr>
<tr>
<td>1.280e+02</td>
<td>257</td>
<td>5.625e-13</td>
<td>4.820e-13</td>
<td>7.039e-13</td>
</tr>
</tbody>
</table>
A similar stability file is used for the reference. Since it represents a measurement of the reference against itself, the Allan deviations of the reference source are multiplied by \( \sqrt{2} \). Simulated overlapping Allan deviation stability data for the reference versus the reference was created by generating 512 points of frequency data with \( \tau = 1 \) second and \( \sigma_y(1) = 1.414e-12 \).

<table>
<thead>
<tr>
<th>Tau</th>
<th>#</th>
<th>Sigma</th>
<th>Min Sigma</th>
<th>Max Sigma</th>
</tr>
</thead>
<tbody>
<tr>
<td>1.000e+00</td>
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<td>1.490e-12</td>
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<tr>
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<td>505</td>
<td>7.631e-13</td>
<td>7.257e-13</td>
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</tr>
<tr>
<td>8.000e+00</td>
<td>497</td>
<td>5.846e-13</td>
<td>5.458e-13</td>
<td>6.329e-13</td>
</tr>
<tr>
<td>1.600e+01</td>
<td>481</td>
<td>3.681e-13</td>
<td>3.349e-13</td>
<td>4.135e-13</td>
</tr>
<tr>
<td>3.200e+01</td>
<td>449</td>
<td>2.451e-13</td>
<td>2.152e-13</td>
<td>2.924e-13</td>
</tr>
<tr>
<td>6.400e+01</td>
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<td>1.637e-13</td>
<td>1.368e-13</td>
<td>2.173e-13</td>
</tr>
<tr>
<td>1.280e+02</td>
<td>257</td>
<td>1.360e-13</td>
<td>1.058e-13</td>
<td>2.285e-13</td>
</tr>
</tbody>
</table>

The corrected UUT and reference stabilities are plotted below:
Here the reference stability is about $1 \times 10^{-12} \tau^{-1/2}$, and the corrected UUT instability is slightly less than the uncorrected values. Note that the B and C columns of corrected stability values both represent the reference source. Appropriate use of the three-cornered hat method to correct stability measurements for reference noise applies where the reference stability is between three to ten times better than that of the unit under test. The correction is negligible for more the latter (see above), and has questionable confidence for less than the former (and a better reference should be used).

The error bars of the individual variances may be set using $\chi^2$ statistics by first determining the reduced number of degrees of freedom associated with the three-cornered hat process [11, 12]. The fraction of remaining degrees of freedom for unit i as a result of performing a three-cornered hat instead of measuring against a perfect reference is given by:

$$\Gamma = \frac{(2 \cdot \sigma^4_i)}{(2 \cdot \sigma^4_i + \sigma^2_a \cdot \sigma^2_b + \sigma^2_a \cdot \sigma^2_c + \sigma^2_b \cdot \sigma^2_c)}.$$

The ratio of the number of degrees of freedom is 0.4 for three units having the same stability, independent of the averaging time and noise type.

The three-cornered hat technique can be extended to M clocks (subject to the same restriction against negative variances) by using the expression

$$\sigma^2_i = \frac{1}{M - 2} \left[ \sum_{j=1}^{M} \sigma^2_{ij} - B \right],$$

where

$$B = \frac{1}{2(M - 1)} \sum_{k=1}^{M} \left[ \sum_{j=1}^{M} \sigma^2_{kj} \right],$$

and the $\sigma^2_{ij}$ are the measured Allan variances for clock i versus j at averaging time $\tau$. Using $\sigma^2_{ii} = 0$ and $\sigma^2_{ij} = \sigma^2_{ji}$, one can easily write closed-form expressions for the separated variances from measurements of M clocks.
1. The term "Three-cornered hat" was coined by J.E. Gray of NIST.

**10.15. Reporting**

The results of a stability analysis are usually presented as a combination of textual, tabular and graphic forms. The text describes the device under test, the test setup, and the methodology of the data preprocessing and analysis, and summarizes the results. The assumptions and analysis choices should be documented so that the results could be reproduced. The report often includes a table of the stability statistics. Graphical presentation of the data at each stage of the analysis is generally the most important aspect of presenting the results. For example, these are often a series of plots showing the phase and frequency data with an aging fit, phase and frequency residuals with the aging removed, and stability plots with noise fits and error bars. Plot titles, subtitles, annotations and inserts can be used to clarify and emphasize the data presentation. The results of several stability runs can be combined, possibly along with specification limits, into a single composite plot. The various elements can be combined into a single electronic document for easy printing and transmittal.
11 Case Studies

This section contains several case studies to further illustrate methodologies of frequency stability analysis.

