Periodically Time-Varying Noise Cancellation for Filtering-by-Aliasing Receiver Front-Ends

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Abstract—This paper presents a periodically time-varying (PTV) noise cancellation technique for filtering-by-aliasing (FA) receivers. The key to the proposed technique is the use of a time-varying transconductance \((G_{\text{v}})\) cell to sense the noise generated by the PTV resistor in an FA receiver while maintaining the sharp filtering offered by FA. A prototype IC fabricated in a 28-nm CMOS process improves the noise figure (NF) by about 3 dB while achieving over 67-dB stopband rejection with a transition bandwidth (BW) of only four times the RF BW. A minimum in-band NF of 3.2 dB and an average in-band NF of 4.2 dB are demonstrated. With an upfront N-path filter to further enhance the linearity, the measured out-of-band IIP\(_3\) is +18 dBm and the blocker 1-dB compression point is +9 dBm. The whole chip, including digital control circuitry, operates under a 0.9-V supply, while consuming 61-mW power at 500-MHz LO.

Index Terms—Periodically time-varying (PTV) circuit, noise cancellation, receiver front-end, FIR filtering, programmable receiver, sampled PTV circuit, software-defined radio.

I. INTRODUCTION

It has been of significant interest in recent years to explore high-programmability SAW-less transceivers for emerging software-defined and cognitive radios applications [1], [2]. However, without the pre-filtering provided by SAW filters, such receivers face great challenges in simultaneously providing sufficient filtering, linearity, and low noise. Some recent approaches include N-path filters (NPFs) [3], mixer-first receivers [4], [5], and discrete-time (DT) charge-domain signal processing [6]. They have demonstrated moderate filtering (usually equivalent to first- or second-order baseband filters), reasonably high small- and large-signal linearity (approximately +20-dBm out-of-band (OOB) IIP\(_3\) and +10-dBm blocker 1-dB compression point (B\(_1\)dB)), good noise performance, and moderate LO and bandwidth (BW) tunability.

On the other hand, the recent filtering-by-aliasing (FA) technique [7]–[10] provides very sharp analog FIR filtering (>70 dB at 4× RF BW\(^1\) offset), good linearity (>+20-dBm OOB IIP\(_3\) and +13-dBm B\(_1\)dB), and comparable or better programmability. The block diagram of a representative active FA receiver is shown in Fig. 1(a), wherein the key component is an input matching resistor that is periodically time-varying (PTV), \(R(t) = R(t + T_s)\). Together with a mixer and the baseband integrate-and-dump circuit, equivalently the input signal \(V_x(t)\) is down-converted to baseband and sees an apparent linear time-invariant (LTI) filter at the sampled output whose impulse response is given by [9]

\[
g(\tau) = \frac{1}{C \left[ R_s + R(-\tau) \right]},
\]

where \(0 \leq \tau \leq T_s\). The FA receiver presents a time-varying impedance to the antenna with an \(S_{11}\) given by

\[
S_{11} = \frac{R(\tau) - R_s}{R(\tau) + R_s},
\]

where \(R_s\) is the antenna impedance (typically 50Ω). Although the desire for a small \(S_{11}\) (typically ~10 dB or less) constrains
the choice of $R(t)$, fairly arbitrary impulse response shape, $g(t)$, and hence, very sharp analog FIR filtering can be achieved by choosing $R(t)$ appropriately. An example of the $R(t)$ variation and the corresponding baseband filter is shown in Fig. 2, which shows that FA is much sharper than a first-order filter. A third-order baseband filter equivalent resulting from a TIA plus biquad combination that is commonly used in traditional mixer-first and $N$-path designs is also shown for the sake of comparison. Note that the in-band droop, filter transition bandwidth, and stop band attenuation of the FA filter can be traded against each other. Furthermore, sharper filtering that also relaxes the $S_1$ constraint was reported using time-interleaving (TI) [10], making it extremely useful for software-defined radio applications. In any case, the overall receiver’s noise is fundamentally limited by the noise contribution from $R(t)$. In fact, during part of each period, its value can get very large (>10$^3$ the 50-Ω antenna resistance), resulting in an overall high noise figure (NF) of >6 dB after considering the NF degradation due to LO harmonics and filter aliasing [9], [10]. This disadvantage prohibits it from being used in more generic RF environments where a low NF may be desired.

On the other hand, noise cancellation (NC) technique has been proven useful to lower the NF of wideband low-noise amplifiers (LNAs) [11]. The frequency-translational NC (FTNC) technique has been successfully extended to mixer-first receivers to cancel the noise contribution of their input matching resistor, and an NF as low as 2 dB was demonstrated in [5]. As shown in Fig. 1(b), generic LTI-NC senses the noise voltage from the input matching resistor at the RF node, $V_s$, with a transconductance ($G_m$) cell and cancels this noise by subtracting the signals at the outputs of the main and NC paths, where a baseband gain factor $k_2$ is used to control the relative gain between the two paths. In principle, such FTNC can be readily extended to FA receivers as well. However, as will be shown in the next Section, such naïve NC will lower the NF of an FA receiver, but it will negate the sharp filtering of FA.

