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Digital Predistortion for High Efficiency Power Amplifier Architectures Using a Dual-input Modeling Approach

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Abstract—In this paper, a novel model is proposed for dual-input high efficiency power amplifier (PA) architectures, such as envelope tracking (ET) and varactor-based dynamic load modulation (DLM). Compared to the traditional single-input modeling approach, the proposed model incorporates the baseband supply voltage/load control as an input. This advantage makes the new approach capable to achieve maximized average power-added efficiency (PAE) and minimized output distortion simultaneously. Furthermore, the new approach has shown to be robust towards time misalignment between the RF input and baseband supply voltage/load control signals, and it can be applied with a reduced-bandwidth baseband supply voltage/load control.

Experiments have been performed in a varactor-based DLM PA architecture to evaluate the new modeling approach. The results show that it can achieve 9 dB and 7 dB better performance than the traditional approaches in terms of adjacent channel leakage ratio and normalized mean square error, respectively. At the same time, the average PAE is maximized. Similar results have been achieved with the proposed model even when reduced-bandwidth baseband load control signal is used or time misalignment between the RF and baseband load control input signals exists. Although the new approach is only tested with DLM architecture in this paper, it is very general and can be applied to ET architectures as well.

Index Terms—Dynamic load modulation, digital predistortion, envelope tracking, efficiency, linearization, power amplifier.

I. INTRODUCTION

RECENTLY, various power amplifier (PA) architectures have been introduced to achieve high average power-added efficiency (PAE) for modulated signals with high peak-to-average power ratio (PAPR) [1]. Two of the most promising examples are envelope tracking (ET) [2], [3] and varactor-based dynamic load modulation (DLM) [4], as shown in Fig. 1. In these architectures, the DC supply and load impedance, respectively, are dynamically controlled along with the input power to maintain high efficiency over a wide range of output

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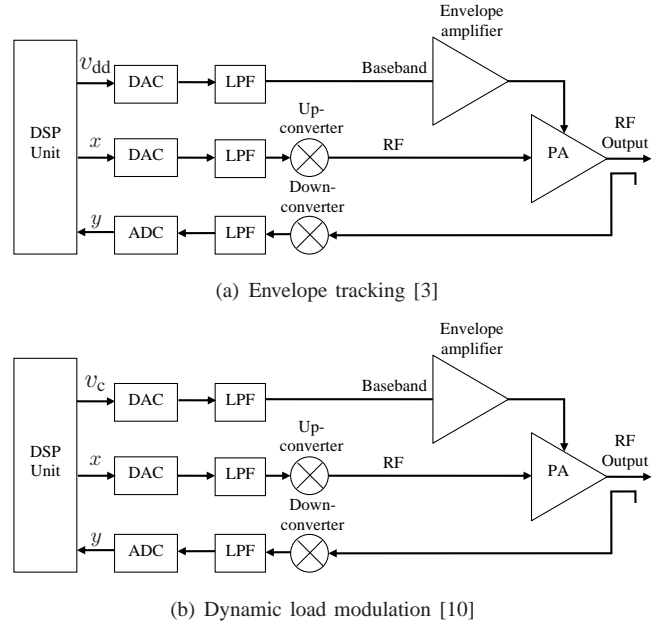


Fig. 1. Simplified block diagrams of high efficiency power amplifier architectures. x is the RF input signal, y is the RF output signal and v_{dd} and v_c are envelope signals for ET and DLM, respectively.

power levels. In this way, the average efficiency can be significantly improved compared to a traditional PA architecture. Details of how to co-control the RF and baseband envelope input signals to achieve optimal average PAE are discussed in [5], [6].

Despite that the ET and DLM PA architectures offer the capability to improve the average PAE, they have two important problems to be considered in practice. First, the bandwidth of the envelope signal is usually 3-4 times larger than that of the modulated RF signal [7]. This brings challenges for hardware designers to design a wideband DC/DC converter or tunable matching network with high efficiency. Second, the two input signals may be difficult to time-align properly [8]. Time-misaligned input signals can affect the linearity performance and degrade the efficiency [9]. Moreover, the dual-input nature of these architectures causes nonlinear distortion effects that need to be compensated for using special linearization techniques.

Dedicated linearization methods for either ET or DLM architectures have been proposed in [3], [10]–[14]. Results

have shown that the nonlinearity introduced by the dynamic supply voltage or tunable matching network is possible to be compensated for by using digital predistortion techniques. A piecewise Volterra-based linearization method has been used for the ET architecture in [3], however, details on how to co-control the RF input and envelope signals to get maximized efficiency is not given. The linearization methods in [10]–[12] use static characterization results as a basis for an efficiency-optimized co-control of the RF and envelope input signals. A conventional single-input digital predistorter (DPD) is then added to compensate for the residual memory effects and nonlinear distortion. However, the inaccuracies in the static models used in these methods will, as will be shown later in this paper, limit the overall linearization performance of transmitter. In [13], [14], nonlinear characteristics of a hybrid-envelope elimination and restoration (EER) transmitter (similar to ET) have been fully investigated. A co-control scheme for the envelope and RF input signals and a new linearization configuration were developed. Experimental results have shown that improved efficiency performance has been achieved with good linearization results. For the bandwidth issue of envelope signal, a bandwidth reduction scheme is proposed in [7], and a 2-D look-up table approach is used to linearize the ET architecture with the reduced-bandwidth envelope signal. Despite the importance of these promising PA architectures and the research that has been done to linearize them, most of the linearization works are still in single-input-single-output sense. No dual/multi-input DPD linearization method has been presented to maximize efficiency and linearity performance, while considering the inevitable practical problems of envelope signal bandwidth reduction or time misalignment sensitivity in practical transmitters.

