

# 1ppm Settling Time Measurement for a Monolithic 18-Bit DAC

When Does the Last Angel Stop Dancing on a Speeding Pinhead?

Jim Williams

## Introduction

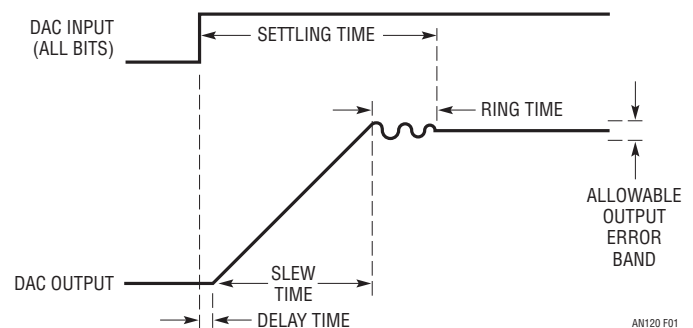
Performance requirements for instrumentation, function generation, inertial navigation systems, trimming, calibrators, ATE, medical apparatus and other precision applications are beginning to eclipse capabilities of 16-bit data converters. More specifically, 16-bit digital-to-analog converters (DACs) have been unable to provide required resolution in an increasing number of ultra-precision applications.

New components (see Components for 18-bit Digital-to-Analog Conversion, page 2) have made 18-bit DACs a practical design alternative<sup>1</sup>. These ICs provide 18-bit performance at reasonable cost compared to previous modular and hybrid technologies. The monolithic DACs DC and AC specifications approach or equal previous converters at significantly lower cost.

## DAC Settling Time

DAC DC specifications are relatively easy to verify. Measurement techniques are well understood, albeit often tedious. AC specifications require more sophisticated approaches to produce reliable information. In particular, the settling time of a DAC and its output amplifier is extraordinarily difficult to determine to 18-bit (4ppm) resolution. DAC settling time is the elapsed time from input code application until the output arrives at, and remains within, a specified error band around the final value. To measure a new 18-bit DAC, a settling time measurement technique has been developed with 20-bit (1ppm) resolution for times as short as 265ns. The new method will work with any DAC. Realizing this measurement capability and its performance verification has required an unusually intense, extensive and protracted effort. Hopefully, the data converter community will find the results useful<sup>2</sup>.

DAC settling time is usually specified for a full-scale 10V transition. Figure 1 shows that DAC settling time has three



**Figure 1. DAC Settling Time Components Include Delay, Slew and Ring Times. Fast Amplifiers Reduce Slew Time, Although Longer Ring Time Usually Results. Delay Time is Normally a Small Term**

distinct components. The *delay time* is very small and is almost entirely due to propagation delay through the DAC and output amplifier. During this interval, there is no output movement. During *slew time*, the output amplifier moves at its highest possible speed towards the final value. *Ring time* defines the region where the amplifier recovers from slewing and ceases movement within some defined error band. There is normally a trade-off between slew and ring time. Fast slewing amplifiers generally have extended ring times, complicating amplifier choice and frequency

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**Note 1.** See Appendix A, "A History of High Accuracy Digital-to-Analog Conversion".

**Note 2.** A historical note is in order. In early 1997, LTC's DAC design group tasked the author to measure 16-bit DAC settling time. The result was published in July 1998 as Application Note 74, "Component and Measurement Advances Ensure 16-Bit DAC Settling Time". Almost exactly 10 years later, the DAC group raised the ante, requesting 18-bit DAC settling time measurement. This constitutes 2 bits of resolution improvement per decade of author age. Since it was unclear how many decades the author (born 1948) had left, it was decided to double jump the performance requirement and attempt 20-bit resolution. In this way, even if the author is unavailable in 10 years, the DAC group will still get its remaining 2 bits.

## COMPONENTS FOR 18-BIT D/A CONVERSION

Components suitable for 18-bit D/A conversion are members of an elite class. 18 binary bits is one part in 262,144—just 0.0004% or 4 parts-per-million. This mandates a vanishingly small error budget and the demands on components are high. The LTC2757 digital-to-analog converter listed in the chart uses Si-Chrome thin-film resistors for high stability and linearity over

temperature. Gain drift is typically 0.25ppm/°C or about 4.6LSBs over 0°C to 70°C. Some amplifiers shown contribute less than 1LSB error over 0°C to 70°C with 18-bit DAC driven settling times of 1.8μs available. The references offer drifts as low as 1LSB over 0°C to 70°C with initial trimmed accuracy to 0.05%

**Short Form Descriptions of Components Suitable for 18-Bit Digital-to-Analog Conversion**

COMPONENT TYPE	ERROR CONTRIBUTION OVER 0°C TO 70°C	COMMENTS
LTC <sup>®</sup> 2757 DAC	≈4.6LSB Gain Drift 1LSB Linearity	Full Parallel Inputs Current Output
LT <sup>®</sup> 1001 Amplifier	<1LSB	Good Low Speed Choice 10mA Output Capability
LT1012 Amplifier	<1LSB	Good Low Speed Choice Low Power Consumption
LT1468/LT1468-2 Amplifier	<8LSB	1.8μs Settling to 18 Bits Fastest Available
LTC1150 Amplifier	<1LSB	Lowest Error. ≈10ms Settling Time. Requires LT1010 Output Buffer. Special Case. See Appendix E
LTZ1000A Reference	<1LSB	Lowest Drift Reference in This Group. 4ppm (1LSB)/Yr. Time Stability Typical
LM199A Reference-6.95V	≈4LSB	Low Drift. 10ppm (2.5LSB) Yr. Time Stability Typical
LT1021 Reference-10V	≈16LSB	Good General Purpose Choice
LT1027 Reference-5V	≈16LSB	Good General Purpose Choice
LT1236 Reference-10V	≈40LSB	Trimmed to 0.05% Absolute Accuracy

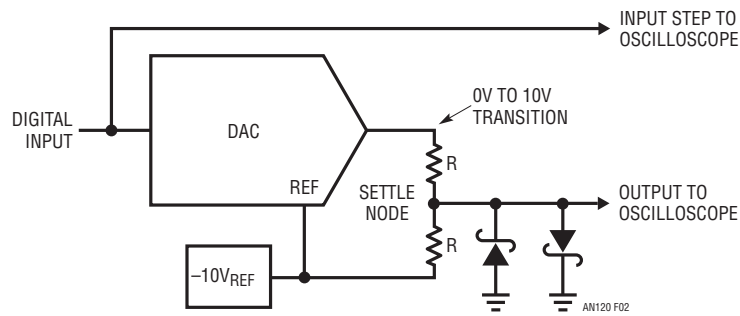
compensation. Additionally, the architecture of very fast amplifiers usually dictates trade-offs which degrade DC error terms<sup>3</sup>.

Measuring anything at any speed to 20-bit resolution (1ppm) is hard. Dynamic measurement to 20-bit resolution is particularly challenging. Reliable 1ppm DAC settling time measurement constitutes a high order difficulty problem requiring exceptional care in approach and experimental technique. This publication's remaining sections describe a method enabling an oscilloscope to accurately display DAC settling time information for a 10V step with 1ppm resolution (10μV) within 265ns. The approach employed permits observation of small amplitude information at the excursion limits of large waveforms without overdriving the oscilloscope.

## Considerations for Measuring DAC Settling Time

Historically, DAC settling time has been measured with circuits similar to that in Figure 2. The circuit uses the “false sum node” technique. The resistors and DAC form a bridge type network. Assuming ideal components, the DAC output will step to  $-V_{REF}$  when the DAC inputs move to all ones. During slew, the settle node is bounded by the diodes, limiting voltage excursion. When settling occurs, the oscilloscope probe voltage should be zero. Note that the resistor divider's attenuation means the probe's output will be one-half of the DAC's settled voltage.

**Note 3.** This issue is treated in detail in latter portions of the text. Also see Appendix D, “Practical Considerations for DAC-Amplifier Compensation”.



**Figure 2. Popular Summing Scheme for DAC Settling Time Measurement Provides Misleading Results. 18-Bit Measurement Causes >800x Oscilloscope Overdrive. Displayed Information is Meaningless**

In theory, this circuit allows settling to be observed to small amplitudes. In practice, it cannot be relied upon to produce useful measurements. The oscilloscope connection presents problems. As probe capacitance rises, AC loading of the resistor junction influences observed settling waveforms. A 10pF probe alleviates this problem but its 10x attenuation sacrifices oscilloscope gain. 1x probes are not suitable because of their excessive input capacitance. An active 1x FET probe will work, but a more significant issue remains.

The clamp diodes at the settle node are intended to reduce swing during amplifier slewing, preventing excessive oscilloscope overdrive. Unfortunately, oscilloscope overdrive recovery characteristics vary widely among different types and are not usually specified. The Schottky diodes' 400mV drop means the oscilloscope may see an unacceptable overload, bringing displayed results into question<sup>4</sup>.

At 10-bit resolution (10mV at the DAC output, resulting in 5mV at the oscilloscope), the oscilloscope typically undergoes a 2x overdrive at 50mV/DIV, and the desired 5mV baseline is just discernible. At 12-bit or higher resolution, the measurement becomes hopeless with this arrangement. Increasing oscilloscope gain brings commensurate increased vulnerability to overdrive induced errors. At 18 bits, there is clearly no chance of measurement integrity.

The preceding discussion indicates that measuring 18-bit settling time requires a high gain oscilloscope that is somehow immune to overdrive. The gain issue is addressable with an external wideband preamplifier that accurately amplifies the diode-clamped settle node. Getting around the overdrive problem is more difficult.

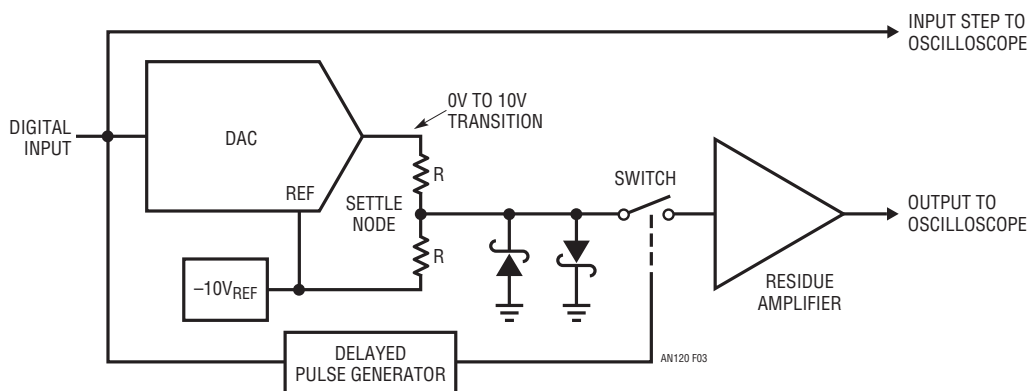
The only oscilloscope technology that offers inherent overdrive immunity is the classical sampling 'scope<sup>5</sup>. Unfortunately, these instruments are no longer manufactured (although still available on the secondary market). It is possible, however, to construct a circuit that utilizes sampling techniques to avoid the overload problem. Additionally, the circuit can be endowed with features particularly suited for measuring 20-bit DAC settling time.

### Sampling Based High Resolution DAC Settling Time Measurement

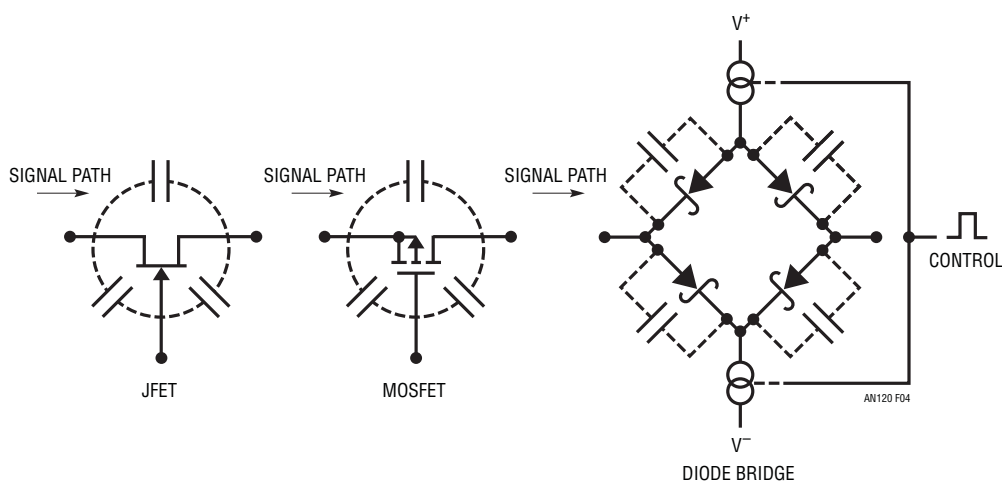
Figure 3 is a conceptual diagram of the 20-bit DAC settling time measurement circuit. This figure shares attributes with Figure 2, although some new features appear. In this case, the preamplified oscilloscope is connected to the settle point by a switch. The switch state is determined by a delayed pulse generator, which is triggered from the same pulse that controls the DAC. The delayed pulse generator's timing is arranged so that the switch does not close until settling is very nearly complete. In this way, the incoming waveform is sampled in time, as well as amplitude. The oscilloscope is never subjected to overdrive—no off-screen activity ever occurs.

**Note 4.** For a discussion of oscilloscope overdrive considerations, see Appendix B, "Evaluating Oscilloscope Overdrive Performance".

**Note 5.** Classical sampling oscilloscopes should not be confused with modern era digital sampling 'scopes that have overdrive restrictions. See Appendix B, "Evaluating Oscilloscope Overload Performance" for comparisons of various type oscilloscopes with respect to overdrive. For detailed discussion of classical sampling oscilloscope operation, see references 17 through 21 and 23 through 25. Reference 18 is noteworthy; it is the most clearly written, concise explanation of classical sampling instruments the author is aware of. A 12-page jewel.



**Figure 3. Conceptual Arrangement Eliminates Oscilloscope Overdrive. Delayed Pulse Generator Controls Switch, Preventing Oscilloscope from Monitoring Settle Node Until Settling is Nearly Complete**



**Figure 4. Conventional Choices for the Sampling Switch Include JFET, MOSFET and Diode Bridge. FET Parasitic Capacitances Result in Large Gate Drive Originated Feedthrough to Signal Path. Diode Bridge is Better; Its Small Parasitic Capacitances Tend to Cancel. Bridge Requires DC and AC Trims and Complex Drive Circuitry**

## Developing a Sampling Switch

Requirements for Figure 3's sampling switch are stringent. It must faithfully pass signal path information without introducing alien components, particularly those deriving from the switch command channel. Figure 4 shows conventional choices for the sampling switch. They include FETs and the diode bridge. The FET's parasitic gate to channel capacitances result in large gate drive originated feedthrough into the signal path. For almost all FETs, this feedthrough is many times larger than the signal to be observed, inducing overload and obviating the switches' purpose. The diode bridge is better; its small parasitic capacitances tend to cancel and the symmetrical, differential structure results in very low feedthrough. Practically, the bridge requires

DC and AC trims and complex drive and support circuitry. This approach, incarnated with great care, can reliably measure DAC settling time to 16-bit resolution<sup>6</sup>. Beyond 16 bits, residual feedthrough becomes objectionable and another approach is needed.

## Electronic Switch Equivalents

A low feedthrough, high resolution "switch" can be constructed with wideband active components. The great advantage of this approach is that the switch control channel can be maintained "in-band"; that is, its transition

**Note 6.** LTC Application Note 74, "Component and Measurement Advances Ensure 16-bit DAC Settling Time" utilized such a sampling bridge and it is detailed in that text.

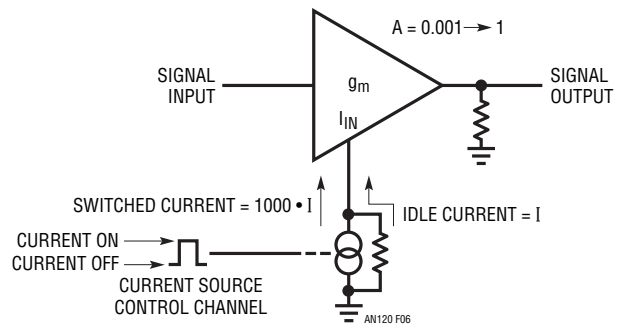
rate is within the circuits' bandpass. The circuit's wide bandwidth means the switch command transition is under control at all times. There are no out-of-band responses, greatly reducing feedthrough. Figure 5 lists some candidates for low feedthrough electronic switch equivalents. A and B, while theoretically possible, are cumbersome to implement. C and D are practical. C must be optimized for low feedthrough on rising and falling control pulse edges because of the multiplier's unrestricted wideband response. D's falling edge feedthrough is inherently minimized by the  $g_m$  amplifier's transconductance collapse when the control pulse goes low. This allows feedthrough to be optimized for the control pulse's rising edge without regard to falling edge effects. This is a significant advantage in constructing an electronic equivalent switch.

## Transconductance Amplifier Based Switch Equivalent

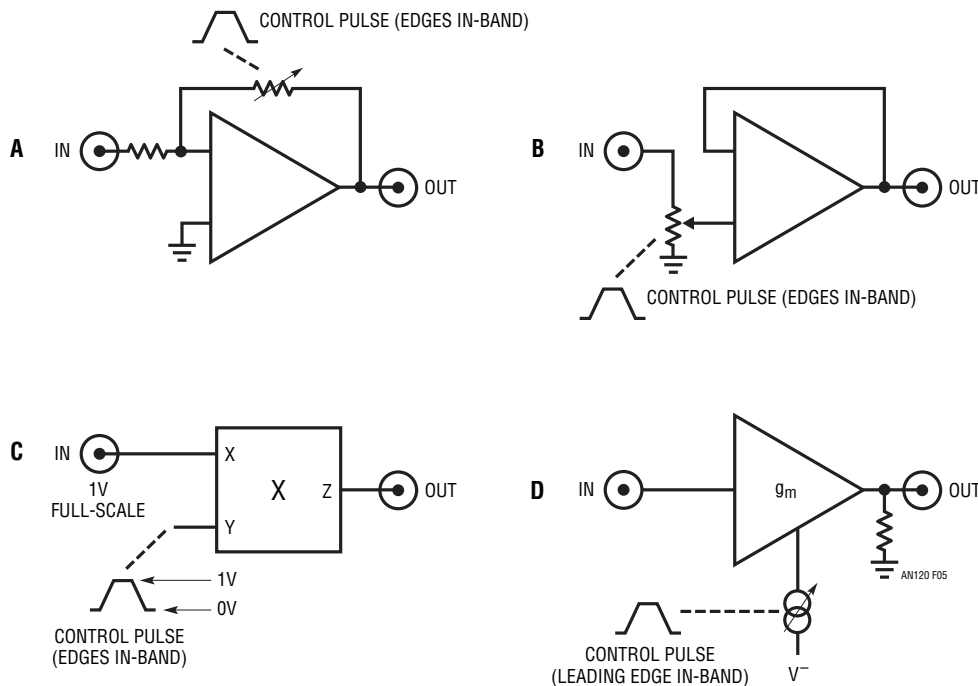
Figure 6 is a conceptual transconductance amplifier based "switch". The wideband control and signal paths faithfully track 1000:1 transconductance change, resulting in exceptionally pure switch dynamics. The switched current source is carefully optimized for lowest feedthrough on the rising control edge without regard to falling edge characteristics. The  $g_m$  amplifier's transconductance

collapse on the falling edge ensures low feedthrough for that condition, preventing oscilloscope overdrive. Figure 7 details the transconductance amplifier-based switch. This design switches signals over a  $\pm 30\text{mV}$  range with peak control channel feedthrough of millivolts and settling times inside 40ns.