In [12], a PTV-NC technique tailored for the FA-based receivers was proposed, which improves the average in-band NF (NF$_{avg,IB}$) by about 3 dB and achieves a minimum in-band NF (NF$_{min,IB}$) of 3.2 dB without noticeable degradation to FA filtering performance, shown in Fig. 1(c). This paper details the design of the PTV-NC in [12] together with supporting theoretical analysis on both filtering and noise. In addition, upfront $N$-path filtering is added to improve the linearity of the front-end [12]. To the authors’ best knowledge, this is the first application of a combination of an upfront $N$-path filter and an FA receiver. The attendant design considerations are also presented in this paper. Section II details how the naïve application of LTI-NC to FA receivers is not useful and then introduces the proposed PTV-NC technique. Section III explains the dynamic range (DR) and linearity issue faced by the proposed technique and describes how an upfront $N$-path filter can be added to the FA receiver to improve DR and linearity without sacrificing the sharp filtering offered by FA. Detailed circuit implementation is presented in Section IV, followed by measurement results in Section V. Finally, the conclusions are drawn in Section VI.

II. NOISE CANCELLATION FOR FA RECEIVER FRONT-ENDS

In this Section, we review the LTI noise cancellation
technique and detail why it is not suited for FA-based receivers, followed by analysis of the proposed PTV-NC.

A. LTI-NC in FA-Based Receivers

A naïve application of LTI-NC to FA leads to the implementation of Fig. 3(a). In the case where \( R(t) = 50 \, \Omega \), it is essentially identical to FTNC if one were to ignore the sampling of the final output [5], [11]. The noise voltage of \( R(t) \), \( V_{\text{nr}}(t) \), leads to two noise currents through the main and auxiliary NC paths, \( i_n(t) \) and \( i_{\text{cm}}(t) \), respectively, which are then converted back to voltage at baseband as shown in Fig. 3(b), where \( Z_{\text{ob}} \) is the baseband current-to-voltage conversion gain (= \( 1/j\omega C \) in the case of FA). It is straightforward to see that, after voltage subtraction, the noise voltage caused by \( R(t) \) is nulled at the output, if \( G_{\text{nr}} \) is selected to be \( k_1 R_1 \), with \( k = k_2 \). If \( R_1 \) and \( R(t) \) are the only noise sources, this renders a perfect 0-dB NF even though \( R(t) \) is a time-varying resistor.

However, unfortunately, the sharp filtering is eliminated: by examining Fig. 3(c), which shows how the input signal, \( V_{\text{in}}(t) \), is processed, we find that the signal current to be integrated, \( i_s(t) \), no longer depends on \( R(t) \)! This is fundamentally different from sampling an active-RC integrator.

B. Proposed PTV Noise Cancellation

Instead of the usage of an LTI \( G_m \) cell, [12] proposed to exploit a PTV \( G_m(t) \) cell to sense \( V_{\text{nr}}(t) \) at node \( V_s \), from which the output current is then down-converted, integrated, and sampled to realize the noise-cancelling FA path (Fig. 1(c)). By replacing the time-invariant \( G_m \) cell in Fig. 3(a) with a time-varying one, the signal and the noise now see different filters. The equivalent signal and noise (due to \( R(t) \) only) flows in the front-end are shown in Fig. 4(a): \( i_{\text{cm}} \) and \( i_n \) are currents from the \( G_m \) cell and through \( R(t) \) flowing into their respective virtual grounds, which are later integrated. Fig. 4(b) shows a simplified model that inspects the signal flow of the noise from \( R(t) \) only. In contrast to setting \( G_{\text{nr}}(t) = k/R_s \), we select

\[
G_{\text{m}}(t) = k/R(t)
\]

Now \( i_{\text{nr}}(t) \), which is the effective noise current caused by \( R(t) \) after cancellation, is not nulled (i.e., the noise from \( R(t) \) is not completely canceled as \( i_{\text{nr}}(t) = 0 \) no longer holds). Instead, it becomes \(-V_{\text{nr}}(t)[1 - k_1 R_1/R(t)]/[R(t) + R(t)]\). However, since the noise currents from the two paths, which have the same polarity, are subtracted, the overall noise after integration is still greatly canceled. On the other hand, as shown in Fig. 4(c), the equivalent signal path is almost identical to the FA case. In fact, it is nothing but a scaled version of the original FA filter (due to our choice of \( G_{\text{m}}(t) \)) with a baseband impulse response of

\[
g(t) = \frac{1 + k_1/k_2}{C[R_s + R(t)(-t)]}
\]

where \( 0 \leq t \leq T_s \). Thus, the FA operation is intact, and sharp filtering is achieved. Note (3) is also the effective filter that the source noise sees.