In this paper, a new linearization approach is proposed for high efficiency PA architectures with two signal paths, such as ET and DLM. Compared to the conventional single-input linearization approaches used in the literature, the new dual-input model uses the envelope signal as an additional input to construct the predistorted RF input signal. We will theoretically show that the new approach can achieve better linearization and efficiency results, in particular when the bandwidth of the envelope signal and time misalignment problems are considered. Experiments are performed on a DLM PA architecture to evaluate the new linearization approach. Although the evaluation is done only on one of the mentioned PA architectures due to the availability of hardware, the new approach is general and can be applied to any other dual-input PA architecture similar to DLM, e.g., ET architecture. To the authors' knowledge, this paper presents, for the first time, a general and complete procedure for linearization of ET/DLM PA architectures

The paper is organized as follows. Section II presents the derivation of the proposed model. Section III presents the dual-input digital predistorter and its application with a reduced-bandwidth envelope signal and time alignment. Experimental results are shown in Section IV and finally conclusions are drawn in Section V.

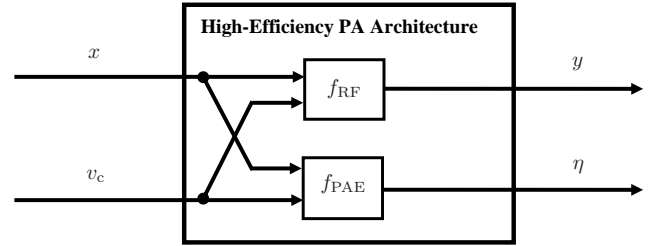


Fig. 2. A general model for high-efficiency power amplifier architectures. x is the RF input signal, v_c is the envelope signal, y is the output RF signal, η is the instantaneous power-added efficiency, f_{RF} and f_{PAE} are transfer functions representing the RF path and output PAE path, respectively.

II. MODELING OF DUAL-INPUT POWER AMPLIFIER ARCHITECTURES

The dual-input characteristic of the ET and DLM PA architectures gives them one extra degree of freedom compared to traditional PA architectures. This allows them to enhance the average PAE and achieve better linearization performance at the same time, if proper design of the RF input and envelope signals is done. A general model representing the main concept of the ET and DLM PA architectures is shown in Fig. 2.

Mathematically, the output instantaneous PAE, η , and output RF signal, y , are functions of the RF input signal, x (or its magnitude $|x|$), and the envelope signal, v_c . The model equations can therefore be written as

$$y = f_{RF}(x, v_c), \quad (1)$$

$$\eta = f_{PAE}(|x|, v_c). \quad (2)$$

Studying (1) and (2) carefully, we see that for a given RF output signal y , there can be many different combinations of x and v_c that satisfy (1). Therefore, we have the freedom to choose the combination of x and v_c that gives us the highest PAE among all possible solutions [6]. This leads to a constrained optimization problem of finding the optimal RF input signal x_{opt} and envelope signal $v_{c,opt}$, that can be expressed as

$$(x_{opt}, v_{c,opt}) = \operatorname{argmax}_{x, v_c} f_{PAE}(|x|, v_c), \quad (3)$$

$$\text{subject to } f_{RF}(x, v_c) = y_{desired} \quad (4)$$

where $y_{desired}$ is the desired output signal of the PA. It can be seen that the maximum PAE and minimum distortion can be achieved by jointly finding the optimal RF input and optimal envelope signals. This is the basis of the static model used for the existing linearization methods [10]–[12], which will be discussed further in the following subsection.

A. Review of Previous Works

Many dedicated linearization papers have been published in recent years for high-efficiency PA architectures, of which all rely on a static model [10]–[12]. Interestingly, for the outphasing PA in [15] which has two RF inputs, a static model similar to the one in [10] is used as parts of the linearization method. All the dedicated linearization methods may be represented by either Fig. 3 or Fig. 4, where f_1 and f_2

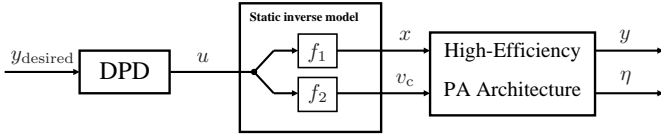


Fig. 3. Linearization method with a static model for high-efficiency PA architectures [10], [12].

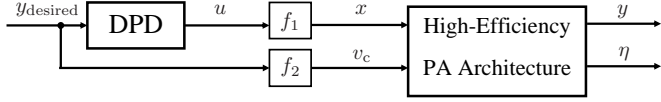


Fig. 4. Modified linearization method for high-efficiency PA architectures [11], [12].

are static polynomial functions for the RF input and envelope signals, respectively.

The static model in [10]–[12] is basically derived by the following steps.