The circuit approximates switch action by varying A1A's transconductance; the maximum gain is unity. At low transconductance, A1A's gain is nearly zero, and essentially no signal is passed. At maximum transconductance, signal



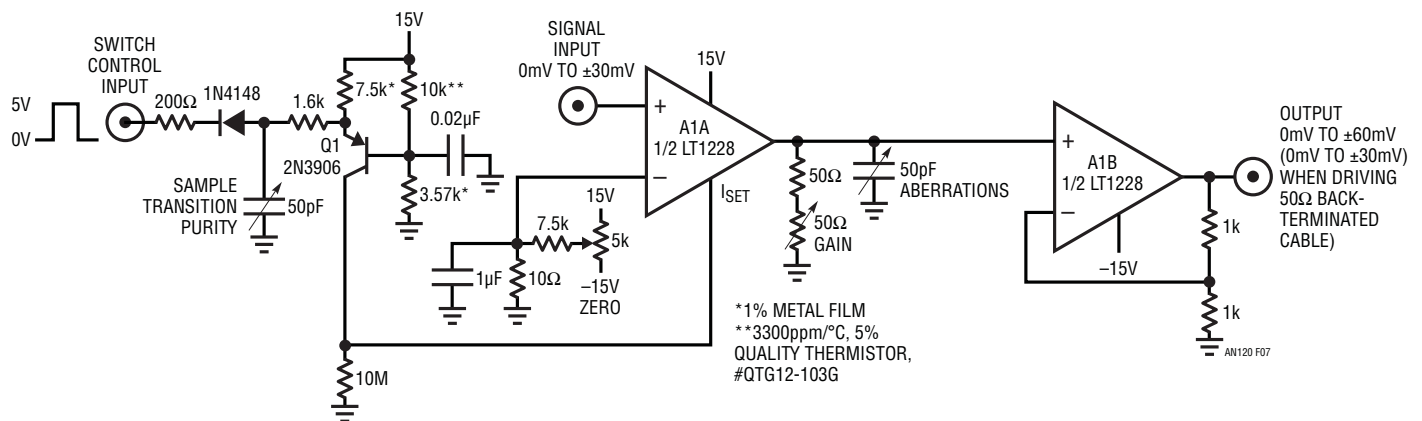
**Figure 6. Transconductance Amplifier Based "Switch" Has Minimal Control Channel Feedthrough. Wideband Control and Signal Paths Faithfully Track 1000:1 Transconductance Change, Resulting in Exceptionally Pure Switch Dynamics**



**Figure 5. Conceptual Low Feedthrough Electronic Switch Equivalents. A and B are Difficult to Implement, C and D are Practical. C Must be Optimized for Low Feedthrough on Rising and Falling Control Pulse Edges. D's Falling Edge Feedthrough is Inherently Minimized by Attendant Bandwidth Reduction**

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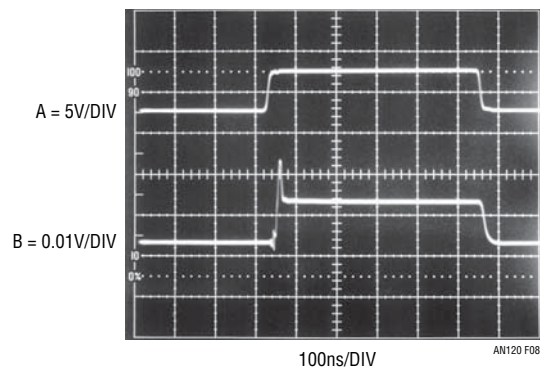
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**Figure 7. Transconductance Amplifier-Based 100MHz Low Level Switch Has Minimal Control Channel Feedthrough. A1A's Unity-Gain Output is Cleanly Switched by Logic Controlled Q1's Transconductance Bias. A1B Provides Buffering and Signal Path Gain**

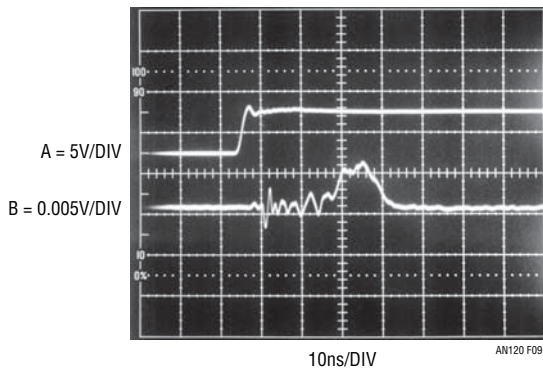
passes at unity gain. The amplifier and its transconductance control channel are very wideband, permitting them to faithfully track rapid variations in transconductance setting. This characteristic means the amplifier is never out of control, affording clean response and rapid settling to the “switched” input's value.

A1A, one section of an LT1228, is the wideband transconductance amplifier. Its voltage gain is determined by its output resistor load and the current magnitude into its “I<sub>SET</sub>” terminal. A1B, the second LT1228 section, unloads A1A's output. As shown, it provides a gain of 2, but when driving a back-terminated 50Ω cable, its effective gain is unity at the cable's receiving end. The back termination enforces a 50Ω environment. Current source Q1, controlled by the “switch control input”, sets A1A's transconductance, and, hence, gain. With Q1 gated off (control input at zero), the 10M resistor supplies about 1.5μA into A1A's I<sub>SET</sub> pin, resulting in a voltage gain of nearly zero, blocking the input signal. When the switch control input goes high, Q1 turns on, sourcing approximately 1.5mA into the I<sub>SET</sub> pin. This 1000:1 set current change forces maximum transconductance, causing the amplifier to assume unity gain and pass the input signal. Trims for zero and gain ensure accurate input signal replication at the circuit's output. The Q1 associated 50pF variable capacitor purifies turn-on switching. The specified 10k resistor at Q1 has a 3300ppm/°C temperature coefficient, compensating A1A's complementary transconductance temperature dependence to minimize gain drift.

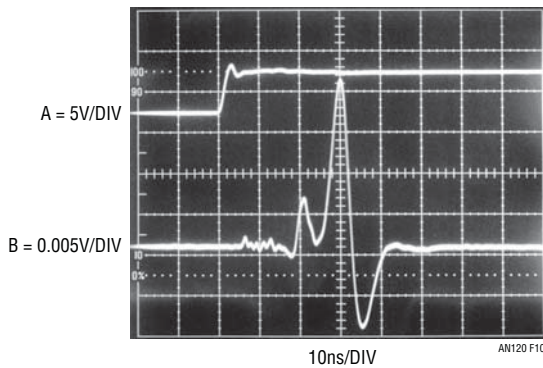


**Figure 8. Control Input (Trace A) Dictates Switch Output's (Trace B) Representation of 0.01V DC Input. Control Channel Feedthrough, Evident at Switch Turn-On, Settles in 20ns. Turn-Off Feedthrough is Undetectable Due to Deceased Signal Channel Transconductance and Bandwidth. C<sub>ABERRATION</sub> ≈ 35pF for this Test**

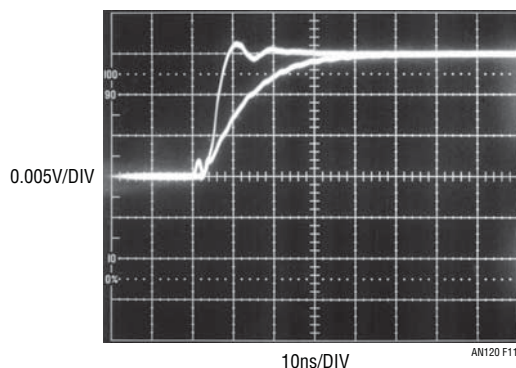
Figure 8 shows circuit response for a switched 10mV DC input and C<sub>ABERRATION</sub> = 35pF. When the control input (trace A) is low, no output (trace B) occurs. When the control input goes high, the output reproduces the input with “switch” feedthrough settling in about 20ns. Note that turn-off feedthrough is undetectable due to the 1000x transconductance reduction and attendant 25x bandwidth drop. Figure 9 speeds the sweep up to 10ns/division to examine zero volt settling detail. The output (trace B) settles inside 1mV 40ns after the switch control (trace A) goes high. Peak feedthrough excursion, damped by C<sub>ABERRATION</sub>, is only 5mV. Figure 10 was taken under identical conditions, except that C<sub>ABERRATION</sub> = 0pF. Feedthrough



**Figure 9. High Speed Delay and Feedthrough for 0V Signal Input. Output (Trace B) Peaks Only 0.005V Before Settling Inside 0.001V 40ns After Switch Control Command (Trace A).  $C_{ABERRATION} \approx 35\text{pF}$  for This Test**



**Figure 10. Identical Conditions as Figure 9 Except  $C_{ABERRATION} = 0\text{pF}$ . Feedthrough Related Peaking Increases to  $\approx 0.02\text{V}$ ; 0.001V Settling Time Remains at 40ns**



**Figure 11. Signal Channel Rise Time for  $C_{ABERRATION} = 0\text{pF}$  (Leftmost Trace) and  $\approx 35\text{pF}$  (Rightmost Trace) Record 3.5ns and 25ns, Respectively. Switch Control Input High for this Measurement. Photograph Utilizes Double Exposure Technique**

increases to approximately 20mV, although settling time to 1mV remains at 40ns. Figure 11, using double exposure technique, compares signal channel rise times for  $C_{ABERRATION} = 0\text{pF}$  (leftmost trace) and approximately 35pF (rightmost trace) with the control channel tied high. The larger  $C_{ABERRATION}$  value, while minimizing feedthrough amplitude (see Figure 9), increases rise time by 7x versus  $C_{ABERRATION} = 0\text{pF}$ .

The transconductance switches' small DC and AC errors nicely accommodate the applications' requirements. The low feedthrough, already sufficient, becomes irrelevant because its small time and amplitude error will be buried in the DAC ring time interval. The transconductance amplifier based "switch" points the way towards practical 1ppm DAC settling time measurement.

## DAC Settling Time Measurement Method

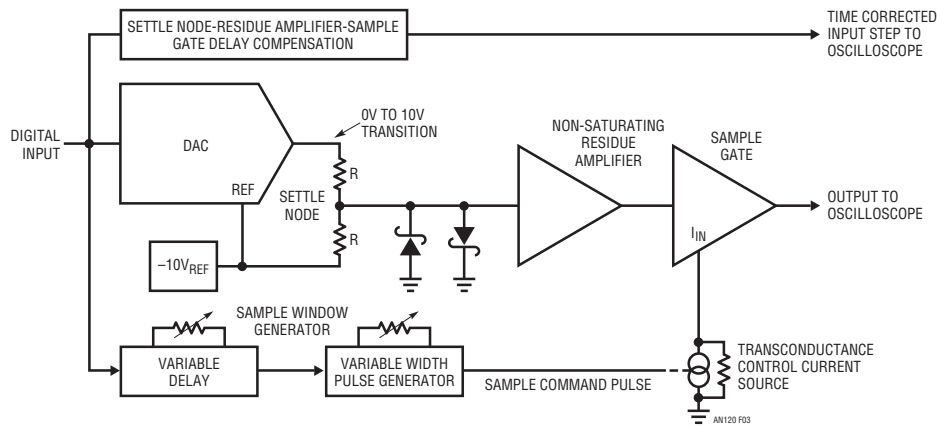
Figure 12, a more complete representation of Figure 3, utilizes the above described sampling switch. Figure 3's blocks appear in greater detail and some new refinements show up. The DAC-amplifier summing area is unchanged. Figure 3's delayed pulse generator has been split into two blocks; a delay and a pulse generator, both independently variable. The input step to the oscilloscope runs through a section that compensates settling time-measurement path propagation delay. This path includes settle node, amplifier and sample gate delays. The transconductance sampling switch ("sample gate"), driven from a non-saturating residue amplifier, feeds the oscilloscope. Placing the sampling switch after the residue amplifier gain further minimizes sample command feedthrough impact.

## Detailed Settling Time Circuitry

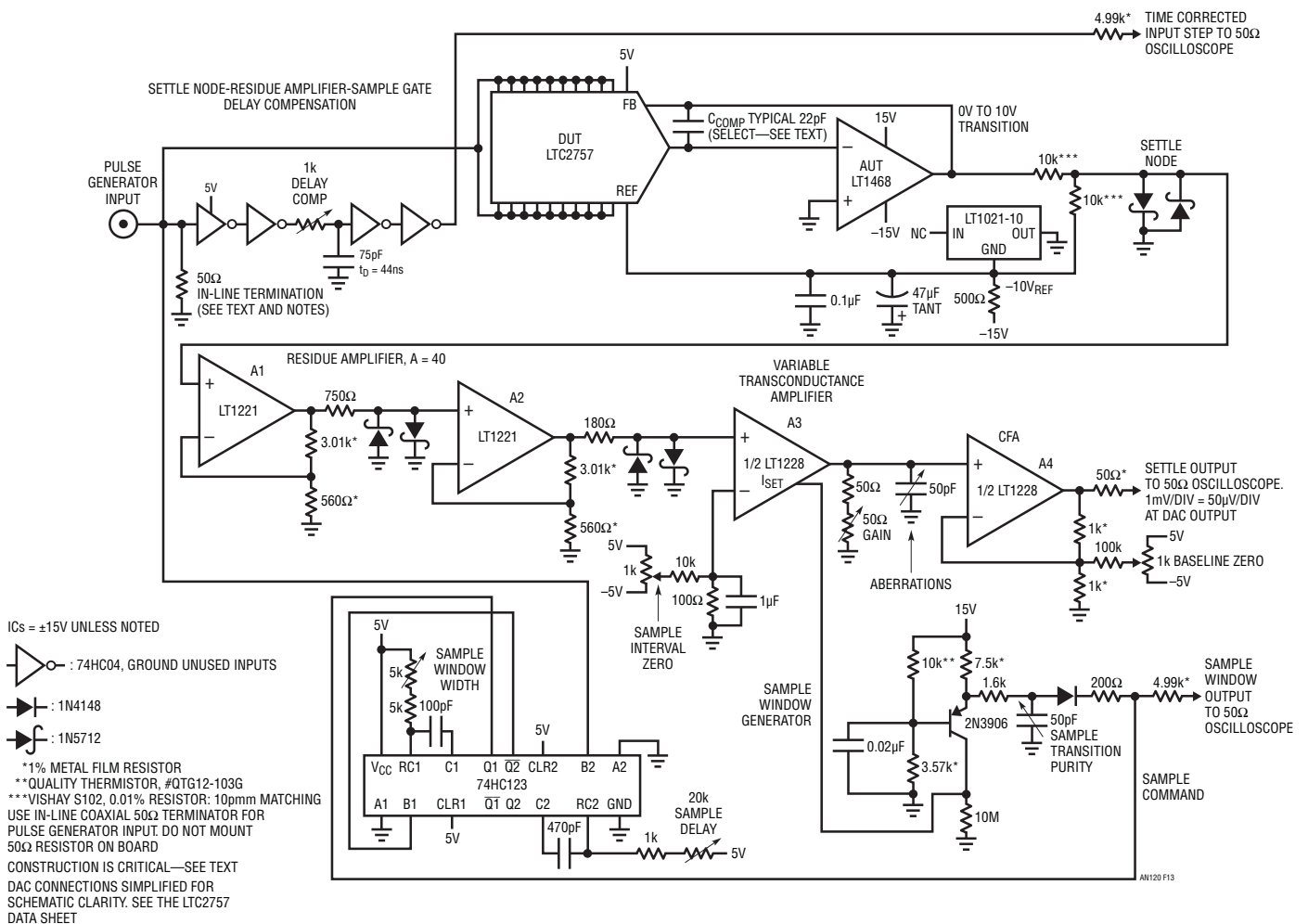
Figure 13 is a detailed schematic of the 20-bit DAC settling time measurement circuitry. The input pulse switches all DAC bits simultaneously and is also routed to the oscilloscope via the delay compensation network. The delay network, composed of CMOS inverters and an adjustable RC network, compensates the oscilloscope's input step signal for the 44ns delay through the circuit measurement path<sup>7</sup>. The DAC-amplifier output is compared against the

**Note 7.** See Appendix C, "Measuring and Compensating Signal Path Delay and Circuit Trimming Procedures".

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**Figure 12. Block Diagram of Sampling-Based DAC Settling Time Measurement Scheme. Placing Transconductance Controlled Sample Gate After Residue Amplifier Minimizes Sample Command Feedthrough Impact, Eliminating Oscilloscope Overdrive. Input Step Time Reference is Compensated for Settle Node, Residue Amplifier and Sample Gate Delays**



**Figure 13. Detailed DAC Settling Time Measurement Circuit Closely Follows Preceding Figure. Optimum Performance Requires Attention to Layout**



LT1021 10V reference via the precision 10k summing resistors. The LT1021 also furnishes the DAC reference, making the measurement ratiometric. The clamped settle node is unloaded by A1, which takes gain. A2 provides additional clamped gain for a total summing node referred amplification of 40. A2's output feeds the sampling switch whose operation is identical to Figure 7's description. The A1-A2 amplifier's clamping and gain are arranged so saturation never occurs—the amplifier is always in its active region.

The input pulse triggers the 74HC123 dual one shot. The one shot is arranged to produce a delayed (controllable by the 20k potentiometer) pulse whose width (controllable by the 5k potentiometer) sets sampling switch on-time. If the delay is set appropriately, the oscilloscope will not see any input until settling is nearly complete, eliminating overdrive. The sample window width is adjusted so that all remaining activity is observable. In this way, the oscilloscope output is reliable and meaningful data may be taken.

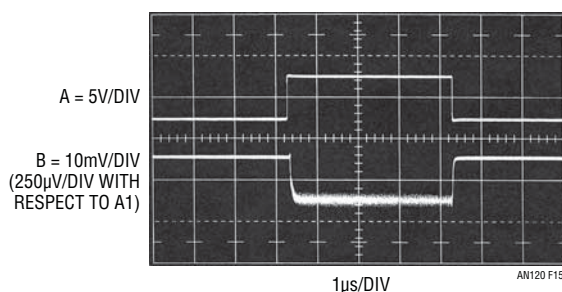
Figure 14 shows circuit waveforms. Trace A is the time corrected input pulse, trace B the sample gate, trace C the DAC-amplifier output and trace D the circuit output. When the sample gate goes high, trace D's switching is clean, the last millivolt of ring time is easily observed and the amplifier settles nicely to final value bounded by broadband noise. When the sample gate goes low, the transconductance switch goes off and no feedthrough is discernible. Note that there is no off-screen activity at any time—the oscilloscope is never subjected to overdrive.