C. Achievable Noise Cancellation

Since perfect noise cancellation is not feasible with this approach, it is instructive to consider the theoretically maximum achievable cancellation. Consider the noise factor contribution from \( R(t) \): it can be calculated by looking at the autocorrelation of the output voltage samples [9], [10]. Here we consider the baseband filter only for brevity. It follows from the Appendix that the overall noise factor due to the source and \( R(t) \) after cancellation is

\[
F_{\text{PTV}} = 1 + F_R = 1 + \lambda \left[ \overline{G - G_{\text{total}}} \right] / \left[ \overline{G_{\text{total}}} (1 + \lambda) \right] F_{\text{PTV}}|_{\lambda = 0}
\]

where \( F_R \) is the noise factor due to \( R(t) \), \( \lambda = k_1/k_2 \) is effectively the gain ratio between the two paths, \( \overline{G} = \text{mean}[1/R(t)] \), \( \overline{G_{\text{total}}} = \text{mean}[1/(R_s + R(t))] \), and \( F_{\text{PTV}}|_{\lambda = 0} = \overline{G_{\text{total}}} \left( R_s \overline{G_{\text{total}}} \right) \) is the noise factor without any NC, i.e., the noise factor given in [9, Section III-B]. It is unclear, simply by inspecting the expression, what the optimum gain ratio \( \lambda \) ought to be, which can be obtained by setting \( \partial F_{\text{PTV}}/\partial \lambda = 0 \). After simplification, we find

\[
\lambda = \frac{k_1}{k_2} = \frac{\overline{G_{\text{total}}}}{\overline{G - G_{\text{total}}}}.
\]

Substituting (5) into (4), the minimum achievable noise factor can be found, given by

\[
F_{\text{PTV(MIN)}} = \frac{\overline{G - G_{\text{total}}}}{\overline{G}} F_{\text{PTV}}|_{\lambda = 0}
\]

It is straightforward to see that, for LTI-NC, \( R(t) = R_s, F|_{\lambda = 0} = 2 \), and (5) suggests \( \lambda = 1 \), plugging which into (4) yields \( F = 1 \), i.e., perfect NC is achieved. For FA, on the other hand, as expected, the optimum gain ratio, achievable amount of noise cancellation, and minimum noise factor depend on \( R(t) \). However, (6) does provide us with a bound on \( F_{\text{PTV(MIN)}} \) due to the minimum value of \( R(t) \) in practical implementations. In [9] and [10], it is about 30 \( \Omega \). Re-write (6), we obtain
In order to minimize the overall noise factor $\gamma$, the baseband filter shape $G$ needs to be designed to be a low pass filter (e.g., Fig. 2), thereby reducing the noise contribution from $G_m(t)$ to be close to zero. However, this will lead to infinite power consumption and infinite baseband capacitor size. Practical selection of $k_1$ and $k_2$ is made by making the contribution of $R(t)$ and $G_m(t)$ to be roughly the same, rendering $k_1 = k_2 = -5$ to 6. Here, we only discussed the noises from circuit elements. Other major sources of NF degradation, namely, aliasing and harmonic folding, will be discussed in Section III-C.

An alternative, and may be more intuitive but not completely mathematically precise, approach to look at how the proposed PTV-NC still effectively cancels the noise is to look at the equivalent filter that $V_{ab}(t)$ sees. For the baseband FA without NC, i.e., main path only, the baseband filter is the same as (1) but with a negative sign. With NC, the baseband filter that $V_{ab}(t)$ sees becomes

$$h(\tau) = -\frac{\lambda R_s - R(-\tau)}{CR(-\tau)[R + R(-\tau)]} = g(\tau) - \frac{\lambda R_s}{R(-\tau)} \times g(\tau),$$

where $\gamma$ is the excess noise factor, needs to be added to (4) for the overall noise factor. Such summation can be done due to the fact that the noise from $G_m(t)$ is not correlated with that of $R(t)$ and only appears in the NC path. Since $\lambda$ is a constant for a given $R(t)$ ($\approx 1$ in most practical cases), in order to minimize $F_{\text{Gain}}$, $k_1$ needs to be made large according to (8).

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In Fig. 5 (a) Noise factor with the PTV-NC for different gain ratio, $\lambda = k_1/k_2$, in three different filter configurations, and (b) the corresponding baseband filter frequency responses with $T_s = 200$ ns. Filter 1: transition BW = 1.5x RF BW, $S_{11} \approx -10$ dB; filter 2: transition BW = 2.5x RF BW, $S_{11} \approx -20$ dB; filter 3: transition BW = 2.5x RF BW, $S_{11} \approx -10$ dB (i.e., same as Fig. 2).

$$F_{\text{PTV(MIN)}} = \frac{\int_0^T dt/R(t)[1 + R(t)/R_s]}{\int_0^T dt/R(t)} F_{\text{PTV}} \bigg|_{t=0}$$

$$< F_{\text{PTV}} \bigg|_{t=0}\left[1 + R(t)/R_s \right]^2 = \frac{5}{8} F_{\text{PTV}} \bigg|_{t=0}$$

for $R_s = 50 \, \Omega$. Given an assumed noise factor of 2 prior to NC, a noise factor better than 1.25 can be achieved after NC. More specifically, for the $R(t)$ variation in Fig. 2, (5) leads to a $\lambda$ of 0.97, which is very close to that in the LTI case. Fig. 5 shows (4) for different values of $\lambda$ in three different filter configurations, where the blue curves correspond to the $R(t)$ variation shown in Fig. 2. A couple of things are apparent. First, the optimum gain ratio is close to unity. Second, the optimum is fairly shallow suggesting that achievable noise cancellation is tolerant to relative gain mismatches between the main and NC paths. Third, the optimum noise factor is actually about 1.1, indicating that about 90% of the noise of $R(t)$ is canceled.