- Perform static power sweep measurements
- Identify the efficiency-optimized settings from (3) based on the measurements
- Polynomial fitting models are used to construct the RF input and envelope signals with the efficiency-optimized settings

As can be noticed, the static model has some disadvantages. In practice, the static model can only invert the nonlinearity of the PA architectures to a limited extent and increasing the order of the polynomial model may lead to over-fitting problems, due to limited number of measurement points. Most importantly, there exist fitting errors at both the RF and baseband branches. The DPD presented in Fig. 3 cannot compensate for these fitting errors, while the DPD in Fig. 4 can only compensate for the distortion at the RF branch, as it has no knowledge of the errors at the baseband branch. Errors at the baseband branch will, however, lead to reduced linearity of the RF output signal. The normal single-input model approach of the existing linearization methods as shown in Fig. 3 and 4 cannot therefore obtain optimal efficiency and linearity results simultaneously.

B. Derivation of A New Model

In this paper, a new model is proposed for ET and DLM PA architectures. The new model does not depend on the static model where the RF and baseband branches are separated, as shown in Fig. 3 and 4. On the other hand, for the new model, the predistorted signal at the RF branch is constructed using the information at both the baseband and RF branches. Moreover, the new model adopts a two-step approach, instead of solving (3) for the optimal RF and envelope input signals jointly. The first step is to choose the envelope signal that can maximize the average PAE. The second step is to choose the RF input signal which minimizes the distortion between the RF output signal y and the desired RF output signal y_{desired} . As we will show, this two-step approach does not imply any loss of generality and it is still an optimal design. The basic

idea of our discussion in the following is that for any given envelope signal v_c , the PA can be properly predistorted as long as the DPD has knowledge of the chosen v_c .¹ It is true that for some choices of v_c the linearization may be more difficult, but we argue that the linearization can still be done with high accuracy for a large class of v_c , and this statement is supported by extensive measurement results later in Section IV.

If an appropriate envelope signal v_c is given, the function f_{RF} in (1) can be seen as a normal transfer function for RF PAs with a one-to-one mapping between the RF input signal x and output signal y . Hence, according to the p -th order inverse theory by Schetzen [16], the function f_{RF} is normally invertible, and the predistorted RF input signal can be written as

$$x = f_{\text{RF}}^{-1}(y, v_c) \quad (5)$$

where f_{RF}^{-1} is the inverse function with respect only to x and y , for a known v_c . Note that the possibility to predistort a PA as in (5) is generally true even for a non-optimally chosen v_c , e.g., v_c may be constant or bandlimited. Using (5) with $y = y_{\text{desired}}$, we can derive the RF signal fulfilling the condition in (4) which can be expressed as

$$x_{\text{opt}} = f_{\text{RF}}^{-1}(y_{\text{desired}}, v_c). \quad (6)$$

Until now, we have derived the optimal RF input signal x_{opt} which is optimized for linearization performance. However, the efficiency is not optimized yet, i.e. the envelope signal, v_c , is a non-optimally chosen signal.

In order to maximize the efficiency, by replacing the variable x in (2) with x_{opt} using (6), the efficiency-optimized envelope signal $v_{c,\text{opt}}$ can be derived as

$$\begin{aligned} v_{c,\text{opt}} &= \operatorname{argmax}_{v_c} f_{\text{PAE}}(|x_{\text{opt}}|, v_c) \\ &= \operatorname{argmax}_{v_c} f_{\text{PAE}}(|f_{\text{RF}}^{-1}(y_{\text{desired}}, v_c)|, v_c). \end{aligned} \quad (7)$$

From (7), we can see that the optimal envelope signal $v_{c,\text{opt}}$ giving maximum efficiency of the PA architecture is only a function of the desired output signal y_{desired} , and can thus be formulated as

$$v_{c,\text{opt}} = g(y_{\text{desired}}). \quad (8)$$

We should notice that although the runtime expression for $v_{c,\text{opt}}$ can be formulated in such a simple way, the identification of the optimal function g can be quite difficult, as is obvious from (7). In the next section, more details about the identification of the function g will be presented.

Once the optimal envelope signal $v_{c,\text{opt}}$ is obtained, the optimal predistorted RF input signal can be resolved directly from (6). With these two optimal-design signals, the optimal average PAE and high linearity can be achieved at the same time. A block diagram representing this new model is shown in Fig. 5.

The main advantage with this proposed model is that it solves the optimization of efficiency and the minimization of distortion in separate steps, while still being fully general.

¹There are, of course, some choices of v_c which make linearization impossible, but those choices can be avoided in the design, as outlined later. For a large class of envelope signals v_c , accurate linearization is indeed possible.

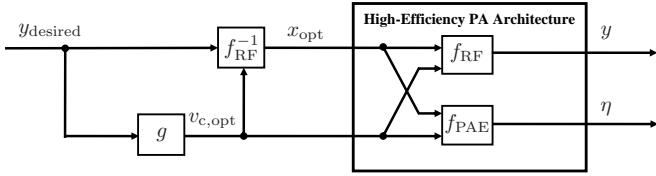


Fig. 5. The new model for linearizing high-efficiency PA architectures.

This makes the derivation of the optimal functions g and f_{RF}^{-1} much simpler, and the linearization results are far better than those of previously proposed methods as will be shown in the experimental results.

III. THE NEW DUAL-INPUT LINEARIZATION METHOD

Based on the derivation in Section II, a new dual-input linearization method is presented in this section. In practice, the ET and DLM PA architectures have the envelope signal bandwidth and time misalignment problems that need to be addressed as mentioned previously. Therefore, the design of the dual-input linearization method should be able to take into account these considerations as well.

