

A Flexible Volterra-Based Adaptive Digital Pre-Distortion Solution for Wideband RF Power Amplifier Linearization

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Base Stations/PA's



\mathbf{R} E A L \mathbf{W} O R L D \mathbf{S} I G N A L \mathbf{P} R O C E S S I N \mathbf{G}^{M}

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The present 3G and other emerging air interfaces use non-constant envelope modulation schemes and are spectrally more efficient than their predecessors

Problem: This technique causes high PAR, necessitating higher PA back-off. This leads to decrease in PA efficiency and increase in cooling and operational costs of a base-station.



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Solution: Drive the PA harder to get more power

Added Problem: Signal distortion occurs

<u>Ultimate Solution</u>: Predict the type of distortion, pre-distort the signal in a reverse manner

<u>Result</u>: Distortion is cancelled out. This extends the linear region of the operation range and produces more output power at an efficiency approaching 40%. Now a smaller amplifier at higher efficiency can be used with DPD to achieve the desired output power

Introduction



•DPD (Digital Pre-Distortion) improves efficiency of PA's
•Most PA's are LDMOS class AB designs and rarely achieve 10% efficiency

•This inefficiency is inherent in the class AB design but also is a result of having to reduce the PA output to deal with signals that exhibit high PAR (Crest Factor) power and to prevent distortion that results in adjacent channel power leakage



Classes of operation of Power amplifier based on transfer characteristics

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Typical PA improvement from this DPD solution

•Reduce PAR's (or Crest Factor) for 3G signals up to 6 dB

•Reduce PAR's (or Crest Factor) for OFDM signals by up to 4 dB

•All while meeting ACPR (Adjacent Channel Power Ratio) and EVM (Error Vector Magnitude) specs

•Correct for up to 11th order non-linearities and PA (Power Amplifier) memory effects up to 200 ns

•Greater than 20 dB ACPR improvement

•Over 4X increase in power efficiency

•As much as 60% reduction in static power consumption



The crest factor or peak-to-average ratio (PAR) or peak-toaverage power ratio (PAPR) is a measurement of a waveform, calculated from the peak amplitude of the waveform divided by the RMS (time-averaged) value of the waveform.

 $C = \frac{|\mathbf{x}| \text{ peak}}{\mathbf{x} \text{ RMS}}$

It is therefore a dimensionless value. While this quotient is most simply expressed by a positive rational number, as shown below, in commercial products it is also commonly stated as the ratio of two whole numbers, e.g., 2:1.

The minimum possible crest factor is 1.



DC voltages have a crest factor of 1 since the RMS and the peak amplitude are equal, and it is the same for a square wave (of 50% duty cycle).





PAR/Crest Factor

This table provides values for some other normalized waveforms:

Wave type	Crest factor (dB)
DC	0.00 dB
Sine wave	3.01 dB
Full-wave rectified sine	3.01 dB
Half-wave rectified sine	6.02 dB
Triangle wave	4.77 dB
Square wave	0.00 dB
QPSK	3.5 - 4 dB
64 QAM	7.7 dB
128 QAM	8.2 dB
WCDMA downlink carrier	10.6 dB



Notes:

1. Crest factors specified for QPSK, QAM, WCDMA are typical factors needed for reliable communication, not the theoretical crest factors which can be larger.

2. Waveform factor is the ratio of DC average to RMS and is used to scale resistors for measurements with DC or AC meters. The waveform factor for the half wave rectified sine wave should be 2.22 as the DC average is VP/Pi



A generalized look up table can be used for pre-distorter gain/phase correction if no memory effects are taken into consideration

Thus we are able to characterize a PA by: 1. Amplitude or AM-to-AM (or Gain Compression) 2. Phase Transfer or AM-to-PM



Performance Analyses of Efficiency Enhancement Techniques of PA's



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Figure 1: Gain compression and AM-PM characteristics for a typical Doherty PA



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Gain and Phase of PA's change with: •Temperature •Voltage •Component ageing

This requires an adaptive control of look-up tables for effective linearization

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•Volterra series and Theorem developed by Vito Volterra in 1887