Note until here the only noise sources are $R_s$ and $R(t)$ for simplicity. In practice, with $R(t)$’s noise mostly canceled, $G_m(t)$’s noise becomes important. To evaluate this, an extra term, derived following the Appendix as well, given by

$$F_{\text{Gain}} = \frac{1}{k_1 R_s} \frac{\gamma G(t)}{(1 + 1/\lambda)^2 G_{\text{total}}(t)},$$

where $\gamma$ is the excess noise factor, needs to be added to (4) for the overall noise factor. Such summation can be done due to the fact that the noise from $G_m(t)$ is not correlated with that of $R(t)$ and only appears in the NC path. Since $\lambda$ is a constant for a given $R(t)$ ($\approx 1$ in most practical cases), in order to minimize $F_{\text{Gain}}$, $k_1$ needs to be made large according to (8).
Recall the fact that $R(t)$ can be very large at times, e.g., Fig. 2, such that $R(t) \gg R_s$, and hence $V_s(t) \approx V_i(t)$. If we assume $V_s(t)$ is a simple sinusoid at an arbitrary frequency of interest with an amplitude of $A$, it is obvious that $V_{x, pk} \approx V_{s, pk} = 2A$. Note this is the case for both in-band and out-of-band signals since the FA-based receiver has no frequency selectivity at the RF node (as suggested by the ideally frequency-independent $S_{11}$, which is only somewhat frequency-dependent in practice because of the parasitic capacitance at the RF node) [9], [13]. Consequently, for a big OOB blocker, $V_s$ suffers from large voltage swings that may be outside the voltage range of core MOSFET devices in advanced nodes. [12] addressed this by employing time-interleaving (TI) and an $N$-path filter. Here, we analyze their effects and design considerations.

A. Time-Interleaved FA

In [10], time-interleaved FA is used to improve both impedance matching and filter performance. By using two interleaved channels, as shown in Fig. 6, the length of $g(t)$ can be doubled, allowing a stopband rejection ($A_{\text{stop}}$) twice as high in theory. In addition, the overall dynamic range of the input resistance seen by the source is less due to the paralleled operation, which is given by $R_{\text{tf}}(t) = R(t)[R(t - T_s)]$. In the non-TI case, the highest value of $R(t)$ can be a few thousands of Ohms [9]. When TI is used, $\max[R(t)]R(t - T_s)]$ is about 300 $\Omega$. Not only does this make the input matching easier without having to sacrifice filter shape much for better $S_{11}$ [10], but it slightly reduces $V_{x, pk}$ for the same blocker level. Referring to (11), the swing at $V_s$ is lowered by about 15% (0.5 V for a +10-dBm blocker). Although this only slightly relaxes the DR problem instead of solving it, TI is nonetheless used in this work to achieve better matching and sharper filter at the cost of higher power and larger chip area for two extra paths (one for the main path and one for the NC path).

B. Upfront $N$-Path Pre-Filtering

An upfront NPF will reduce the swing on $V_s$ and relax the linearity requirements of the NC path [3], [14]. A block diagram of such a combination of an FA receiver, with a time-varying resistor, $R(t)$, and an upfront NPF is shown in Fig. 7(a).

For in-band signals, the NPF presents a high impedance and has minimal effect on $V_s$, or the current flowing into the PTV resistor, $R(t)$. In contrast, for signals well beyond the NPF’s BW, the NPF presents a low impedance, approximately $R_{sw}$, as shown in Fig. 7(b). Since $R_{sw} \ll R(t)$, $V_s$ is effectively much smaller than without the NPF:

$$V_s(t) \approx V_i(t) \frac{R_{sw}}{R_s + R_{sw}} = V_i(t) A_{\text{NPF}},$$

as desired, where $A_{\text{NPF}} = R_{\text{sw}}/(R_s + R_{sw})$ is the rejection provided by the NPF. However, the small $R_{sw}$ siphons away much of the signal current from $R(t)$ in a time-varying manner, greatly degrading the effective FA filter shape.

Both effects are readily illustrated using an example $\tilde{R}(t) = R(t)$ designed for 60-dB $A_{\text{stop}}$, 10-MHz RF BW, and 20-MHz transition bandwidth with and without an NPF with 30-MHz BW and 15-dB $A_{\text{NPF}}$. As shown in Fig. 8, the NPF reduces the swing at $V_s$ for a 10-dBm blocker from 4 V to just 0.77 V. However, as is evident from the effective filter responses, without and with the NPF, plotted in Fig. 9, the transition bandwidth is almost doubled even as a higher overall $A_{\text{stop}}$ is achieved.

Fortunately, the filter shape degradation can be corrected...
simply by choosing
\[ R(t) = \beta \left[ R + R(t) \right], \]  
(13)
where \( R(t) \) is the PTV resistor variation that ensures the desired filter shape for an FA receiver without the upfront NPF, and \( \beta \) is a constant scaling factor.