A. Behavioral Model of Efficiency-Optimized Function

The efficiency-optimized function g in Fig. 5 (or in (8)) can be modeled by a polynomial function as

$$v_c(n) = g(y(n)) = \sum_{k=0}^K a_k |y(n)|^k, \quad (9)$$

where y is the RF output signal, a_k is the model coefficient and K is the nonlinear order. To identify the optimal function g or extract the model coefficients \mathbf{a} , static CW measurements are first utilized to sweep both the amplitudes of the RF and envelope input signals. Static output amplitudes and efficiencies are then recorded. At each output power level, the combination of input power and envelope input signal that results in highest efficiency is used. This procedure guarantees that the PA can be linearized, since the efficiency-optimized envelope signal, $v_{c,\text{opt}}$, is chosen among the set of v_c that can generate the desired output power. The resulting efficiency maximized relationship between the envelope voltage and the output power is implemented and used as the optimal function g . It should be noted that, since in our case static CW measurements rather than modulated measurements are used to identify the optimal function g , it is not guaranteed that maximum efficiency is obtained. However, in our experience, and for the power levels considered in this work, the degradation of efficiency is negligible. The linearity is not compromised by the choice of g either, since the envelope signal v_c is used in generation of the RF input signal in (6).

B. Behavioral Model of Linearity-Optimized Function

The function f_{RF}^{-1} in Fig. 5 can be seen as a dual-input single-output behavioral model for the high-efficiency PA architectures. The dual-input model can be extended from regular single-input single-output RF PA behavioral models by

adding one real-valued dimension corresponding to the envelope signal. As we know, the generalized memory polynomial (GMP) model [21] is known to have excellent performance in terms of linearity versus complexity [22]. In this work, considering the complexity issue, a modified 2-D GMP model is developed based on the original GMP model and used for the dual-input single-output behavioral model. The digital predistortion function f_{RF}^{-1} can then be written as

$$\begin{aligned} x(n) = f_{\text{RF}}^{-1}(y(n), v_c(n)) = & \\ & \sum_{n_1=1}^{N_1} \sum_{m_1=0}^{M_1} \sum_{m_v=0}^{M_v} \sum_{p=0}^P h_{n_1 m_1 m_v p} y_{m_1} |y_{m_1}|^{n_1} v_{c,m_v}^p \\ & + \sum_{n_2=1}^{N_2} \sum_{m_2=0}^{M_2} \sum_{l_1=1}^{L_1} \sum_{m_v=0}^{M_v} \sum_{p=0}^P h_{n_2 m_2 l_1 m_v p} y_{m_2} |y_{m_2+l_1}|^{n_2} v_{c,m_v}^p \\ & + \sum_{n_3=1}^{N_3} \sum_{m_3=0}^{M_3} \sum_{l_2=1}^{L_2} \sum_{m_v=0}^{M_v} \sum_{p=0}^P h_{n_3 m_3 l_2 m_v p} y_{m_3} |y_{m_3-l_2}|^{n_3} v_{c,m_v}^p \end{aligned} \quad (10)$$

where h are the model coefficients for the 2-D GMP model, N is the nonlinear order, L is the lagging/leading memory depth, M is the RF input signal memory depth, P is the nonlinear order for the envelope signal v_c , M_v is the envelope signal memory depth, y_m is the RF output signal with delay m , $y(n-m)$, and v_{c,m_v} is the envelope signal with delay m_v , $v_c(n-m_v)$.

This new behavioral model has one more input—the envelope signal—than the normal single-input PA behavioral models. In this case, the knowledge of the envelope signal can be used by the dual-input model when constructing the predistorted RF input signal. Thus, even if there exist fitting errors when deriving the optimal envelope signal, this dual-input model can take into account those effects and compensate for them through the predistorted RF signal. This advantage enables the proposed model to have the potential to outperform existing linearization methods for ET and DLM PA architectures. In regards to the identification of this behavioral model, although having an extra input compared to traditional single-input models, it can be noticed that the output is still linear with respect to the model coefficients. Therefore, the conventional least squares method can still be used to identify the dual-input behavioral model.

In practical implementations, the RF and baseband branches have different delays [9], [17]. If the predistorted RF input and envelope signals are not time aligned, even in the scale of less than one integer sample, the linearity and efficiency will be seriously affected [17]. Hence, fractional sample delay estimation is required in order to achieve the best performance offered [9]. The new dual-input behavioral model in this paper can, however, relax the time delay estimation requirement, since the envelope signal with memory can be included in the model as an input. In this case, integer sample delay estimation is enough. This can be verified by the experimental results in the next section.

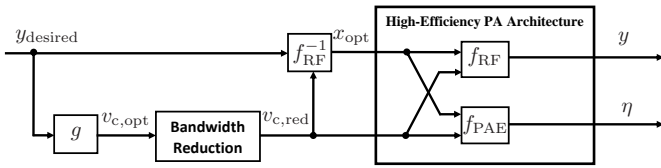


Fig. 6. The new linearization method for high-efficiency PA architectures with bandwidth reduction for envelope signal $v_{c,opt}$.

C. Bandwidth Reduction of The Envelope Signal

The bandwidth expansion of the envelope signal in ET/DLM PA architectures is creating severe practical challenges for the hardware design of the DC/DC converter in ET or the tunable matching network in DLM. In particular for ET, there is a direct trade-off between the envelope signal bandwidth and envelope amplifier efficiency [7]. Different methods have, therefore, been proposed to reduce the bandwidth of the envelope signal for ET architecture [7], [18], [19]. The results shown in [7] is promising, they reported that the bandwidth of the envelope signal can be reduced from 20 MHz to 5 MHz, while the average PAE is reduced from 54.1% to 49.5%. The existing linearization methods in [10], [11] cannot be directly used with reduced-bandwidth envelope signal, since the predistorted RF input signal does not have the knowledge of the envelope signal. Hence, when the characteristics of the envelope signal change (e.g., the envelope signal is bandwidth-reduced), the existing linearization methods are not optimal for linearity any more.