11.1. Flicker Floor of a Cesium Frequency Standard

The purpose of this analysis is to examine the flicker floor of a commercial cesium beam tube frequency standard. An instrument of this kind can be expected to display a white FM noise characteristic out to long averaging times. At some point, however, the unit will typically “flicker out”, with its stability plot flattening out to a flicker FM noise characteristic, beyond which its stability will no longer improve. Determining this point requires a lengthy and expensive run, and it is worthwhile to use an analytical method that provides the best information at long averaging factors. The effort of a more elaborate analysis is far easier than extending the measurement time by weeks or months.

This is the five-month frequency record for the unit under test. It has already been “cleaned up” for any missing points, putting in gaps as required to provide a time-continuous data set. The data look very “white” (see Section 6.1), the frequency offset is very small (+7.2x10^{-14}), and there is no apparent drift (+1.4x10^{-16}/day). Overall, this appears to be a very good record.

An overlapping Allan deviation plot shows a white FM noise level of about 8.2x10^{-12}\tau^{-1/2} out to about 5 days, where the stability levels off at about 1.2x10^{-14}. While this is very respectable behavior, one wonders what the stability actually is at the longer averaging times. But to gain meaningful confidence using ADEV there, the run would have to be extended by several months.
The total deviation can provide better confidence at the longer averaging times. It seems to indicate that the stability continues to improve past 10 days, where it drops below $1 \times 10^{-14}$. But the results are not conclusive.

Théo1 can provide even better long-term information, at the expense of a longer calculation time. This Théo1 plot seems to show even more clearly that the stability continues to improve at longer averaging times, well into the $10^{-13}$ range. It assumes white FM noise (no Théo1 bias removal).

A ThéoH analysis with automatic bias removal, which combines the AVAR and ThéoBR statistics, requires an unacceptably long time for the original data set. By averaging the data by a factor of 5 ($\tau = 500$ sec), the analysis can be performed in several hours. The results are essentially the same as for Théo1 above. The Théo bias is 1.676, intermediate between white and flicker FM noises for medium to large averaging factors.

One can conclude that this cesium beam frequency standard reaches a stability slightly better than $1 \times 10^{-14}$ at an averaging time on the order of 1 month.
11.2. Hadamard Variance of a Source with Drift

These frequency data simulate a typical rubidium frequency standard (RFS) with a combination of white and flicker FM noise, plus significant frequency drift.

If an Allan variance analysis is performed directly on these data without drift removal, the stability at the longer averaging times is degraded. In particular, the stability plot has a $\tau^{-1}$ slope beyond $10^5$ seconds that corresponds to the drift (i.e., about $1 \times 10^{-12}$ at 5 days).

If the linear frequency drift is removed before performing the AVAR stability analysis, the stability plot shows a white FM noise ($\tau^{-1/2}$) characteristic changing to flicker FM noise ($\tau^0$) at longer averaging times. It is usually best to use a stability plot only to show the noise, and analyzing and removing the drift separately.
The Hadamard variance is insensitive to linear frequency drift. It can therefore be used to perform a stability analysis on the original data without first having to remove the drift. The HDEV results are essentially identical to those of the drift-removed ADEV. This can be more convenient, especially when analyzing a combination of sources having differing amounts of drift (e.g., cesium and rubidium units).

11.3. Phase Noise Identification

Consider the problem of identifying the dominant type of phase noise in a frequency distribution amplifier. Assume that time domain stability measurements have been made comparing the amplifier’s input and output signals. How should these data be analyzed to best determine the noise type? Simulated white and flicker PM noise at equal levels are analyzed below in several ways to demonstrate their ability to perform this noise identification.

