

Evaluating TLP Transients and HBM Waveforms

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Abstract –Transmission Line Pulsing (TLP) is capable of determining a true impedance function in the s-domain, including time-dependent transients, as $Z=Z_1(1+as+bs^2+\dots)$, with nonzero s-coefficients. We show how to determine the Z_1 and a terms and how to find more terms if desired, for long and short TLP pulses, drawing on network theory and approximation methods. We apply these to examples of inductive delay in devices. The same kind of methods are used to remove the effects of measurement tools themselves, including finite rise time of the TLP pulse and voltage droop of a current transformer. Then the true RC time constant of an HBM waveform is extracted as a network coefficient, through methods developed in this work.

I. Introduction

For about the last 25 years, semiconductor TLP characterization of ESD behavior [1] has centered on steady-state I-V, equivalent to finding, point-by-point, an equivalent DUT impedance. While works like [1] touched on finding capacitance or inductance from i-t and v-t plots, the focus was on steady-state response to the TLP step. But structured methods are available for characterizing DUT transients, first by finding first-order frequency dependence and setting the stage for higher-order characterization.

This work presents systematic TLP methods for finding DUT impedance $Z_1(1+as)$, where Z_1 is the usual dc impedance, s the complex frequency $\sigma+j\omega$, and aZ_1 is the s-coefficient, describing a one-pole response. Z_1 and aZ_1 can be calculated through 0th and 1st order moments (integrals and centroids) of the TLP waveforms, respectively, although it will be shown how familiar step-response TLP to steady state requires only integration of the waveform in order to extract the s-coefficient. Essentially the same observations apply to the admittances Y_1 and aY_1 , in shunt networks complementary to the presumed series network with impedance $Z_1(1+as)$. Higher order terms in the expansion of

Z_1 or Y_1 in s can then be found from the higher order waveform moments.

The methods can also be extended to very fast TLP (VF-TLP) cases where the waveforms are too short to find a clear steady state. With proper calibration pulses for comparison, much essential modeling information can still be extracted. Convolution techniques and theorems are used to formulate the methods and prove the relationships.

These methods are applied to some examples of time dependent response. One is ferrite inductance and resistance as extracted from step pulse response, and the other is a model of diode-triggered SCR overshoot response to a TLP step. The very same concepts can be applied to circuits with a shunt capacitance.

Finally, we discuss methods of removing the effects of the measurement tools. An example is the finite rise time of the TLP step, in cases where the pulse speed competes with the device under test. Another example is correcting for the voltage droop of a current transformer, as used in HBM waveform measurements. The true RC time constant of an HBM tester is thus extracted.

II. Step Response Solution

1. Waveform Analysis and Networks

Traditional Transmission Line Pulsing (TLP) could be described as examining a device's step response out to a time where steady voltage V_d and current I_d are achieved, giving a steady-state $Z_1 = V_d/I_d$. But this is just the first term of what is presumed to be a (locally) linear, time-invariant network, and the true impedance function in the s-domain, describing the transients as well, would be $Z = Z_1(1 + as + bs^2 + \dots)$, $s = \sigma + j\omega$, with nonzero s-coefficients. An objective in this paper is to examine the a term, for basic inductive or capacitive response, and to show how to find more terms if desired.

In the TDR-TLP method [2,3], although the others could be used too, we obtain device voltage and current from

$$V(t) = V_+(t) + V_-(t), \quad I(t) = \frac{V_+(t) - V_-(t)}{Z_0}, \quad (1)$$

while still knowing our outgoing pulse from V_+ . The prototype circuits for this characterization are in Fig. 1, showing impedances Z , admittances Y , and step voltage and current sources that may not be ideal.

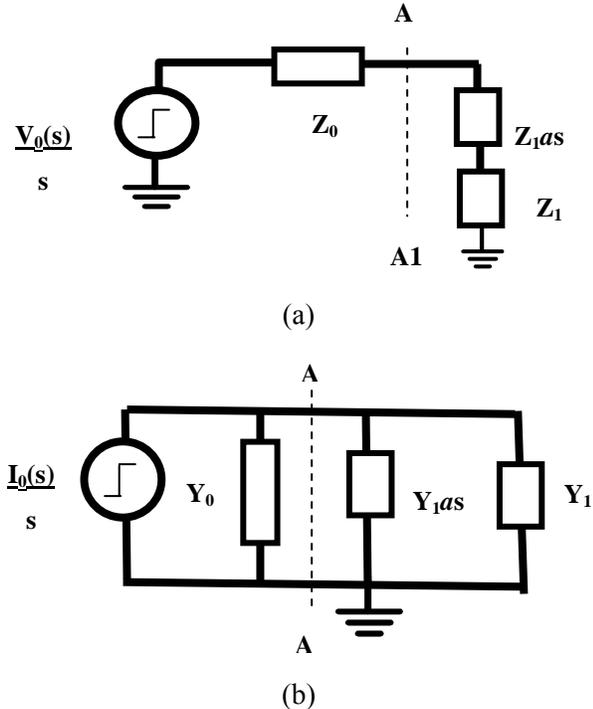


Figure 1: a) Thevenin equivalent step-driven TLP circuit with transient. b) Norton equivalent. Device is at plane A-A1.