•It is used to predict non-linear response of a system to a given input

•Similar to Taylor series but Volterra has ability to capture "memory" effects



Y(n) = Y1(n) + Y2(n) + Y3(n) + Y4(n) + Y5(n) + ... + v(n) (1) Where, $Y1(n) = \sum^{i=0:M1} h1(i).x(n-i)$

 $Y2(n) = \sum^{i_{1}=0:M_{2}} \sum^{i_{2}=0:M_{2}} h2(i_{1},i_{2}).x(n-i_{1}).x(n-i_{2})$

Y3(n) = ∑^{i1=0:M3} ∑^{i2=0:M3} ∑^{i3=0:M3} h3(i1,i2,i3).x(n-i1).x(n-i2).x(n-i3)

Y4(n) = ∑^{i1=0:M4} ∑^{i2=0:M4} ∑^{i3=0:M4} ∑^{i4=0:M4} h4(i1,i2,i3, i4).x(n-i1).x(n-i2).x(n-i3).x(n-i4)

 $Y5(n) = \sum_{i=0:M5} \sum$

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Simplify



WOW!! We need to simplify this!

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Memory Polynomial Model

This technique constrains the Volterra Series so that everything except the diagonal terms in the kernels are zero, thus giving a memory polynomial model:

 $\mathbf{Y}(\mathbf{n}) = \sum_{k=0:K} \sum_{i=0:M} \mathbf{h} \mathbf{k}(i) \cdot \mathbf{x}(\mathbf{n}-i) |\mathbf{x}(\mathbf{n}-i)|^k$

This simplification method has been proven to effectively model PA: 1. Thermal effects 2. Active matching network 3. Bias circuits due to slowly varying, non-constant amplitude of PA input signal



Figure 2: DPD System Diagram



A complete digital pre-distortion transmit signal chain with the GC5322SEK (System Evaluation Kit) based on TI high-speed analog and DSP technology



Figure 2A: GC5322 Diagram



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Linear Equalizer Non-Linear DPD Feedback Non-Linear Compensator and Smart Capture Buffers

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We used a combination of algorithmic and model reduction approaches:

- 1. The number of terms in (1) significantly reduced by eliminating redundancies associated with various index permutations.
- 2. Volterra coefficients assumed to be symmetric
- 3. Real input signal to the PA x(n) expressed in terms of its complex baseband representation significantly reducing the number of terms. For band-limited systems we are only interested in frequency components close to the carrier frequency fo
- 4. Even order inter-modulation terms lie far away from frequency band of interest, allowing us to further drop half the terms in (1)



Equation Reduction/Simplification Techniques Used (cont'd)

We used a combination of algorithmic and model reduction approaches:

- The model is rotationally invariant, this simplifies things since a phase shift on the input of the PA produces exactly the same phase shift on the output. This allows (1) to be reduced to terms involving products of the signal and powers of its magnitude squared.
- 6. The PA is causal, so we assume the linear portion of the PA is minimum phase. This further restricts Volterra terms
- 7. Since PA implementations perform the processing in stages, this also helps simplify the model into cascade sections with each matched to the needs of compensating the distortions induced by the particular PA stage. 25



We get the model for the Linear Equalizer block by restricting the Volterra Series to only linear terms with memory M1:

 $Y1(n) = \sum_{i=0:M1}^{i=0:M1} h1(i).x(n-i)$

(2)

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We get the Non-Linear DPD block by restricting the Volterra Series to only the non-linear terms with memory M2, and dropping even terms we get:

$$\begin{split} Y(n) &= \sum_{i=0:M2}^{i=0:M2} h3(i,i,i).x(n-i).|x(n-i)|^2 + \\ &\sum_{i=0:M2}^{i=0:M2} h5(i,i,i,i,i).x(n-i).|x(n-i)|^4 + \\ &\sum_{i=0:M2}^{i=0:M2} h7(i,i,i,i,i,i,i).x(n-i).|x(n-i)|^6 \end{split}$$



+ other higher order terms depending on the polynomial modeling accuracy requirements of the adaptation algorithm. (3)