Rationale: This choice can be intuitively explained by contrasting Fig. 7(b) with Fig. 1(a), which represents the desired FA operation without an NPF. It is easy to see that for a signal well beyond the NPF’s BW, the current through \( R(t) \) in Fig. 7(b) is just a scaled version of the current through \( R(t) \) in Fig. 1(a), given by
\[ i(t) \approx V(t) \frac{R_{in}}{R_{in} + R(t)} \frac{1}{R(t)} = \frac{R_{in}}{R_{in} + R} \frac{V(t)}{\beta \left[ R(t) \right]}. \]  
(14)
Consequently, the OOB filter shape remains effectively the same as what would be achieved without the NPF.

The simulated overall filter shape with (13) is shown in Fig. 10(a). As is evident, most of the filter sharpness is restored (transition bandwidth is only extended by ~6 MHz instead of over 20 MHz while \( A_{stop} \) is a few dB higher than the original FA). The filter shape is primarily defined by FA while the NPF adds extra rejection at high offset frequencies. Note that this approach results in a slightly larger in-band filter droop, as seen in Fig. 10(b), which is generally acceptable.

Choice of NPF BW and Switch Size: In this work, the NPF BW is chosen to be slightly larger than the desired RF BW of FA. It is however worth noting that the actual NPF BW can be chosen to provide more rejection for close-in blockers by using narrower BW in order to meet certain blocker profiles. This comes at having larger overall filter droop, but it can be remedied by re-designing FA to compensate for the extra signal loss by sacrificing some transition BW or \( A_{stop} \). The NPF switch size choice is driven by the tolerable swing at the RF input node, \( V_s \), and the desired OOB IIPm. Assuming a perfectly linear NPF\(^3\), with a rejection of \( A_{NPF} \), the intercept point amplitude is approximately
\[ i_{IP} = \sqrt{3G_m/2G_{m3}}/2, \]  
where \( G_{m3} \) is the third-order polynomial coefficient of the transconductance. Therefore, the OOB IIP\(^m\) will be roughly improved by \( A_{NPF} \). In this work, we chose an equivalent 2.5-\( \Omega \) switch resistance and a BW of 30 MHz for a 10-MHz RF BW configuration to keep the swing at RF within 0.9 V for a +10-dBm blocker and improve the OOB IIP\(^m\) by about 15 dB at 80-MHz offset.

Note also that the NPF changes the effective \( S_{11} \) of the FA receiver slightly. A frequency-domain analysis with conversion matrices has been employed before to compute the \( S_{11} \) and the effective in-band impedance of an FA receiver [13], [17], [18]. The same approach was extended to the FA + NPF combination described here, after which the scaling factor, \( \beta \), was chosen to fine-tune the impedance matching.

C. Upfront NPF + TI-FA

Extending the design analysis to TI-FA can be achieved in a similar manner. By recognizing the low impedance presented by the NPF, the OOB signal current flowing through each time-varying resistor can be calculated even when two TI resistors are involved (similar to (14)), and the resistor variations can therefore be modified to keep the impulse response a scaled version of the original filter for OOB signals as well. Here, we omit the details for the sake of brevity.

D. Simplified NF Analysis for the Overall Front-End

The noise sources in the NPF (from its switch resistance) are much smaller than other noise sources and can be negligible. Since the overall filter shape is primarily determined by FA [recall Fig. 10(a)], the NPF’s effect on the noise spectrum can be ignored. Behavioral simulation results suggest that a discrepancy of less than 0.1 dB in the averaged in-band NF is seen by ignoring these effects.

If TI were not employed, the noise analysis presented in Section II-C could be directly used here simply by replacing \( R(t) \) in (4)–(7) with \( R(t) \). The effect of TI can be easily approximated by ignoring interactions between the two paths, which is still reasonably viable because the TI channel interac-

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\(^3\) This is a fair assumption since the NC path dominates the non-linearity.
tion happens only when \( \tilde{R}(t) \equiv \tilde{R}(t - T_s) \), which is a short period of time [10], and the correlation between two output samples is small. In this case, the two TI channels contribute independent noise of identical statistics in a TI manner. When combined together, it is as if a single non-TI FA receiver employing a time-varying resistor \( \tilde{R}(t) \) [10]. Accordingly, the baseband noise figure can be determined as \( NF_{\text{baseband}} = 10 \log (d(F_{\text{PTV}} + F_{\text{Gain}})) \).

However, the TI channels do interact, resulting in correlation between samples of the output noise. This effect has been described in [10] for a system without NC. Similar calculations can be performed here (see Appendix) to calculate the precise NF contribution of the input matching resistors and \( G_m \) cells. The resultant \( NF_{\text{baseband}} \) is now a function of the baseband frequency, \( \Delta f \), where \( |\Delta f| \leq 1/(2T_s) \). Fig. 11(a) plots the calculated NF with and without NC for both the cases where the TI channel interaction is considered (solid curves) and ignored (dashed curves). The approximation ignoring the TI interaction only introduces a small error on \( NF_{\text{baseband}} \), albeit making it frequency-independent. The average in-band NF error is less than 0.2 dB with approximation. In either case, NC is observed and the average \( NF_{\text{baseband}} \) is lowered by about 3 dB. In our calculation, the baseband amplifiers’ noises are taken care of in two means: 1) the amplifiers in the main path present themselves as part of \( \tilde{R}(t) \) and \( \tilde{R}(t - T_s) \) [10], and their noises get mostly canceled; 2) those in the NC path still present their noises but they are suppressed due to the gain of the \( G_m \) cells just as in [5] (moreover, the baseband amplifiers in this work are designed to have large \( g_m \)). The passive mixers’ switch noises are considered similarly: 1) main-path mixer’s switch noise gets mostly canceled; 2) the NC-path mixer contributes little noise.