The circuit requires trimming to achieve this level of performance<sup>8</sup>. Figure 15 shows a typical display resulting from poor "Sample Interval Zero" adjustment. This

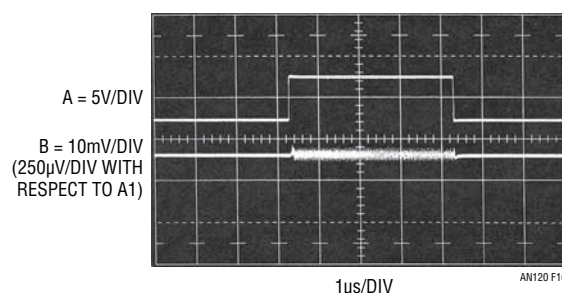
adjustment, corrected in Figure 16, results in a continuous baseline. Sample command feedthrough is just visible at trace B's leading edge. Figure 17 shows output response (trace B) to the sample command (trace A) turn-on before trimming "aberrations" and "transition purity"<sup>9</sup>. Delay is approximately 20ns with aberrations peaking 350μV and

**Note 8.** To maintain text flow and focus, trimming procedures are not presented here. Detailed trimming information appears in Appendix C.

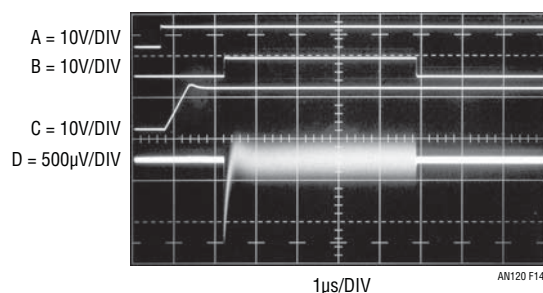
**Note 9.** A1's positive input was grounded via 5kΩ (precision 10k resistors disconnected) for Figure 17 and 18's tests.



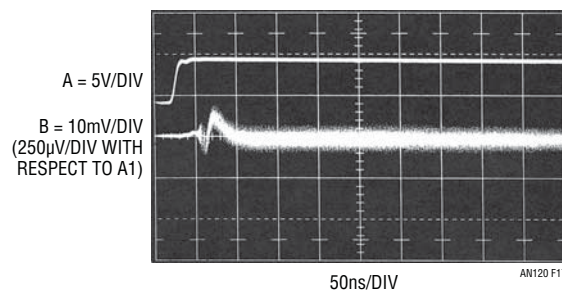
**Figure 15. Poor Sample Interval Zero Adjustment Causes Shifted Output Baseline (Trace B) During Trace A's Sample Interval**



**Figure 16. Trimmed Sample Interval Zero Has No Output Baseline Deviation (Trace B) During Sample Interval (Trace A). Sample Command Feedthrough is Just Visible at Trace B's Leading Edge**



**Figure 14. Settling Time Circuit Waveforms Include Time Corrected Input Pulse (Trace A), Sample Command (Trace B), DAC Output (Trace C) and Settling Time Output (Trace D). Sample Window Delay and Width are Variable**



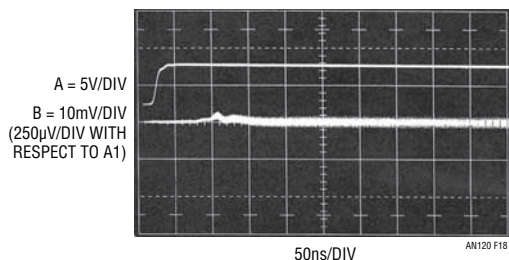
**Figure 17. Output Response (Trace B) To Sample Command (Trace A) Turn-On Before Trimming Aberrations and Transition Purity. Delay is ≈20ns. Aberrations Peak 350μV, Settle in 50ns. A1's Positive Input Grounded via 5kΩ for This and Succeeding Figures**

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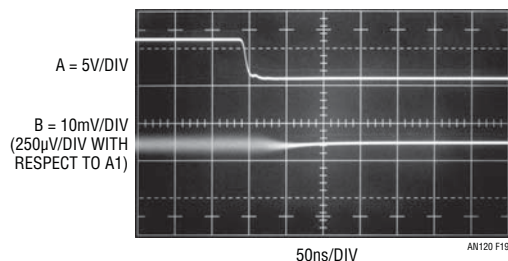
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settling in 50ns. Figure 18 shows post trim response to sample command turn-on. Delay increases to 70ns but aberrations peak only 50 $\mu$ V, settling in 50ns. Figure 19 shows output response (trace B) to sample command (trace A) turn-off. The 1000:1 transconductance drop ensures a clean transition independent of the turn-on optimized trims.

Circuit gain is adjusted with the indicated potentiometer.



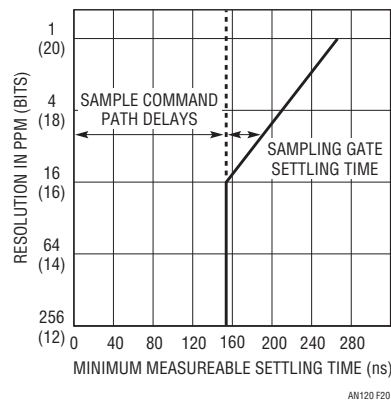
**Figure 18. Post-Trim Output Response (Trace B) To Sample Command Turn-On, Trace A. Delay Increases to 70ns but Aberrations Peak Only 50 $\mu$ V, Settling in 50ns**



**Figure 19. Output Response (Trace B) To Sample Command (Trace A) Turn-Off. 1000:1 Transconductance Drop Ensures Clean Transition, Independent of Trim State**

## Settling Time Circuit Performance

Figure 20 summarizes settling time circuit performance. The graph indicates the minimum measurable settling time for a given resolution. Speed limitations are imposed by sample command path delays and sample gate switching residue profile<sup>10</sup>. Minimum measurable settling time below 160ns is available to 16-bit resolution. Beyond this point, the sample gate's switching residue profile dictates increased minimum measurable settling time to about 265ns at 20 bits. Circuit noise limitations are imposed by the DAC/amplifier, summing resistors, and residue amplifier/sampling switch with about equal weighting. Because of this, resolution beyond approximately 15ppm requires filtering or noise averaging techniques.



**Figure 20. Minimum Measurable Settling Time vs Resolution. Limits are Imposed by Sample Command Path Delays and Sample Gate Settling Profile. Resolution Beyond  $\approx$ 15ppm Requires Filtering or Noise Averaging**

## Using the Sampling-Based Settling Time Circuit

It is good practice to “walk” the sampling window backwards in time from the settled region up to the last 100 $\mu$ V or so of amplifier movement so ring time cessation is observable. The sampling-based approach provides this capability and it is a very powerful measurement tool. Additionally, slower amplifiers may require extended delay and/or sampling window times. This may necessitate larger capacitor values in the 74HC123 one-shot timing networks.

Figure 21 shows DAC settling in an unfiltered bandpass. The DAC settles (trace B) to 16 bits 1.7 $\mu$ s after trace A's time corrected input step<sup>11</sup>. Sample gate feedthrough is undetectable, indicating higher resolution is possible without overdriving the oscilloscope. Noise is the fundamental measurement limit. Figure 22 attenuates noise by reducing measurement bandwidth to 250kHz. Trace assignments are as in the previous photo. 18-bit settling (4ppm) requires approximately 5 $\mu$ s. The reduced bandwidth permits higher resolution although the indicated settling time is likely pessimistic due to the filter's lag. Figure 23, decreasing bandwidth to 50kHz, permits 19-bit (2ppm) resolution with indicated settling in about 9 $\mu$ s. Again, the same filtering which permits high resolution almost certainly lengthens observed settling time.

**Note 10.** Driving the sample command path (74HC123 B2 input) with a phase-advanced version of the pulse generator input largely eliminates sample command path delay induced error, considerably improving minimum measurable settling time. This benefit is not germane to the present efforts purposes and was not implemented

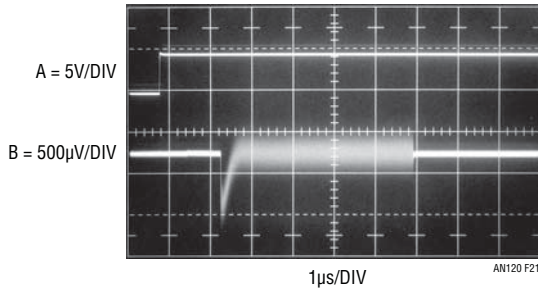
**Note 11.** Settling time is significantly affected by the DAC-amplifier compensation capacitor. See Appendix D, “Practical Considerations for DAC-amplifier Compensation” for tutorial.

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Figure 24 uses noise averaging techniques to measure settling time to 20 bits (1ppm-10 $\mu$ V) without the band limiting filter's time penalty<sup>12</sup>. Photo A shows the DAC-amplifier adjusted for overdamped response, B and C

underdamped and optimum responses, respectively. Averaging eliminates noise, permitting determination of settling time due to DAC dynamics<sup>13</sup>. Settling time ranges from 4 $\mu$ s to 6 $\mu$ s with fractional LSB tailing evident.

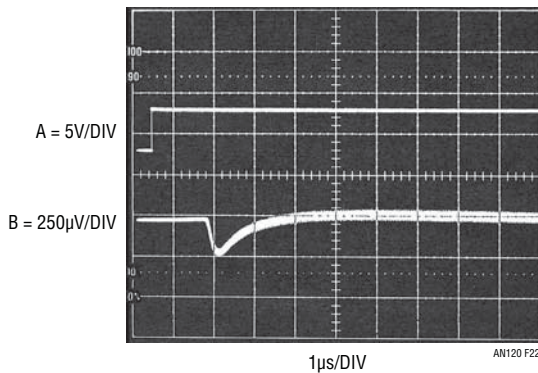
*Note: This application note was derived from a manuscript originally prepared for publication in EDN magazine.*



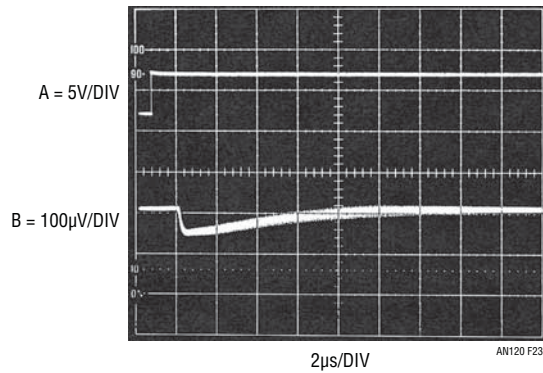
**Figure 21. 0V to 10V DAC Settling in Unfiltered Bandpass. DAC Settles (Settle Output, Trace B) to 16 Bits (15ppm) <2 $\mu$ s After Trace A's Time Corrected Input Step. Sample Gate Feedthrough is Well Controlled, Indicating Higher Resolution is Possible Without Overdriving Oscilloscope. Noise Limits Measurement**

**Note 12.** Most oscilloscopes require preamplification to resolve Figure 24's signal amplitudes. See Appendix I, "Auxiliary Circuits" for an example.

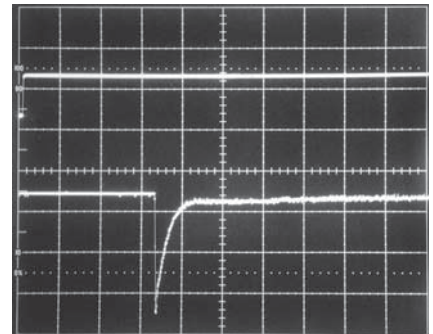
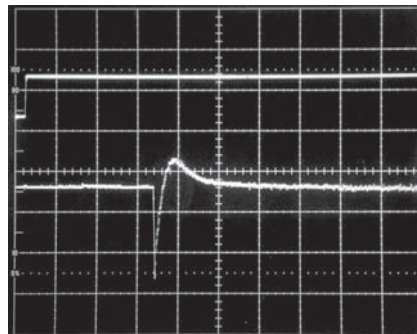
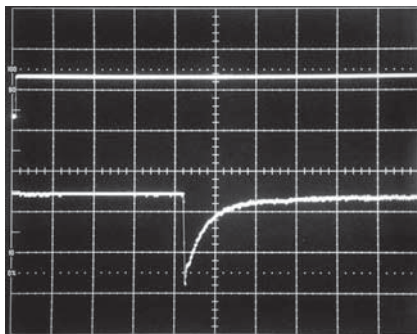
**Note 13.** More properly, this measurement determines DAC settling time due solely to step input initiated dynamics. For this reason, Figure 24's averaged results may be considered somewhat academic. Noise limits measurement certainty at any given instant to approximately 100 $\mu$ V. It is not unreasonable to maintain that this 100 $\mu$ V of noise means the DAC never settles inside this limit. The averaged measurement defines settling time with noise limitations removed. Hopefully, this disclosure will appease technolawyers among the readership.



**Figure 22. Same Trace Assignments as Previous Photo; Measurement Taken in 250kHz Bandpass. Settling to 18 Bits (4ppm) Requires  $\approx$  5 $\mu$ s. Filtering Permits Increased Resolution Although Indicated Settling Time Increases**



**Figure 23. 19 Bit (2ppm) Settling is Discernable About 9 $\mu$ s After Input Command in 50kHz Bandwidth**



ALL PHOTOS  
TRACE A = 5V/DIV  
TRACE B = 25 $\mu$ V/DIV (AVERAGED)  
HORIZ = 1 $\mu$ s/DIV

**A**

**B**

**C**

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**Figure 24. Noise Averaging Oscilloscope Permits 1ppm, (10 $\mu$ V) Settling Time Measurement Without Bandlimiting Filter Time Penalty. Photo A Shows Overdamped Response, B and C Underdamped and Optimum Responses, Respectively. Averaging Eliminates Noise, Permitting Determination of Settling Time Due To DAC Dynamics. Settling Times Range From 4 $\mu$ s to 6 $\mu$ s; Fractional LSB Tailing is Evident**

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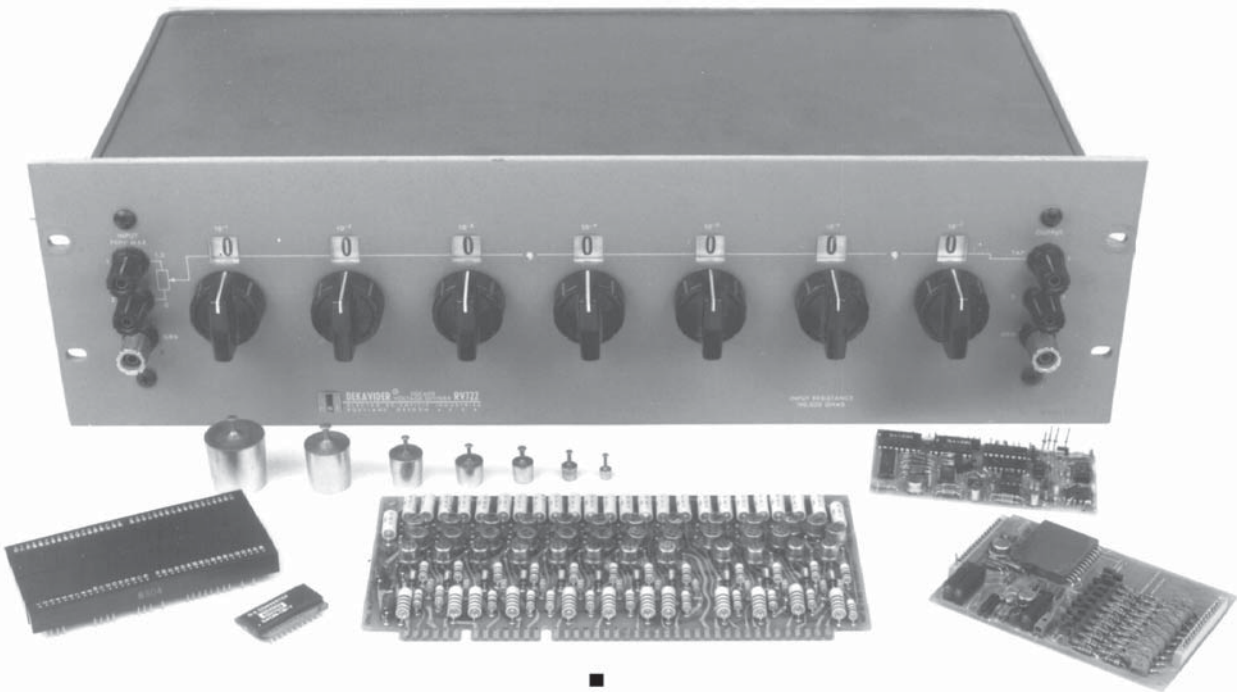
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## APPENDIX A

## A HISTORY OF HIGH ACCURACY DIGITAL-TO-ANALOG CONVERSION

People have been converting digital-to-analog quantities for a long time. Probably among the earliest uses was the summing of calibrated weights (Figure A1, left) in weighing applications. Early electrical digital-to-analog conversion inevitably involved switches and resistors of different values, usually arranged in decades. The application was often the calibrated balancing of a bridge or reading, via null detection, some unknown voltage. The most accurate resistor-based DAC of this type is Lord Kelvin's Kelvin-Varley divider (Figure, large box). Based on switched resistor ratios, it can achieve ratio accuracies of 0.1 ppm (23+ bits) and is still widely employed in standards laboratories. High speed digital-to-analog conversion resorts to electronically switching the resistor network. Early electronic DACs were built at the board level using discrete precision resistors and Germanium transistors (Figure, center foreground, is a 12-bit DAC from a Minuteman missile D-17B inertial

navigation system, circa 1962). The first electronically switched DACs available as standard product were probably those produced by Pastoriza Electronics in the mid 1960s. Other manufacturers followed and discrete-and monolithically-based modular DACs (Figure, right and left) became popular by the 1970s. The units were often potted (Figure, left) for ruggedness, performance or to (hopefully) preserve proprietary knowledge. Hybrid technology produced smaller package size (Figure, left foreground). The development of Si-Chrome resistors permitted precision monolithic DACs such as the LTC2757 (Figure, immediate foreground). In keeping with all things monolithic, the cost-performance trade-off of modern high resolution IC DACs is a bargain. Think of it! An 18-bit DAC in an IC package. What Lord Kelvin would have given for a credit card and LTC's phone number.



