

WP9.5: Phase Monitoring

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Phase noise basics (1)

- Phase noise is usually given in terms of dBc/Hz for oscillators and associated devices.
- That is, we are given the power in the components of the noise spectrum relative to the carrier.
- In order to find the total noise of the system, this noise needs to be integrated over the frequencies of interest.

Phase noise basics (2)

- Suppose we have an oscillator outputting some noisy signal:

$$V(t) = \sin(\omega_0 t + \phi_N(t))$$

$$V(t) = \sin(\omega_0 t) \cos(\phi_N(t)) + \cos(\omega_0 t) \sin(\phi_N(t))$$

$$V(t) \approx \sin(\omega_0 t) + \cos(\omega_0 t) \phi_N(t)$$

- The phase noise, $\phi_N(t)$, can be rewritten as an integral:

$$\phi_N(t) = \int a_\omega \sin(\omega t + \theta_\omega) d\omega$$

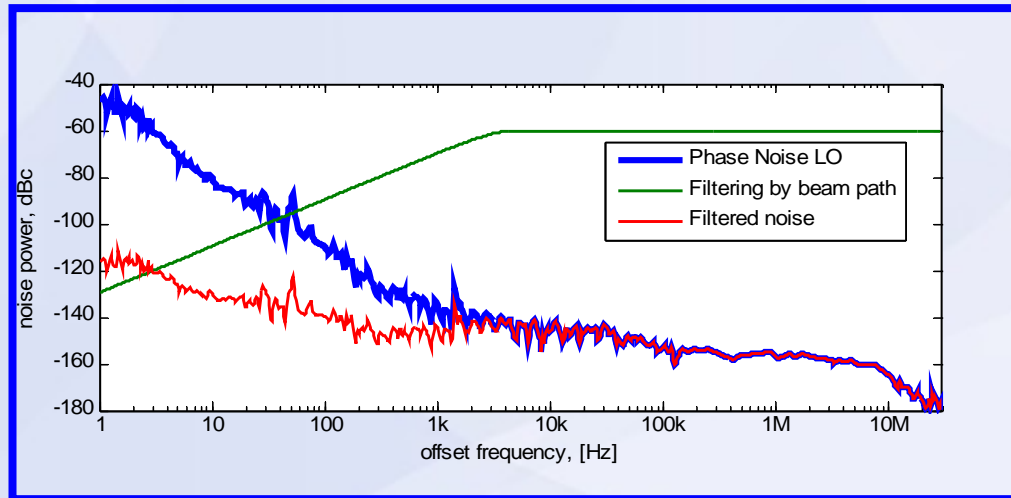
- The power in one of these components, compared to the power in the carrier, is just a_ω^2 . The noise power spectrum is just these coefficients:

$$N_P(\omega) = 10 \log_{10}(a_\omega^2)$$

Phase noise basics (3)

- The upper bound of our for the frequencies we include in our noise calculation will be given by the lowest bandwidth in the oscillator signal path.
- Good oscillators very quickly get down to the thermal noise floor (-174 dBm) where filtering can no longer help.
- The lower frequency bound will be given by how long we need to keep accurate time for.
- The lower bound is usually much more important as we find higher noise power the closer we get to the carrier frequency.
- Keeping time accurately for a long time is *difficult!*

Phase noise example



- We see that the phase noise increases significantly towards low frequencies.
- In the CLIC case, we know that we have to keep time for at most $160 \mu\text{s}$.
- This will impose a high pass filtering function on the system

Total noise

- To calculate total noise:
- Take our noise function, $N_P(\omega)$, and the system filter function $H^2(\omega)$
- Then calculate $N_{total} = \sqrt{\int_{\omega_L}^{\omega_U} 10^{H^2(\omega)N_P(\omega)/10} d\omega}$
- This is the RMS voltage noise, compared to the 1 we get from the carrier
- This is thus directly the phase noise in Radians
- If we would like to know it in terms of time, compute: $\frac{T N_{total}}{2\pi}$ where T is the period of the oscillator

How to improve noise performance

- Averaging, averaging, averaging
- Must ensure that the various copies of the signal are added together in phase
- Noise add in squares
- Signal to noise ratio improvement goes with the square-root of the number of devices.