In addition to \( NF_{\text{baseband}} \), two additional sources also contribute to the overall NF, namely the aliasing of the source noise as a result of sampling at the output and harmonic folding due to the \( N \)-path operation. They are given by [9]

\[
NF_{\text{aliasing}}(\Delta f) = \frac{PSD_{R_s}(\Delta f)}{2kT_R} \left| \frac{G(\Delta f + nf_s)}{G(\Delta f)} \right|^2, \\
NF_{\text{harmonics}} = \frac{1}{\sin^2(\Delta f/2)} \approx 0.91 \text{dB},
\]

where \( PSD_{R_s}(\Delta f) \) is the power spectral density due to noise from \( R_s, f_s = 1/T_s \), the sampling rate at the output, \( G(f) \) is the frequency response of the filter, \( g(t) \), and \( N \) is the number of paths in the NPF and mixer, which is 4 in our implementation. Note \( PSD_{R_s}(\Delta f) \) can also be computed by looking at the autocorrelation of the output sampled voltages [10].

Finally, the overall NF at a certain \( \Delta f \) for the complete system can be derived as

\[
NF_{\text{total}}(\Delta f) = NF_{\text{baseband}}(\Delta f) + NF_{\text{aliasing}}(\Delta f) + NF_{\text{harmonics}}.
\]

The calculated NF with PTV-NC and NPF is shown in Fig. 11(b) in comparison with the simulated results4. With both NC and NPF, the calculated \( NF_{\text{min,IN}} \) is 1.8 dB and \( NF_{\text{avg,IB}} \) is 3.2 dB. The simulation results agree well with calculation, except for an extra NF degradation of about 0.8 dB mostly due to the loss caused by input parasitic capacitance [13]. The peaking close to dc is due to flicker noise, and as a consequence of the TI operation, half of the flicker noise power being shifted up to \( f/2 \) is observed, similar to [10].

IV. CIRCUIT IMPLEMENTATION

Fig. 12 shows the block diagram of the implemented PTV-NC receiver front-end. The values of \( k_1 \) and \( k_2 \) are programmable but are roughly set to 5–6 by design for best tradeoff between \( G_m \)’s noise contribution and area. The receiver front-end consists only of switches, inverter-based amplifiers, digital circuits, and passive devices (namely, resistors and capacitors).

\( R_1(t) \) are implemented as two 13-bit binary resistor DACs (RDACs) (Fig. 13(a)). The switches are implemented by transmission gates with equally sized pFETs and nFETs, and the resistors are made of high-resistive polysilicon. Both are binary scaled in the RDACs. In this work, the linearity is primarily limited by the NC path. Therefore, in contrast to prior implementation of FA-based receivers where it is sized such that the resistance ratio between the transmission gate and the polysilicon resistor is 1:4, this work uses a ratio of 1:1, and minimum-sized (both width and length) transmission gates are used, lowering the parasitic capacitance due to the RDACs. The RDACs are designed to have a minimum resistance of 30 \( \Omega \). \( G_m(t) \) are formed by inverter-based \( G_m \) cells binarily turned on/off by switches \( (G_m \text{DACs}) \), shown in Fig. 13(b), which is similar to that in [19], albeit being a higher-power one. Switches in the \( G_m \text{DACs} \) act as source-degeneration resistors to the \( G_m \) cells and are therefore designed to contribute less than 10% of the effective \( g_m \) in each cell for noise consideration. Non-minimum-length devices are used in the \( G_m \) cells (one unit cell has an equivalent \( \text{W/L} \) of 140 nm/55 nm) for lower \( \gamma \) and higher output resistance at the cost of more degradation on both \( S_{11} \) and NF at high frequencies. The finite output resistances of the \( G_m \) cells degrade the NF slightly and would be more if minimum-length devices are used. The linearity is primarily limited by the \( G_m \) cells, and no special techniques were employed to reduce the non-linearity of the \( G_m \) cells except to keep the baseband input impedance and mixer switch resistance small in order to reduce the effects from \( g_{th} \) non-linearity, similar to [5]. The RDACs and \( G_m \text{DACs} \) all vary at the rate of a clock frequency, \( f_{clk} \), which is also used to generate all sampling and reset control clocks.

The bias of the entire chain is set to about half of the supply voltage, \( V_{DD} \), by resetting baseband amplifiers, which also define the voltages at the input and the output of the \( G_m \text{DACs} \) via the main and NC paths, respectively. No dedicated biasing circuitry is used. These baseband amplifiers are sized to have 125-mS \( g_m \) and 35-dB dc gain each with ping-pong capacitor banks around them for sampling. Similar to [9] and [10], the baseband amplifier’s \( g_m \) is in fact part of \( R(t) \) and is thus made large. The baseband integrator capacitors are tunable from 10 to 70 pF in the main path and 50 to 350 pF in the NC path.