**Figure A1. Historically Significant Digital-to-Analog Converters Include: Weight Set (Center Left), 23+ Bit Kelvin-Varley Divider (Large Box), Hybrid, Board and Modular Types, and the LTC2757 IC (Foreground). Where Will It All End?**

## APPENDIX B

### EVALUATING OSCILLOSCOPE OVERDRIVE PERFORMANCE

The settling-time circuit is heavily oriented towards eliminating overdrive at the monitoring oscilloscope. Oscilloscope recovery from overdrive is a murky area and almost never specified. How long must one wait after an overdrive before the display can be taken seriously? The answer to this question is quite complex. Factors involved include the degree of overdrive, its duty cycle, its magnitude in time and amplitude and other considerations. Oscilloscope response to overdrive varies widely between types and markedly different behavior can be observed in any individual instrument. For example, the recovery time for a 100x overload at 0.005V/DIV may be very different than at 0.1V/DIV. The recovery characteristic may also vary with waveform shape, DC content and repetition rate. With so many variables, it is clear that measurements involving oscilloscope overdrive must be approached with caution.

Why do most oscilloscopes have so much trouble recovering from overdrive? The answer to this question requires some study of the three basic oscilloscope types' vertical paths. The types include analog (Figure B1A), digital (Figure B1B) and classical sampling (Figure B1C) oscilloscopes. Analog and digital 'scopes are susceptible to overdrive. The classical sampling 'scope is the only architecture that is inherently immune to overdrive.

An analog oscilloscope (Figure B1A) is a real-time, continuous linear system<sup>1</sup>. The input is applied to an attenuator, which is unloaded by a wideband buffer. The vertical preamp provides gain, and drives the trigger pick-off, delay line and the vertical output amplifier. The attenuator and delay line are passive elements and require little comment. The buffer, preamp and vertical output amplifier are complex linear gain blocks, each with dynamic operating range restrictions. Additionally, the operating point of each block may be set by inherent circuit balance, low frequency stabilization paths or both. When the input is overdriven, one or more of these stages may saturate, forcing internal nodes and components to abnormal operating points and temperatures. When the overload ceases, full recovery of the electronic and thermal time constants may require surprising lengths of time<sup>2</sup>.

The digital sampling oscilloscope (Figure B1B) eliminates the vertical output amplifier, but has an attenuator buffer and amplifiers ahead of the A/D converter. Because of this, it is similarly susceptible to overdrive recovery problems.

The classical sampling oscilloscope is unique. Its nature of operation makes it inherently immune to overload. Figure B1C shows why. The sampling occurs before any gain is taken in the system. Unlike Figure B1B's digitally sampled 'scope, the input is fully passive to the sampling point. Additionally, the output is fed back to the sampling bridge, maintaining its operating point over a very wide range of inputs. The dynamic swing available to maintain the bridge output is large and easily accommodates a wide range of oscilloscope inputs. Because of all this, the amplifiers in this instrument do not see overload, even at 1000x overdrives, and there is no recovery problem. Additional immunity derives from the instrument's relatively slow sample rate—even if the amplifiers were overloaded, they would have plenty of time to recover between samples<sup>3</sup>.

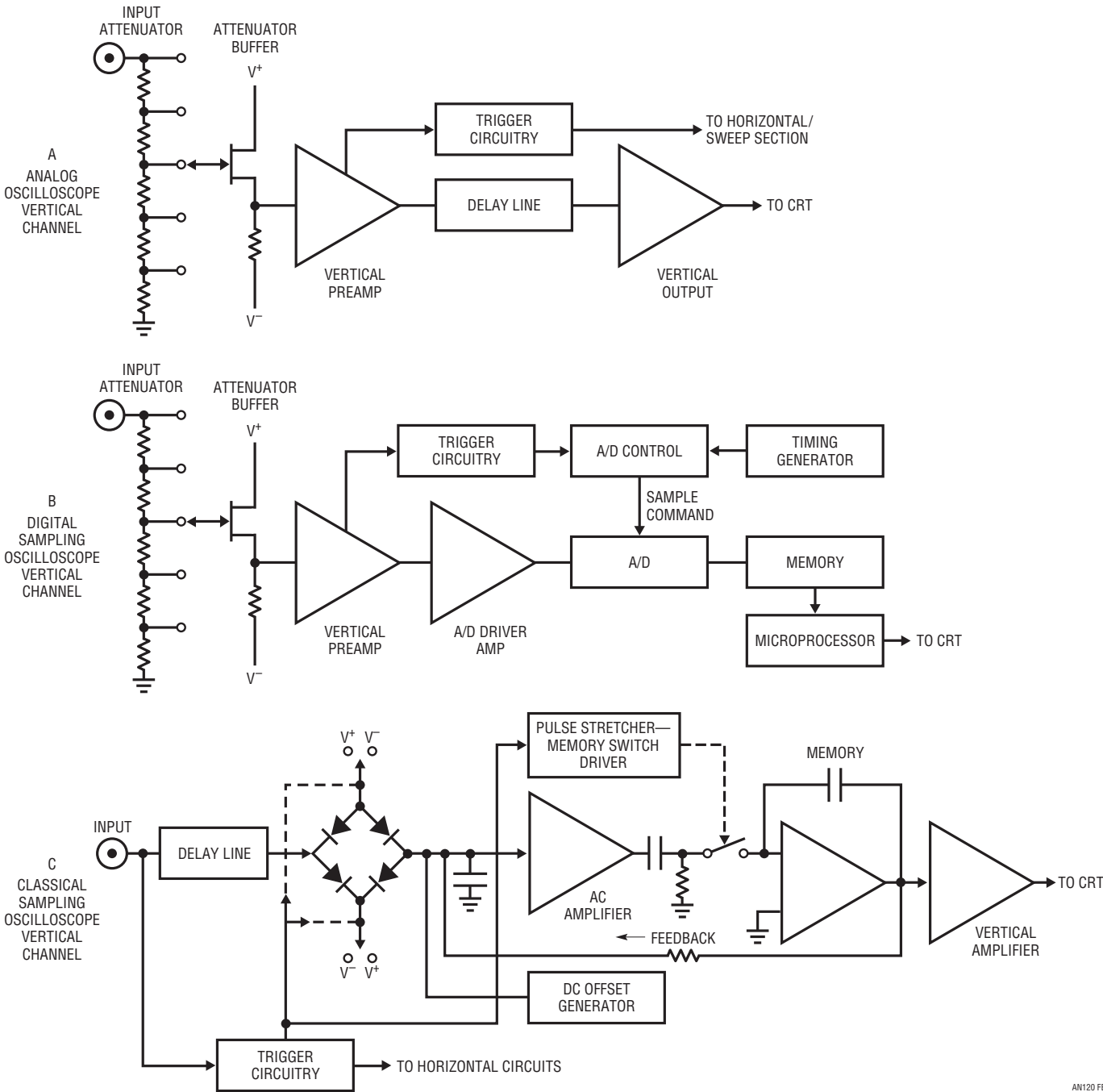
The designers of classical sampling 'scopes capitalized on the overdrive immunity by including variable DC offset generators to bias the feedback loop (see Figure B1C, lower right). This permits the user to offset a large input, so small amplitude activity on top of the signal can be accurately observed. This is ideal for, among other things, settling time measurements. Unfortunately, classical sampling oscilloscopes are no longer manufactured, so if you have one, take care of it!

Although analog and digital oscilloscopes are susceptible to overdrive, many types can tolerate some degree of this abuse. The early portion of this Appendix stressed that measurements involving oscilloscope overdrive must be approached with caution. Nevertheless, a simple test can indicate when the oscilloscope is being deleteriously affected by overdrive.

**Note 1:** Ergo, the Real Thing. Hopelessly bigoted residents of this locale mourn the passing of the analog 'scope era and frantically hoard every instrument they can find.

**Note 2:** Some discussion of input overdrive effects in analog oscilloscope circuitry is found in reference 13.

**Note 3:** Additional information and detailed treatment of classical sampling oscilloscope operation appears in references 17-20 and 23-25.



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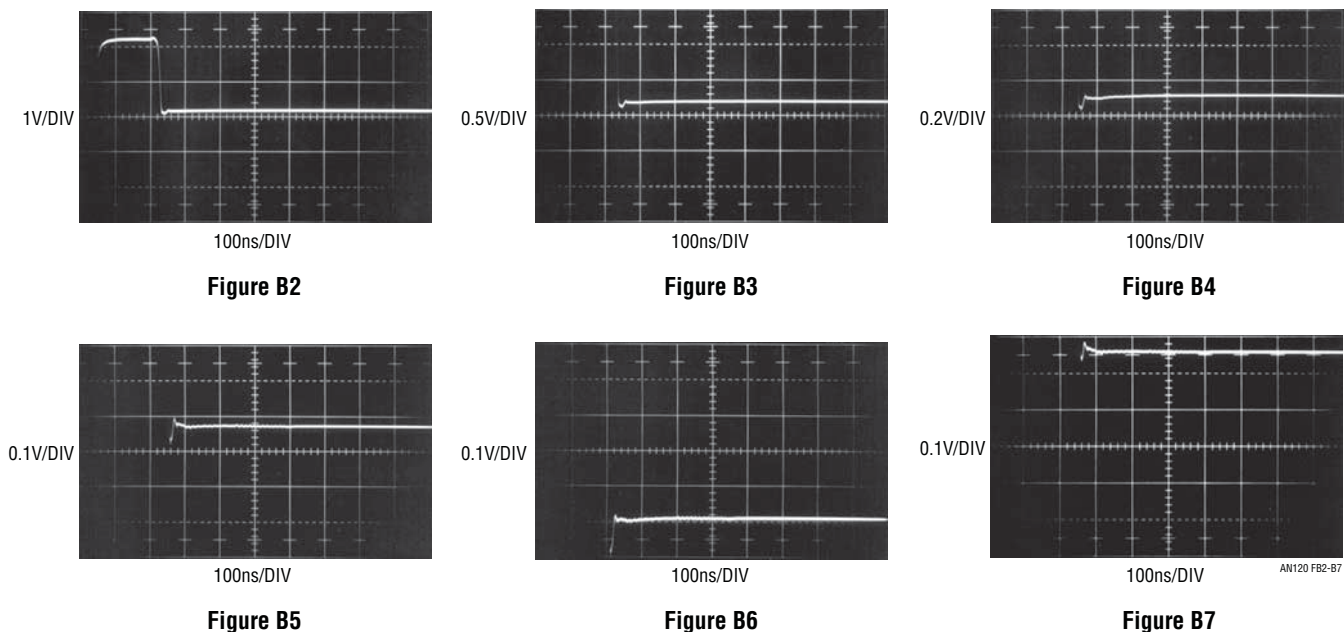
**Figure B1. Simplified Vertical Channel Diagrams for Different Type Oscilloscopes. Only the Classical Sampling Scope (C) Has Inherent Overdrive Immunity. Offset Generator Allows Viewing Small Signals Riding on Large Excursions**

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The waveform to be expanded is placed on the screen at a vertical sensitivity that eliminates all off-screen activity. Figure B2 shows the display. The lower right hand portion is to be expanded. Increasing the vertical sensitivity by a factor of two (Figure B3) drives the waveform off-screen, but the remaining display appears reasonable. Amplitude has doubled and waveshape is consistent with the original display. Looking carefully, it is possible to see small amplitude information presented as a dip in the waveform at about the third vertical division. Some small disturbances are also visible. This observed expansion of the original waveform is believable. In Figure B4, gain has been further increased, and all the features of Figure B3 are amplified accordingly. The basic waveshape appears clearer and the dip and small disturbances are also easier to see. No new waveform characteristics are observed. Figure B5 brings some unpleasant surprises. This increase in gain causes definite distortion. The initial negative-going peak, although larger, has a different shape. Its bottom appears

less broad than in Figure B4. Additionally, the peak's positive recovery is shaped slightly differently. A new rippling disturbance is visible in the center of the screen. This kind of change indicates that the oscilloscope is having trouble. A further test can confirm that this waveform is being influenced by overloading. In Figure B6, gain remains the same but the vertical position knob has been used to reposition the display at the screen's bottom.<sup>4</sup> This shifts the oscilloscope's DC operating point which, under normal circumstances, should not affect the displayed waveform. Instead, a marked shift in waveform amplitude and outline occurs. Repositioning the waveform to the screen's top produces a differently distorted waveform (Figure B7). It is obvious that for this particular waveform, accurate results cannot be obtained at this gain.

**Note 4:** *Knobs* (derived from Middle English, "knobbe", akin to Middle Low German, "knubbe"), cylindrically shaped, finger rotatable panel controls for controlling instrument functions, were utilized by the ancients.



**Figures B2-B7. The Overdrive Limit is Determined by Progressively Increasing Oscilloscope Gain and Watching for Waveform Aberrations**



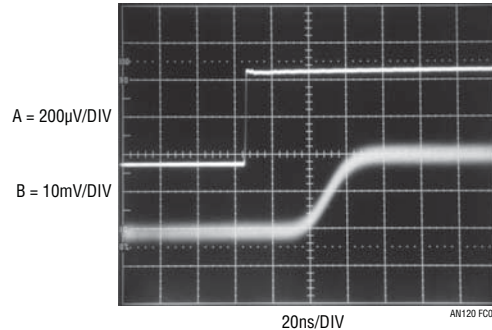
## APPENDIX C

### MEASURING AND COMPENSATING SIGNAL PATH DELAY AND CIRCUIT TRIMMING PROCEDURES

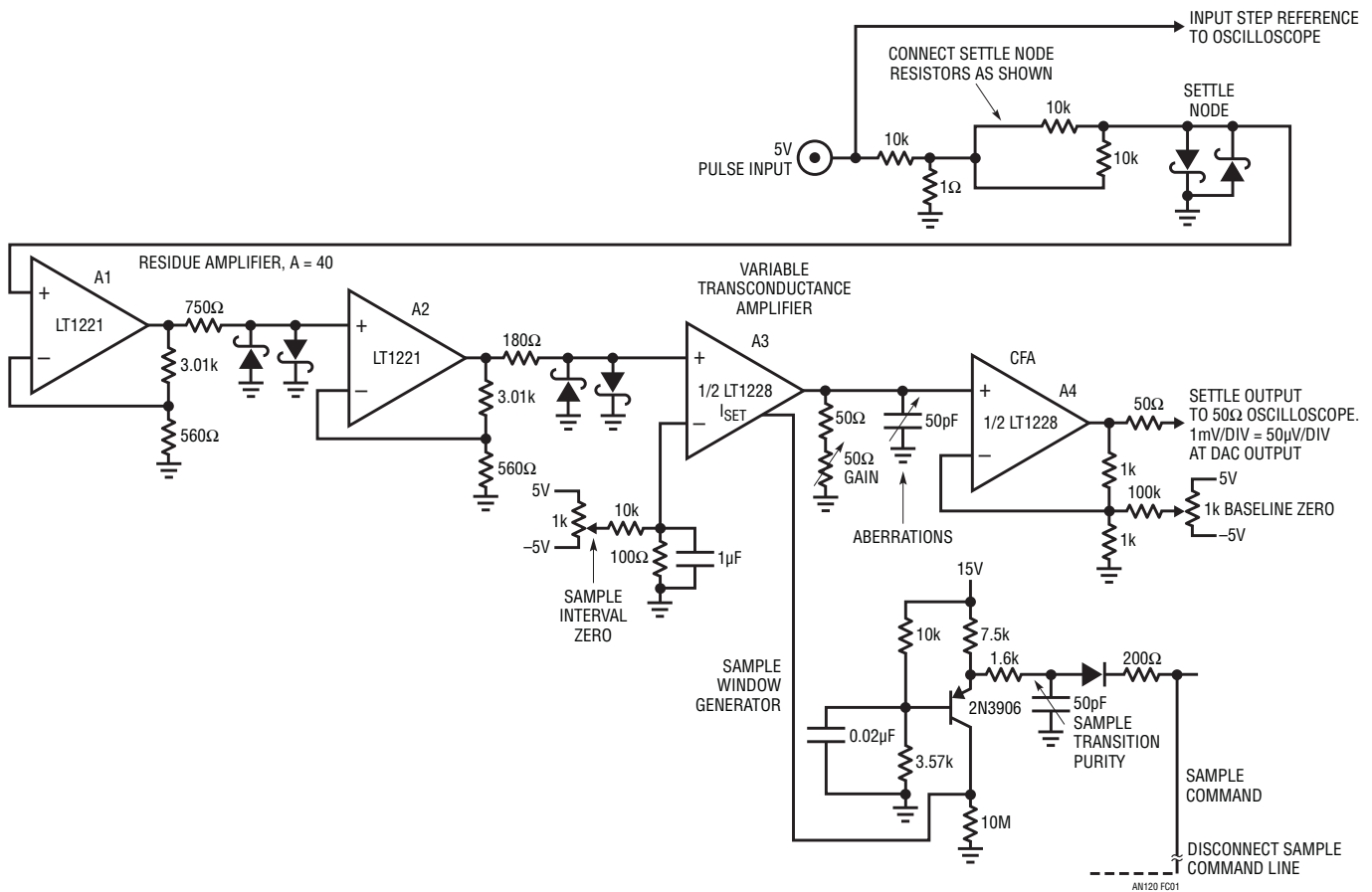
#### Delay Compensation

The settling time circuit utilizes an adjustable delay network to time correct the input pulse for delays in the signal-processing path. Typically, these delays introduce errors of a few percent, so a first-order correction is adequate. Setting the delay trim involves observing the network's input-output delay and adjusting for the appropriate time interval. Determining the "appropriate" time interval is somewhat more complex. Measuring the settling time circuit's signal path delay involves modifications to Figure 13, shown in Figure C1. These changes lock the circuit into its "sample" mode, permitting an input-to-output delay measurement under signal-level conditions similar to normal operation. In Figure C2, trace A is the

pulse-generator input at 200 $\mu$ V/DIV (note 10k-1 $\Omega$  divider feeding the settle node). Trace B shows the circuit output at A4, delayed by about 44ns. This delay is a small error, but is readily corrected by adjusting the delay network for the same time lag. If appendix F's serial interface is utilized, 10ns should be added to the delay correction. Similarly, if appendix I's post amplifier is used, the delay correction must be increased by 17ns.



**Figure C2. Sampling Circuit Input-Output Delay Measures About 44ns**



**Figure C1. Partial Text Figure 13 Schematic Shows Modifications for Measuring Signal Path Delay. Changes Lock Circuit into Sample Mode, Permitting Input-to-Output Delay Measurement**

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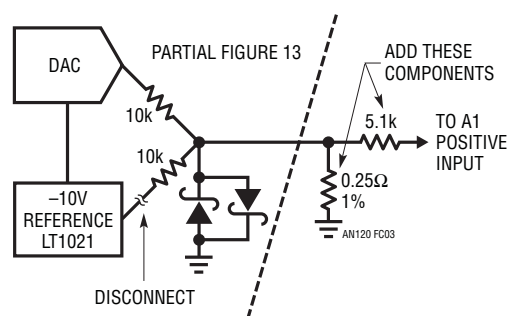
## Circuit Trimming Procedure

The following procedure, given in numerical order, trims the settling time circuit for optimum performance. It is advisable to execute trimming in the order given, avoiding out-of-sequence adjustments.