<table>
<thead>
<tr>
<th>White PM</th>
<th>Flicker PM</th>
</tr>
</thead>
<tbody>
<tr>
<td>Examination of the phase data is a good first step. An experienced analyst will immediately notice the difference between the white and flicker noise plots, but will find it harder to quantify them.</td>
<td></td>
</tr>
</tbody>
</table>

In contrast, the frequency data shows little noise type discrimination, because the differencing process whitens both. Examination of the frequency data would be appropriate to distinguish between white and flicker FM noise, however.
The Allan deviation is not able to distinguish between white and flicker PM noise. Both have slopes of about 1, as shown by the superimposed noise fit lines.

Even better discrimination is possible with the autocorrelation function. The lag 1 ACF is 0.006 and 0.780 for these white and flicker PM noise data, and is able to quantitatively estimate the power law noise exponents as +1.99 and +0.93, respectively. That ID method is quite effective even for mixed noise types.

The modified Allan deviation, by virtue of its additional phase averaging, is able to distinguish between white and flicker PM noise, for which the slopes are –1.5 and –1.0, respectively.
11.4. Detection of Periodic Components

Frequency stability data can contain periodic fluctuations due to external interference, environmental effects, or the source itself, and it can therefore be necessary to detect discrete spectral components when performing a stability analysis. This example uses a set of 50,000 points of $\tau = 1$ second simulated white FM noise at a level of $1 \times 10^{-11} \tau^{-1/2}$ that contains sinusoidal interference with a period of 500 seconds at a peak level of $1 \times 10^{-12}$. The latter simulates interference that might occur as the result of air conditioner cycling. Several analytical methods are used to detect this periodic component.

The interference level is too low to be visible in the full frequency data plot.
By zooming in, there is just a hint of the interference (10 cycles over 5000 data points).

The interference is quite visible in an “all tau” stability plot as a null that occurs first at an averaging time of 500 seconds (the period of the interference). Here the stability is equal to the underlying white FM noise level.

The interference is clearly visible in the power spectral density (PSD), which has a bright component at 2 mHz, corresponding to the 500 sec period of the interference.
The interference is less visible in an autocorrelation plot, where cyclic behavior is barely noticeable with a 500-lag period. The PSD has equivalent information, but its log scale makes low-level components more apparent.

It is a good analysis policy to examine the power spectral density when periodic fluctuations are visible on a stability plot and periodic interference is suspected.

11.5. Stability Under Vibrational Modulation

This plot shows the stability of an oscillator with a combination of white PM noise and a sinusoidal component simulating vibrational modulation. Nulls in the Allan deviation occur at averaging times equal to the multiples of the 20 Hz sinusoidal modulation period, where the stability is determined by the white PM noise. Peaks in the Allan deviation occur at the modulation half cycles, and have a $\tau^{-1}$ envelope set by the vibrational phase modulation.
11.6. White FM Noise of a Frequency Spike

The Allan deviation of a frequency record having a large spike (a phase step) has a $\tau^{-1/2}$ characteristic [1]. Thus adding a single large (say $10^6$) central outlier to the 1000-point test suite of section 12.4 will give a data set with $\sigma_y(\tau) = \left[(10^6)^2/(1000-1)\right]^{1/2} = 3.16386\times10^4$, as shown in this stability plot.


11.7. Composite Aging Plots

A composite plot showing the aging of a population of frequency sources can be an effective way to present this information visually, providing a quick comparison of their behaviors. The following figure shows the stabilization period of a production lot of rubidium clocks.

Plots for 40 units are shown with their serial numbers, all plotted with the same scales. In particular, these plots all have a full x-scale time range of nine weeks (1 week/div) and a full y-scale frequency range of $1.5\times10^{-11}$ (1x$10^{-12}$/div). A diagonal slope downward across the plot corresponds to an aging of about $-2.4\times10^{-13}$/day. All the data have $\tau = 900$ seconds (15 minutes). A figure like this immediately shows that (a) all the units have negative frequency aging of about the same magnitude that stabilizes in about the same way, (b) there are occasional gaps in some of the records, (c) all of the units have about the same short-term noise, but some of the records are quieter than others in the longer term, (d) some of the units take longer than others to reach a certain aging slope.

The plots, although small, still contain enough detail to allow subtle comparisons very quickly, far better than a set of numbers or bar graphs would do. The eye can easily see the similarities and differences, and can immediately select units based on some criterion, which would be harder to do using a set of larger plots on separate pages. Closer inspection of even these small plots can reveal a lot of quantitative information if one knows the scale factors. Color coding, although not used here, could be used to provide additional information.