Switches in the 4-path mixer and NPF are sized for on-resistances of 2.5 and 5 \( \Omega \), respectively. Here, for having

4 The simplified analysis model without actually introducing the NPF is used for calculation only, not for simulations.
better linearity on the NPF and reduce the effective switch resistance, the bottom-plate mixing version of NPF [14] is used in both simulations and implementation of this work at the cost of higher input parasitic capacitance. The effective switch resistance of the NPF is therefore 2.5 Ω by design. Since both the $G_m$ cells and the NPF add additional capacitance to the input node that limits $S_{11}$ at high carrier frequencies, whereas the former is also observed in the LTI-NC case, the top-plate mixing NPF may be used instead to improve high-frequency performance but worsens the OOB IIP3 by a couple dBm according to simulations. Nevertheless, it still resolves the OOB DR and linearity issue. Note that in this work, the mixer switches in the main path are large to accommodate the variation of $R(t)$, while in LTI-NC, since the noise from these mixers is canceled, they can be much smaller to save LO power. The NPF and the mixers are driven by the same set of 25% duty-cycle clocks at $f_{LO}$.

V. MEASUREMENT RESULTS

The implemented test chip was fabricated in a 28-nm CMOS process. Fig. 14 shows the die photo of the chip. The active area is 3.75 mm², 90% of which is occupied by baseband capacitors. Note the capacitor area can be significantly reduced when designed for operations with only higher RF bandwidths. The supply voltage of the whole chip is 0.9 V. At $f_{LO} = 500$ MHz, the entire chip consumes 61-mW power. Each baseband amplifier consumes about 2.7 mA, the LO divider and switch drivers consume about 16 mA, in which the NPF drivers consume about 1 mA, and the digital control circuitry dissipates 5.2 mA at a nominal $f_{LO}$ of 1 GHz. It has been verified to work with an $f_{LO}$ up to 2 GHz. On average, each $G_m$DAC consumes roughly 2-mA current. The power increases with $f_{LO}$ due to LO divider and switch drivers being more power-hungry at higher frequencies. The sampled outputs are buffered externally, converted to digital signals by off-chip ADCs, and then processed digitally for signal summation and subtraction, with relative gain correction similar to that in [20]. The filter responses are generated by providing tonal inputs and then measuring the downconverted and aliased signals at baseband after sampling, similar to other FA works [8]–[10] (see Section V of [8] for more details). The RDACs and $G_m$DACs are dc calibrated at startup [9].

Figs. 15(a) and (b) show the measured in-band NF at 500-MHz $f_{LO}$ and the measured NF with and without NC at different LO frequencies, respectively. With both NPF and NC,
The NFmin,IB is 3.2 dB and the averaged NF over [0, fLO/2], NFavg,IB, is 4.2 dB. The increase of in-band NF at higher offset frequencies is due to filter droop, as seen in Fig. 11. It is also observed that the NPF does have minimal impact on the NF. Without the NPF, NFavg,IB is about 0.15-dB better. In contrast, both NFmin,IB and NFavg,IB are about 3-dB worse without NC.

The measured Astop for the overall filter is greater than 67 dB for a transition bandwidth of 40 MHz with a gain of ~30 dB, as shown in Fig. 16(a) for 10-MHz RF BW. The achieved Astop and transition BW are similar to [10] with 46- and 58-dB rejection at 22- and 30-MHz offset (using the same 67-dB Astop configuration). This indicates that filter performance is preserved well with PTV-NC and NPF. The filter can be programmed to have 5–40-MHz RF BW, shown in Fig. 16(b). Fig. 16(c) shows the filter responses with fLO varied from 100 MHz to 1 GHz.

The measured IIP3 and B1dB with and without NPF for fLO = 500 MHz, and (b) OOB IIP3 at 49-MHz offset for different LO frequencies.

The linearity performance, i.e., B1dB and IIP3, against different frequency offset and fLO of the receiver in the 10-MHz RF BW configuration are depicted in Fig. 17. An OOB IIP3 of +18 dBm and an OOB B1dB of +9 dBm are achieved even with upfront GaN DACs and 0.9-V supply thanks to the NPF. Without the NPF, both OOB B1dB and IIP3 evidently degrade by about 9 dB. The measured S11 and blocker NF are given in Fig. 18. The NFavg,IB is 12 dB with a 0-dBm continuous-wave (CW) blocker placed at 30-MHz offset. It is primarily limited by the phase noise of the LO divider, which has a simulated phase noise of −164 dBc/Hz at 30-MHz offset. Better blocker NF should be achievable by burning more power in the LO divider.