1. Turn off input pulses.
2. Trim “Baseline Zero” for 0V out at oscilloscope at 10mV per division or less.
3. Disconnect precision 10k resistors and ground settle node via 5.1k $\Omega$ .
4. Set sample delay to mid-range, sample window width to minimum.
5. Drive pulse generator input with 40kHz square wave.
6. Adjust “Sample Interval Zero” for no offset between the sample interval and the unsampled baseline<sup>1</sup>.
7. Adjust “Sample Transition Purity” and “Aberration” trims for minimum amplitude disturbances when the sample gate opens with oscilloscope horizontal at 50ns per division and vertical sensitivity of 10mV per division.
8. Reconnect precision 10k resistors and remove 5.1k $\Omega$  unit from the settle node.
9. Adjust “Delay Compensation” for 44ns delay from the pulse generator input to the time corrected output pulse.

10. Turn off input pulses. Disconnect the pulse generator and its 50 $\Omega$  termination. Apply 5V DC to the pulse input.
11. Connect Figure C3’s network to the settle node. The added components shown furnish a 250 $\mu$ V DC gain calibration source when the input pulses are replaced by a 5V level. Under the figure’s conditions, the DAC assumes a 10V output with the 5.1k resistor mimicking the 10k $\Omega$  divider output impedance at A1. Figure 13’s “Gain” trim is adjusted for a 10mV DC deflection at the oscilloscope. This completes the trimming procedure and the circuit is ready for use.

**Note 1.** The “Sample Interval Zero” trim is unnecessary if Appendix I’s optional auto-zero circuitry is used.

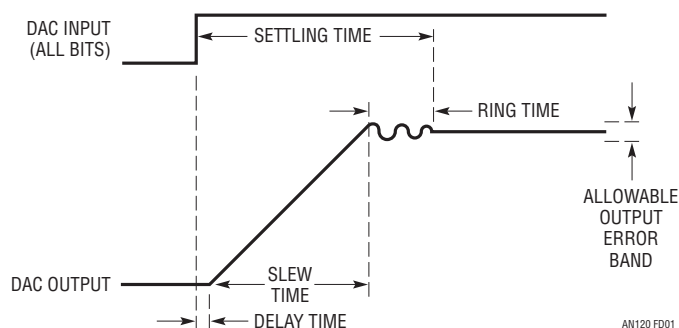


**Figure C3.** Added Components Furnish 250 $\mu$ V Gain Calibration Source with Input Pulses Replaced by 5V Level. DAC Output Assumes 10V Reference Potential Under These Conditions; 5.1k Resistor Mimics 10k $\Omega$  Divider Output Impedance at A1

## APPENDIX D

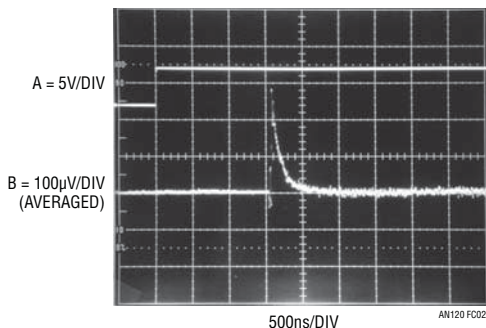
### PRACTICAL CONSIDERATIONS FOR DAC-AMPLIFIER COMPENSATION

There are a number of practical considerations in compensating the DAC-amplifier pair to get fastest settling time. Our study begins by revisiting text Figure 1 (repeated here as Figure D1). Settling time components include delay, slew and ring times. Delay is due to propagation time through the DAC-amplifier and is a very small term. Slew time is set by the amplifier’s maximum speed. Ring time defines the region where the amplifier recovers from slewing and ceases movement within some defined error band. Once a DAC-amplifier pair have been chosen, only ring time is



**Figure D1.** DAC-Amplifier Settling Time Components Include Delay, Slew and Ring Times. For Given Components, Only Ring Time is Readily Adjustable

readily adjustable. Because slew time is usually the dominant lag, it is tempting to select the fastest slewing amplifier available to obtain best settling. Unfortunately, fast slewing amplifiers usually have extended ring times, negating their brute force speed advantage. The penalty for raw speed is, invariably, prolonged ringing, which can only be damped with large compensation capacitors. Such compensation works, but results in protracted settling times. The key to good settling times is to choose an amplifier with the right balance of slew rate and recovery characteristics and compensate it properly. This is harder than it sounds because amplifier settling time cannot be predicted or extrapolated from any combination of data sheet specifications. It must be measured in the intended configuration. In the case of a DAC-amplifier, a number of terms combine to influence settling time. They include amplifier slew rate and AC dynamics, DAC output resistance and capacitance, and the compensation capacitor. These terms interact in a complex manner, making predictions hazardous<sup>1</sup>. If the DAC's parasitics are eliminated and replaced with a pure resistive source, amplifier settling time is still not readily predictable. The DAC's output impedance terms just make a difficult problem more messy. The only real handle available to deal with all this is the feedback compensation capacitor,  $C_F$ .  $C_F$ 's purpose is to roll off amplifier gain at the frequency that permits best dynamic response. Normally, the DAC's current output is unloaded directly into the amplifier's summing junction, placing the DAC's parasitic capacitance to ground at the amplifier's input. The capacitance introduces feedback phase shift at high frequencies, forcing the amplifier to "hunt" and ring about the final value before settling. Different DACs have different values of output capacitance. CMOS DACs have the highest output capacitance, typically 100pF, and it varies with code.

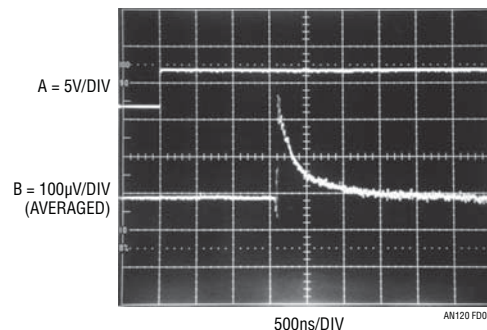


**Figure D2. Optimized Compensation Capacitor Permits Nearly Critically Damped Response, Faster Settling Time,  $t_{SETTLE} = 1.8\mu s$  to 0.0004% (18 Bits)**

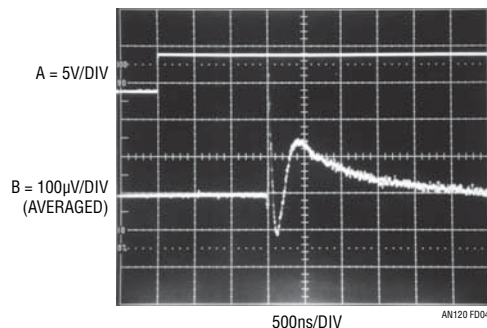
Best settling results when the compensation capacitor is selected to functionally compensate for all the above parasitics. Figure D2, taken with an LTC2757/LT1468 DAC-Amplifier combination, shows results for an optimally selected (in this case, 20pF) feedback capacitor. Trace A is the DAC input pulse and trace B the amplifier's settle signal. The amplifier is seen to come cleanly out of slew and settle very quickly.

In Figure D3, the feedback capacitor is too large (27pF). Settling is smooth, although overdamped, and a 300ns penalty results. Figure D4's feedback capacitor is too small (15pF), causing a somewhat underdamped response with resultant excessive ring time excursions. Settling time goes out to 2.8µs. Note that the above compensation values for 18-bit settling are not necessarily indicative of results for 16 or 20 bits. Optimal compensation values must be established for any given desired resolution. Typical values range from 15pF to 39pF.

**Note 1.** Spice aficionados take notice.



**Figure D3. Overdamped Response Ensures Freedom from Ringing, Even with Production Component Variations. Penalty is Increased Settling Time.  $t_{SETTLE} = 2.1\mu s$  to 0.0004% (18 Bits)**



**Figure D4. Underdamped Response Results from Undersized Capacitor. Component Tolerance Budgeting Will Prevent This Behavior.  $t_{SETTLE} = 2.8\mu s$  to 0.0004% (18 Bits)**

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When feedback capacitors are individually trimmed for optimal response, DAC, amplifier and compensation capacitor tolerances are irrelevant. If individual trimming is not used, these tolerances must be considered to determine the feedback capacitor's production value. Ring time is affected by DAC capacitance and resistance, as well as the feedback capacitor's value. The relationship is nonlinear, although some guidelines are possible. The DAC impedance terms can vary by  $\pm 50\%$  and the feedback capacitor is typically a  $\pm 5\%$  component. Additionally, amplifier slew rate has a significant tolerance, which is stated on the data sheet. To obtain a production feedback capacitor value,

determine the optimum value by individual trimming with the production board layout (board layout parasitic capacitance counts too!). Then, factor in the worst-case percentage values for DAC impedance terms, slew rate and feedback capacitor tolerance. Combine this information with the trimmed capacitors measured value to obtain the production value. This budgeting is perhaps unduly pessimistic (RMS error summing may be a defensible compromise), but will keep you out of trouble<sup>2</sup>.

**Note 2:** The potential problems with RMS error summing become clear when sitting in an airliner that is landing in a snowstorm.

## APPENDIX E

### A VERY SPECIAL CASE—MEASURING SETTLING TIME OF CHOPPER-STABILIZED AMPLIFIERS

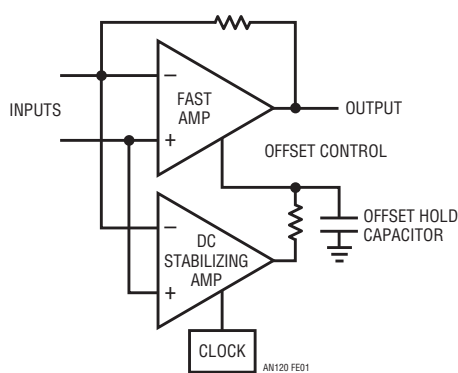
The text box section (page 2) lists the LTC1150 chopper-stabilized amplifier. The term “special case” appears in the “comments” column. A special case it is! To see why requires some understanding of how these amplifiers work. Figure E1 is a simplified block diagram of the LTC1150 CMOS chopper-stabilized amplifier. There are actually two amplifiers. The “fast amp” processes input signals directly to the output. This amplifier is relatively quick, but has poor DC offset characteristics. A second, clocked, amplifier is employed to periodically sample the offset of the fast channel and maintain its output “hold” capacitor at whatever value is required to correct the fast amplifier's offset errors. The DC stabilizing amplifier is clocked to permit it to operate (internally) as an AC amplifier, eliminating its DC terms as an error source<sup>1</sup>. The clock

chops the stabilizing amplifier at about 500Hz, providing updates to the hold capacitor-offset control every 2ms<sup>2</sup>.

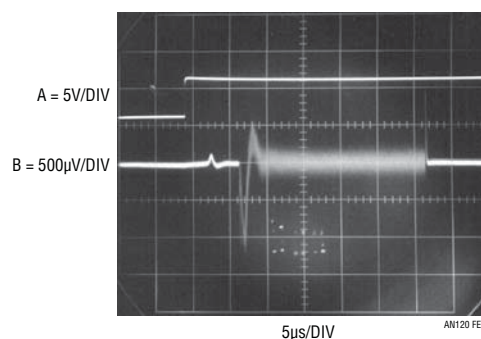
The settling time of this composite amplifier is a function of the fast and stabilizing paths response. Figure E2 shows amplifier short-term settling. Trace A is the DAC input pulse and trace B the settle signal. Damping is reasonable and the 10 $\mu$ s settling time and profile appear typical. Figure E3 brings an unpleasant surprise. If the DAC slewing interval happens to coincide with the amplifier's sampling cycle, serious error is induced. In Figure E3, trace A is the amplifier output and trace B the settle signal. Note the slow horizontal scale. The amplifier initially settles quickly (settling is visible in the 2nd

**Note 1.** This AC processing of DC information is the basis of all chopper and chopper-stabilized amplifiers. In this case, if we could build an inherently stable CMOS amplifier for the stabilizing stage, no chopper stabilization would be necessary.

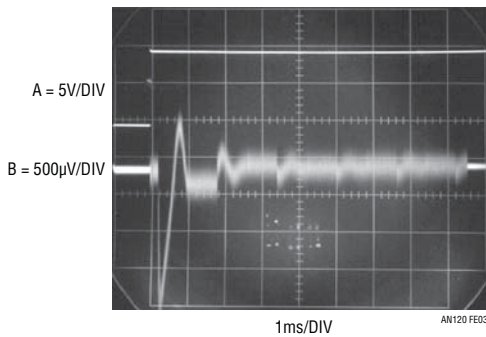
**Note 2.** Those finding this description intolerably brief are commended to reference 31.



**Figure E1. Highly Simplified Block Diagram of Monolithic Chopper-Stabilized Amplifier. Clocked Stabilizing Amplifier and Hold Capacitor Cause Settling Time Lag**



**Figure E2. Short-Term Settling Profile of Chopper-Stabilized Amplifier Seems Typical. Settling Appears to Occur in 10 $\mu$ s**



**Figure E3. Surprise! Actual Settling Requires 700× More Time Than Figure E2 Indicates. Slow Sweep Reveals Monstrous Tailing Error (Note Horizontal Scale Change) Due to Amplifier’s Clocked Operation. Stabilizing Loop’s Iterative Corrections Progressively Reduce Error Before Finally Disappearing Into Noise**

vertical division region) but generates a huge error 200µs later when its internal clock applies an offset correction. Successive clock cycles progressively chop the error into the noise but 7 *milliseconds* are required for complete recovery. The error occurs because the amplifier sampled offset when its input was driven well outside its bandpass. This caused the stabilizing amplifier to acquire erroneous offset information. When this “correction” was applied, the result was a huge output error.

This is admittedly a worst case. It can only happen if the DAC slewing interval coincides with the amplifier’s internal clock cycle, but it can happen<sup>3,4</sup>.

**Note 3.** Readers are invited to speculate on the instrumentation requirements for obtaining Figure E3’s photo.

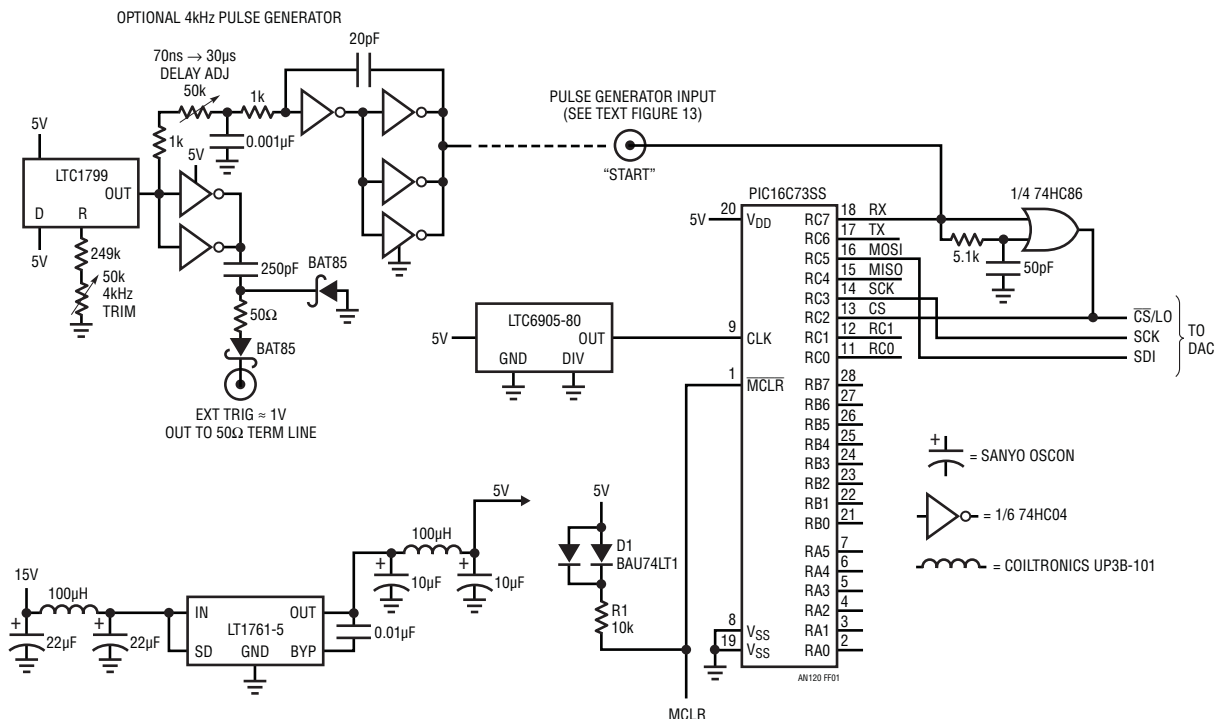
**Note 4.** The spirit of Appendix D’s footnote 2 is similarly applicable in this instance.

## APPENDIX F

### SETTLING TIME MEASUREMENT OF SERIALLY LOADED DACS

Measuring serially loaded DACs settling time requires additional circuitry. This circuitry must provide a “start” pulse to the settling time measurement circuit after serially

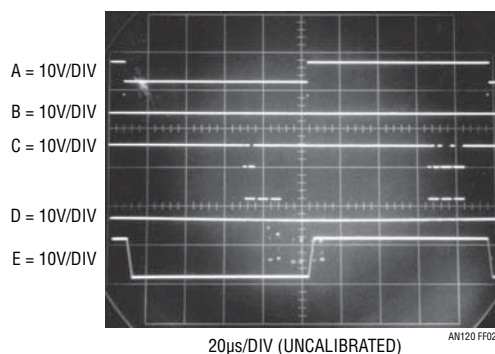
loading a full-scale step into the DAC. Figure F1’s processor based circuitry, designed and constructed by LTC’s Mark Thoren, does this. The “start” pulse (trace A, Figure F2) initiates the measurement. Traces B, C and D are  $\overline{CS}/LD$ , SCK, and SDI, respectively. Trace E, the resultant DAC output, is measured for settling time in (what should be by now) familiar fashion. Figure F3 is a complete processor software code listing.