These plots are inspired by Edward Tufte’s book *The Visual Display of Quantitative Information*. 

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Software is necessary to perform a frequency stability analysis, because those calculations generally involve complex specialized algorithms and large data sets needed to interactively perform and document a complete stability analysis. It is convenient to use an integrated software package that combines all of the required analytical tools, operates on current computer hardware and operating systems, includes the latest analytical techniques, and has been validated to produce the correct results.

### 12.1. Software Validation

Considerable effort is needed to ensure that the results obtained from frequency stability analysis software are correct. Several suggested validation methods are shown. Mature commercially available software should be used whenever possible instead of developing custom software. User feedback and peer review is important. There is a continuing need to validate the custom software used to analyze time domain frequency stability, and the methods listed below can help ensure that correct results are obtained.

Several methods are available to validate frequency stability analysis software:

1. **Manual Analysis:** The results obtained by manual analysis of small data sets (such as in NBS Monograph 140Annex 8.E) can be compared with the new program output. This is always good to do to get a “feel” for the process.
2. **Published Results:** The results of a published analysis can be compared with the new program output. One important validation method is comparison of the program results against a test suite such as the one in References [1, 2]. Copies of those test data are available on-line [3].
3. **Other Programs:** The results obtained from other specialized stability analysis programs (such as that from a previous generation computer or operating system) can be compared with the new program output.
4. **General Purpose Programs:** The results obtained from industry standard, general purpose mathematical and spreadsheet programs such as MathCAD®, Matlab® and Excel® can be compared with the new program output.
5. **Consistency Checks:** The new program should be verified for internal consistency, such as producing the same stability results from phase and frequency data. The standard and normal Allan variances should be approximately equal for white FM noise. The normal and modified Allan variances should be identical for an averaging factor of 1. For other averaging factors, the modified Allan variance should be approximately one-half the normal Allan variance for white FM noise, and the normal and overlapping Allan variances should be approximately equal. The overlapping method provides better confidence of the stability estimates. The various methods of drift removal should yield similar results.
6. **Simulated Data:** Simulated clock data can also serve as a useful cross check. Known values of frequency offset and drift can be inserted, analyzed, and removed. Known values of power-law noise can be generated, analyzed, plotted, and modeled.
12.2. Test Suites

The following tables summarize the values for several common frequency stability measures for both the classic NBS data set and a 1000-point portable test suite.

12.3. NBS Data Set

A "classic" suite of frequency stability test data is the set of nine 3-digit numbers from Annex 8.E of NBS Monograph 140 shown in Table I. Those numbers were used as an early example of an Allan variance calculation. This frequency data is also normalized to zero mean by subtracting the average value, and then integrated to obtain phase values. A listing of the properties of this data set is shown in Table II. While nine data points are not sufficient to calculate large frequency averages, they are, nevertheless, a very useful starting point to verify frequency stability calculations, since a small data set can easily be entered and analyzed manually. A small data set is also an advantage for finding "off-by-one" errors where an array index or some other integer variable is slightly wrong and hard to detect in a larger data set.

Table I. NBS Monograph 140, Annex 8.E Test Data

<table>
<thead>
<tr>
<th>#</th>
<th>Frequency</th>
<th>Normalized Frequency</th>
<th>Phase (τ=1)</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>892</td>
<td>103.11111</td>
<td>0.00000</td>
</tr>
<tr>
<td>2</td>
<td>809</td>
<td>20.11111</td>
<td>103.11111</td>
</tr>
<tr>
<td>3</td>
<td>823</td>
<td>34.11111</td>
<td>123.22222</td>
</tr>
<tr>
<td>4</td>
<td>798</td>
<td>9.11111</td>
<td>157.33333</td>
</tr>
<tr>
<td>5</td>
<td>671</td>
<td>-117.88889</td>
<td>166.44444</td>
</tr>
<tr>
<td>6</td>
<td>644</td>
<td>-144.88889</td>
<td>48.55555</td>
</tr>
<tr>
<td>7</td>
<td>883</td>
<td>94.11111</td>
<td>-96.33333</td>
</tr>
<tr>
<td>8</td>
<td>903</td>
<td>114.11111</td>
<td>-2.22222</td>
</tr>
<tr>
<td>9</td>
<td>677</td>
<td>-111.88889</td>
<td>111.88889</td>
</tr>
<tr>
<td>10</td>
<td></td>
<td></td>
<td>0.0000</td>
</tr>
</tbody>
</table>