Fig. 19 shows the measured LO leakage power and
worst-case image filter (normalized to the peak of the corresponding desired filter frequency responses) at different LO frequencies. Due to the N-path operation of the NPF and the mixers plus the lack of an isolating LNA after the antenna, the LO leakage power is about −65 to −70 dBm, which is similar to other N-path-based or mixer-first architectures [21]. The image filter is caused by the TI path mismatches and degrades at higher LO frequencies due to LO clock skews [10]. The worst-case image rejection is better than 30 dB, sufficient for most SNR requirements. Other than filter shapes, other metrics do not vary appreciably for different configurations.

Table I compares this work with the state-of-the-art. While it maintains very sharp filtering with narrow transition band and high $A_{\text{stop}}$ of FA [10], the NF compares more favorably against other works compared to [10]. Good OOB linearity is demonstrated with a 0.9-V supply while all other works use higher supply voltages, mostly in the range of 1.2−1.6 V. Since this work relies on NPF to improve the linearity rather than linear resistors, close-in linearity is worse than [10]. It may be noted that unlike traditional architectures [5], [14], [15], [22], [23], where further filtering can be done on adjacent channels, the FA system as presented here allows folding of transition band signals into passband without full suppression, which is detrimental in a congested spectrum. To prevent such folding, the FA filter can be designed to have a lower BW while maintaining the sampling rate to ensure that the passband is free of folding artifacts, but this places an upper limit on the allowable transition bandwidth (and hence $A_{\text{stop}}$). To increase the allowable transition bandwidth and therefore higher rejection, more time-interleaving is generally needed.

VI. CONCLUSION

In this paper, we detailed a PTV noise cancellation technique for FA-based receiver. It realizes both sharp filtering and low noise figure by employing periodically time-varying resistors and $G_m$ cells. Both minimum and averaged in-band NFs are improved by about 3 dB by the proposed technique. The stopband rejection is better than 67 dB for a transition bandwidth of 4 times the RF bandwidth. Out-of-band linearity is preserved well by introducing an NPF that helps the NC path better handle the OOB blockers, while in-band linearity is worsened due to the presence of active devices at RF.

APPENDIX

GENERALIZED NOISE ANALYSIS FOR BASEBAND FA

We present a generalized analysis to calculate the noise factor contribution from each circuit component here. The fundamental principles are the same as in [9], but more general.

For a particular noise voltage, $V_{n}(t)$, consider it goes through an equivalent baseband filter by multiplying $V_{n}(t)$ with a periodically time-varying conductance, $D_{m}(t)$, and then integrating the current with a capacitor $C$. The equivalent model is depicted in Fig. 20. The sampled output becomes

$$V_{\text{out}}[n] = \int_{t_{n-1}}^{t_{n}} V_{n}(t) D_{m}(t) dt$$

from which the autocorrelation of the output voltage samples, $R_{\text{out}}[m,n]$, can be calculated. As $D_{m}(t)$ is periodic with a period of $T_s$, $R_{\text{out}}[m,n]$ is wide-sense stationary and is given by [9]

$$R_{\text{out}}[m,n] = R_{\text{out}}[m-n] = R_{\text{out}}[l] = E[V_{\text{out}}[m] V_{\text{out}}[n]]$$

$$= E\left[ \frac{1}{C} \int_{t_{n-1}}^{t_{n}} V_{n}(t_{1}) V_{n}(t_{2}) D_{m}(t_{1}) D_{m}(t_{2}) dt_{1} dt_{2} \right]$$

(18)

where $D_{in}(t)$ and $D_{out}(t)$ are the time-varying conductances for $V_{out}[m]$ and $V_{out}[n]$, respectively. For non-TI-FA, they are the identical, but for TI-FA, as will be shown later, they are not necessarily the same. If we assume the noise source to be white
Gaussian with autocorrelation

\[ R_m(t_1, t_2) = \mathbb{E}[V(t_1) V(t_2)] = 2kT R_m(t_1) \delta(t_1 - t_2), \]

where \( k \) is the Boltzmann constant, \( T \) is the temperature in Kelvin, and \( \delta(t) \) is the Dirac delta function. Then we find that (18) can be expressed as

\[ R_m[l] = \frac{2kT}{C} \int_{v_{iR1}(lT)}^{v_{iR1}(lT+T_s)} R_m(t) D_m(t_1) D_m(t_2) dt_1 dt_2, \]

which for the non-TI case can be simplified into

\[ R_m[0] = \frac{2kT}{C} \int_{v_{iR1}(0T)}^{v_{iR1}(0T+T_s)} R_m(t) \left[ D_m(t_1) \right]^2 dt_1, \]

and \( R_m[l] = 0 \) when \( l \neq 0 \). The only remaining unknown factor \( D(t) \) can be found with the equivalent models like Fig. 4(c) using simple KCL/KVL analyses. From (21), the overall output noise voltage autocorrelation can be easily computed with superposition since the noise sources, i.e., \( R_m \), \( D(t) \), and \( G_m(t) \), are independent. For the circuit in Fig. 4(a), we consider the noise sources to be white Gaussian with autocorrelations

\[ R_n(t_1, t_2) = \mathbb{E}[V(t_1) V(t_2)] = 2kT \delta(t_1 - t_2), \]

and \( R_n[l] = 0 \) when \( l \neq 0 \). The corresponding PSD can be found

\[ S_n(e^{j\omega}) = R_n[0]. \]

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