Rearranging Equation 3



Rearranging terms in Equation 3 gives:

$$\begin{split} Y(n) &= \sum_{i=0:M^2} \{ h3(i,i,i) | x(n-i)|^2 + h5(i,i,i,i,i) [|x(n-i)|^2]^2 + h7(i,i,i,i,i,i,i,i,i) [|x(n-i)|^2]^3 + higher order terms \}.x(n-i) \\ &= \sum_{i=0:M^2} LUT(|x(n-i)|^2).x(n-i) \end{split}$$

Equation 4

Simplifying Equation 1 Volterra Series terms:

$$\begin{split} Y(n) &= \sum^{i=0:M3} h3(i,i,0).|x(n-i)|^{2}.x(n-i) + \\ &\sum^{i=0:M3} h5(i,i,0,0,0).|x(n-i)|^{2}.|x(n)|^{2}.x(n) + \\ &\sum^{i=0:M3} h5(i,i,i,i,0).|x(n-i)|^{4}.x(n) + \\ &\sum^{i=0:M3}, i \neq j \sum^{j=0:M3} h5(i,i,j,j,0).|x(n-i)|^{2}.|x(n-j)|^{2}.x(n) + \\ &\sum^{i=0:M3} h7(i,i,0,0,0,0,0).|x(n-i)|^{2}.|x(n)|^{4}.x(n) + \\ &\sum^{i=0:M3} h7(i,i,i,i,0,0,0).|x(n-i)|^{4}.|x(n)|^{2}.x(n) + \\ &\sum^{i=0:M3} h7(i,i,i,j,0,0,0).|x(n-i)|^{2}.|x(n-j)|^{2}.|x(n)|^{2}.x(n) + \\ &\sum^{i=0:M3} i \neq j \sum^{j=0:M3} h7(i,i,j,j,0,0,0).|x(n-i)|^{2}.|x(n-j)|^{2}.x(n) + \\ &\sum^{i=0:M3}, i \neq j \sum^{j=0:M3} h7(i,i,i,j,j,0).|x(n-i)|^{4}.|x(n-j)|^{2}.x(n) + \\ &\sum^{i=0:M3}, i \neq j \sum^{j=0:M3} h7(i,i,j,j,0).|x(n-i)|^{4}.|x(n-j)|^{2}.x(n) + \\ &\sum^{i=0:M3}, i \neq j \sum^{j=0:M3} h7(i,j,j,j,0).|x(n-i)|^{4}.|x(n-j)|^{2}.x(n) + \\ &\sum^{i=0:M3}, i \neq j \sum^{j=0:M3} h7(i,j,j,j,0).|x(n-i)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{2}.x(n) + \\ &\sum^{i=0:M3}, i \neq j \sum^{j=0:M3} h7(i,j,j,j,0).|x(n-i)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n-j)|^{4}.|x(n$$



Third Stage---Feedback Non-Linear Compensator and Smart Capture Buffers

Feedback signal from PA used to compute the instantaneous error, which along with reference transmit signal can be captured by a pair of on-chip memories.

DSP processor reads back these captured signals and implements the adaptation algorithms for the pre-distorter blocks.



A Direct Learning architecture is used in the pre-distortion algorithm implemented on the DSP.

A model of the pre-distorter is maintained in software---its parameters optimized to minimize the error signal captured in the hardware.



Figure 3: GC5322 evaluation platform system diagram



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Table 1: GC5322 evaluation platform system parameters

Evaluation System Configuration					
RF Card Version	WiMax	WCDMA			
DUC Input Sample Rate	11.2 MSPS (WiMax)	3.84MSPS (WCDMA), 1.28MSPS(TD-SCDMA), 4.333MSPS (MC-GSM), 30.72MSPS (LTE)			
CFR Sample Rate	67.2MSPS (WiMax)	61.44MSPS (WCDMA, TD- SCDMA, LTE), 69.333MSPS (MC-GSM)			
DPD Sample Rate	112 M SPS	122.88MSPS			
DAC	DAC 5682 @ 672 M Hz Complex	DAC 5682 @ 737.28MHz Complex			
ADC	ADS5444 @ 224MHz Real	ADS5444 @ 245.76MHz Real			
IF Frequency	168MHz	184.32 M Hz			
RF Frequency	2.123GHz	2.139GHz			
IQ Modulator	TRF3703				
Mixer	HM C 2 14				
LO	TRF3761				
PLL	CDCM7005				