Table II. NBS Monograph 140, Annex 8.E Test Data Statistics

<table>
<thead>
<tr>
<th>Averaging Factor</th>
<th>1</th>
<th>2</th>
</tr>
</thead>
<tbody>
<tr>
<td># Data Points</td>
<td>9</td>
<td>4</td>
</tr>
<tr>
<td>Maximum</td>
<td>903</td>
<td>893.0</td>
</tr>
<tr>
<td>Minimum</td>
<td>644</td>
<td>657.5</td>
</tr>
<tr>
<td>Average</td>
<td>788.8889</td>
<td>802.875</td>
</tr>
<tr>
<td>Median</td>
<td>809</td>
<td>830.5</td>
</tr>
<tr>
<td>Linear Slope</td>
<td>-10.20000</td>
<td>-2.55</td>
</tr>
<tr>
<td>Intercept</td>
<td>839.8889</td>
<td>809.25</td>
</tr>
<tr>
<td>Standard Deviation [1]</td>
<td>100.9770</td>
<td>102.6039</td>
</tr>
<tr>
<td>Normal Allan Deviation</td>
<td>91.22945</td>
<td>115.8082</td>
</tr>
<tr>
<td>Overlapping Allan Dev</td>
<td>91.22945</td>
<td>85.95287</td>
</tr>
<tr>
<td>Modified Allan Dev</td>
<td>91.22945</td>
<td>74.78849</td>
</tr>
<tr>
<td>Time Deviation</td>
<td>52.67135</td>
<td>86.35831</td>
</tr>
<tr>
<td>Hadamard Deviation</td>
<td>70.80607</td>
<td>116.7980</td>
</tr>
<tr>
<td>Overlap Hadamard Dev</td>
<td>70.80607</td>
<td>85.61487</td>
</tr>
<tr>
<td>Hadamard Total Dev</td>
<td>70.80607</td>
<td>91.16396</td>
</tr>
<tr>
<td>Total Deviation</td>
<td>91.22945</td>
<td>93.90379</td>
</tr>
<tr>
<td>Modified Total Dev</td>
<td>75.50203</td>
<td>75.83606</td>
</tr>
<tr>
<td>Time Total Deviation</td>
<td>43.59112</td>
<td>87.56794</td>
</tr>
</tbody>
</table>
12.4. 1000-Point Test Suite

The larger frequency data test suite used here consists of 1000 pseudo-random frequency data points. It is produced by the following prime modulus linear congruential random number generator:

\[ n_{i+1} = 16807 \, n_i \mod 2147483647 \]

This expression produces a series of pseudo-random integers ranging in value from 0 to 2147483646 (the prime modulus, \(2^{31} - 1\), avoids a collapse to zero). When started with the seed \(n_0 = 1234567890\), it produces the sequence \(n_1 = 395529916, n_2 = 1209410747, n_3 = 633705974, \) etc. These numbers may be divided by 2147483647 to obtain a set of normalized floating-point test data ranging from 0 to 1. Thus the normalized value of \(n_0\) is 0.5748904732. A spreadsheet program is a convenient and reasonably universal way to generate these data. The frequency data may be converted to phase data by assuming an averaging time of 1, yielding a set of 1001 phase data points. Similarly, frequency offset and/or drift terms may be added to the data. These conversions can also be done by a spreadsheet program.

The values of this data set will be uniformly distributed between 0 and 1. While a data set with a normal (Gaussian) distribution would be more realistic, and could be produced by summing a number of independent uniformly distributed data sets, or by the Box-Muller method [5], this simpler data set is adequate for software validation.