Recently, authors in [20] have proposed a linearization method which employs the reduced-bandwidth envelope signal as an input. Their measurement results have shown that the pronounced memory distortion is possible to be compensated for by such a linearization method. However, since no memory terms are used for either the RF or the envelope signals in their linearization method, some memory effects are still present and the linearization performance is limited.

In this paper, we incorporate the bandwidth reduction scheme in [7] with the new linearization method. The block diagram can be easily modified by adding a bandwidth reduction block to Fig. 5, as shown in Fig. 6. The resulting reduced-bandwidth envelope control signal and its spectrum are shown in Fig. 7, where the reduced-bandwidth of the envelope signal is chosen to be 5.6 MHz² as an example. The beauty of the proposed method is that the bandwidth reduction can be directly incorporated as part of the normal operation of the linearization, without any structural changes.

As we know, the reduced-bandwidth envelope signal may introduce additional memory effects to the PA [7], [20]. However, the dual-input DPD can be used to mitigate these memory effects as the predistorted RF input signal is now constructed by the reduced-bandwidth envelope signal with memory in (10). In other words, the constructed predistorted RF input signal is still optimal in terms of linearity. The average PAE, on the other hand, may be degraded since the reduced-bandwidth envelope signal is not the efficiency-

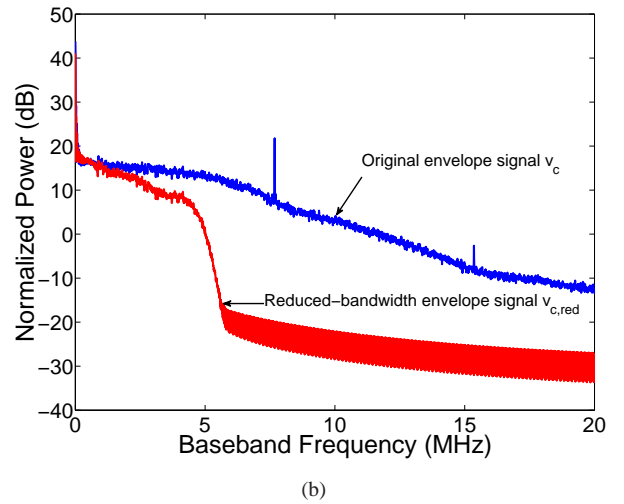
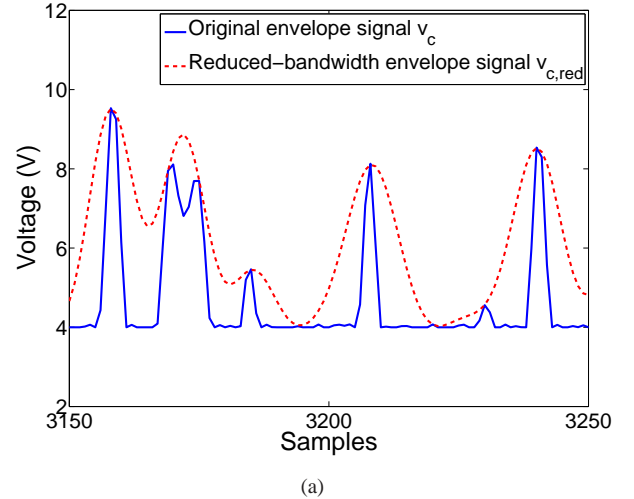


Fig. 7. Time and frequency domain plots of the original and reduced-bandwidth envelope signals for a 3.84 MHz single carrier WCDMA signal. The bandwidths of the original and reduced-bandwidth envelope signals are 12.5 MHz and 5.6 MHz, respectively. (a) Time domain waveforms of the reduced-bandwidth envelope signal $v_{c,red}$ and original envelope signal v_c . (b) Power spectra of $v_{c,red}$ and v_c .

optimized signal any more, but rather a slow time-varying function of it.

D. Complexity

In terms of complexity, the number of coefficients of the proposed dual-input model increases with the nonlinear order P and memory term M_v of the envelope signal. Therefore, it has $M_v \times (P + 1)$ more coefficients than a regular DPD, as the one used in [10], [11]. However, experimental evaluation has shown that only low orders of P and M_v are needed ($0 \leq P \leq 2$ and $0 \leq M_v \leq 2$) to get good linearization results with the new model. The added complexity of the new model is, therefore, acceptable considering its large performance improvements.

IV. EXPERIMENTAL RESULTS

In this section, measurements are performed to evaluate the new linearization method. The measurement setup is

²The bandwidth is defined by 99.99% of the signal energy.

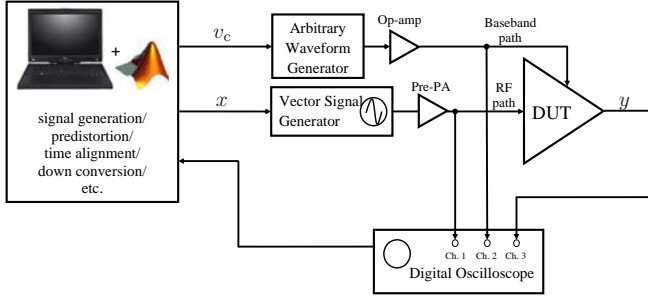


Fig. 8. Measurement setup with a dynamic load modulation power amplifier architecture as the DUT.

