

# Mathematical Appendix to Rohde & Schwarz Application Note 1GP45

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## Oversampling for ARB with Interpolation Filter

Let  $W_I$  be the bandwidth of the interpolation filter and  $W_S$  the bandwidth of the modulated signal. To avoid cutting the signal with the interpolation filter:

$$W_I \geq W_S \quad (1)$$

This equation can be written as:

$$\frac{O \cdot f_{sym}}{f_{sample}} \cdot W_I \geq W_S \quad (2)$$

with  $f_{sample}$  being the sample rate,  $O$  the oversampling factor and  $f_{sym}$  the sample rate of the signal. (Remember that  $f_{sample} = O \cdot f_{sym}$ ) This gives:

$$O \cdot \frac{W_I}{f_{sample}} \geq \frac{W_S}{f_{sym}} \quad (3)$$

For a W-CDMA signal with a  $\sqrt{\cos}$  filter,  $\alpha = 0.22$ :

$$\frac{W_S}{f_{sym}} = \frac{1 + \alpha}{2} = 0.61 \quad (4)$$

The interpolation filter has the standardized bandwidth:

$$\frac{W_I}{f_{sample}} = 0.375 \quad (5)$$

This gives:

$$O \geq \frac{0.61}{0.375} = 1.63 \quad (6)$$

## Effect of non-ideal I/Q Signals

We will discuss this for a single CW carrier with an offset from the RF center frequency, i.e. at  $\omega_0 + \omega_M$ .

### Ideal I/Q Signal

The ideal I/Q signal for this scenario is:

$$I(t) = \cos \omega_M t \quad (7)$$

$$Q(t) = \sin \omega_M t \quad (8)$$

Then - if we assume that the I/Q modulator itself is ideal - the modulated RF signal will be:

$$\begin{aligned} s(t) &= \Re \{ (I(t) + iQ(t)) e^{i\omega_0 t} \} \\ &= \cos \omega_M t \cdot \cos \omega_0 t - \sin \omega_M t \cdot \sin \omega_0 t \\ &= \cos (\omega_0 + \omega_M) t \end{aligned} \quad (9)$$

### Non-ideal I/Q signal

In reality both the I/Q Modulator and the I/Q input signal are not ideal. This can be described as small deviations in amplitude and phase of the Q signal:

$$I(t) = \cos \omega_M t \quad (10)$$

$$Q(t) = (1 + \epsilon) \sin (\omega_M t + \varphi) \quad (11)$$

with  $\epsilon \ll 1, \varphi \ll 1$ . For  $\varphi$  the following approximations are valid:

$$\sin \varphi \approx \varphi, \quad \cos \varphi \approx 1 \quad (12)$$

$\epsilon$  can result either from different magnitudes for I and Q of the input signal, or from different gain in the I and Q channel of the modulator.  $\varphi$  can be caused by an I/Q modulator with quadrature error (phase between I and Q channel is not 90 degrees), or by a delay between I and Q of the input signal. In the second case, the resulting *varphi* depends on the frequency of the input signal, according to

$$Q(t) = (1 + \epsilon) \sin (\omega_M (t + \Delta t)) \quad (13)$$

$$= (1 + \epsilon) \sin (\omega_M t + \varphi) \quad (14)$$

with

$$\varphi = \omega_M \Delta t \quad (15)$$

Expanding  $Q(t)$  and using (12) gives:

$$Q(t) = \sin \omega_M t + \varphi \cos \omega_M t + \epsilon \sin \omega_M t + \varphi \epsilon \cos \omega_M t \quad (16)$$

The last term in (16) is of second order nature and can be neglected, so:

$$Q(t) = \sin \omega_M t + \varphi \cos \omega_M t + \epsilon \sin \omega_M t \quad (17)$$

The RF signal is again calculated with:

$$s(t) = \Re \{ (I(t) + iQ(t)) e^{i\omega_0 t} \} \quad (18)$$

This leads to the following result :

$$\begin{aligned} s(t) &= \cos(\omega_0 + \omega_M) t \\ &\quad - (\varphi \cos \omega_M t + \epsilon \sin \omega_M t) \sin \omega_0 t \end{aligned} \quad (19)$$

which can be written as:

$$\begin{aligned} s(t) &= \cos(\omega_0 + \omega_M) t \\ &\quad - \frac{\varphi}{2} [\sin(\omega_0 + \omega_M) t + \sin(\omega_0 - \omega_M) t] \\ &\quad + \frac{\epsilon}{2} [\cos(\omega_0 + \omega_M) t - \cos(\omega_0 - \omega_M) t] \end{aligned} \quad (20)$$

The first terms in the second and third row can be neglected compared to the undisturbed signal (first row), especially if the signal is measured with a spectrum analyzer that usually has a logarithmic scale. Thus:

$$\begin{aligned} s(t) &= \cos(\omega_0 + \omega_M) t - \frac{\varphi}{2} \sin(\omega_0 - \omega_M) t - \frac{\epsilon}{2} \cos(\omega_0 - \omega_M) t \\ &= \cos(\omega_0 + \omega_M) t - A \sin[(\omega_0 - \omega_M) t + \phi] \end{aligned} \quad (21)$$

with:

$$A = \frac{1}{2} \sqrt{\epsilon^2 + \varphi^2} \quad (22)$$

$$\tan \phi = \frac{\varphi}{\epsilon} \quad (23)$$

Note that the disturbances increase with increasing frequency of the input signal if a delay between I and Q of the input signal is present. With (15) follows

$$A = \frac{1}{2} \sqrt{\epsilon^2 + \omega_M^2 \cdot \Delta t^2} \quad (24)$$

$$\tan \phi = \frac{\omega_M \Delta t}{\epsilon} \quad (25)$$