The equations for networks (a) and (b) in Fig. 1 are similar and have complementary solutions, so we will focus on (a), best suited to an inductive transient. The device voltage V_d at plane A-A1 responding to a step voltage is, in the complex frequency or s-domain,

$$V_d(s) = \frac{Z_1 + Z_1as}{Z_0 + Z_1 + Z_1as} \frac{V_0(s)}{s} \\ = \frac{X_{div}(1 + as)}{1 + X_{div}as} \frac{V_0(s)}{s}, \quad X_{div} = \frac{Z_1}{Z_1 + Z_0}. \quad (2)$$

Expanding the ratio to first order we have

$$V_d(s) = X_{div} [1 + (1 - X_{div})as + \dots] \frac{V_0(s)}{s} \quad (3)$$

The coefficient of $V_0(s)/s$ in (3) is the transfer function; call it $X_1(s)$. We differentiate to get the impulse response, which is

$$sV_d(s) = \mathcal{L}\left(\frac{dV_d(t)}{dt}\right) \\ = [X_{div} + X_{div}(1 - X_{div})as + \dots] V_0(s). \quad (4)$$

We use $\mathcal{L}(\)$ to denote the Laplace transform of $dV_d(t)/dt$. According to the convolution theorems in [4] (p. 111), the time-domain areas of convolved functions, as in (4), multiply, and the time domain centroids add, so that

$$V_d(t \rightarrow \infty) = X_{div} V_0 = \frac{Z_1 V_0}{Z_1 + Z_0} \quad (5), \text{ and}$$

$$\langle t \rangle \frac{dV_d}{dt} = \langle t \rangle_{X_1} + \langle t \rangle_{V_0},$$

$$\text{where } \langle t \rangle_f = \frac{\int_0^{\infty} tf(t)dt}{\int_0^{\infty} f(t)dt}. \quad (6)$$

The leading term, Z_1 , computes as usual. Note that $\langle t \rangle_{V_0}$ is the centroid of the differentiated incoming step.

In work pioneered by Elmore [5] and continued in more recent works [6,7], it is shown that a waveform $h(t)$ is transformed into the Laplace domain by expanding the exponent in the transform as follows:

$$\begin{aligned}
H(s) &= \int_0^{\infty} h(t)e^{-st} dt \\
&= \int_0^{\infty} h(t) \left[1 - st + \frac{s^2 t^2}{2} - \frac{s^3 t^3}{6} + \dots \right] dt \\
&= \sum_{k=0}^{\infty} \frac{(-1)^k}{k!} s^k \int_0^{\infty} t^k h(t) dt. \quad (7)
\end{aligned}$$

Equation (5), above, is about the first term, the waveform integral or 0th moment. The second term (i.e., the first moment $k=1$ in the moment expansion) determines the s -coefficient of $H(s)$. Thus we can find the DUT impedance, $Z_1(s)=Z_1(1+as)$.

Integration by parts [8] states that if $f(t)$ and $g(t)$ are two continuously differentiable functions, then given an interval with endpoints a, b , one has

$$\int_a^b f(t)g'(t)dt = [f(t)g(t)]_a^b - \int_a^b f'(t)g(t)dt. \quad (8)$$

Our main case is $a=0, b=t_1$ (integration to somewhere in the flat part of $V_d(t)$), $f(t)=t, g(t)=V_d(t)$, and $V_d(0)=0$. Thus

$$\int_0^{t_1} t \frac{dV_d(t)}{dt} dt = t_1 V_d(t_1) - \int_0^{t_1} V_d(t) dt. \quad (9)$$

To get the centroid, we normalize as in Eq. (6) by dividing (9) by

$$\int_0^{t_1} \frac{dV_d(t)}{dt} dt = V_d(t_1) = X_{div} V_0 = \frac{Z_1 V_0}{Z_1 + Z_0}. \quad (10)$$

The same kind of thing is done to $V_0(s)$ to calculate its centroid from $V_+(t)$, which flattens out at $V_0/2$. More about correcting for $V_0(s)$ is in Section IV.