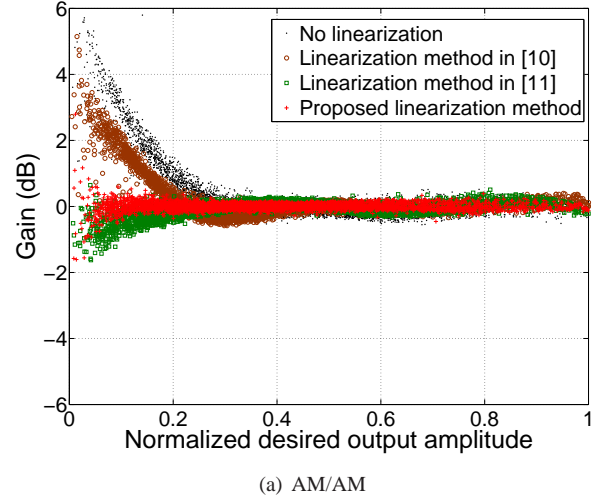
shown in Fig. 8, where an Agilent E4438C vector signal generator is used to generate the RF input signal, a Tabor electronics WW2572A arbitrary waveform generator (AWG) is used to generate the envelope signal and an Agilent 54845A oscilloscope is used as a receiver to capture the RF output signal. The device under test (DUT) is a varactor-based DLM PA architecture, operating at 2.65 GHz [11]. The peak output power is around 38 dBm. In order to achieve the full dynamic swing of the envelope voltage, a LM 7171A op-amp from National Semiconductor Corporation is used also. The signal used is a single carrier WCDMA signal with 7 dB PAPR. Time alignment between the RF and envelope input signals has been performed using the technique in [17] before they are uploaded.

Four different experimental scenarios have been defined to evaluate the feasibility of the proposed dual-input linearization method. First, normal operation of the DUT is used, i.e. no time misalignment and full envelope signal bandwidth are used. In the second scenario, the linearization and efficiency performance with different reduced-bandwidths of the envelope signal are investigated. In the third scenario, different time misalignments between the RF and envelope input signals are tested. In the final scenario, time misalignment sensitivity of the new model is investigated with full and reduced envelope signal bandwidths.

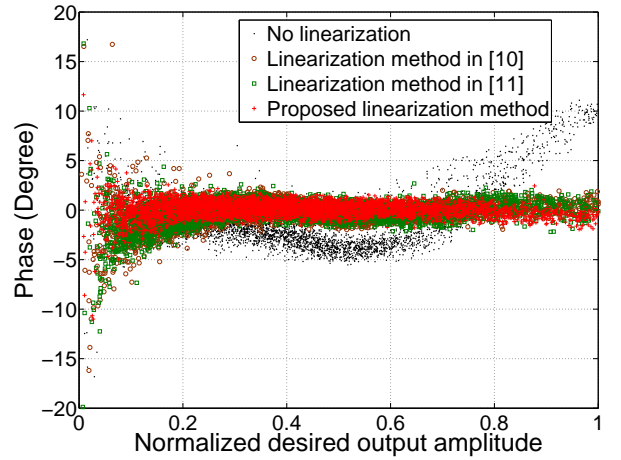
A. Normal Operation

In the experiment, we have tested with different model orders. Finally, we select the nonlinear order 7 and memory depth 4 for the RF input signal which gives the best performance. In the following experiments, these orders for the RF input signal are used if not otherwise specified. Note that, the memory depth of the envelope signal is set to zero for this application and only nonlinear order $P = 1$ is used, since the RF input and envelope signals are already time aligned and full bandwidth envelope signal is used.

Fig. 9 shows the AM/AM and AM/PM of the proposed dual-input DPD. The performance of the other linearization techniques proposed in literature are included as references for comparison. We can clearly see that the proposed DPD has better performance than all other tested linearization methods. The residual gain and phase error, after linearization with the new model, is less than 0.5 dB and 2 degrees, respectively. This indicates that the envelope signal as an additional input to



(a) AM/AM



(b) AM/PM

Fig. 9. AM/AM and AM/PM performance when different linearization methods are applied.

the DPD can indeed help to improve the linearity performance. The reason is that the polynomial fitting errors in v_c are taken into consideration by the dual-input model and compensated for when generating an optimal RF input signal that linearizes the PA. Fig. 10 shows the linearized output spectra of the proposed method and the techniques in [10], [11]. It is noticed that the spectrum of the proposed method is already very close to the noise floor of the measurement setup which is the linearity performance limits.