Table III. 1000-Point Frequency Data Set

<table>
<thead>
<tr>
<th>Averaging Factor</th>
<th>1</th>
<th>10</th>
<th>100</th>
</tr>
</thead>
<tbody>
<tr>
<td># Data Points</td>
<td>1000</td>
<td>100</td>
<td>10</td>
</tr>
<tr>
<td>Maximum</td>
<td>9.957453e-01</td>
<td>7.003371e-01</td>
<td>5.489368e-01</td>
</tr>
<tr>
<td>Minimum</td>
<td>1.371760e-03</td>
<td>2.545924e-01</td>
<td>4.533354e-01</td>
</tr>
<tr>
<td>Median</td>
<td>4.798849e-01</td>
<td>5.047888e-01</td>
<td>4.807261e-01</td>
</tr>
<tr>
<td>Linear Slope [2,3]</td>
<td>6.490910e-06</td>
<td>5.979804e-05</td>
<td>1.056376e-03</td>
</tr>
<tr>
<td>Log Fit [2,4], a=</td>
<td>5.577220e-03</td>
<td>5.248477e-03</td>
<td>7.138988e-03</td>
</tr>
<tr>
<td>(y(t)=a \cdot \ln (bt+1)+c, b=)</td>
<td>9.737500e-01</td>
<td>4.594973e+00</td>
<td>1.420429e+02</td>
</tr>
<tr>
<td>(y'(t)=ab/(bt+1), c=)</td>
<td>4.570469e-01</td>
<td>4.631172e-01</td>
<td>4.442759e-01</td>
</tr>
<tr>
<td>Slope at end</td>
<td>5.571498e-06</td>
<td>5.237080e-05</td>
<td>7.133666e-04</td>
</tr>
<tr>
<td>Overlap Allan Dev [8]</td>
<td>2.922319e-01</td>
<td>9.159953e-02</td>
<td>3.241343e-02</td>
</tr>
<tr>
<td>Mod Allan Dev [7,8]</td>
<td>2.922319e-01</td>
<td>6.172376e-02</td>
<td>2.170921e-02</td>
</tr>
<tr>
<td>Time Deviation [8]</td>
<td>1.687202e-01</td>
<td>3.563623e-01</td>
<td>1.253328e-00</td>
</tr>
<tr>
<td>Hadamard Deviation</td>
<td>2.943883e-01</td>
<td>1.052754e-01</td>
<td>3.910861e-02</td>
</tr>
<tr>
<td>Overlap Had Dev</td>
<td>2.943883e-01</td>
<td>9.581083e-02</td>
<td>3.237638e-02</td>
</tr>
<tr>
<td>Hadamard Total Dev</td>
<td>2.943883e-01</td>
<td>9.614787e-02</td>
<td>3.058103e-02</td>
</tr>
<tr>
<td>Total Deviation</td>
<td>2.922319e-01</td>
<td>9.134743e-02</td>
<td>3.406530e-02</td>
</tr>
<tr>
<td>Modified Total Dev</td>
<td>2.418528e-01</td>
<td>6.499161e-02</td>
<td>2.287774e-02</td>
</tr>
<tr>
<td>Time Total Deviation</td>
<td>1.396338e-01</td>
<td>3.752293e-01</td>
<td>1.320847e-00</td>
</tr>
</tbody>
</table>
Table III Notes:
[1] Expected value = 0.5.
[2] All slopes are per interval.
[4] Exact results will depend on iterative algorithm used. Data not suited to log fit.
[5] Sample (not population) standard deviation. Expected value = $\frac{1}{\sqrt{12}} = 0.2886751$.
[8] Calculated with listed averaging factors from the basic $\tau = 1$ data set.

Table IV. Error Bars for n=1000 Point =1 Data Set with Avg Factor=10

<table>
<thead>
<tr>
<th>Allan Dev Type</th>
<th>Sigma Value</th>
<th>Conf Interval</th>
<th>$\chi^2$ for</th>
</tr>
</thead>
<tbody>
<tr>
<td>#Pts</td>
<td>Noise Type &amp; Ratio</td>
<td>Remarks</td>
<td>95% CF</td>
</tr>
<tr>
<td>99</td>
<td>W FM [1], B1=0.870</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Overlapping</td>
<td>9.159953e-02</td>
<td>Max $\sigma_0(\tau)$=1.014923e-01 [7]</td>
<td>119.07</td>
</tr>
<tr>
<td>981</td>
<td>W FM</td>
<td>Max $\sigma_0(\tau)$=1.035201e-01[8]</td>
<td>114.45</td>
</tr>
<tr>
<td></td>
<td># $\chi^2$ df=146.177</td>
<td>Min $\sigma_0(\tau)$=8.223942e-02 [8]</td>
<td>181.34</td>
</tr>
<tr>
<td>Modified [4]</td>
<td>6.172376e-02</td>
<td>Max $\sigma_0(\tau)$=7.044412e-02 [7]</td>
<td>72.64</td>
</tr>
<tr>
<td>972</td>
<td>W FM[5], R(n)=0.384</td>
<td>Max $\sigma_0(\tau)$=7.224944e-02 [8]</td>
<td>69.06</td>
</tr>
<tr>
<td></td>
<td># $\chi^2$ df=94.620</td>
<td>Min $\sigma_0(\tau)$=5.419961e-02 [8]</td>
<td>122.71</td>
</tr>
</tbody>
</table>