The integration limit t_1 in Equations (9-10) can stop anywhere on the flat (derivative is zero) part of $V_d(t)$ because nothing is added to the centroid in that region. Now we have

$$\begin{aligned}
\langle t \rangle_{\frac{dV_d}{dt}} &= t_1 - \frac{Z_1 + Z_0}{Z_1 V_0} \int_0^{t_1} V_d(t) dt, \\
\langle t \rangle_{V_0} &= t_1 - \frac{2}{V_0} \int_0^{t_1} V_+(t) dt. \quad (11)
\end{aligned}$$

This gives

$$\langle t \rangle_{X_1} = \frac{2}{V_0} \int_0^{t_1} V_+(t) dt - \frac{Z_1 + Z_0}{Z_1 V_0} \int_0^{t_1} V_d(t) dt. \quad (12)$$

These integrals also cancel beyond a t_1 when the voltages V_+ and V_d go flat. But from Eqs. (4) and (7), we know that $\langle t \rangle_{X_1} = -\frac{Z_0 a}{Z_0 + Z_1}$; therefore the s -coefficient or effective inductance is

$$\begin{aligned}
Z_1 a &= L_{eff} \\
&= \frac{Z_1(Z_0 + Z_1)}{Z_0} \left[\frac{Z_1 + Z_0}{Z_1 V_0} \int_0^{t_1} V_d(t) dt - \frac{2}{V_0} \int_0^{t_1} V_+(t) dt \right] \\
&= \frac{(Z_0 + Z_1)^2}{Z_0 V_0} \int_0^{t_1} [V_-(t) - \rho_0 V_+(t)] dt \quad (13)
\end{aligned}$$

This theorem can also be put in terms of Laplace transforms: For a network function (impulse response) $N(s) = a_0 + a_1 s + a_2 s^2 + \dots$, the step response gives $a_0/s + a_1 + a_2 s + \dots$. Subtract the dc offset and the a_0 term is gone; integrate this new waveform (i.e., apply $1/s$) and the new dc offset gives the first moment a_1 . Higher-order moments can be extracted by successive reduction of waveforms in this fashion.

For the above inductor in a series circuit with total resistance $Z_0 + Z_1$, the impact on Z_1 rise time (Elmore Delay) is

$$\begin{aligned}
t_D &= \frac{L_{eff}}{Z_0 + Z_1} \\
&= \frac{Z_1}{Z_0} \left[\frac{Z_1 + Z_0}{Z_1 V_0} \int_0^{t_1} V_d(t) dt - \frac{2}{V_0} \int_0^{t_1} V_+(t) dt \right] \\
&= \frac{(Z_0 + Z_1)}{Z_0 V_0} \int_0^{t_1} [V_-(t) - \rho_0 V_+(t)] dt. \quad (14)
\end{aligned}$$

ρ_0 is $(Z_1 - Z_0)/(Z_0 + Z_1)$. Note that there is no difficulty for $Z_1 = Z_0$ or for Z_1 approaching zero. For the case of negligible rise time or overshoot on the input pulse, these equations simplify to

$$L_{eff} = \frac{(Z_1 + Z_0)^2}{V_0 Z_0} \left[\int_0^{t_1} V_d(t) dt - \frac{V_0 Z_1 t_1}{Z_1 + Z_0} \right] \quad (15)$$

Now the coefficient of t_1 is $V_d(t_1)$, the final voltage, and it's even clearer why the integration limit t_1 is not critical. The time delay, or inductance, is the difference of two areas on a voltage-time plot. However, for our prototypical transient modeled by a single inductor or capacitor plus a resistor, the rising slope of the overshooting quantity (voltage for an inductor, current for a capacitor) will follow the input

waveform. Thus the graphical view of Eq. 14, in Fig. 2, shows matched rising edges for the two waveforms whose areas will be subtracted. The inductance and time delay is then computed from the overshoot-only area as shown.

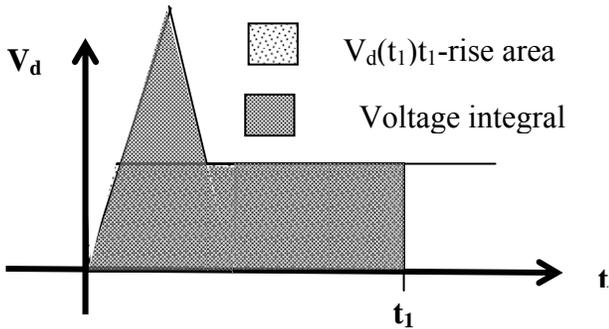


Figure 2: Time delay or inductance as derived from the difference of two areas, as in Eq. (14).

2. Ferrite Example

An example of this is found in the analysis of Figures 3 and 4, showing an overshooting TLP voltage pulse resulting from a step input to an 11-ohm resistor with a ferrite slipped over one of the resistor leads. This creates an inductance and, as we shall see, series and parallel resistances. Figure 3 shows the result at a short time scale (2.5 nsec/div), with overshoot due to the wire inductance, while Figure 4, at 1 μ sec/div, shows overshoot due to the (considerably larger) ferrite inductance.