AM-PM conversion

- If we are trying to build a high resolution phase measuring system, it is not just noise that concern us. We must also be able to treat the RF containing phase information with sufficiently low distortion.

| Carrier | Phase Modulation (quadrature) | Amplitude Modulation (inline) |
|--------------------|---|---------------------------------------|
| $\sin(\omega_0 t)$ | $\varphi_0 \sin(\omega_d t) \cos(\omega_0 t)$ | $a \cos(\omega_d t) \sin(\omega_0 t)$ |

$$\begin{array}{cc}
 \swarrow & \searrow \\
 -\frac{\varphi_0}{2} \{ \sin([\omega_0 + \omega_d]t) - \sin([\omega_0 - \omega_d]t) \} & \frac{a}{2} \{ \sin([\omega_0 + \omega_d]t) + \sin([\omega_0 - \omega_d]t) \}
 \end{array}$$

AM-PM amplitude flatness

- Suppose we have a device that affect the amplitude of the two frequencies differently:

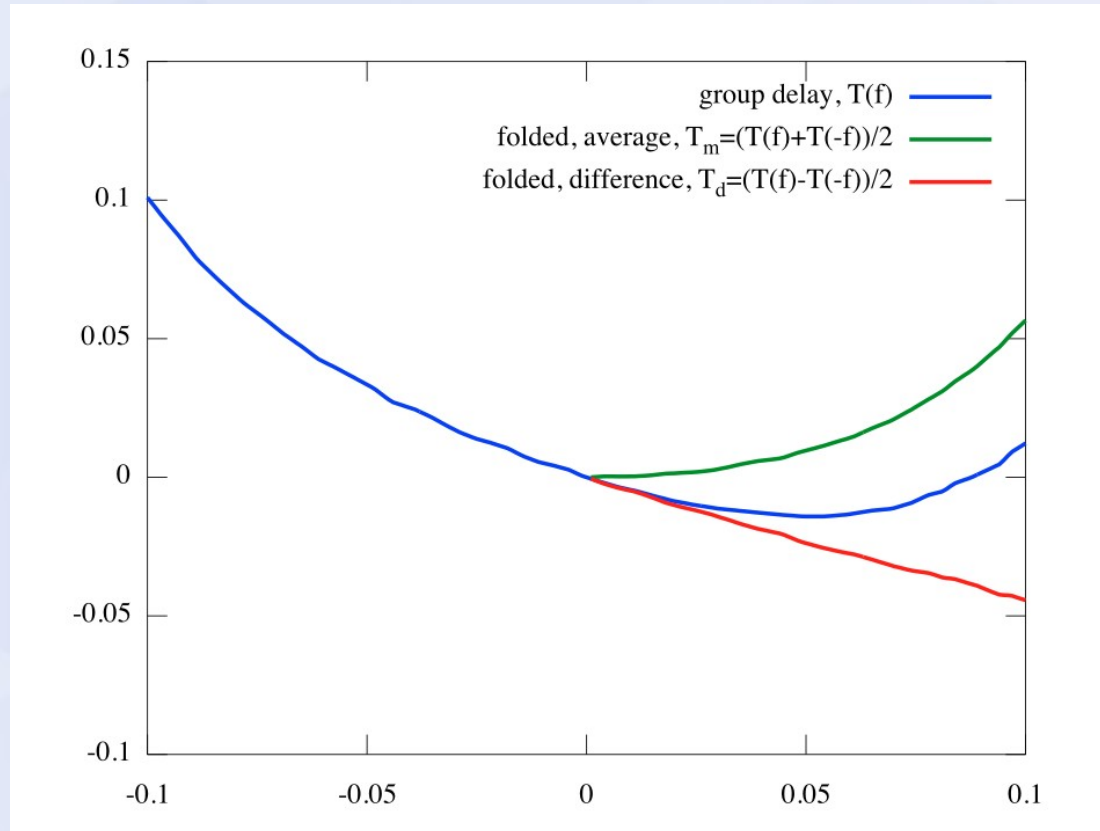
$$a_1 \sin([\omega_0 + \omega_d]t) + a_2 \sin([\omega_0 - \omega_d]t) \quad \text{Rewrite as sums and diffs}$$

$$\frac{a_1 + a_2}{2} \sin([\omega_0 + \omega_d]t) + \frac{a_1 + a_2}{2} \sin([\omega_0 - \omega_d]t) \quad \text{Amplitude Modulation}$$

$$+ \frac{a_1 - a_2}{2} \sin([\omega_0 + \omega_d]t) - \frac{a_1 - a_2}{2} \sin([\omega_0 - \omega_d]t) \quad \text{Phase Modulation}$$