Table. I summarizes the comparison of the different linearization methods. Note that the calculation of PAE does not include the power consumption of the LM 7171A op-amp, since it is negligible in the experiments. The results demonstrate that the proposed linearization method can be used effectively to minimize the distortion and maximize the average PAE simultaneously for high-efficiency PA architectures. This is due to that the envelope signal is optimized for efficiency, while the predistorted RF input signal is optimized for linearity at the same time. Concerning the complexity of the proposed linearization method, it has twice the number

TABLE I
COMPARISON OF MODULATED MEASUREMENT RESULTS WITH DIFFERENT LINEARIZATION METHODS

Methods	Average PAE [%]	Average Output Power [W]	NMSE [dB]	ACLR1 [dB]	Number of Coefficients
Linearization method in [10]	48	1.2	-32	-41	100
Linearization method in [11]	49	1.2	-34	-44	100
Proposed linearization method	49	1.2	-39	-50	200

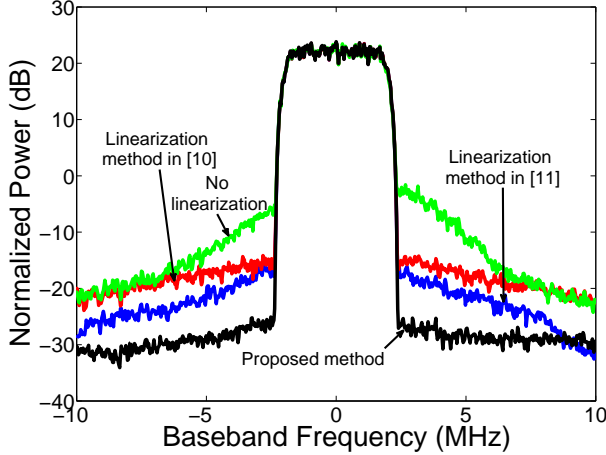


Fig. 10. Output power spectra when different linearization methods are applied.

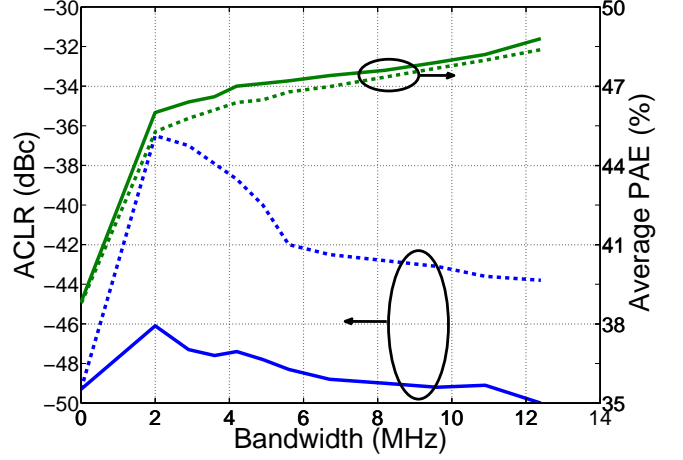


Fig. 11. Average PAE and ACLR vs. bandwidth of the envelope signal u_c . Solid lines denote the proposed linearization method; dashed lines denote the linearization method in [11].

of coefficients compared to the existing linearization methods. However, considering the improvement of performance and if a much less complex behavioral model for the DPD is used, the proposed linearization method is a promising candidate for the high-efficiency PA architectures having two inputs.

B. Reduced-Bandwidth Envelope Signal

In the second experiment, we evaluate the effectiveness of the new dual-input DPD when the envelope signal bandwidth is reduced. This experiment also demonstrates the generality of the proposed method. The memory terms of the envelope signal are used in this application to compensate for the additional memory effects introduced by the reduced-bandwidth envelope signal and is set to $M_v = 2$. Fig. 11 shows the ACLR and average PAE vs. different envelope signal bandwidths. The results indicate that the proposed DPD can achieve good linearization results even when the bandwidth of the envelope signal is reduced from 12.5 MHz to 2 MHz. At the same time, the average PAE is only degraded by less than 3%. It should be noted that the existing linearization methods in [10], [11] cannot be successfully used for the reduced-bandwidth envelope signal. As shown in Fig. 11, when the envelope signal bandwidth is reduced, the ACLR of the linearization method in [11] becomes worse. When the envelope signal bandwidth is less than 5.6 MHz, the linearity degrades more severely. Also, the average PAE of the linearization method in [11] is worse than that of the proposed model. These results prove that the additional envelope signal input to the DPD can indeed

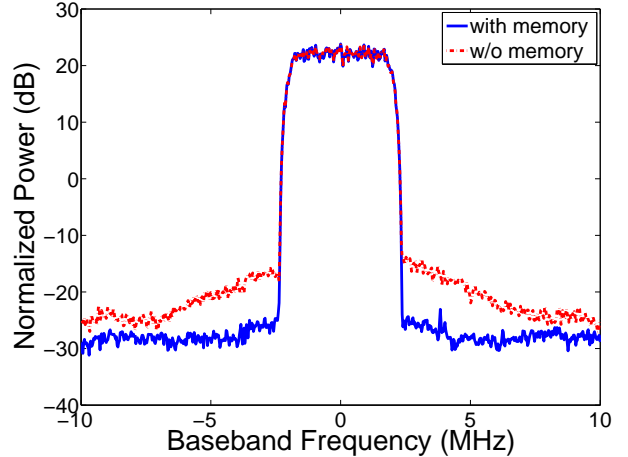


Fig. 12. Output power spectra of the proposed method with and without memory compensation. With memory compensation, the memory depths M and M_v used for the RF input signal and envelope signal are 4 and 2, respectively.

improve the linearization performance, even the property of the envelope signal changes.

The zero bandwidth of the envelope signal is also included in Fig. 11, which corresponds to traditional PA operation with DLM disabled. In this case, we can see that the average PAE drops severely, while the linearization results become better. This is the expected behavior, since this case corresponds to the normal single-input single-output PA and DPD operations.