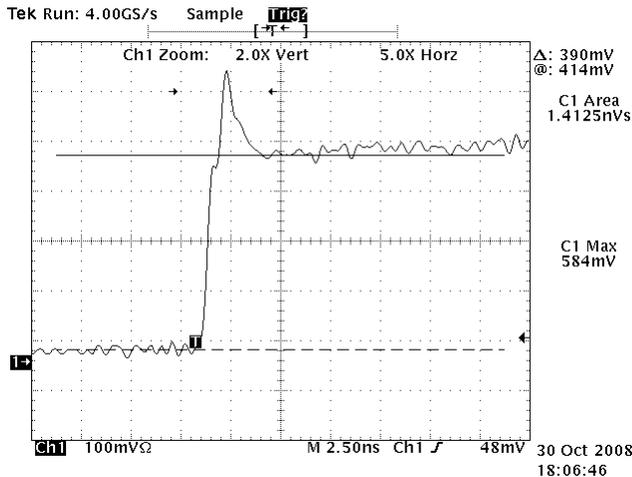


Figure 3: Voltage waveform of 11 ohm resistor plus ferrite on a resistor lead. Short time scale shows wire inductance (30 nH) and additional resistance values.

In Fig. 3, we find that 390 mV (after 100X attenuation) compares with 760 mV open circuit, giving a long-term Z_1 impedance of 52.7 ohms. This means that the ferrite adds 41.7 ohms to the circuit, plus its inductance. The overshoot peaks at 560 mV, short of the open circuit value, approximated by the

added inductance being in parallel with about 98 ohms. It is approximate because finite rise time has an effect, to be discussed later. Fig. 4 shows a long-term decay of voltage due to ferrite inductance. We integrate the overshoots to find inductances and complete the resulting network (Figure 5). The inductance of 30.2 nH, from the fast waveform, is

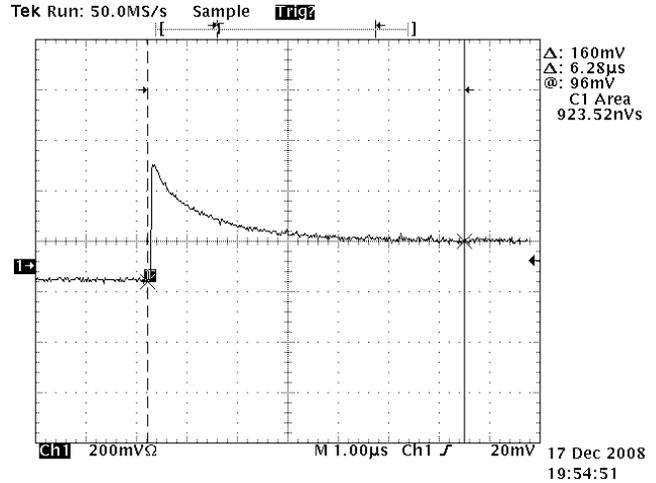


Figure 4: Voltage waveform of 11 ohm resistor plus ferrite on a resistor lead. Longer time scale shows ferrite inductance (25 μ H) and confirms resistance values.

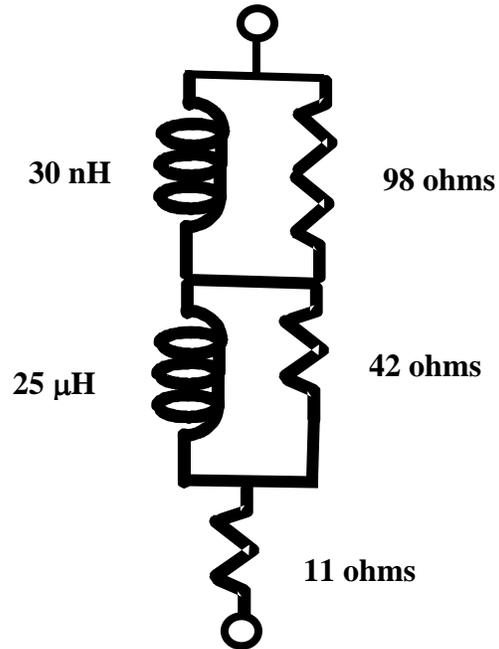


Figure 5: Circuit model of 11 ohm resistor plus ferrite, from waveform analysis at two time scales.

about right for the resistor wire itself. Over a period of 5.5 μ sec, the sustaining voltage declines to a level appropriate to the remaining 11 ohm resistor, as expected. This gives 25.4 μ H in parallel with the 42 ohms as the ferrite contribution, as in Fig. 5.

3. Diode Example

The forward-biased diode is of course the most common ESD protection device. The best protection diodes respond quickly to fast pulses but some overshoot of the final voltage has been seen if a fast enough step was applied. As VF-TLP has advanced, rise times have shortened and the overshoot of well-designed ESD protection diodes and associated circuits are being more clearly measured. Figure 6, from this same conference [9], is a good example of overshoot from a diode-triggered silicon controlled rectifier (DTSCR), incorporating both the diode and SCR delay.