Table IV Notes:
[1] Theoretical B1=1.000 for W FM noise and 0.667 for F and W PM noise.
[2] Simple, noise-independent CI estimate = $\sigma_0(\tau)/\sqrt{N}=1.001594e-02$.
[3] This CI includes $\kappa(\alpha)$ factor that depends on noise type:

<table>
<thead>
<tr>
<th>Noise</th>
<th>$\alpha$</th>
<th>$\kappa(\alpha)$</th>
</tr>
</thead>
<tbody>
<tr>
<td>W PM</td>
<td>2</td>
<td>0.99</td>
</tr>
<tr>
<td>F PM</td>
<td>1</td>
<td>0.99</td>
</tr>
<tr>
<td>W FM</td>
<td>0</td>
<td>0.87</td>
</tr>
<tr>
<td>F FM</td>
<td>-1</td>
<td>0.77</td>
</tr>
<tr>
<td>RW FM</td>
<td>-2</td>
<td>0.75</td>
</tr>
</tbody>
</table>

[4] BW factor $2\pi f_0 \tau_0 = 10$. Applies only to F PM noise.
[5] Theoretical R(n) for W FM noise = 0.500 and 0.262 for F PM noise.
[6] Double-sided 68 % confidence interval: p = 0.158 and 0.842.
[7] Single-sided 95 % confidence interval: p = 0.950.
[8] Double-sided 95 % confidence interval: p = 0.025 and 0.975.

12.5. IEEE Standard 1139-1999

References for Software

13 Glossary

The following terms are used in the field of frequency stability analysis

**Aging**
The change in frequency with time due to internal effects within the device.

**Allan Variance**
The two-sample variance $\sigma^2(\tau)$ commonly used to measure frequency stability.

**Averaging**
The process of combining phase or frequency data into samples at a longer averaging time.

**Averaging Time**
See Tau.

**BW**
Bandwidth, hertz.

**Confidence Limit**
The uncertainty associated with a measurement. Often a 68% confidence level or error bar.

**Drift**
The change in frequency with time due to all effects (including aging and environmental sensitivity).

**Frequency Data**
A set of fractional frequency values, $y[i]$, where $i$ denotes equally-spaced time samples.

**Hadamard Variance**
A three-sample variance, HVAR, that is similar to the two-sample Allan variance. It uses the second differences of the fractional frequencies, and is unaffected by linear frequency drift.

£($f$)
\[ £(f) = 10 \cdot \log \left( \frac{1}{2} \cdot S_{\phi}(f) \right), \]
the ratio of the SSB phase noise power in a 1 Hz BW to the total carrier power, dBC/Hz. Valid when noise power is much smaller than the carrier power.

**MJD**
The Modified Julian Date is based on the astronomical Julian Date, the number of days since noon on January 1, 4713 BC. The MJD is the Julian Date - 2,4000,000.5.

**Modified Sigma**
A modified version of the Allan or total variance that uses phase averaging to distinguish between white and flicker PM noise processes.