**Figure F1. The Serial Interface. Processor Responds to Input Pulse, Directs DAC to Perform 10V Steps**

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**Figure F2. Serial Interface Operation Includes Input Start Pulse (Trace A),  $\overline{CS}/LD$  (Trace B), SCK (C), SDI (D) and Resultant DAC Output (E), Digital Data Lines are Static During Measurement Interval, Precluding Crosstalk Induced Corruption**

```
/*  
Serial DAC step program. Makes controlling the serial DAC as easy as the old way of  
tying all the digital lines of a parallel DAC to a pulse generator.
```

```
the serial DAC CS/LD signal is the output of an XOR gate edge detector that gives  
a 1µs pulse on either the rising edge or falling edge of CONTROL signal.  
Program enters main loop when CONTROL is high. When CONTROL goes low, the code  
for +5V is sent. When CONTROL goes high, the code for -5V is sent. Thus the  
timing of the load pulse accurately follows the input signal by about 20ns.
```

```
A delay of 60µs is inserted after the load pulse so that you can look at  
settling details without having to worry about digital feedthrough.
```

```
*/
```

```
#include <16F73.h>  
#include "pcm73a.h"  
#use delay(clock=20000000) // 20 meg clock  
#fuses HS, NOWDT, PUT//, MCLR
```

```
// Defines for DAC addresses  
#define DACA 0  
#define DACB 2  
#define DACC 4  
#define DACD 6
```

```
#define PM10 0x03  
#define PM5 0x02
```

```
// Control input  
#define CONTROL PIN_C7
```

```
void init(void);
```

```
void main()
```

```
{  
    init(); // set up hardware
```

```
    // This just allows the program to sync up to a pulse generator that  
    // may not have a clean output on power-up. You need to see at least one rising  
    // and one falling edge before continuing.  
    while(!input(CONTROL)){} delay_us(2); // wait for rising edge  
    while(input(CONTROL)){} delay_us(2); // wait for falling edge  
    delay_ms(100);  
    while(!input(CONTROL)){} delay_us(2); // wait for rising edge  
    while(input(CONTROL)){} delay_us(2); // wait for falling edge
```

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```

// Okay, now we're all synchronized.
// Since program does not have direct access to the CS/LD line, you
// have to rely on the externally applied pulse.
while(!input(CONTROL)){ delay_us(2); // wait for rising edge
while(input(CONTROL)){ delay_us(2); // wait for falling edge
spi_write(0x6F); // Set all DACs to +/-10V range
spi_write(0x00);
spi_write(PM5);

while(!input(CONTROL)){ delay_us(2); // wait for rising edge
spi_write(0x70 | DACB); // Set DACB to -10 volts
spi_write(0x00);
spi_write(0x00);

while(input(CONTROL)){ delay_us(2); // wait for falling edge
spi_write(0x70 | DACC); // Set DACC to 0 volts
spi_write(0x80);
spi_write(0x00);

while(!input(CONTROL)){ delay_us(2); // wait for rising edge
spi_write(0x70 | DACD); // Set DACD to +10 volts
spi_write(0xFF);
spi_write(0xFF);

while(1)
{
while(input(CONTROL)){ delay_us(80); // wait for falling edge
spi_write(0x70 | DACA); // Set DACA to 0 volts
spi_write(0xFF);
spi_write(0xFF);

while(!input(CONTROL)){ delay_us(80); // wait for rising edge
spi_write(0x70 | DACA); // Set DACA to +10 volts
spi_write(0x00);
spi_write(0x00);
}
}