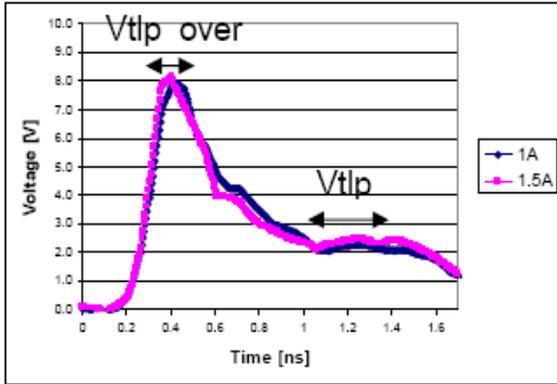


Figure 6: VF-TLP measurement of the overshoot transient behavior of a DTSCR [9].

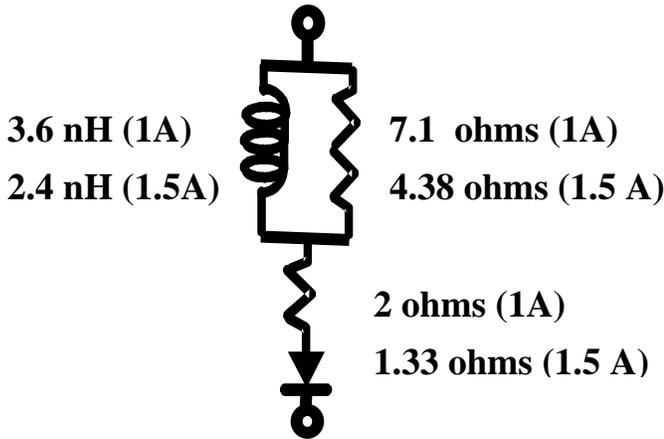


Figure 7: Circuit model for DTSCR as computed from waveforms in Fig. 6.

The overshoot shown in Fig. 6 is almost the same for the 1A and 1.5A cases (50 ohms, for 53V and 79V line charge), while the 8V peak indicates a noticeable parallel resistance in the equivalent circuit, exclusive of rise time effects. This means that the equivalent circuit will depend on pulse voltage and current, as shown in Figure 7. This is not surprising, given that the delay times and overshoots are semiconductor

phenomena, nonlinear and nonmagnetic. Even so, models such as in Fig. 7 are useful in ESD circuit simulation.

How much this circuit model, or even device behavior itself, depends on the measurement system is not known. As the TLP impedance Z_0 affects the Elmore Delay (Eqs. 14-15), the waveform could change for 1A or 1.5A final current if the load line Z_0 is different. This could be an interesting item for future study.

4. Other Conditions

For the common case of current overshoot and voltage gradually approaching a limit, one will apparently have negative inductances and time delays, which are still meaningful. But the model in Fig. 1b, entirely complementary, would give positive aY_1 , or C_{eff} , and positive $t_D = C_{eff}/(Y_0 + Y_1)$, and could be more agreeable for data records. Figure 2 still applies if the device current is plotted instead of voltage, although integration of the voltage undershoot due to capacitance is an equally acceptable way to calculate it [10], as would be integrating the current undershoot to find inductance. Indeed, the Elmore Delay is most commonly represented as an R-C phenomenon, as in [10]. The present work's treatment shows exactly the same moment-matching method in equivalent and complementary terms, and thus demonstrates the power of the technique for both inductive and capacitive transients. If one substitutes

$$V_0 \rightarrow I_0$$

$$Z_0 \rightarrow Y_0$$

$$Z_1 \rightarrow Y_1$$

$$V_d \rightarrow I_d$$

$$V_{+,-} \rightarrow I_{+,-}$$

in Eq. 13 and uses the Norton network (Fig. 1b), one gets $C_{eff} = Y_1 a$ and a method of finding C_{eff} .

The ease of determining L_{eff} or C_{eff} with the above methods could allow a needle probe pair as in [11] to be modeled as a high-Z transmission line by finding L_{eff} for a shorting load and C_{eff} for an open load. Equation 13, and its equivalent for capacitance, describes the proper waveform area differences, and accounts for the finite rise time of the source. With L_{eff} and C_{eff} , the transmission line transfer function could be used for de-embedding various devices being probed, and would comprehend load-dependent effects. Thus if needle probes of high impedance

$Z_{line} = \sqrt{L_{eff} / C_{eff}}$ and electrical length l connect to a load Z_L , the impedance looking into the needle probes followed by load Z_L is

$$Z_{in} = Z_{line} \left[\frac{1 + jr \tan \beta l}{r + j \tan \beta l} \right], r = \frac{Z_{line}}{Z_L},$$

$$\beta l = \omega \sqrt{L_{eff} C_{eff}} = \frac{\omega l}{c}, \quad (16)$$

c the propagation velocity on the needle probe line. In s-domain terms, $\tan(\beta l)$ could be replaced by $\tanh(ks)$, approximated by $ks - (ks)^3/3 + \dots$. Eq. 16 thus gives all we need to find Z_L (and not just low-impedance loads) given a first-order measurement of $Z_{in} = Z_L(1 + as)$. The latter is derived by measuring step response and transients as described earlier.