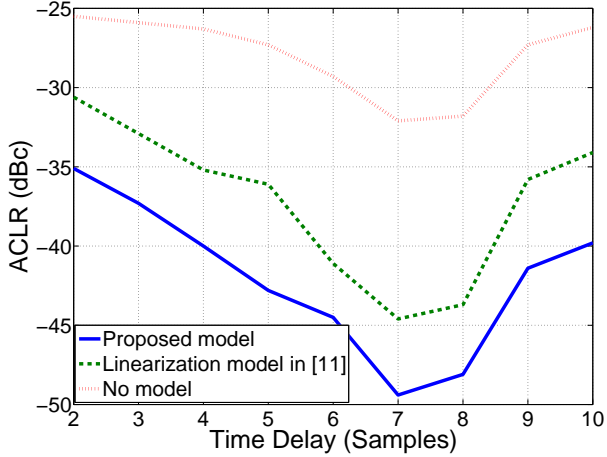


Fig. 13. ACLR vs. delay samples between the predistorted RF input signal x and envelope signal v_c .

As stated previously in Section III, a reduction of the envelope signal bandwidth will result in severe memory effects. In order to justify the need for memory compensation in the DPD, the proposed DPD has therefore also been evaluated without any memory terms in the RF and envelope signals, i.e. $M = M_v = 0$ in (10). In this experiment, the bandwidth of the reduced-bandwidth envelope signal is 6.7 MHz. The output power spectra with and without memory compensation are shown in Fig. 12. From the output spectra, it can be clearly noticed that the ACLR is only around -40 dBc when there is no memory compensation in the DPD, the nonlinear distortion can be further suppressed by around 9 dB by including the memory terms. This indicates that the memory compensation in the DPD is necessary for the linearization of practical high-efficiency PA architectures with finite envelope signal bandwidths.

C. Time Misalignment

Different time delays between the RF and baseband branches are manually added in the DUT in order to test the sensitivity to time misalignment when the new linearization method is used. As mentioned before, the memory depth M_v of the envelope signal is set to 2 throughout these experiments. Fig. 13 shows the performance of the proposed DPD with different time delays. We can see that for an ACLR requirement of -45 dBc, which is the standard for the single carrier WCDMA signal [23], the time delay estimation requirement can be significantly relaxed compared to the existing linearization methods. In fact, a fractional time delay estimation is not needed and at least one sample delay mismatch is still tolerable. This can further reduce the complexity of the transmitter implemented, since the sinc-interpolation for estimating the fractional time delay [17] can thus be avoided.

D. Reduced-Bandwidth and Time Misalignment

In the final experiment, the proposed dual-input DPD is used to linearize a DLM PA when both time mismatches and

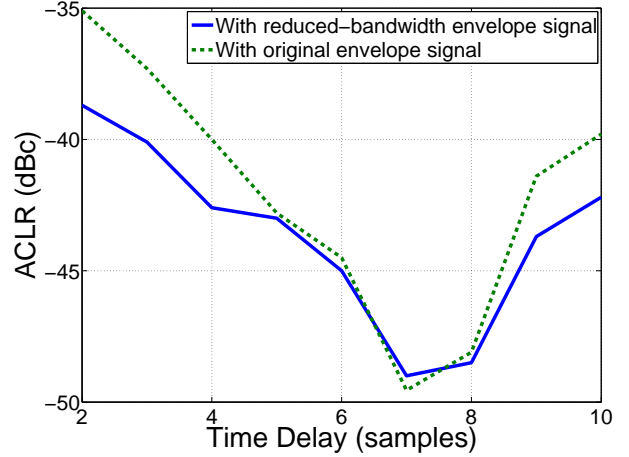


Fig. 14. ACLR vs. delay samples between the predistorted RF input signal x and reduced-bandwidth envelope signal $v_{c,red}$.

reduced-bandwidth envelope signal is used. The bandwidth of the envelope signal in this experiment is reduced to around 6.7 MHz. Fig. 14 shows that if there is time mismatches, the linearization performance with the reduced-bandwidth envelope signal is better than the case in which full envelope signal bandwidth is used. The reason is that when the bandwidth of the envelope signal is reduced, the envelope changes more slowly. In this case, the model becomes less sensitive to time mismatches between the RF and envelope input signals. This further proves that the proposed DPD is very general, robust, and effective in a variety of practical applications.

V. CONCLUSIONS

A new linearization method for high-efficiency dual-input PA architectures, such as ET and DLM, has been proposed in this paper. A two-step approach is first used to derive the optimal envelope and RF input signals in order to obtain optimal efficiency and linearity performance at the same time. A new dual-input DPD is then developed based on the theoretical derivations and its various applications, e.g., bandwidth reduction for envelope signal and time misalignment are presented. Compared with existing linearization techniques in literature, the proposed method separates the problems of efficiency maximization and distortion minimization, and constructs a more accurate RF input signal for the given envelope signal, thereby making large improvements possible. Experimental results have shown when using the proposed linearization method in the normal operation, the ACLR1 can achieve around 9 dB improvement compared to the existing linearization methods, while the average PAE is still as high as 49%. When the bandwidth of the envelope signal is reduced, the ACLR1 can still be less than -45 dB over a wide bandwidth range. The time mismatch sensitivity is also relaxed if the proposed linearization method is used, which avoids the need for fractional samples delay estimation. Although the proposed linearization method is evaluated on the DLM PA architecture, it can be also applied to other dual-input PA architectures which have similar properties with the DLM

architecture.

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