void init()
{
setup_adc_ports(NO_ANALOGS);
setup_adc(ADC_CLOCK_DIV_2);
setup_spi(SPI_MASTER|SPI_L_TO_H|SPI_CLK_DIV_4|SPI_SS_DISABLED);
CKP = 0; // Set up clock edges - clock idles low, data changes on
CKE = 1; // falling edges, valid on rising edges.
setup_counters(RTCC_INTERNAL,RTCC_DIV_2);
setup_timer_1(T1_DISABLED);
setup_timer_2(T2_DISABLED,0,1);
setup_ccp1(CCP_OFF);
setup_ccp2(CCP_OFF);
set_tris_a(0b00000000);
set_tris_b(0b00000000);
set_tris_c(0b10000100); // Make sure control signal is input
}

```

**Figure F3. Software Listing for PIC16C73SS Processor. Code Directs DAC to Step 10V at Each Input Pulse Transition**

## APPENDIX G

### BREADBOARDING, LAYOUT AND CONNECTION TECHNIQUES

The measurement results presented in this publication required painstaking care in breadboarding, layout and connection techniques. Wideband, 10 $\mu$ V resolution measurement does not tolerate cavalier laboratory attitude. The oscilloscope photographs presented, devoid of ringing, hops, spikes and similar aberrations, are the result of an exhaustive (and frustrating) breadboarding exercise<sup>1</sup>. The breadboard was rebuilt numerous times and required weeks of layout and shielding experimentation before obtaining a noise/uncertainty floor worthy of 20-bit measurement. In particular, extreme measures were required to minimize sample command signal feedthrough. Layout techniques include minimization and restriction of radiative paths, ground plane current management and mounting the LT1228 “switch” upside down, allowing its V-referred substrate to approximate a monolithic shield for the IC’s internal circuitry.

#### Ohm’s Law

It is worth considering Ohm’s Law is a key to successful layout<sup>2</sup>. Consider that 1mA running through 0.1 $\Omega$  generates 100 $\mu$ V—almost 3LSB at 18 bits! Now, run that milliampere at 5ns to 10ns rise times (approximately 75MHz) and the need for layout care becomes clear. A paramount concern is disposal of circuit ground return current and disposition of current in the ground plane. The impedance of the ground plane between any two points is not zero, particularly as frequency scales up. This is why the entry point and flow of “dirty” ground returns must be carefully placed within the grounding system. In the sampler-based breadboard, the approach was separate “dirty” and “signal” ground planes, tied together at the supply ground origin.

A good example of the importance of grounding management involves delivering the input pulse to the breadboard. The pulse generator’s 50 $\Omega$  termination must be an in-line coaxial type, and it cannot be directly tied to the signal ground plane. The high speed, high density (5V pulses through the 50 $\Omega$  termination generate 100mA current spikes) current flow must return directly to the pulse

generator. The coaxial terminator’s construction ensures this substantial current does this, instead of being dumped into the signal ground plane (100mA termination current flowing through 1m $\Omega$  of ground plane produces approximately 3LSB of error!). The 50 $\Omega$  termination is physically distanced from the breadboard via a coaxial extension tube (visible in Figure H7)<sup>3</sup>. This further ensures that pulse generator return current circulates in a tight local loop at the terminator and does not mix into the signal plane.

It is worth mentioning that every ground return in the entire circuit must be evaluated with these concerns in mind. A paranoiac mindset is quite useful.

#### Shielding

The most obvious way to handle radiation-induced errors is shielding. Various following figures show shielding. Determining where shields are required should come *after* considering what layout will minimize their necessity. Often, grounding requirements conflict with minimizing radiation effects, precluding maintaining distance<sup>4</sup> between sensitive points. Shielding<sup>5</sup> is usually an effective compromise in such situations.

A similar approach to ground path integrity should be pursued with radiation management. Consider what points are likely to radiate, and try to lay them out at a distance from sensitive nodes. When in doubt about odd effects, experiment with shield placement and note results, iterating towards favorable performance<sup>6</sup>. *Above all, never rely on filtering or measurement bandwidth limiting to “get rid of” undesired signals whose origin is not fully understood.* This is not only intellectually dishonest, but may produce wholly invalid measurement “results”, even if they look pretty on the oscilloscope.

---

**Note 1.** “War” is perhaps a more accurate descriptive.

**Note 2.** I do not wax pedantic here. My abuse of this postulate runs deep.

**Note 3.** Strictly considered, this technique introduces mis-termination originated transmission line reflections but no appreciable error results at the bandwidth of interest.

**Note 4.** Distance is the physicist’s approach to reducing radiation induced effects.

**Note 5.** Shielding is the engineer’s approach to reducing radiation induced effects.

**Note 6.** After it works, you can figure out why.

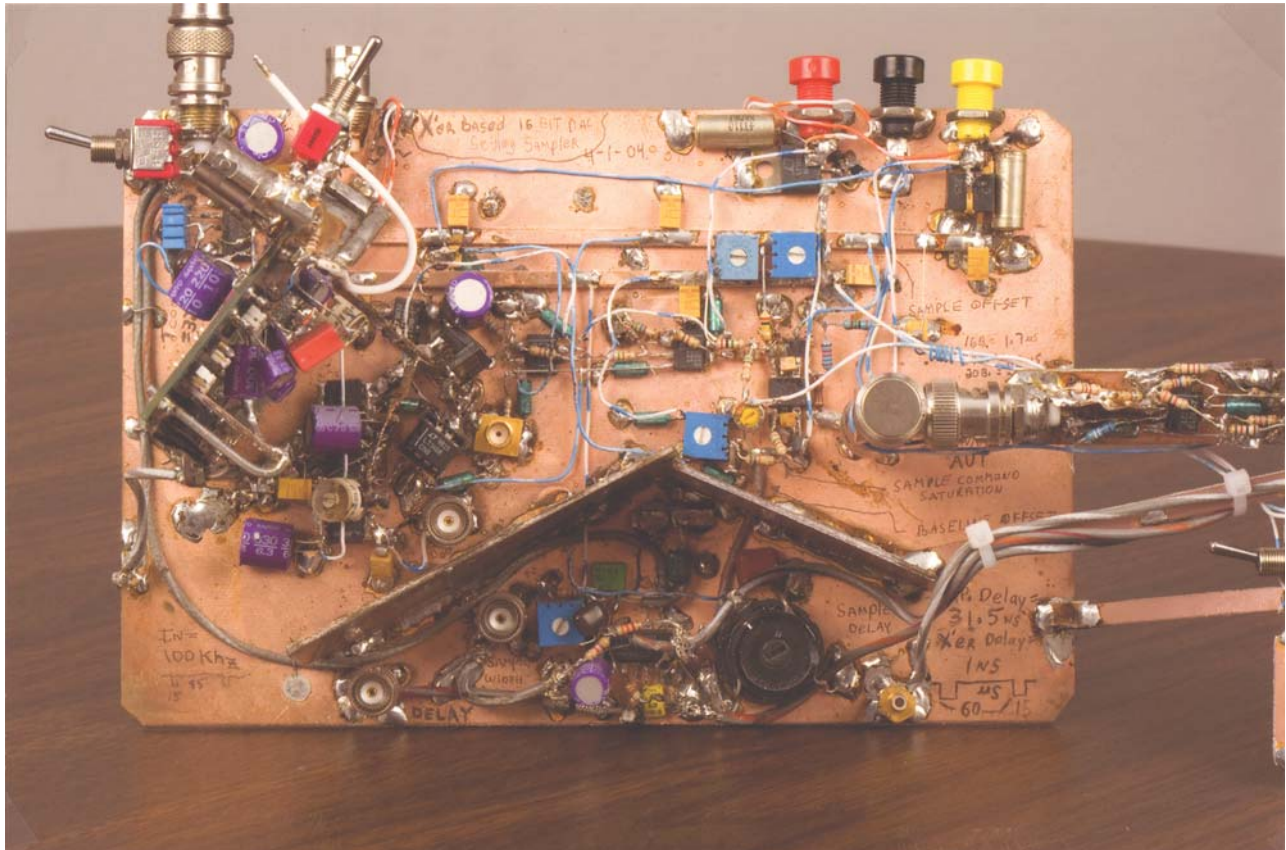


## Connections

All signal connections to the breadboard must be coaxial. Ground wires used with oscilloscope probes are forbidden. A one inch ground lead used with a 'scope probe can easily generate several LSBs of observed "noise". Use

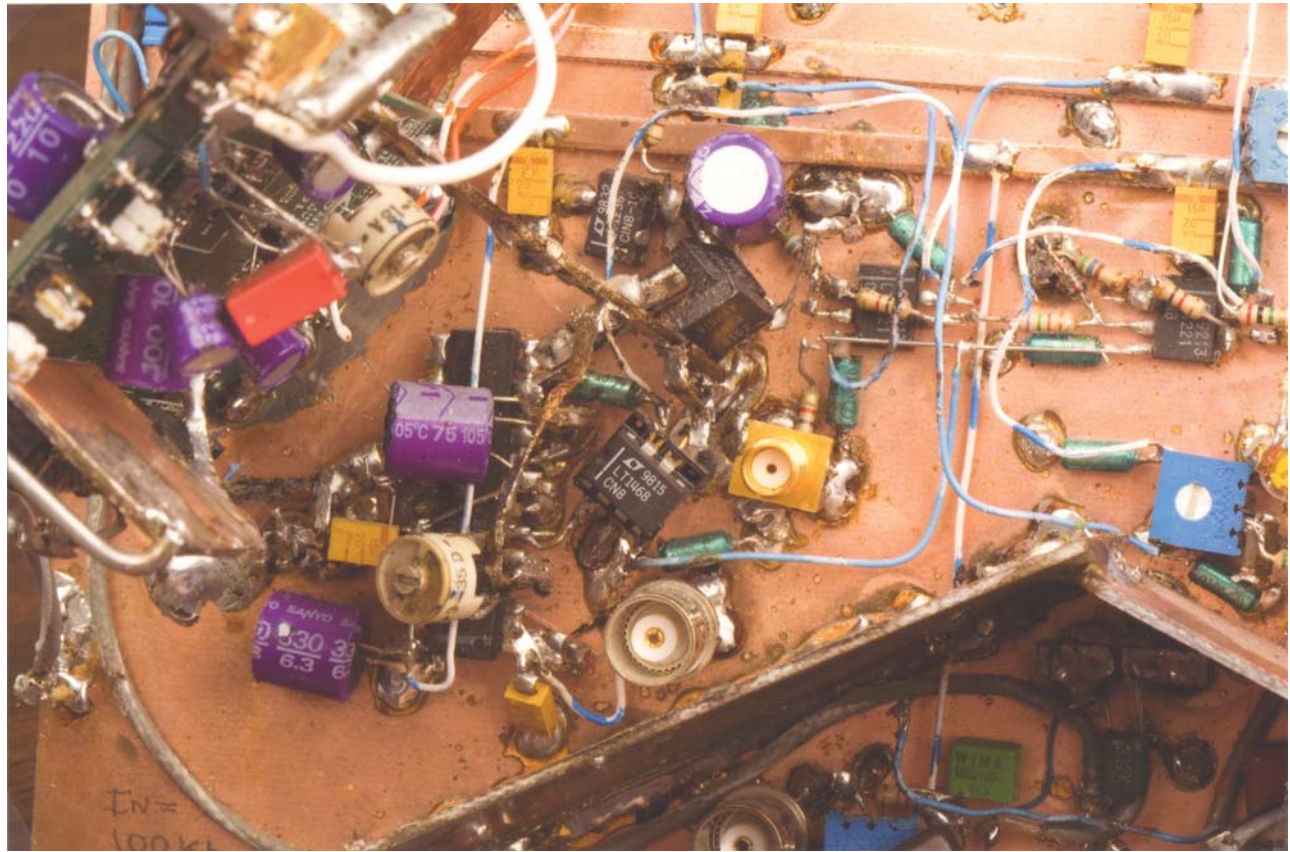
coaxially mounting probe tip adapters<sup>7</sup>. Figures G1 to G5 restate the above sermon in visual form while annotating the text's circuits.

**Note 7.** See reference 28 for additional nagging along these lines.



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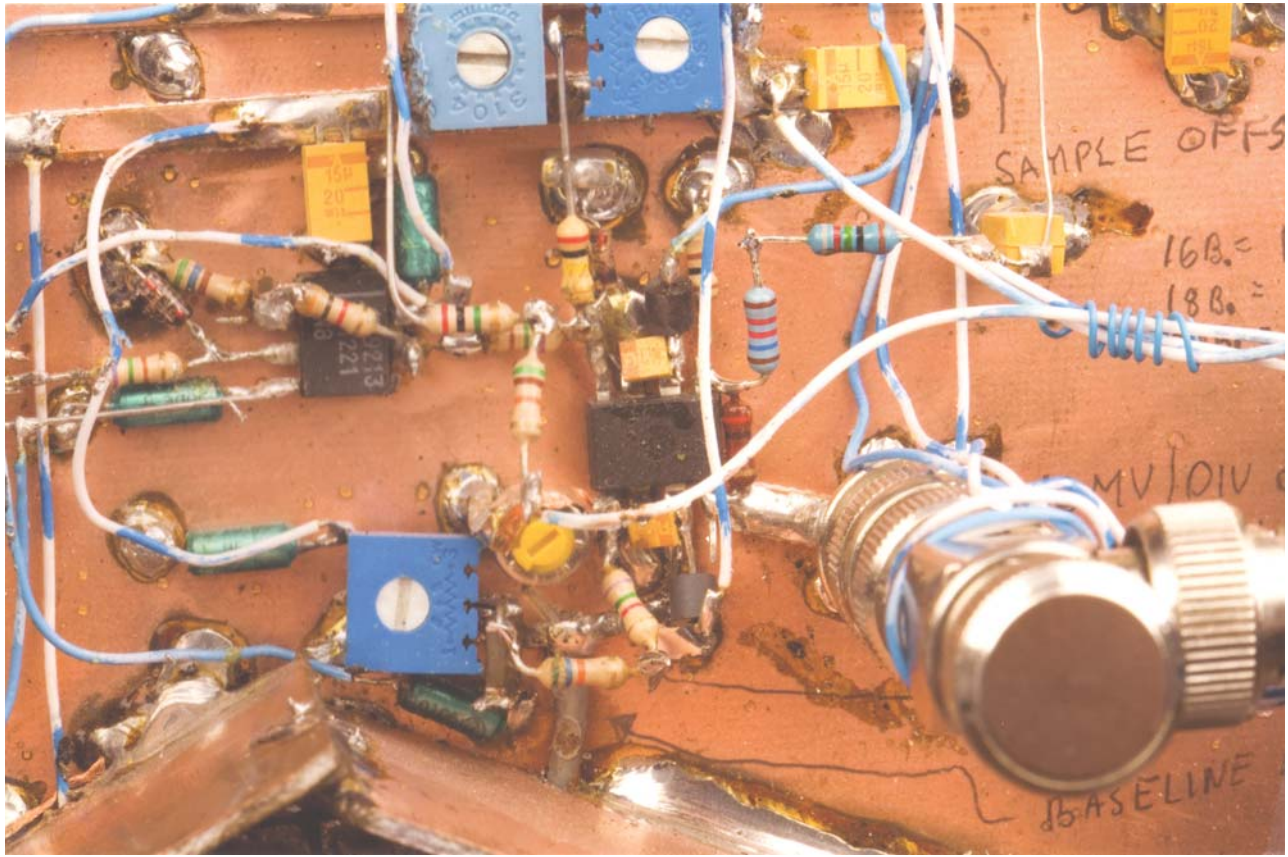
**Figure G1. Settling Time Breadboard Overview.** Pulse Generator Input Enters Top Left—50Ω Coaxial Terminator Mounted On Extension Tube (Not Visible—See Appendix H, Figure H7) Minimizes Pulse Generator Return Current Mixing Into Signal Ground Plane. DAC-amplifier and Support Circuitry are at Left. Sampler Circuitry Occupies Board Right. Sampler Digital Support Circuitry is Contained Within Large Shield (Board Lower). Nonsaturating Amplifier Occupies Board Center. Partially Visible X10 Post Amplifier (See Appendix I) is BNC Fixtured, Thin Board at Photo Right. Auto-Zero Circuit (see Appendix I), Mounted on Thin Strut (Lower Right), is Not Visible



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**Figure G2. DAC-Amplifier Detail. DAC and Output Amplifier are at Photo Center Left. Precision Summing Resistors (Box-Shaped, Just Below Large Round Capacitor Near Photo Center) are Oriented to Restrict Radiative Coupling While Minimizing Summing Node Capacitance. Variable Capacitor (Lower Left Center) Sets Amplifier (Photo Center) Compensation. Non-Saturating Amplifier Appears at Right. Shield (Bottom) Restricts Sampler Digital Section's Radiation. Vertically Mounted Board (Extreme Upper Left) is Optional Serial DAC Interface (See Appendix F). Coaxial Connectors (Center Lower) Facilitate High Purity Signal Extraction**

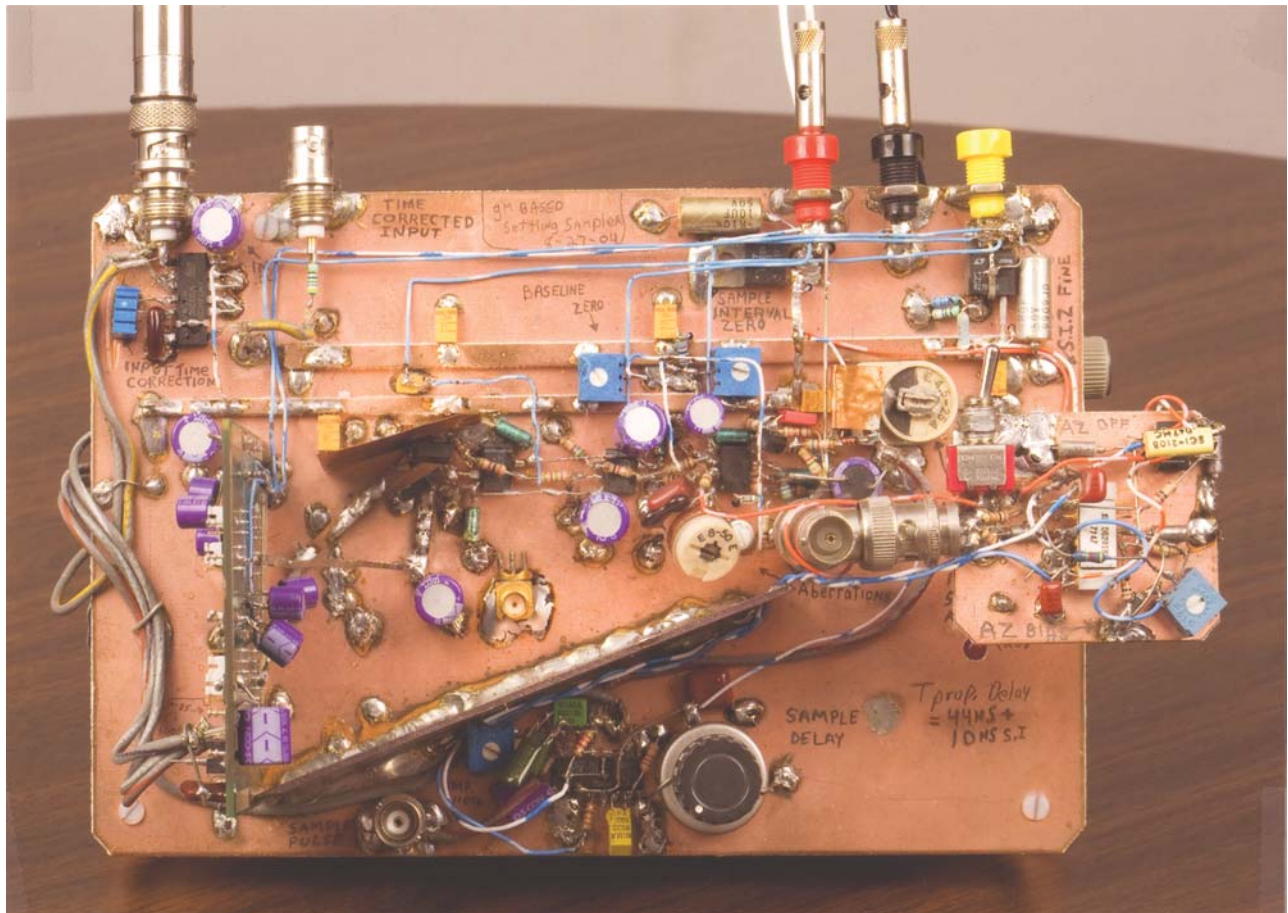
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AN120 FG03

**Figure G3. LT1228 Sampling “Switch” (Photo Center) Is Mounted Upside Down, Permitting  $V^-$  Referred Die Backside To Shield Residual Radiative Coupling, Reducing Sampling Switch Drive Feedthrough. Switch Signal Channel Is Fed From Non-Saturating Amplifier (Photo Left). Sample Command From Shielded Digital Section (Lower) Arrives Via Coaxial Cable Tunneling Through Shield (Lower Center)**

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**Figure G4. A Dedicated, Serially Interfaced Settling Time Breadboard. Serial Interface Digital Board is Obscured Beneath Visible Analog Board (See Appendix H, Figure H7). Note Insulating Nylon Screws (Right and Left Lower Corners and Upper Edge Between BNC Connectors) Used to Attach Digital and Analog Boards. Digital Board Ground is Single-Point Connected at Analog Board Ground Entry Point (Middle Banana Jack). Serial Signals Access Vertical DAC Board Via Coax (Photo Left). Shields Isolate Sample Switch Digital Section (Lower) and Summing Node (Left Center). Non-Saturating Amplifier and Sampling Switch are as in Figure G1. Optional Auto-Zero Board (See Appendix I) Mounts from BNC Fitting at Center Right. Upper Left Corner Components are Input Pulse Time Correction**

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**Figure G5. Serially Interfaced Settling Time Breadboard Signal Path Detail. DAC Board is at Left. Precision Resistors Feed Summing Node and Non-Saturating Amplifier (Photo Upper Center). Shield Protects Summing Node from Input Pulse Originated Radiation. LT1228 Sampling “Switch” (Far Right) is Mounted Upside Down, Permitting  $V^-$  Referred Die Backside to Shield Radiative Coupling, Reducing Switch Drive Feedthrough. Switch Drive Level Shift-Current Source (Extreme Right Upper) Receives Sample Command Via Coax, Partially Visible at Lower Right. Large Vertical Shield Confines Sample Command Digital Section Radiation (Lower). SMA Connector (Center) Enables Test and Calibration Signal Connection**

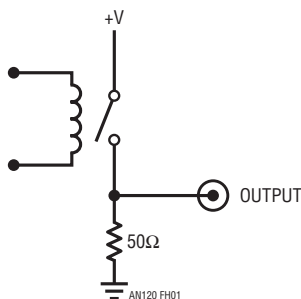
## APPENDIX H

### HOW DO YOU KNOW IT WORKS? Settling Time Circuit Performance Verification

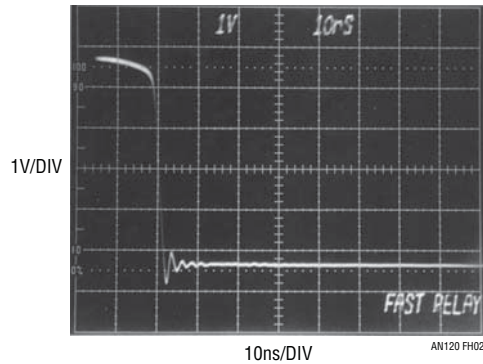
#### High Purity Pulse Generator

Any prudent investigation requires performance verification of the test method. Strictly considered, it may not be possible to furnish indisputable proof that the circuit in question is functioning at its design limits, particularly in a state-of-the-art measurement. However, a reasonable level of confidence is a realistic goal. Performance verification for the settling time test circuit requires a high purity pulse generator that transitions and settles to 1ppm as quickly as possible. This is a high order difficulty requirement and the author is unaware of any electronic means of achieving this capability. Fortunately, electro-mechanical technology offers a solution.

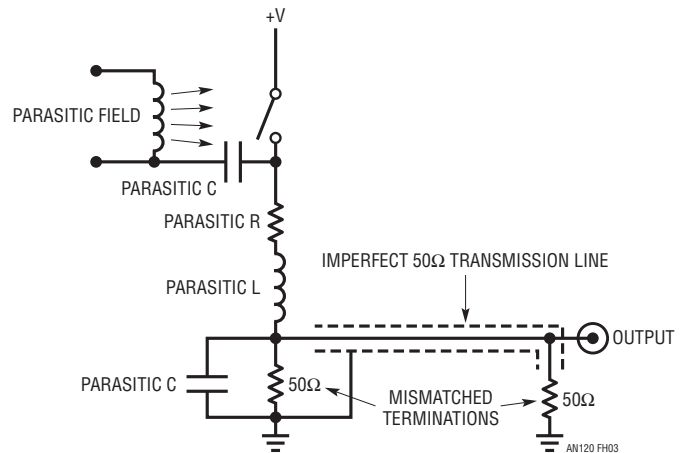
Figure H1 shows a conceptual mercury wetted reed relay pulse generator. Theoretically, when the contacts open, an infinitely fast falling edge appears across the 50Ω termination with zero settling time to 0V. Figure H2 reveals this to be not the case. This photograph, taken with a typical commercially available relay, shows <math><5\text{ns}</math> transition time with a 500MHz ring off over 10ns<sup>1</sup>. These imperfections are not surprising when Figure H3's parasitic terms are considered. Figure H1's deceptively simple schematic is revealed to have a number of unintentional terms which severely limit performance. These terms include, but are not limited to, parasitic resistance, inductance, and capacitance as well as undesirable field interaction within the relay. Additionally, the connection through the relay to the output terminal constitutes an ill-defined transmission line which promotes additional vagaries. The parasitic terms