III. Short Pulse Solution

TLP as step response is the most comfortable situation for analyzing the transients. Even if VF-TLP methods are used [3], if the steady-state is achieved in a short response time, the above analysis can be used. But sometimes transients will not finish and falling pulse edges will have to be comprehended along with the usual rising edges.

In such a case we make use of the fact that the pulses return to zero, and therefore have their own moment series without taking derivatives. The short incoming pulse is, in effect, an approximation of an impulse. The responding device voltage thus approximates an impulse response or transfer function. To quantify this, we return to Fig. 1 and replace the step voltage source with a $V_0(s)/s$ representing the near-rectangular VF-TLP pulse. In the s-domain, the perfect rectangular ‘‘boxcar’’ function of length Δt is

$$\frac{V_0(s)}{s} = \frac{V_0(1 - e^{-\Delta t s})}{s}, \quad (17)$$

also expressible in $\sin x/x$ form. Let this correspond to $V_0(t)$. In any case, Eq. 3 applies, where now

$$V_d(s) = [X_{div} + X_{div}(1 - X_{div})as + \dots] \frac{V_0(s)}{s} \quad (18)$$

and following Eqs. 5-7, the 0th and first moments can be matched to determine the coefficients of $X_1(s)$, the coefficient of $V_0(s)/s$ in (18):

$$\int_0^{\infty} V_d(t) dt = X_{div} \int_0^{\infty} V_0(t) dt, \quad (19)$$

$$\langle t \rangle_{V_d} = \langle t \rangle_{X_1} + \langle t \rangle_{V_0}. \quad (20)$$

Short pulse waveform integrals as in (19) determine Z_1 , while $Z_1 a$ now results from actual computation of

the centroids, with units of time, as in (20). In accordance with (7) and (20),

$$\langle t \rangle_{X_1} = -\frac{Z_0 a}{Z_0 + Z_1} = \langle t \rangle_{V_d} - \langle t \rangle_{V_0}, \quad (21)$$

$$\text{and } Z_1 a = L_{eff} = \left[\langle t \rangle_{V_0} - \langle t \rangle_{V_d} \right] \frac{Z_1 (Z_0 + Z_1)}{Z_0}. \quad (22)$$

Positive L_{eff} results when the overshoot on V_d produces a centroid value less than that of the incoming waveform V_0 .

IV. Waveform Corrections

1. Inverse Filtering and Deconvolution

TLP transients will sometimes be on the same time scale as the TLP rise time. For example, in Fig. 3 and Fig. 6, a detailed TLP calibration waveform into a 50 ohm load should help find a more nearly ideal step response for the device under test.

In earlier sections, the generalized s-domain TLP step is given as $V_0(s)/s$. With $V_0(s)$, we can invert it, multiply the original waveform (in the s-domain; call it $W(s)$), and transform to the time domain for the corrected waveform $W_c(t)$. A frequency domain representation of $V_0(s)$ is sufficient; consider this simple example.

Starting with digital data of a calibration step into 50 ohms (call it $V_{cal}(t)$), we differentiate to get

$$\frac{dV_{cal}(t)}{dt} = V_0(t) \Leftrightarrow \frac{sV_0(s)}{s} = V_0(s). \quad (23)$$

$V_0(t)$ looks like a short pulse; with methods described earlier to find the centroid τ_r of this pulse, we can transform $V_0(t)$ into $V_0(s)$ after approximating $V_0(t)$. The centroid τ_r is the rise time of a 1-pole exponential approach expression for the $V_{cal}(t)$ waveform, and if $a = 1/\tau_r$, we have

$$V_{cal}(t) = V_0 \left(1 - \exp\left(-\frac{t}{\tau_r}\right) \right)$$

$$\Leftrightarrow V_{cal}(s) = \frac{V_0}{s(s+a)} = \frac{V_0(s)}{s}. \quad (24)$$

Thus to restore our ‘‘perfect’’ step function V_0/s , we multiply $V_{cal}(s)$ by $s+a$, and similarly with $W(s)$ to find $W_c(s)$ and eventually $W_c(t)$. The convolution theorem cited earlier [4] provides guidance.