**MTIE**
The maximum time interval error of a clock.

**Normalize**
To remove the average value from phase or frequency data.

**Phase Data**
A set of time deviates, $x[i]$ with units of seconds, where $i$ denotes equally-spaced time samples. Called “phase” data to distinguish them from the independent time variable.
<table>
<thead>
<tr>
<th>Term</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Phase Noise</strong></td>
<td>The spectral density of the phase deviations.</td>
</tr>
<tr>
<td><strong>Sampling Time</strong></td>
<td>See Tau.</td>
</tr>
<tr>
<td><strong>Sigma</strong></td>
<td>The square root or deviation of a variance, often the two-sample or Allan deviation, $\sigma_y(\tau)$.</td>
</tr>
<tr>
<td><strong>Slope</strong></td>
<td>The change in frequency per tau interval.</td>
</tr>
<tr>
<td><strong>SSB</strong></td>
<td>Single sideband.</td>
</tr>
<tr>
<td><strong>$S_x(f)$</strong></td>
<td>The one-sided spectral density of the phase deviations, rad$^2$/Hz.</td>
</tr>
<tr>
<td><strong>$S_x(f)$</strong></td>
<td>The one-sided spectral density of the time deviations, sec$^2$/Hz.</td>
</tr>
<tr>
<td><strong>$S_y(f)$</strong></td>
<td>The one-sided spectral density of the fractional frequency deviations, 1/Hz.</td>
</tr>
<tr>
<td><strong>Tau</strong></td>
<td>The interval between phase measurements or the averaging time used for a frequency measurement.</td>
</tr>
<tr>
<td><strong>Total</strong></td>
<td>A variance using an extended data set that provides better confidence at long averaging times.</td>
</tr>
<tr>
<td><strong>Théo1</strong></td>
<td>Theoretical variance #1, a variance providing stability data out to 75 % of the record length.</td>
</tr>
<tr>
<td><strong>ThéoBR</strong></td>
<td>A biased removed version of Théo1.</td>
</tr>
<tr>
<td><strong>ThéoH</strong></td>
<td>A hybrid combination of ThéoBR and the Allan variance.</td>
</tr>
<tr>
<td><strong>TIE</strong></td>
<td>The time interval error of a clock. Can be expressed as the rms time interval error TIE rms or the maximum time interval error MTIE.</td>
</tr>
<tr>
<td><strong>Total Variance</strong></td>
<td>A two-sample variance similar to the Allan variance with improved confidence at large averaging factors.</td>
</tr>
<tr>
<td><strong>$x(t)$</strong></td>
<td>The instantaneous time deviation from a nominal time, $x(t) = \phi(t)/2\pi v_0$, seconds, where $v_0$ is the nominal frequency, hertz. This dependent time variable is often called &quot;phase&quot; to distinguish it from the independent time variable $t$.</td>
</tr>
<tr>
<td><strong>$y(t)$</strong></td>
<td>The instantaneous fractional frequency deviation from a nominal frequency, $y(t) = [v(t)−v_0]/v_0 = x'(t)$, where $v_0$ is the nominal frequency.</td>
</tr>
</tbody>
</table>
References for Glossary

14 Bibliography

- Notes
  1. These references are arranged by topic and sorted by date.
  2. All of the FCS and UFFC references are available on-line for IEEE UFFC members.
  3. Only the 1990 and later PTTI proceedings are now available on-line.
  4. None of the EFTF or older IEEE I&M proceedings are available on-line.
  5. NIST papers are available on-line at http://tf.nist.gov/cgi-bin/showpubs.pl.

- General

• **Standards and Specifications**
  2. MIL-PRF-55310, Oscillators, Crystal, General Specification For.

• **Classic (Pre-Allan variance)**

• **Allan Variance**

- **Modified Allan Variance**

- **Time Variance and Time Error Prediction**

- **Hadamard Variance**
• Modified Hadamard Variance

• Total Variance

• Modified Total Variance

• Time Total Variance

• Thêo1, ThêoBR, and ThêoH

- **MTIE**

- **Multi-Variance Analysis**

- **Dynamic Stability**
• **Confidence Intervals**
  7. F. Vernotte and M. Vincent, "Estimation of the Uncertainty of a Mean Frequency Measurement", 

• **Drift Estimation and Removal**
  6. L.A. Breakiron, “A Comparative Study of Clock Rate and Drift Estimation”, 

- **Bias Functions**
  3. W.J. Riley, "Confidence Intervals and Bias Corrections for the Stable32 Variance Functions", Hamilton Technical Services

- **Noise Identification**

- **Simulation**
• **Dead Time**

• **3-Cornered Hat**

• **Domain Conversions**


**Robust Statistics**


**Computation and Algorithms**


**Measurements**


6. Data Sheet, Model 5110A Time Interval Analyzer, Symmetricom, Inc. (formerly Timing Solutions Corp.), Boulder, CO 80301 USA.


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