**Figure H1. Conceptual Mercury Wetted Reed Relay Pulse Generator Produces Infinitely Fast Falling Edge Across 50Ω Termination with Zero Settling Time to 0V**



**Figure H2. Mercury Wetted Reed Relay Opens in <math><5\text{ns}</math>, Settles Quickly to Zero. 500MHz Ring Off Derives from Source-Termination Impedance Mismatch**



**Figure H3. Parasitic Terms Limit Achievable Performance. Transmission Line, Required to Route Output Pulse, Adds Termination Mismatch Errors and Line Related Infidelities**

interact in a haphazard and unpredictable way, resulting in alien terms at the pulse output. What is really needed is a relay specifically designed and constructed for inclusion into a wideband 50Ω system.

#### A True 50Ω, Wideband Mercury Wetted Reed Relay

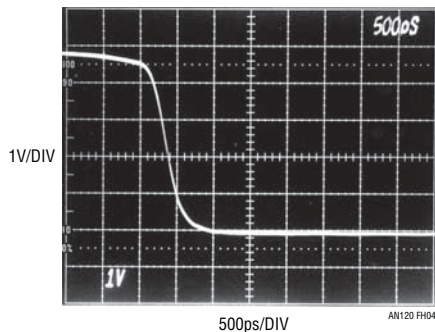
In the 1960s, Tektronix manufactured the type 109 mercury wetted reed relay, intended for use as a pulse generator. In its preferred configuration, energized charge lines are

**Note 1.** Tangential to this discussion, but nonetheless interesting, is the corner rounding at the pulse top just before its rapid fall. This may be due to "teasing" of the mercury, causing its resistance to increase just before it fully opens. John Willison of Stanford Research Systems suggests the mechanism may be charge displacement in the capacitor formed as the relay contacts just open. Scott Hamilton, Manchester University (UK), has raised the possibility of quantum tunneling across the brief, small contact gap *a la* scanning tunneling microscope operation. Comments from the readership are welcome.

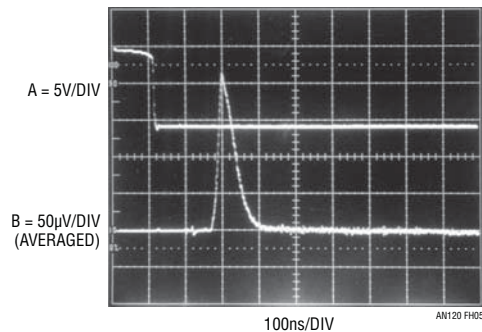
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switched by the relay into a  $50\Omega$  termination, resulting in a 250ps risetime pulse. Here, charge lines are not employed; rather, the device is used as a simple switch. Advantage is taken of the exquisite care at manufacture to make the relay transparent in a  $50\Omega$  system. Figure H4's large scale transition reveals 800ps fall time in a 1GHz bandwidth. Actual fall time is probably somewhat faster as the monitoring oscilloscope has a 350ps risetime<sup>2</sup>. The transition is singularly clean and devoid of discontinuities with the exception of the previously noted pre-fall corner rounding. Figure H5's remarkable photograph uses sampling switch techniques similar to those described in the

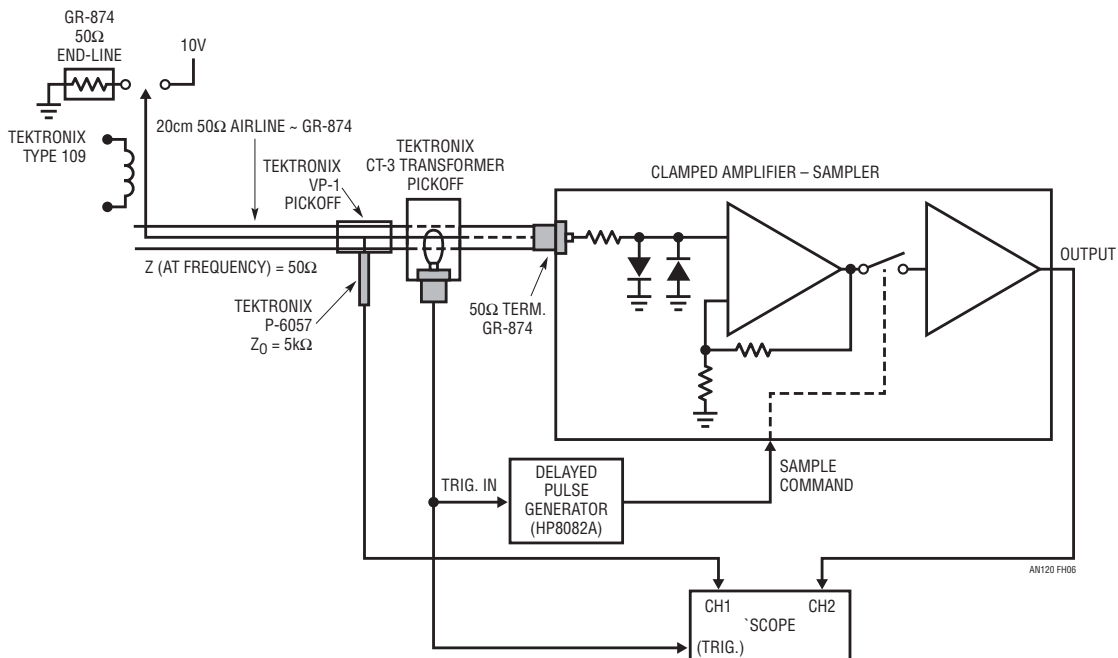
text to measure type 109 fall-time purity to microvolts<sup>3</sup>. The 109 output (trace B) moves the final  $220\mu\text{V}$ , settling inside  $10\mu\text{V}$  approximately 265ns after the relay contacts open (trace A). Actual 109 settling time may be faster as the measurement is likely sampling circuit limited<sup>4</sup>. Figure H6 is a simplified version of the test circuit which



**Figure H4. High Grade Mercury Wetted Reed Relay (Tektronix Type 109) Falls in 800ps Viewed in 1GHz Real Time Bandwidth. Strict Attention To Parasitic Minimization in Relay Structure and Transmission Path Produces High Fidelity Transition Without Alien Components**



**Figure H5. Sampling Switch Techniques Permit Measuring Type 109 Falling Edge Residue to Microvolts in 20MHz Bandpass. Trace A is 109 Falling Edge, Trace B the Last  $220\mu\text{V}$  of Movement Before Settling to 1ppm in Indicated 265ns**



**Figure H6. Simplified Test Circuit. Attention to Hg Wetted Reed Relay and Transmission Line Allow 1ppm Residue  $\approx 265\text{ns}$  After Contacts Open**

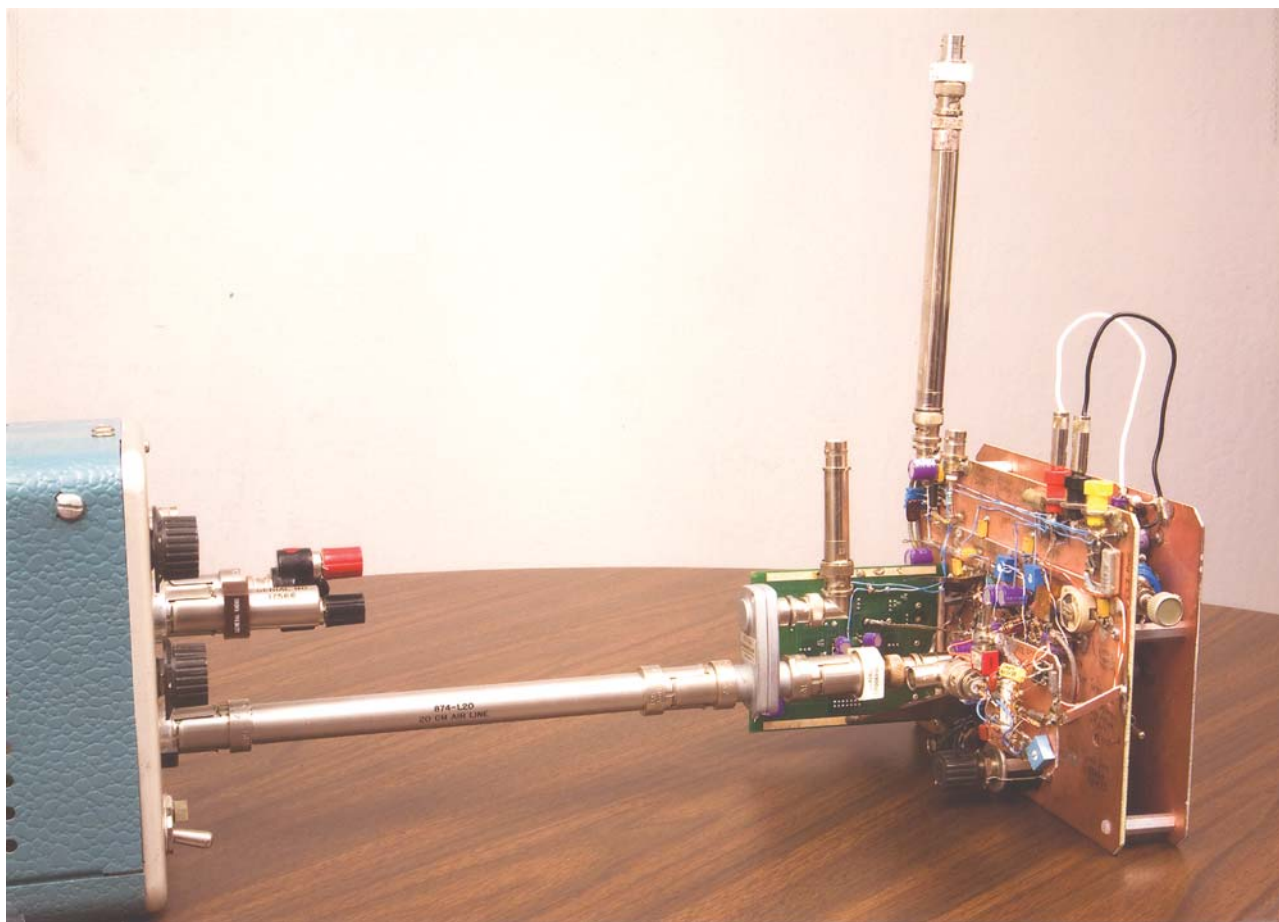
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produced these results. The type 109 drives 20 centimeters of 50Ω GR-874 airline into a high quality GR-874 50Ω termination at the clamped amplifier-sampling switch. The delayed pulse generator and oscilloscope are set up similarly to the main text description. The Tektronix pickoff components noted allow signal extraction from the airline without degrading transmission line integrity. It is interesting to note that airline is a non-negotiable requirement. The highest quality Teflon 50Ω cable produced impure response, albeit minor.

Figure H7's photograph shows the high purity step generator connected to Figure G4's settling time breadboard. The

type 109 (photo left) delivers its pulse via a General Radio 20 centimeter airline. The Tektronix CT-3 coaxial transformer (right end of airline) supplies the trigger pick-off via the vertically mounted 50Ω BNC connector. A GR-874 50Ω load (right side of CT-3) terminates the airline and supplies the pulse to the breadboard. Pulse amplitude is set by a DC voltage applied at the 109 via banana inputs. The unused 109 contact is terminated with an endline GR-874 fitting. All connections must be polished and mechanically secured to ensure a high fidelity 50Ω environment. Any component substitution or mechanical connection imperfection will degrade Figure H5's results.



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**Figure H7. High Purity Step Generator Connected to Figure G4's Settling Time Breadboard. Tektronix Type 109 Mercury Wetted Relay Based Pulse Generator (Photo Left) Delivers Pulse Via General Radio 20 Centimeter Airline. Tektronix CT-3 Coaxial Transformer (Right End of Air Line) Supplies Trigger Pick-Off Via 50Ω BNC Vertical Connector. GR-874 Coaxial 50Ω Load (Right Side of CT-3) Terminates Line And Supplies Pulse to Breadboard. Pulse Amplitude is Set by DC Voltage Applied to Tektronix 109 Via Banana Input Adapter. Unused Type 109 Contact is Terminated with End-Line GR-874 Fitting. All Connections Must Be Polished and Mechanically Secure to Ensure High Fidelity 50Ω Environment. Long Coaxial Extension Tube On Breadboard Isolates Input Pulse Termination (Tube Top) Current from Board Ground Plane (see Text and Figure G1 For Commentary). Lower Board Contains Digital Serial Interface (See Appendix F and Figure G4)**

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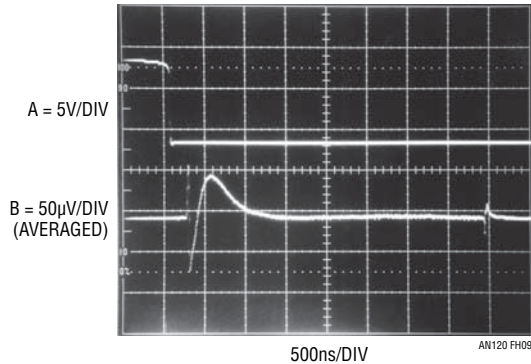


## “Pretty Good” Mercury Wetted Reed Relay Pulse Generator

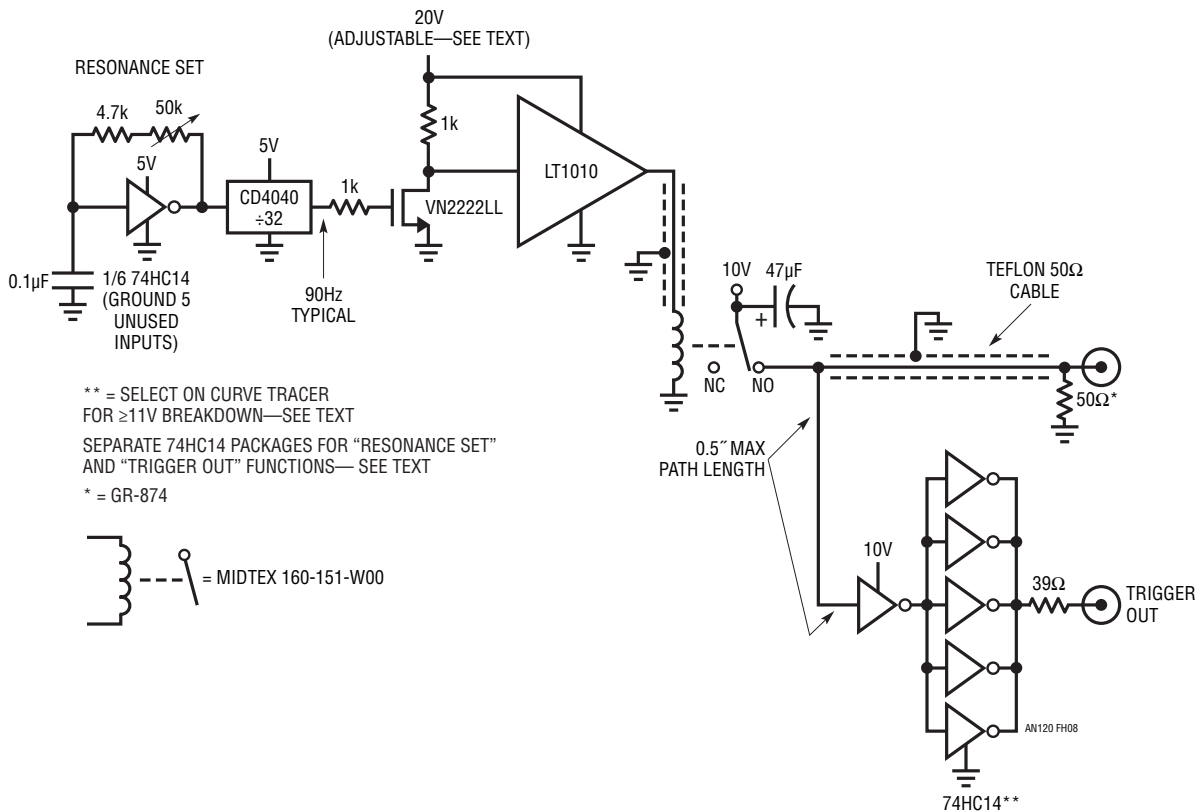
It may be unrealistic for readers to duplicate the Tektronix 109 based results. The specified exotic apparatus is difficult and expensive to obtain and the set up requires arduous labor and almost fanatical attention to detail. In this spirit, Figure H8’s “pretty good” mercury wetted reed relay pulse generator is offered. Its performance, while falling well short of Figure H6, still furnishes a 10V step which settles to 1ppm in 950ns<sup>5</sup>. A simple clock (“resonance set”) furnishes low frequency drive to the relay via the transistor level shift and the LT1010 power buffer. Trigger pick-off is provided by the paralleled CMOS inverters. Separate packages for the “resonance set” and trigger functions must be used to avoid output pulse contamination. Figure H9 shows results for the “pretty good” step generator. Its slower settling and alien residue components compared to the Tektronix 109 approach are apparent. The event at the 10th vertical division is sample gate turn-off feedthrough related.

Circuit calibration involves adjusting “resonance set” until the relay emits a reasonably pure audible tone. Next, set the 20V supply to a value which promotes the cleanest settling characteristics.

**Note 5.** Footnote 4’s 70ns timing allowance applies here. Figure H8 likely settles in  $\approx 880$ ns.



**Figure H9.** “Pretty Good” Step Generator Output (Trace A) Settles (Trace B) to 1ppm (10µV) in 950ns. Ill-Defined Relay Impedance Results in Approximately 3.6× Slower Settling and Alien Residue Components Compared to Figure H5’s Tektronix 109 Based Results. Event at 10th Vertical Division is Sample Gate Turn-Off Feedthrough



**Figure H8.** “Pretty Good” Substitute for Figure H6’s Tektronix 109, Constructed from Commercially Available Components, Settles 10V Step to 1ppm in 950ns

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## APPENDIX I

### Auxiliary Circuits

Several auxiliary circuits have been found useful in the DAC settling time work described in the text. Figure I1 is a simple, wideband, X10 preamplifier for oscilloscopes lacking the required sensitivity for 1ppm (10 $\mu$ V) settling time resolution. This preamplifier should be placed directly (no cable) at the oscilloscope input and connected via 50 $\Omega$  terminated BNC cable to the settling time fixture output.

Figure I2, an auto-zero circuit, locks the sample interval zero value to the non-sampled region baseline. It eliminates the need for periodic readjustment of the "Sample Interval Zero" trim when working at the highest levels of resolution over a protracted time. Synchronously switched A1 compares sample interval and non-sampled region zero values and applies an appropriate offset, closing a correction loop around the LT1228. M1's extended pulse precludes settling activity from influencing the sample interval zero value. The "Auto-Zero Bias" trim corrects

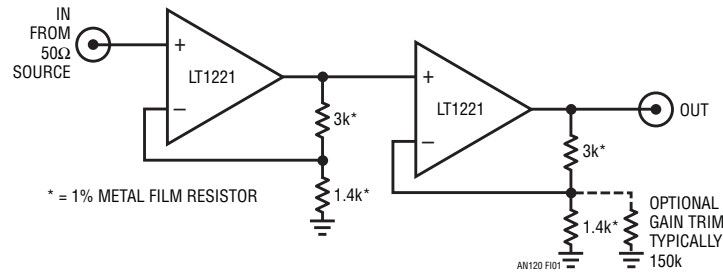


Figure I1. Simple X10 Pre-Amplifier for Oscilloscopes Lacking Required Sensitivity for 1ppm (10 $\mu$ V) Settling Time Resolution

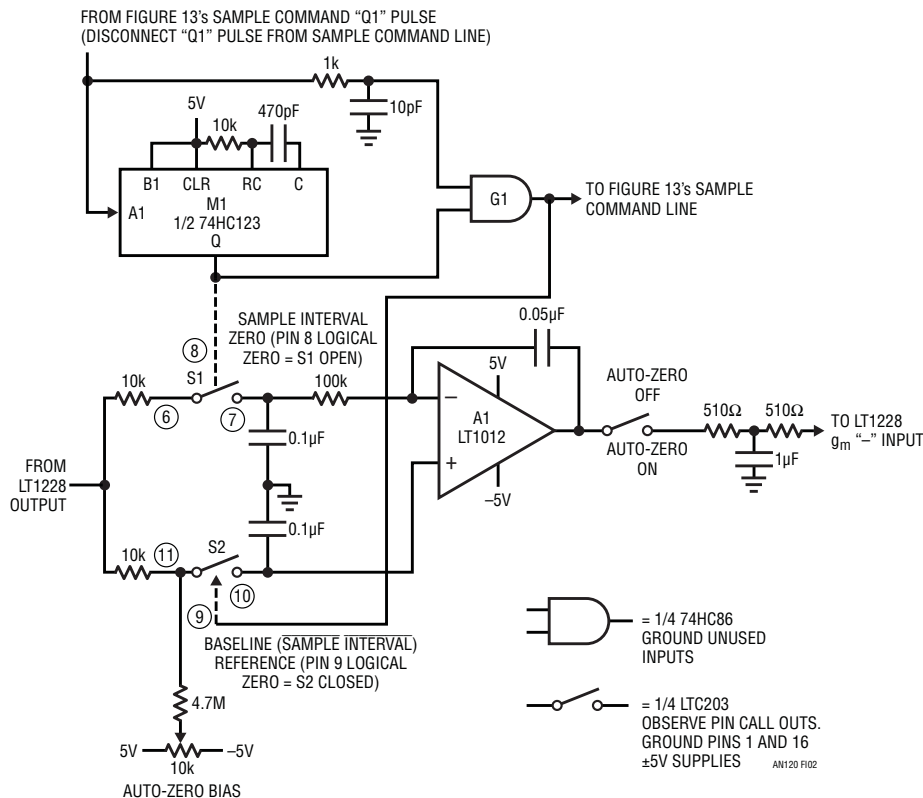
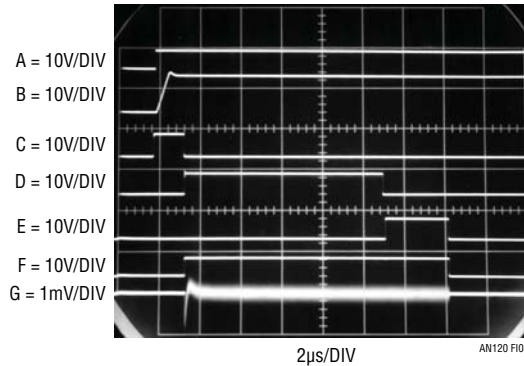


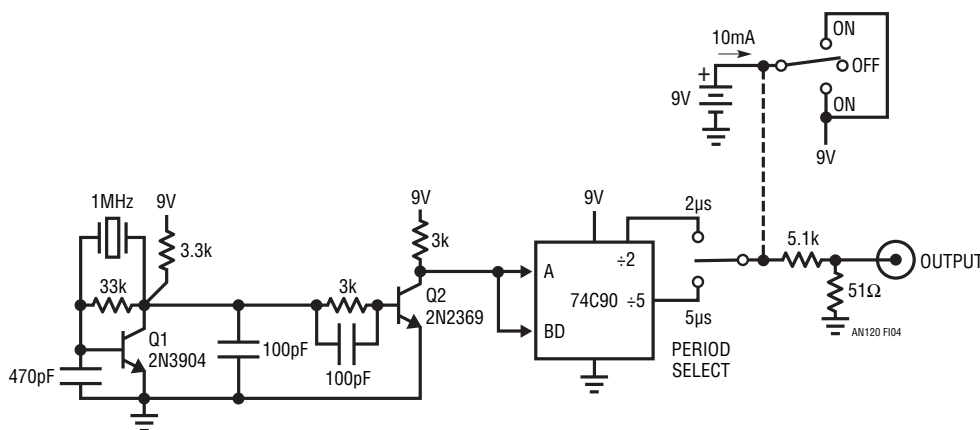
Figure I2. Auto-Zero Locks Sample Interval Zero Value to Non-Sampled Region Baseline. Synchronously Switched A1 Compares Sample Interval and Non-Sampled Region Values and Applies Appropriate Offset, Closing Correction Loop Around LT1228. M1 Precludes Settling Activity from Influencing Sample Interval Zero Value

for slight errors and should not require readjustment once set to equalize the sample interval zero value and the non-sample region baseline. The commented schematic provides information for the auto-zero's interconnection to the settling time circuit. Figure 13 shows auto-zero related waveforms. They include the time corrected input pulse (trace A), DAC output (B), sample delay (C), M1 input (D), M1's sample interval zero pulse (E), G1's sample command (F), and settle signal output (G). M1's delayed output maintains the sample interval zero value independent of the settling signature.

Figure 14 is a simple time calibrator used to verify oscilloscope time base accuracy. Q1 and Q2 form a 1MHz quartz oscillator. The 74C90 provides switch selectable output periods of 2µs and 5µs and the attenuator supplies a 50Ω output impedance. The period values have been selected for calibration points appropriate for expected DAC settling times. Other periods are available by varying oscillator frequency, division ratio or both. 9V battery drain is about 10mA.



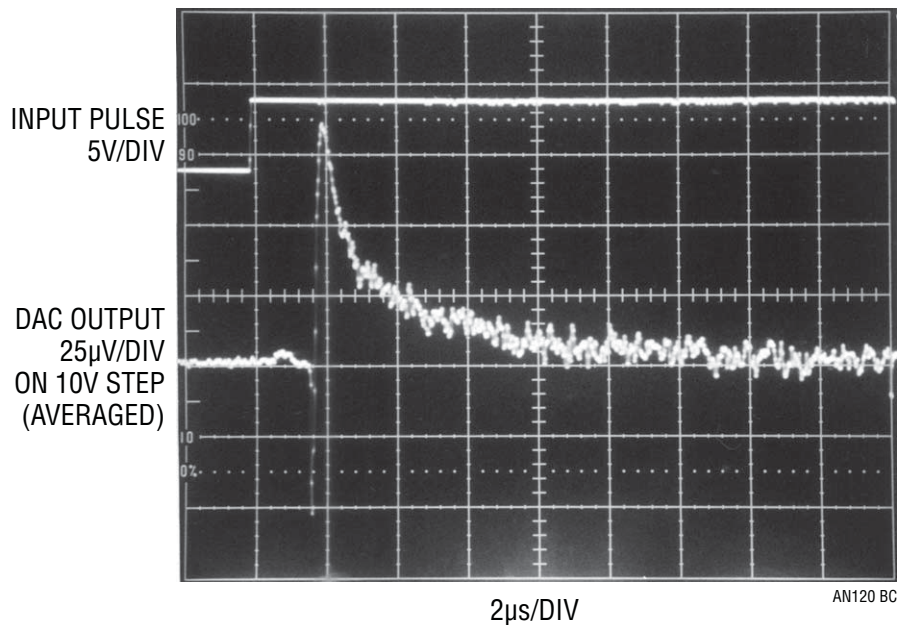
**Figure 13. Auto-Zero Related Waveforms Include Time Corrected Input Pulse (Trace A), DAC Output (B), Sample Delay (C), M1's Input (D), M1's Sample Interval Zero Pulse (E), G1's Sample Command (F) and Settle Signal Output (G). M1's Delayed Output Maintains Sample Interval Zero Value Independent of Settling Signature**



**Figure 14. Battery-Powered Oscilloscope Time Base Verifier Has 2µs and 5µs Period Outputs. Quartz Oscillator Q1 Is Buffered by Q2; Digital Divider Supplies Outputs Via Attenuator**

# Application Note 120

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“A part-per-million is a part-per-million. It’s magic. It’s the brass ring. It’s the holy grail of every measurement artist. It will mesmerize you, it will goad you, it will drive you crazy and, if you’re lucky, it will reward you. A part-per-million is a part-per-million.”

Jerrold R. Zacharias

M.I.T. physicist, mentoring a young,  
very naïve investigator.

— 1971