Multiplying by $s+a$ means to differentiate the step response $W(t)$ (s is the differential operator) and add to it an amount of $W(t)$ proportional to a , the reciprocal of the rise time. A fast rise time will thus

give a $W_c(t)$ that is much the same as $W(t)$, with little correction due to the derivative. Conversely, a step response dominated by TLP rise time will show more of a true step response, of the device only, consistent with noise limits of the measuring system.

Notice that our one-pole approximation of $V_{cal}(t)$ was a “simple example”, arbitrarily chosen. We could have chosen to take more moments to find a multi-pole fit for $V_0(s)$ and $V_{cal}(s)$, resulting in a more complicated inverse filter expression, and ever-improving approximations of the waveform, ever-closer to the noise limit of the channel. Indeed, we could take the digital calibration data and do a computer-based fast Fourier transform (FFT) to find out as much as possible about $V_0(s)$. That would take us to **Wiener deconvolution** [12,13], a long-established signal processing method. This would make maximum use of the calibration data in order to remove its influence from the final measurement.

To correct the above waveform $W(s)$ with a Wiener filter $G(s)$ so that $G(s)W(s)=W_c(s)$, the Wiener deconvolution theorem [12,13] states that

$$G(s) = \frac{1}{V_0(s)} \left[\frac{|V_0(s)|^2}{|V_0(s)|^2 + \frac{1}{SNR(s)}} \right], \quad (25)$$

where $SNR(s)$ is the frequency-dependent signal-to-noise ratio. In the absence of noise, for infinite $SNR(s)$, this is exactly the $1/V_0(s)$ inverse filter that we used above. But it is clear that any $(s+a)$ filter needs to pay attention to noise, at least at low and high frequency, and roll off accordingly.

Many modern oscilloscopes have the digital signal processing (DSP) and math functions required for complete inversion of a measurement channel transfer function like $V_0(s)$. This allows measured waveforms to be processed by a user-defined finite impulse response (FIR) filter [14]. The FFT math function alone (present in many digital oscilloscopes) should allow a first moment to be measured by taking a slope at low frequency. As these tools become more common, it is very likely that ESD waveform measurement and calibration will make use of them more often.

2. Practical Example, HBM Waveform

An interesting yet simple case of inverse filtering would be to examine the 0-ohm ~ 150 nS decay constant of the human body model (HBM) ESD tester according to ESDA and JEDEC standards [15,16]. The required current transformer is or resembles the Tektronix CT-1, which is specified with a minimum

L/R droop time of 6.35 microseconds; call it τ_{xf} . The result is a step response (readily measured with TLP) resembling Figure 8, generally true of any transformer.

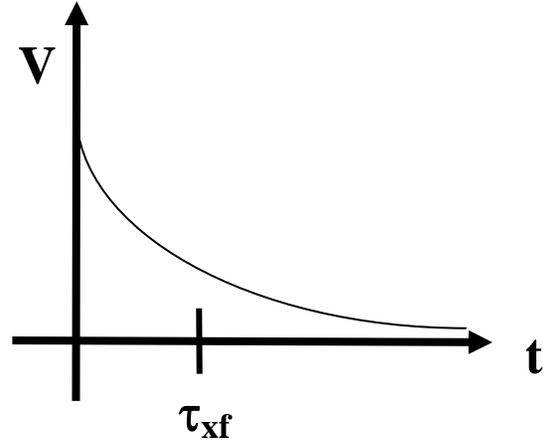


Figure 8: Step response of a transformer; τ_{xf} is the $1/e$ decay time.

Impulse response would be the derivative of Fig. 8, meaning a spike at zero followed by a negative gradual return to zero. On the 150 nS time scale of HBM, the negative contribution to the convolution with the “real” waveform is small but not negligible. Note that the step response is the familiar 1-pole exponential decay, $1/(s+a)$ again, where $a=1/\tau_{xf}$. But this is the step response; the impulse response is $s/(s+a)$, so the inverse filter is simply $1/a/s$. The measured HBM waveform $h(t)$ can therefore be corrected to

$$h_{corr}(t) = h(t) + \frac{\int_0^t h(\tau) d\tau}{\tau_{xf}}, \quad (26)$$

as $1/s$ is the integration operator. Note that the correction is not only inversely proportional to the L/R time constant, as expected, but also grows as the waveform progresses because of history. Thus the 36.8% point in the waveform should be noticeably affected, adding several nanoseconds to the measured decay constant for the τ_{xf} of 6.35 μ S, cited above.

We applied (26) to some measured HBM waveform data, which had the expected drop below zero for long times (>700 nS in this case). The 36.8% decay point rose from 136.2 nS to 142.8 nS with the correction, about as expected and closer to the nominal 150 nS decay constant. The rest of the difference from 150 nS should be related to the fact that RC is the sum of the time constants for a 2-pole or even multi-pole RLC model of the HBM waveform [6, 17], with the result that a long rise time shortens the decay constant, for a given RC product. This last topic will

be treated at greater length in a future work, but now that we have a corrected HBM waveform, we should at least find the RC time constant.