- Phase modulation thus induced scales with the modulation depth and the defects in device amplitude flatness.
- There is a similar phase to amplitude conversion, which will scale the phase modulation by some amount

AM-PM dispersion



AM-PM dispersion

- Suppose we have a device that is dispersive, so that the frequency components of an amplitude modulation have different transit times

$$\sin(\omega_0 t) + a \{ \sin([\omega_0 + \omega_d][t + T_m(\omega_d) + T_d(\omega_d)]) + \sin([\omega_0 - \omega_d][t + T_m(\omega_d) - T_d(\omega_d)]) \}$$

- Rewrite

$$\sin(\omega_0 t)$$

$$+ a \cos(T_m \omega_0 + T_d \omega_d) \cos(T_m \omega_d + T_d \omega_0 + \omega_d t) \sin(\omega_0 t) \text{ Amplitude Mod.}$$

$$+ a \sin(T_m \omega_0 + T_d \omega_d) \cos(T_m \omega_d + T_d \omega_0 + \omega_d t) \cos(\omega_0 t) \text{ Phase Modulation}$$

- The AM-PM conversion is thus $a \sin(T_m \omega_0)$ with $\omega_0 \gg \omega_d$

that is, proportional to the size of the amplitude modulation and the average transit time difference of the two modulation components

AM-PM dispersion (2)

- The resulting modulation term

$$\cos(T_m \omega_d + T_d \omega_0 + \omega_d t) \approx \cos(T_d \omega_0 + \omega_d t)$$

causes distortion of the signal envelope

- T_d is of course dependent on the modulation frequency ω_d . The linear part of $T_d(\omega_d)$ causes only a time-shift of the modulation, any non-linear part distorts the signal
- The phase modulation term works exactly the same way. That is, some phase modulation will turn into amplitude modulation, and some distortion of the phase signal will occur

AM-PM – some numbers

- Devices can be specified to 0.1dB flatness, but not 0.01dB. (Influence of connectors and cable become too large for reliable measurements to be made)
- With 0.1 dB flatness, we get 0.0033° per percent of amplitude modulation
- 0.1 dB flatness will scale a phase modulation by 0.5%
- At 12 GHz, with 1% amplitude modulation, we get 0.04° per ps

AM-PM device non-linearities

$$A \sin(\omega_1 t) + B \sin(\omega_2 t)$$

$$\Rightarrow A \sin(\omega_1 t) + a_2 A^2 \sin^2(\omega_1 t) + a_3 A^3 \sin^3(\omega_1 t) + \dots$$

$$+ B \sin(\omega_2 t) + b_2 B^2 \sin^2(\omega_2 t) + b_3 B^3 \sin^3(\omega_2 t) + \dots$$

$$+ A B c_{1,1} \sin(\omega_1 t) \sin(\omega_2 t) + A^2 B c_{2,1} \sin^2(\omega_1 t) \sin(\omega_2 t) + A B^2 c_{1,2} \sin(\omega_1 t) \sin(\omega_2 t)$$

$$\omega_1, \omega_2, 2\omega_1, 2\omega_2, 2\omega_1 - \omega_2, \omega_1 - 2\omega_2, \dots$$

- By device non-linearities we get a large set of frequencies in the output that were not in the input.
- These grow with the amplitude to the power of term order of the input signals.
- By lowering input power we can have a smaller fraction of the output of these frequencies

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AM-PM device non-linearities

- In phase detection devices, there are in particular a second order term that mixes to baseband and is indistinguishable from real phase.
- We must thus lower input power until this contribution becomes small enough.
- This reduces our signal to noise ratio.
- Recover better SNR by using multiple devices in parallel and averaging their output.
- We get an improvement of square root of N for N devices.