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Figure 4-A: Adjacent channel ACLR Vs. Pout at different PAR levels & test signals, pre & post DPD



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Figure 4-B: Alternate channel ACLR Vs. Pout at different PAR levels & test signals, pre & post DPD



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Figure 5-A: Pre-DPD spectrum at 46.75dBm Pout and 6dB PAR (TM1-64 data)



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Figure 5-B: Post-DPD spectrum at 46.75dBm Pout and 6dB PAR (TM1-64 data)



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Figure 6-A: PCDE Vs. Pout at different PAR levels and test signals, pre and post DPD



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Figure 6-B: EVM Vs. Pout at different PAR levels and test signals, pre and post DPD



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Figure 7: PA drain power efficiency Vs. output power



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Figure 8-A: WCDMA: Pre-DPD spectrum at 42.75dBm Pout and 6dB PAR (TM1-64 data)





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Figure 8-B: WCDMA: Post-DPD spectrum at 42.75dBm Pout and 6dB PAR(TM1-64 data)



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Figure 9-A: WiMax: Pre (red) and post (blue) DPD spectrums at 43.75dBm Pout and 8.5dB PAR



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Figure 9-B: WiMax: Clockwise from top left: Pre-DPD Constellation, Post-DPD Constellation, Post-DPD Error vector spectrum, Pre-DPD Error Vector Spectrum plots for 43.75dBm Pout, 8.5dB PAR



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Figure 10-A: TD-SCDMA: Pre (blue) and post (green) DPD spectral plots at 46dBm Pout and 8dB PAR



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Figure 10-B: TD-SCDMA: Pre-DPD (left) and post-DPD (right) Constellation plots for 46dBm Pout, 8dB PAR



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Figure 11: MC-GSM: Pre (blue) and post (green) DPD spectral plots at 42dBm Pout and 6.3dB PAR



1	-22.71 dBm /	30.00 kHz	600.0 kHz	30.00 kHz	-37.00	-59.70	-36.53	-59.24
2	-22.64 dBm /	30.00 kHz	800.0 kHz	30.00 kHz	-46.38	-69.09	-46.17	-68.88
3	-22.54 dBm /	30.00 kHz	1.000 MHz	30.00 kHz	-48.99	-71.70	-48.27	-70.97
4	-22.44 dBm /	30.00 kHz	1.200 MHz	30.00 kHz	-40.20	-62.90	-38.92	-61.63
			1.800 MHz	100.0 kHz	-39.76	-62.47	-38.57	-61.27
			6.000 MHz	100.0 kHz	-72.22	-94.93	-71.31	-94.02
_		() ()						

Post - DPD							
Total Carrier Power -16.370 dBm/ 120.00 kHz		ACP-IBW		RRC Filter : Off			
Carrier Power		Offset Freq	Lower Integ BW dBc dB		wer dBm	Upper dBc dBm	
1	-22.36 dBm / 30.00 kHz	600.0 kHz	30.00 kHz	-70.66	-92.04	-70.51	-92.89
2	-22.37 dBm / 30.00 kHz	800.0 kHz	30.00 kHz	-72.27	-94.66	-72.84	-95.22
3	-22.38 dBm / 30.00 kHz	1.000 MHz	30.00 kHz	-71.15	-93.54	-70.53	-92.92
4	-22.46 dBm / 30.00 kHz	1.200 MHz	30.00 kHz	-70.50	-92.88	-70.34	-92.73
		1.800 MHz	100.0 kHz	-67.24	-89.62	-66.98	-89.37
		6.000 MHz	100.0 kHz	-72.41	-94.79	-72.25	-94.64

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Figure 12: LTE: Pre(red) and post(blue) DPD spectrums at 43.5dBm Pout and 7.5dB PAR



The pre-distortion scheme presented here is shown to be highly efficient at improving amplifier linearity and power efficiency.

The GC5322 integrated transmit solution presented here not only provides a significant environmental benefit, but also provides a substantial cost savings both in capital expenditure and operational expenditure for next generation base stations.

By providing an integrated DUC-CFR-DPD signal processing hardware solution, along with optimized DSP-based adaptation software and a proven reference RF board design, faster time to market can be achieved.