For a 2-pole series RLC HBM equivalent circuit, the admittance in the s-domain is

$$Y(s) = \frac{Cs}{LCs^2 + RCs + 1} \quad (27)$$

Discharge of the charged capacitor amounts to finding the step response of this circuit, so the centroid of the 0-ohm HBM waveform gives RC. This is true even for the 4-pole HBM circuit model as in [18], despite the more complicated Y(s) expression.

The integration by parts theorem and its Laplace equivalent in Eqs. 8-13 inspire us to find RC, through the same kind of integration and reduction process, now using the HBM current waveform $h_{corr}(t)$. After total charge Q_t , and therefore C, are found by integrating the current, the next integral gives us RC. Figure 9 shows the integrated 1kV HBM waveform as a starting point; the shaded area A gives RC when divided by Q_t . This can be stated as

$$RC = \frac{\int_0^{\infty} \left[Q_t - \int_0^t h_{corr}(\tau) d\tau \right] dt}{Q_t} \quad (28)$$

It is the Elmore Delay of the integrated HBM waveform. The result for the case in Fig. 9 was C=110 pF, R=1511 ohms, RC=166 nS. This is the same HBM waveform as above that gave 136.2 nS as the decay constant with the uncorrected current probe through standard methods. It now appears we can improve on those methods [15,16] so that we don't span the entire 130-170 nS allowed range with the various alternatives. Digital oscilloscopes and ability to do this modest amount of data processing are now common enough to explore the methods outlined here.

Conclusions

We now can report first-order DUT transient information from TLP data, with a $Z_1(1+as)$ for each pulse condition. This is done by considering the zeroth and first moments (integrals and centroids) of the waveforms or their derivatives, depending on whether short or long TLP pulses are used. This practice corresponds to "moment matching" of dc and first order transient response. The result is an equivalent series R+Ls or shunt admittance G+Cs, capturing voltage and current overshoot, respectively. The waveform moments are shown to be obtainable from successive waveform integration, with each

moment found as a dc offset, then removed and the process continued to find the next moment.

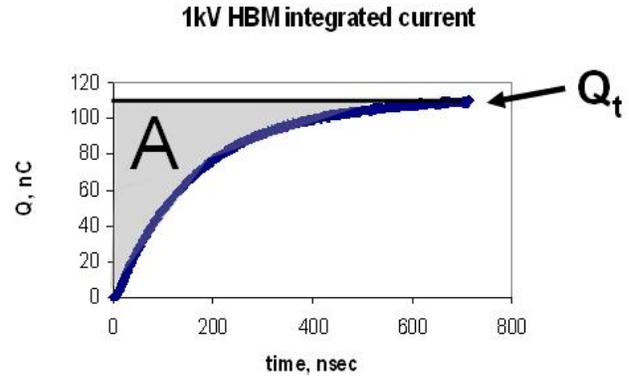


Figure 9: Integrated 0-ohm 1kV HBM waveform, showing convergence to total charge $Q_t=110$ nC. The centroid of the original waveform is its time constant $RC=A/Q_t$, and here is also the time at 63.2% of Q_t , the expected result for 1-pole Elmore Delay [6,10].

In this work, a circuit with a resistor plus ferrite bead was measured and characterized, with first order circuit models found sufficient to capture a 4-element model that operates at two distinct time scales.

We applied the same methods to formulate a circuit model for the transient behavior of a diode-triggered SCR as reported at this conference [9]. The linearized models are voltage/current dependent, as can be expected with semiconductor phenomena, but useful for circuit modeling and for capturing the essentials.

The effect of finite rise time of a TLP system is often not negligible, particularly if the transients are on the same time scale as the TLP rise time. For example, the foregoing measurements and model extraction for ferrite bead and DTSCR would be somewhat enhanced by correcting for the TLP's finite rise time. Often, even a one-pole model of the transient from TLP calibration data helps.

We also applied the same time-and-frequency domain Laplace transform methods to show how to correct HBM waveforms for droop in the CT-1 current transformer. The corrected waveform had closer to the expected decay constant value, and was also suitable for moment extraction and accurate measurement of the RC time constant and its constituents. The work explained much of why the HBM decay times come out lower than the expected 150 nS, while the actual RC is higher.

The correction methods are examples of inverse filtering and deconvolution, and those discussed here can be achieved simply with digital waveform data and a spreadsheet program. Modern DSP methods and their growing use in oscilloscopes can help

achieve this even more easily and are expected to help with this activity in the future.

The lesson for all of us in this work is that TLP response is a step response, and results can be treated as an s-domain network function. With step response as our data, we are inspired to characterize our devices, circuits, and even measurement systems in a structured manner, using the many tools of network theory, time-and-frequency transformations, and their related approximation methods.

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